GUITAR RECORDING AND SOUND ISSUE

THE PROJECT RECORDING & SOUND MAGAZINE

1996 • ISSUE TWO

NEW GUITAR GEAR IN REVIEW: ART TUBE MIC PREAMP KORG PANDORA PROCESSOR HUGHES & KETTNER TUBEMAN TECH 21, MIDI MOUSE

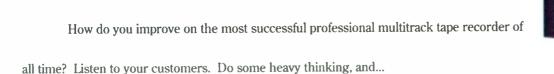
2 I E

•

AORE TIPS & TECHNIQUES: FENDER'S NEW MIXER GREAT DIRECT SOUND BLUES TRAVELER LIVE HENDRIX: STUDIO DAYS STEROIDS FOR GUITAR ENGINEER WOES COUNTRY STRINGS LIVE GUITAR GD-ROMS

STUDIO

\$3.95 Can. \$4.95 U.K. £2.60 February 1996



Think Agai

8 7

Make the transport four times faster and put it under constant software control. Incorporate advanced onboard digital

s AL

bouf



editing with track copy, auto punch, track delay, tape offset, 10-point autolocator, rehearse mode and more. Use the latest oversampling converters for the ultimate in digital audio quality. Design a beautiful vacuum fluorescent display

0

recor

ALESIS

that provides all the critical information. Wrap all this well-thought-out technology in an utterly professional six-pound solid die-cast aluminum chassis. Of course, make it 100% compatible with over 70,000 ADATs already in use worldwide. Introducing the new, definitely improved **ADAT-XT' 8 Track Digital Audio Recorder**. Consider it a think tank for your creativity. See your Alesis dealer. Don't think twice.



adat xt

CK DIGITAL AUDIO RECORDER

THE STREET

s Corporation 3630 Holdrege Avenue Los Analysis Co

ers CA 90016 310-841-2272 alecorp@albsis1.usa.com



adat











ON THE COVER: Edward Van Halen at his 5150 Studio. Photo by Karen Miller.

FEATURES

EQ LIVE

PETE ANDERSON: SIX STRINGS ON STAGE By Liana Jonas	97
BLUES TRAVELER: THE EVENING BLUES By Richard Vink	104
NEW GEAR FOR YOUR NEXT GIG	10 <mark>8</mark>
ROAD TEST: FENDER MX-5232 MIXING CONSOLE By Zenon Schoepe	110
ROAD TEST: AUDIO TOYS INC. PRO6 By Wade McGregor	114

TECHNIQUES/WORKSHOPS

CHAS CHANDLER: GUIDING JIMI HENDRIX By Patrick Stapley	50
ENGINEERS VS. GUITARISTS By Bennet Spielvogel	58
ACOUSTIC GUITAR MIKING TECHNIQUES By Edward Gerhard	6 <mark>2</mark>
DIY: STEROIDS FOR ELECTRIC GUITAR By Craig Anderton	118

COLUMNS/DEPARTMENTS

MI INSIDER. THE ULTIMATE DIRECT GUITAR SOUND By C	raig Anderton42
MAINTENANCE: TUBE AMPLIFIER TIPS By Eddie Ciletti.	
ACROSS THE BOARD: TALES OF A BLITHERING MIDIOT By	Roger Nichols146
EDITORIAL	ROOM WITH A VU: MICK GUZAUSKI
LETTERS TO EQ	EQ TRACKS: DAR WILLIAMS "MORTAL CITY"
EQ&A	FIRST LOOK: TL AUDIO INDIGO SERIES
PRODUCT VIEWS	IN REVIEW: VALLEY AUDIO DYNAMAP
STUDIOWARE	AD INDEX

EQ (ISSN 1050-7868) is published monthly by Miller Freeman PSN Inc., 2 Park Ave., Ste. 1820, New York, NY 10016. Second dass postage paid at New York, NY and additional mailing offices. POSTMASTER: Send address changes to EQ, P.O. Box 0532, Baldwin, NY 11510-0532. SUBSCRIPTIONS: U.S. 1 yr. \$24.95, 2 yrs. \$39.95, 3 yrs. \$59.95; CANADA add \$10.00 per year for surface; other countries add \$15.00 per yr. for surface; All add \$30.00 per yr. for Airmail. Back-issues \$5. Printed in the U.S.A.

THE BEST DYNAMIC MICROPHONES IN THE WORLD

And that's *not* just our opinion. Why do 70% of world tours specify the M 88 TG in their equipment schedules? Pure quality of sound and dependable performance.

The M 88 TG's stunning performance together with a construction tough enough to cope with the road make it a favourite with artists and sound engineers.

beverdynamic)

On vocals, the M 88 TG's low proximity effect and hypercardioid response combine to give outstanding performance with fantastic feedback rejection while the high SPL capability provides ((()) a clean, clear signal with a unique sound.

For instrument use on stage, check out the M 69 TG. Its low proximity effect, high sensitivity, and extended frequency response give excellent results in almost every environment.

CIRCLE 12 ON FREE INFO CARD

FREE HEADPHONE!

Buy an M 88 TG or M 69 TG mic between Nov. 1 and Jan. 31, 1996 and get a <u>FREE</u> pair of DT 211 stereo headphones— a \$49.00 value. See your dealer for details. Now priced within everyone's reach!

For more information or a brochure, call 1-800-293-4463



A Miller Freeman PSN Publication Vol. 7, No. 2 February 1996

PAUL G. GALLO Publisher

KATHLEEN MACKAY Associate Publisher

TIM WETMORE Editorial Director

HECTOR LA TORRE Executive Director

MARTIN PORTER

ANTHONY SAVONA Managing Editor

CRAIG ANDERTON Technology Editor

STEVE LA CERRA Senior Editor

LIANA JONAS Associate Editor

DAVID MILES HUBER, EDDIE CILETTI, DAVID FRANGIONI, BOB LUDWIG, HOWARD MASSEY, WADE MCGREGOR, ROGER NICHOLS, MARTIN POLON

Contributing Editors

Editorial/Design Consultants

MATT CHARLES

ANDREA BERRIE, CHRISTINE CALI, KAREN GODGART, MICHAEL PHELPS, TARA ESPOSITO Advertising Soles

AMY HERRION Marketing Assistant

RIVA DANZIG Creative Director

MARK ALHADEFF Art Director

SUSAN FALCO Assistant Ar Director

MELISSA HALKA

JOY ZACCARIA Reprint Coordinator

FRED VEGA Production Manager

Editorial Offices 939 Port Washington Blvd. Port Washington, NY 11050 Tel: (516) 944-5940, Fax: (516) 767-1745

Administrative/Sales Offices 2 Park Avenue, Suite 1820 New York, NY 10016 Tel: (212) 213-3444, Fax: (212) 213-3484



EQ (ISSN 1050-7368) is published manthly by Miller Freeman PSN Int. 2 Park Avenue, Suite 1820, New York, NY 0014 Secand class postage paid at New York, NY and additional realing offices, POST-

Additional realing offices. POST-MASTER: Send address changes to EC, PO Back 0532, Boldwin, NY 11510-0532, SUBSCRIPTIONS U.S. 1 yr. 524 95, 2 yr. 539 95, 3 yr. 559 95, CANADA and 310 per yr for unface other countries add 315 per yr for unface, Al add 320 per yr for Armel Back Issus 55. Add praduct information is wybact to change. Dor ber assumes na responsibility for such changes. All laied madel numbers and praduct names are manufacturen' regulared Content of the Content of the comment. Printed in the U.S.A.

The Axe is Back!

LETTER FROM THE EDITOR

Although guitars lost some of the spotlight when the MIDI revolution put synthesizers on the map, the axe is back — in a big way. Whether you're talking the angst of alternative or the taut rhythms of worldbeat, people are rediscovering the magic an accomplished player can coax from six pieces of metal stretched nearly to the breaking point over a plank of wood. From the sensuous timbres of nylon-string classical to the brain-bashing crunch of a heavy metal power chord, the guitar occupies a unique place in our hearts, as well as in our ears.

And it also occupies a unique place on stage or in the studio. The funky, noise-prone, unpredictable effects and amps of the past — while coveted for some applications — are giving way to multieffects, predictable tube emulation, superclean sounds, better tube design, and physical modeling. For capturing the music, hard disk recording gives guitarists the kind of editing options formerly reserved for the MIDI world, while digital tape preserves the guitar's sound quality better than analog tape ever did.

The issue of miking amps and guitars has been dissected time and time again, but comparatively little attention has been given to guitars in the project studio — where subjects such as direct injection, amp emulation, and level/impedance matching have become an important part of the mix. And there's more to come, since the guitar renaissance we've seen recently has not yet peaked. Soon we'll see more guitarists embrace hard disk recording, extensive MIDI control of multieffects, guitar "synthesizers," and much more.

This issue is the first salvo in *EQ*'s coverage of the recording and sound problems, and opportunities, faced by today's guitarist. Future articles will examine how one guitarist made the transition from heavy metal idol to multimedia mogul, what hard disk recording means to guitarists, and how to get true analog tape compression effects with digital recording (tape or hard disk, and we're not kidding!). If you're a guitarist — or an engineer trying for the ultimate guitar sound — keep reading.

Craig Anderton Technology Editor (and Resident Target of Guitarist Jokes)

AND THE MEMORABLE ADVANTAGES OF **AUTO-**Mation.

Pro Mix



A library of pre-programmed eq curves, dynamic settings and effect presets lets you work faster and more efficiently by drawing on the experience of mixing professionals.

R emember that great live gig last month? You know – the one where the mix was perfect. Every instrument eq'd just right. The punchy dynamics and tasteful effects.

ProMix Ol does.

Or how about that jingle you cut last year? What happens when the client calls and wants to change the voice-over. Can you precisely recall and reset the entire mix? Levels, eq. panning, compressors, effects and everything else? ProMix Ol can.

Redefining the affordable mixer. When it comes to mixing, the buzz is automation. It's designed to make your life easier and improve your mix. In the not-so-distant past, technology began allowing you to memorize and automate your mix. With only one catch: you needed access to a really sophisticated console costing a few hundred thousand dollars! Or you had to be willing to link your mixer to a pricey outboard automation system that still left many functions under manual control. Limited options at best

control. Limited options at best. ProMix Ol changes all that with memory and real-time automation of all mix settings – at a price you can afford. Before we tell you about ProMix automation, let's define the terms "automation" and "memory." What they actually mean; what they do for you; and what you really need.

Memorable moments.

Total Recall – Talk about buzz. This gives you the ability to go back in time. Total Recall tells you where the knobs were set on your mixer at a given moment in time. But you still have to manually reset most knobs and functions to a list of displayed values. Very timeconsuming if you consider eq, pan pots and aux sends. And this still doesn't take into account your outboard processing like gates, compressors, effects and routing. Total Reset - While Total Recall tells you where the controls were, Total Reset automatically returns the controls to the memorized positions for you. Very fast, very cool, and (until ProMix OI) very expensive.

New levels of automation.

Fader and Mute Automation – These are the standard automated functions. They allow hands-free real-time control over channel volume and/or channel on and off status. But remember – there's more to life than faders and mutes. What about eq. effects, dynamics and routing?

Onboard or External Automation-Where does the actual automation occur? Does the audio stay cleanly inside the console, or do you have to patch into an external automation box? And do those extra boxes and cables have an adverse effect on your sound?

Knowing these facts, suppose truly comprehensive memory and automation were available in a mixer you could actually afford. Would anything be left out? What would you get?

ProMix OI-The new definition of mixing with memory.

Snapshot Memory – As defined above, ProMix features total recall and total reset of every mixing parameter. It's like taking a picture of every mixer setting – the complete capture of a moment in time. That's why it's called snapshot memory. And with ProMix's total reset, you just press the Recall button and that moment comes back in an instant. In fact, ProMix Ol gives you the ability to save 50 of these snapshots, called scenes, right on board. Library Memories - Not only does ProMix Ol save, recall and reset your settings, but it also comes with the built-in experience and memories of working professionals. These onboard libraries include pre-programmed eq curves, dynamic settings and effect presets that help you get working quickly and efficiently. And as you develop favorite settings of your own, ProMix Ol even has room to store these custom settings in memory as well

Total Real-Time Automation - Add an outboard MIDI sequencer, and you'll have real-time control of every parameter inside ProMix OL That means everything - not just faders and mutes! So imagine eq, panning, dynamics, effects and more, all changing in real time. And not only can you imagine it, now you can afford it too. Complete Sonic Integrity - With its 20-bit A/D converters and its 24-bit internal digital processing, all of these features and functions happen inside ProMix Ol - without any compromise to your sound.

So after all this talk of automation and memory, what's the most important thing for you to remember? *ProMix 01.* And to jog your mind even further, **call 1-800-937-7171, ext. 570 for your free copy of the ProMix 01 Application Guide and Video.** Once you've read the book and seen the movie, you'll want to have your first lasting memory by visiting your nearest Yamaha dealer. Then get your hands on a ProMix 01. It might just remind you why you got into this business in the first place.

©1995 Yamaba Corporation of America, Pro Audio Products, P.O. Box 6600, Banns Park, CA 90522 (714)522-9011 Yamaha Camula Munic LTD, 135 Milhert Acemue, Scarborough, Ontario M18 3R1 (916) 298-1311 Store every setting in your mix as a single electronic snapshot, or "scene." Then push the Recall buttor, and you can instamly return to these settings – days, weeks or even months later

Ext. 1-800-937-7171



YAMAHA Smart Mixing CIRCLE 84 ON FREE INFO CARD



Call the 800 number below for your free ProMix 01 Video and Application Guide. LETTERS TO EQ

ON THE RIGHT 4-TRACK?

With boards such as the Mackie 1202 and Spirit Folio at such affordable prices, I am amazed that TASCAM and Fostex have yet to offer 4-track cassette transports to complement these mixers. I have seen the budget 4-track devices, and it would seem that these are lacking in features or quality in the mixing section only. "No headroom" is the most common complaint I hear. The heads are fine, right TASCAM and Fostex?

It would be nearly revolutionary to develop a 4-track transport with balanced in/out jacks and the usual complement of controls. Take a cue from Mackie and make it tough, pro, and affordable. If you want to satisfy your corporate interests with big profits, add an option to make two or more of these operate in sync.

I have been considering this for years, and every musician and engineer to whom I have spoken says they would buy it, especially as a Mackie 1202 companion. With the option to upgrade by synching these units for more tracks, I believe this would be an incredible value.

Can anyone tell me why this has not been done? It seems almost a nobrainer. Is there a glaring technical problem I have failed to consider?

> Kevin Pa**e**z Freelance Mastering Engineer Dallas, TX

MEEK YOUR MAKER

Re: Joe Meek December, '95

I read this article because I had no idea who Joe Meek was until I got to page 112 when you casually mentioned that he was a double murderer. How could anyone put this heinous man on a pedestal. Quote "..so wherever Joe Meek currently resides he must be having a bit of a laugh." I'm a sinner, not the Judge, but I know that where he resides presently, there is no laughter. *Charlie Solak*

via America Online

[For those who called asking for more info on getting Meek's album, contact the record label, Razor & Tie, at P.O. Box 585, Cooper Station, NY 10276. Tel: 800-443-3555.]

WELL RED

As the designer of the MP16, I was

pleased to see Wade McGregor's review in the December 1995 issue of *EQ*. I would like to make a couple of things known to Wade and *EQ* readers so they might be able to put the technology in the MP16 to use.

The MP16 is built by BEC Technologies under a nonexclusive license agreement from me and my company, RED Designs. What does this mean to EQ readers? It means that other manufacturers who would like to use this remote control mic preamp architecture can certainly do so, by also licensing the design. RED Designs is also able to make custom remote control mic preamp solutions - so if Wade wants more splits, I can accommodate him. While RED Designs has been primarily a design consulting firm, we are looking to start manufacturing products targeted toward the project studio and modest touring applications.

Once again, thanks for the review, Wade. And if anyone would like to contact RED Designs to discuss manufacturing rights or custom mic preamp solutions they would like, they may contact us via e-mail: REDDE-SIGNS@aol.com.

> Rick Downs Dallas, TX

CREATIVE COMMENTS

I generally like the direction that your magazine is going. I would like to strongly urge, on the behalf of myself and several other people who feel the same way, that you offer more creative personalities, and be more in-depth with them. One of the most common criticisms that I hear from people who don't subscribe to your magazine is that there are not enough creative recording tips in your magazine.

Creative tips can include: mic placement, mixing, equipment use and abuse, great tips at getting more than the equipment was made for, how to get weird sounds, creative equipment set-up, differing philosophies of the whole recording process, and so on.

And getting back to the creative personalities: what about articles, *indepth* articles in which the person discusses equipment, creative ideas, and philosophies at length? Interview people who are creative and interesting — Daniel Lanois, Brian Eno, Rick Rubin, Flood, Pierre Marchand, Michael Brook, Peter Gabriel, Jimmy Page, Trevor Horn (slick, but damn good productions), Steve Albini, Butch Vig, and David M. Allen (The Cure). And what about all those artists who produce themselves that none of the recording publications seem to pay much attention to? Future Sound of London, The Orb, Portishead, Tricky, Us3, and one of the finest field recordings I've ever heard, *Sundanese Degung-Mojang Priangan* (Music of West Java) on Interra Records?

This is what we are interested in. The things you have in your magazine are interesting. New digital technology and sequencing and digital editing and outboard gear are all interesting. But how can we be even more creative with them? Details! We want details! Details and depth!

> Ken Lee Eleven Shadows (Hyperium Recording Artist) via America Online

MOIST AND DRY

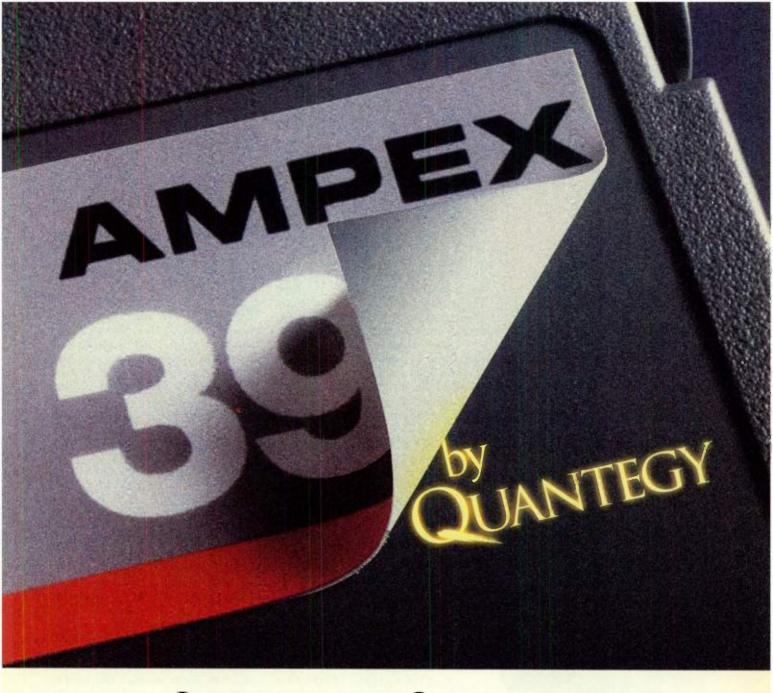
In regards to "Baking Tips," (EQ&A in the December '95 issue):

I have an excellent method of removing moisture from most anything without the use of heat. I have not had the occasion to use this method on tape as was the need in this case (restoring moisture-damaged, reel-to-reel tape), however, I see no reason why my method would not work. And if it doesn't work, nothing lost — you can still bake the tape afterwards. And the equipment required has countless uses.

What you need is a dehumidifier. The most common use for dehumidifiers is to dry out damp basements. However, put a dehumidifier in any room and close the doors and windows, and it will dry out anything and everything in the room. For quickest results use in a small room — a closet is perfect. The dehumidifier has a small heater in it

WRITE TO US

EQ Magazine Editorial Offices 939 Port Washington Blvd. Port Washington, NY 11050 Fax: 516-767-1745 E-mail: EQMagazine@aol.com



INTRODUCING QUANTEGY. THE NEW COMPANY THAT'S BEEN MAKING AMPEX TAPE FOR OVER 35 YEARS.

Nothing's changed, really.

You still get the audio mastering tapes that go gold more than all other brands combined.

The same top quality video tapes used by broadcast and creative professionals around the world.

The same market-leading instrumentation tapes used by aerospace and government.

And the same manufacturing, technical support and sales people. You even get the same Ampex¹¹ brand name.

The difference is that we're now the only media company dedicated exclusively to you, the recording professional.



44 1-734 302240

49-69-8007540

So call us today and we'll tell you more about Quantegy."

After 35 years, we're just getting started.

Quantegy Worldwide Sales Offices (201) 472-4100 I RUW YORK

M 5 Af unit (Mu hington D.C.) (31) 530-8880 Footbaart (Munhington D.C.) 774) 491-7112 Mident (Chican) 17001 591-5100

Seuto Cantral (D.I. an) (21 J) 620 90.3 (510) 881 7341 Southwent (Lins Anc) Fran & Month A rul (Parin) 33-1-4731-71 1 All trackmarks are the presently of their resp

UK & Ist and London) Italy & Iber I (Rome) Germany & Austri Cuntral Europe (London)

E netux (N (mrger)) 5 and ray a Citor shalm) 44-1-734 302208 ext 209 Canada (Toronta)

lue, Mourtain View, CA 94043 (415) 903-1100

Middle East/Africa (Londen) 44-1-734 .02208 ext 21 31-24 3730484 46-8 590 100

Latin America & Carlobian (Sur Francisco) Alla Trisici Kennel

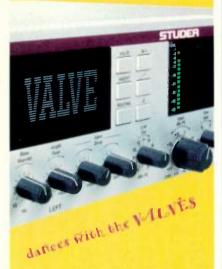
(415: 903: 11:2 152.77 18 m E1-2 ### (%C)

1/96

STUDER D19 Series Mic VALVE

Valve Dignified Digital Mic Preamp

 Top MIC and LINE Preamps - High performance modular 20-bit A/D converters - Switchable Valve Dignifier stage - 3 adjustable valve sound parameters: Bass Warmth, Angel Zoom & Valve Drive
 Highest Audio Quality & Individual Valve Sound Tuning - DSP Dithering and Noise Shaping for best quality on 16-bit media - Analogue and AES/EBU outputs - balanced pre/post valve inserts
 Optional ADAT™ or TDIF™ 8-CH outputs with output routing - A harmonious combination of traditional preamplifier circuitry, valve technology and state-of-the-art digital techniques !



for more information call Switzerland (worldwide) + 41 | 870 /5 |1, Austra + 43 | 866 54-0, France + 33 | 45 | 47 86, Germany + 49 30 72 39 34-0, UK + 44 | 707 66 50 00, Canada + 1 416 510 13 47, Japan + 81 3 24 65 22 |1, Singapore + 65 481 56 88, USA + 1 615 391 33 99



PROFESSIONAL AUDIO EQUIPMENT

A Harman International Company

CIRCLE 73 ON FREE INFO CARD

LETTERS TO EQ

to heat the room air before it is blown across a refrigerated coil. The difference in temperature causes the moisture in the air to condense on the coil and drip into a bucket which must be emptied when it becomes full. The amount of heat added to the room is small.

Dehumidifiers have a control on them that turns the dehumidifier off when a set humidity level has been reached, and back on when the humidity level rises back up above this level, much like a thermostat. I'm not sure of its technical name, but I suppose "humidistat" will work! For quickest drying, this control can be set to "constant," which means it is always on.

Most dehumidifiers will accommodate a hose or plumbing on the drain so that you can remove the bucket and not have to worry about manually draining it. Plumbed thus, a dehumidifier can be made a permanent installation in, for example, your studio control room to keep harmful, corrosive humidity under control - especially if you live in a high humidity area like 1 do here in S.E. Kansas, Adjust the "humidistat" so that the dehumidifier only comes on when the humidity is high, or your electric bill will be quite high! (Dehumidifiers are available for \$200 from JC Penny, Sears, etc., last I checked - it's been a while!)

Tom Rhodes Pittsburgh, KS

MR. IMPATIENT

I thought I'd drop you some mail and let you in on a few opinions 1 have about your magazine.

I have been a reader of EQ for the past year or so and have found only one major fault with it (and with many other mags too): I'm ready for the next issue three hours after purchasing the latest. The problem is I can't get the next issue for at least four or five more weeks.

I spend most of the time scanning the ads and reading reviews. I'm a budding musician and a confessed "bell and whistle" freak. I enjoy reading the Product Views and Room With A Vu sections.

I just read the December '95 "Across the Board" and have several questions for Roger Nichols:

1. How can any self-respecting musician be satisfied with just a onehour shopping spree at Sweetwater Sound? It would take me at least three hours just to load the stuff, plus another hour for the junk.

2. As long as we are being greedy, why not go for a pair of \$8K Genelec monitors?

3. How old must a piece of gear be before it is considered "vintage"?

4. How old must an engineer be before he or she is considered "vintage"?

Anyway, thanks for producing your periodical, even if I can't get the next issue in three hours.

> David L. Agne Jr. Silverdale, WA

FUN AND PROFFITT

I've read Paul Dean's letter in the December EO, and I'm sill trying to figure out whether some vital information was left out or if he simply doesn't understand Pro Tools. His idea that, since he has one internal hard drive and two external drives, he has to use up an authorization for each of them is clearly misguided. We have several drives, and we usually have at least ten projects in progress on the system. How many authorizations do we use? One, on the internal drive. If we want to take a project with us, we just grab the disk that contains the sessions and go. The problem resides on the internal drive, along with its authorization.

Now, assume I go work on someone else's system, and they are running a previous version. Then, I take my disks with me, install the program on the other, and *uninstall* it when I leave. I own the upgrade and the right to use it on a system. What I do not own is the license to take three authorizations and run the program simultaneously on various machines. That is, if I buy a used Pro Tools system, I also need to transfer the license, unless I'm going to be traveling with the second system. In that case, I still have three authorizations for one system at a time.

There's no doubt that Digidesign is a growing company and often suffers from the downtime due to training new tech support that growing companies endure. I've had to wait on the support lines like everybody else. However, I haven't yet had to cancel a session due to a lost authorization. I've been running the system for several years, and maybe I've just been lucky, but I've always been able to get a replacement disk if the unthinkable happened and I was running on my last authorization.

> K.K. Proffitt The Proffitt Center Hendersonville, TN



for your ears

You will love the Wizard M2000. This digital multieffects processor is specifically designed for the artist within you. Based on the unequaled DARC[™] chip, the two independent engines deliver uncom-

promising effects, meeting the high performance demands of your ears. The clarity, density and feel is beyond anything you have ever experienced before. It will lift your music to the highest quality level.

TC Electronic Inc., 705-A Lakefield Road, Westlake

We realize that you are busy creating music. TC's engineers are artists themselves, and have therefore gone to great lengths to make the M2000 intuitive and easy to operate. For example, with the 'Wizard' function you can find the best presets in any given situation; and all parameters are maximum one menu level away - no more searching through multiple menu levels.



M2000 - Combine your artistic skills with science and create magic.

Please also refer to the brochure included with this issue of EQ. For information about your nearest TC dealer, please call 1-800-798-4546.

Wizard M2000

Art + Science = Magic





DROP OUT

QMy ADAT will at different times drop out of record while synched to a SMPTE signal. I have checked my SMPTE signal to make sure it is not dropping out. Also, I have used a "SMPTE Jammer," which, when engaged, just keeps on generating SMPTE. Even so, my ADAT will still drop out of record in one or more places and usually in the same places. If I rewind back passed the dropout place, I can punch in and, in most cases, record past these dropout places. Only on occasion can I then not stay in record. Do you have any ideas that can help me?

> Len Weisman Audio Recording Studios Los Angeles, Ca.

Several things can cause an ADAT Ato pop out of record mode while being synchronized to external recorders using SMPTE timecode and an Alesis BRC, Alesis AI-2, or JLCooper Data Master (or when using the Fostex RD-8, which has a built-in SMPTE synchronizer). In all digital recording, time is of the essence. Bits have to wind up in exactly the right spot on the tape. When a digital recorder, like an ADAT, is recording while synchronizing to a nondigital source, its sampling rate must be derived from the incoming timecode. When things don't match up, the system is designed to drop out of record to avoid damaging data or recording nonsynchronously. The different SMPTE synchronizers for the ADAT system each have different features that may affect your problem. For the sake of this answer, I'll assume you're using a BRC (Alesis Big Remote Control).

1. Wow and Flutter in Timecode: Analog audio and video recorders are much less stable from a timing standpoint than digital recorders; they're plagued with small speed variations known as wow and flutter. Many VCRs are particularly bad in this regard; if SMPTE is recorded on a longitudinal audio track, the wow and flutter can be considerable, even if there are no dropouts and the level is good. If the BRC's "Clock Source" menu is set to "SMPTE," it must continuously adjust the sampling rate of the ADAT to match the timecode. If the timecode has too much flutter, the ADATs will drop out of record and you'll momentarily see "Er 8" on the ADAT's display.

The cure for this is to run a stable video source (preferably from a black burst generator, but you can try using a composite video output of the VCR itself) to the Video Sync Input of the BRC, and set Clock Source to "Video." You might try recording future video tapes with VITC (timecode recorded in the vertical interval of the picture itself), which is more stable, but this requires a VITC reader to extract the timecode to send to the BRC.

If you're trying to synchronize to an analog multitrack, try to reduce its wow and flutter by cleaning its capstan, buying a new pinch roller, and having it checked out by an experienced technician. Ideally, you should buy a synchronizer for your analog recorder, and let it be the slave to the ADAT master, which has no appreciable wow and flutter.

2. Lock & Release Mode: If you are already using Video as your clock source, nongenlocked timecode could be the problem. Many people make the mistake of recording SMPTE onto a video tape "wild," that is, without simultaneously running a video signal to the timecode generator. As a result, the timecode and the video frames will slowly fall out of alignment and the BRC will pull the ADATs out of record. The same will be true if you set the BRC's clock source to "Internal" while the locate reference is set to SMPTE.

A software feature of the BRC is designed to make it less sensitive to this drift. In the External Sync menu, change Ext Sync Mode from "Frame Lock" to "Lock & Release." Frame Lock will stop recording if the clock source is more than one frame away from the SMPTE reference; Lock & Release allows the two references to drift up to five frames apart before it will stop recording.

3. Tape Damage or Dirty Heads: If you see the ADAT's Advanced Information Indicator (the period to the right of the seconds display in the ADAT, or the "sunburst" on the ADAT- XT screen) flashing, the ADAT is having trouble reading its own tape. If the tape is at fault, simply making a digital backup to a new tape will cure the problem. If the ADAT's heads are dirty, have them cleaned. Often, the simple act of rewinding with the heads engaged will clean the heads enough to get through a session, which may account for why your ADAT didn't stop recording on the second try.

There are some other possibilities you might explore, such as crosstalk in your 9-pin sync cable (common when people use computer cables with ADAT instead of the supplied Alesis cable), unusually high or low SMPTE timecode levels (remember the BRC can supply either +4 or -10 outputs), or a balanced timecode source that has problems when connected to an unbalanced input like the BRC. In any case, these symptoms are easily cured with a little detective work.

Dan Tinen Technical Communications Manager Alesis

RIP-OFF OR NOT?

QI need some quick advice on my TASCAM 38 (yeah! for analog). Channel 8 on record and playback disappears from time to time. I brought it into a local repair place and they said it was a problem with intermittent relay connections and would need some board connection work costing \$250-300. My questions are:

1) Does this sound right? Are board connections something that wear out? They said that other connections (the other 7 channels) will go out and should be fixed.

> 2) Does this price sound right? Author Unknown

A The TASCAM 38 suffers from several intermittent problems and several service tech stumpers like, "How do I get to that !@#\$!\$#@ thing?"

Yes, the problem can be relays; it can also be cold solder joints on the motherboard. I owned a 38 for years and serviced a few dozen machines over the course of seven years. Assuming the problem is not massive amounts of dirt on the heads and assuming your external wiring isn't suspect, the problem must be narrowed down to:

The Alesis BRC[™] Master Remote Control...

the ultimate power tool for the ADAT[®] Digital Recording System.



- Control up to 16 ADATs (128 ADAT tracks) from one location
 Synchronize ADAT to SMPTE time code, or MIDI systems to ADAT using SMPTE, MMC, MTC or MIDI Clock
- Create digitally-assembled productions using bars/beats or second/frame/sample display references
- Store 460 locate points, 420 with user-definable names
- Copy and paste digital audio between ADATs with single-sample accuracy
- Auto Punch In/Out, Pre/Post-Roll, Rehearse Mode, Track Delay, Digital Crossfading, Auto Loops, Session Data Backup and more
- Completely compatible with the new Alesis ADAT-XT

For more information about the BRC, see your Authorized Alesis Dealer or call 310-841-2272. @ Alesis and ADAT are registered trademarks, BRC and ADATXT are trademarks of Alesis Corporation. Alesis Corporation 3630 Holdrege Avenue Los Angeles CA 90016 310-841-2272 alecorp@alesis1.usa.com



ULTIVATE

AUDIO Recording, Duplicating & Packaging Supplies



- Signal Processors
- Studio Monitors and Amplifiers
- Synthesizers and Samplers
 - Drum Machines
 - PA Systems
 - Microphones
 - Computer Software and Hardware
 - Tape and Cabling All major brands from A-Z

250 Old Towne Road Birmingham, AL 35216 205-979-1811 Fax 72662.135 @ Compuserve.Com





a. The output reed relay (this looks like an IC) on the I/O card. I always replace these with a piece of wire because they only "mute" the turn on/off thumps.

b. The record/playback relay. TASCAM used nonsealed relays. I always replaced all of these with the sealed OMRON type.

c. The motherboard. This is a real mother to do because the only way to solve the problem is to remove the momma and redo every connection. Remember, if you have one problem, you will have others.

The best way to troubleshoot the problem is to:

Remove the bottom metal panel.

2. Remove the back plastic cover.

3. Using either "Y" cables or system wiring, route an oscillator to all eight tracks.

4. Gently turn the machine upside down on the bench so that you can easily access the I/O cards.

5. Record tone on all eight tracks for 15 minutes.

6. Confirm that all tracks actually recorded in both sync and repro modes.

7. Tap on the cards while monitoring both the meters and the outputs in both repro and sync modes.

8. Place the machine in input and again tap on the cards.

Important: Be sure to monitor each track while listening to each track. Do not swap cards because doing so will flex the momma board and temporarily "fix" the problem.

Sometimes the meters will cut out but not the audio. This is because the meter signal goes off the I/O board onto the motherboard and back. This is easily repaired by each card.

To extract the cards, it is first necessary to remove the support/guard rail that goes from front to back. Once removed, take the screw that held the rear of the rail and put it into the rail from the other side. This will allow single screw removal without taking off the back plastic panel. Well, almost. It is not possible to access the last two cards (7 & 8) because of the back cover. I used a small power jig saw to cut away the plastic.

Pulling out the cards requires a little bit of side-to-side wiggling. Sometimes a pair of long-nose pliers helps, but be careful not to damage circuit traces near the edge. Replacing the cards requires that you use "the force;" that is, Don't use force! There are two dozen+ pins that must all line up. If they don't, you will bend or break the connector on the card. Use plenty of light and be very careful.

Yes, \$250 to \$300 bucks is fair — provided they do a thorough job.

Eddie Ciletti Manhattan Sound Technicians NYC, NY

ASK AKAI

Q I use my Akai GX-R60 cassette deck to record music for presentation videos. The unit stopped working and I'm having a hard time getting it repaired. Most repairmen won't look at it and others tell me they have no way of getting parts. Can you supply me with either Akai's address and phone number or a service center that could handle the job?

> Lance F. Karp via America Online

Akai has a slew of repair centers in Anearly all the States and beyond. However, you may be having trouble because you are trying to reach Akai directly. You need to contact their distributor to receive information on service centers in your region. International Music Corporation (IMC), located at 1316 East Lancaster, Fort Worth, TX., tel.: 817-336-5114, is the distributor of Akai gear. That address is also the site of the Akai National Service Center. Take note that some repair sites don't handle tape equipment, so specify your equipment model number and what you need when you call the main service center for information.

Hector G. La Torre Executive Director EQ Magazine

Send your queries to: EQ Editorial Offices, 939 Port Washington Blvd., Port Washington, NY 11050 Fax: 516-767-1745 E-mail: EQMagazine@AOL.COM

The Best Sounding Amp Is Now The Most Powerful PowerLight

The new PowerLight 4.0 introduces

amplifiers that have already become

PowerWave™ Switching Technology

(patent pending), at the heart of the

PowerLights, is the power supply

technology for the future. The

4.0 clearly demonstrates why ...

The 4.0 achieves remarkable audio

to the output section is powerful

and stiff. There are virtually no sags

que ty because voltage supply

following transients or long peak

power demands. You won't hear

typical modulation from AC ripple

or garbling during clipping. The bass

is ultra-clean and fat. Highs are pure

A PowerWave transformer "flies"

at 115 kHz (2000 times the frequency

of a conventional transformer). Waste

heat from resistance is almost nil. The

supply is 90% efficient! And the 4.0

Class H linear output circuitry for

maximum audio quality and efficiency.

features the latest three-step

MAXIMUN EFFICIENCY

and unstressed.

SUPERIOR AUDIO

a benchmark for excellent audio

massive power to a series of

performance and reliability.

PowerWave "" Switching Technology allows for the high-speed transformer to be small and dramatically light weight.

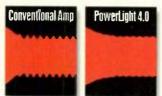
INSIDE THE PL4.0

> The charged capacitor bank distributes current as needed before the transformer—delivering continuous, stiff supply voltage.

> > A temperaturecontrolled, continuously variable speed fan blows heated air rear to front, keeping the amp and the rack cool.

NO-FAIL PERFORMANCE

QSC's renowned protection circuitry is enhanced by the PL 4.0's "smart" supply which matches power delivery to changing musical demand. The 4.0's thermal management keeps the amp cool under extreme conditions (even 2 ohm loads). And with the 4.0, *EMI and RFI are absent*, making the amp safe for wireless mics.



Less power sag, less clipping intermodulation, more bass impact, cleaner transients...PowerLight.

INCOMPARABLE AMP

You won't see or hear another amp with the power and performance of the PowerLight 4.0. *Call your QSC Dealer or QSC direct at* 714-754-6175 for complete details.

> The front panel of the PL 4.0 features comprehensive LED status arrays, detented gain controls with soft touch knobs and 2 dB steps for easy resetting

Made in USA.

An aluminum heat-sink tunnel designed with the maximum possible surface area transfers and dissipates heat.

➡ 4 PowerLight Models

BHD IL	8.8*	1 42.	2 2 **
PowerLight 1.0	200 Watts	325 Watte	500 Watts
PowerLight 1,4	300 Watts	500 Watts	700 Watts
PowerLight 1.8	400 Watts	650 Watts	900 Watts
PowerLight 4.0	900 Watts	1400 Watts	2000 Watts

PowerLight 4.0 Key Features

- FowerWave''' Switching Technology
- Advanced Thermal Management System
- > DC, Sub Audio and Thermal Overload Protection
- Data Port for Use With QSC MultiSignal Processors
- Neutrik Combo and "Tool-less" Binding Post Connectors.
- Extreme Light Weight—30 lbs!
- ▶ 6 Year Warranty Available







© Copyright 1995 QSC Audio Products, Inc. "Power/Light," "Power/Vave," and "Hear the Power of Technology" are trademarks of QSC Audio Products, Inc. "QSC" and the QSC logic are registered with the U.S. Patent and Trademark Office. QSC Audio Products, Inc. 1675 MacArthur Bivd, Costa Mesa, CA 92626-1468 Ph (714)754-6175 Fax (714)754-6174 E-mail: tiffany_walker@qs:zaudic.com QSC OnLine BBS: (800) 856-6003 or (714) 668-7567 (In Canada) S.F. Marketing 6161, Cypihol SL Laurent, Quebec, Canada H4S 1R3 Ph (514) 856-1919 Fax (514) 856-1920

CIRCLE 54 ON FREE INFO CARD

the hottest digital recorder for post production.

For starters, virtually every production studio has at least one. If your business is post-production audio for dialog, sound effects and music nothing is faster, nothing is more reliable and nothing sounds better than the TASCAM DA-88. Fact is, when deadlines are looming, the last thing you want is a machine that spends more time in the shop than in the studio. Plus you need a machine that synchronizes and locks up fast — typically 2 seconds or less. What's more, only the DA-88 can record an entire feature on a single tape. That's why the TASCAM DA-88 is the choice of production professionals. And that's how it became the undisputed industry standard.



POWER

VAR SPEED

DIGITAL IN

FENOTE

VASCAM DA-88

a DA-88 in your

Because without one your studio could be wasting time and losing money. The DA-88 seamlessly integrates into any production studio to directly interface with SMPTE, MIDI and Sony 9-pin equipment — TASCAM's expert software makes it possible. Its blazingly fast transport means you spend less time waiting and more time producing. Plus, a truly modular design makes expanding and enhancing a DA-88 with third-party products quick and easy and makes it simple to service.

you need to get production studio.

STOP

Q

INSERT RIKSL AUTO

) F FWD

AUTO

ALL

INPUT

CASSETTE IN ----- EJECT

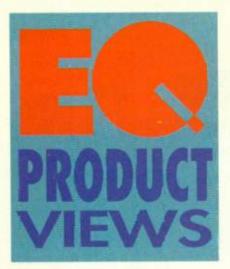
RECORD

Fact — the DA-88 delivers what production professionals demand — economy, reliability, speed and synchronization that really works. That's why the DA-88 is the only modular digital multitrack to win the coveted *Emmy Award*. So, whether you already own a DA-88, or are waiting to buy your first, why wait another minute?





 1996 TEA: America, Inc. 7733 Telegruph Road: Montrubelin, CA 99640 (213) 726-030 TEAC CANAGA LTD: 340 Brunel Rd: Minissunge, Ontimo L422C2 Canada (965) 890 996 World Radio History: CIRCLE 76 ON FREE INFO CARD



SET YOUR SIGHTS ON THIS

echnical Laser Company has created Speaker Sights to address speaker alignment problems. Speaker Sights has separate switching and LED indicators for both left and right lasers, and is powered by a battery pack no larger than the palm of your hand. An



optional A/C power adapter is also available. By positioning the Sight over the driver and activating the system, two bright red dots produced by the lasers can be shown on a flat surface placed at the mixing position. An optional sighting grid is available if needed. For more information, contact Technical Laser Company, 25242 G Steinbeck Ave., Stevenson Ranch, CA, 91381. Tel: 805-255-5500. Circle EQ free lit. #102.

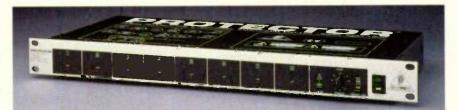
EASY AS ABD

udio By Design (ABD) currently uses Focal, Audax, Morel, JBL, and Klipsch professional drivers. All speakers use conventional cone bass, midbass and midrange drivers, and silk dome tweeters to reduce sibilance, nasality and droning. All models use the exact same drivers to help maintain consistency in sound integrity regardless.For more details, contact ABD. 139 East 9th Street, Huntington Station, NY 11746-1846. Tel: 516-271-3348. Circle EQ free lit. #104.



YEAR OF THE AARDVARK

ardvark has created the AardScape, a unit that utilizes new True Analog[™] technology to endow digital recordings with the sound and feel of analog tape. The True Analog process simulates the warm sound of recording directly to analog tape without the tape hiss. The AardScape gives you control over saturation (soft, medium, or hard) and brilliance (full, clean, bright) and overall warmth. The unit supports +4 dBu I/O on balanced, XLR connectors and -10 dBv I/O on unbalanced, 1/4-inch connectors. For further details, contact Aardvark, 202 E. Washington, Ste. 306, Ann Arbor, MI 48104. Tel: 313-665-8899. Circle EQ free lit. #101.



DO YOU HAVE PROTECTION?

ehringer has introduced the Protector MDX 1800, designed to protect recordings from distortion, expand headroom and dynamic range, "energize" the sound of mixes, and optimize the resolution of digital equipment. The Protector is an 8-channel program/peak limiter that works with any analog or digital multitrack recorder, and protects digital inputs from overloading. The Protector features Behringer's dual-action IGC (Interactive Gain Control) circuitry, LED "limit" indicators on all channels, and operating level switchable from –10 dBv to +4 dBu. The Protector is also equipped with servo-balanced inputs. To find out more about the Protector, contact Samson Technologies Corp., P.O. Box 9031, Syosset, NY 11791-9031. Tel: 516-364-2244. Circle EQ free lit. #103.



YOU BETA YOU BETA YOU BET ith the introduction of two new models, the Shure line of Beta microphones has grown to a point where five distinct offerings can be used to fulfill the needs of an entire live performance. The Beta 52 is a dynamic mic designed to deliver the low-end punch required for kick drums and other bass instruments. The Beta 52 incorporates Shure's Dynamic Locking adjustment mounting assembly and a hardened grille. Shure's new Beta 56 utilizes a consistent supercardiod pattern to supply a high degree of gain-before-feedback and is centered around a frequency response suited for drums and instruments. Other members of Shure's Beta family include the Beta 57A, the redesigned Beta 58A which now offers a smoother response and extended high frequency performance, and the Beta 87. For more information on the entire Shure family of microphones, contact Shure Brothers Incorporated, 222 Hartrey Avenue, Evanston, IL 60202-3696. Tel: 800-25-SHURE. Circle EQ free lit. #105.



	***********				1 .E 3-	
77.79	4. 4. 4.		3. S. A.	dia at	Ne C	17, 210 mm
	sate sate so		1. A	245 24	a sala a	
	The statist	7		Sta Ist	1.51/2 J	5.4
0						1 . S X
		i				

HAPPINESS IS A WARM TRACK

Tawmer has released the 1962 digital vacuum preamp that combines Drawmer's vacuum tube technology with two low-noise preamps and two 24-bit A/D converters. The 1962 incorporates matched pairs of preamps and D/A converters, while a switchable "zero overshoot" transparent limiter enables the full dynamic range to be utilized. The 1962 can be supplied as an analog-only device because its digital hardware is modular and can be upgraded at a later time. Other features include: variable high- and low-pass filters, fine tune equalization, dynamic enhancement, variable tube drive, selectable digital resolution (16, 18, 20, and 24 bits), provisions for mixing up to eight channels, and AES/EBU, S/PDIF, and TDIF outputs. For more details, contact QMI, 7 October Hill Road, Holliston, MA 01746. Tel: 508-429-6881. Circle EQ free lit. #106.

CUT THROUGH

SS recently released the DPR-901 II Dynamic Equalizer. The unit offers a new "Split Band Switch" feature capability, which allows the DPR-901 to be used as a single 4-band device or as two separate 2-band devices, permitting one section to cover the low- and lowmid ranges, and the second to cover the high-mid and high ranges. Also new on the DPR-901 II is the "Side Chain Monitor" feature, which allows the engineer to tune in to the exact frequency area that he or she wants. For more info., contact Harman Pro North America, 8500 Balboa Boulevard. Northridge, CA 91329. Tel: 818-830-8278. Circle EO free lit. #107.

TUBE TECH

igiTech's new VTP-1 incorporates a vacuum tube mic preamp, tube line amp/DI, EQ section, and an 18-bit A/D converter in a 2U chassis. Also included is



12AX7 vacuum tube input circuitry to add warmth to digitally recorded vocal tracks. Each preamp control section offers front-panel switches for mic/line input select, phase invert, 20 dB pad and 48-volt phantom power. The VTP-1 also functions as a tube DI box, allowing synthesizers, guitars or bass to be run directly into a recording device. The unit's EQ section consists of four bands (two fixed and two sweepable) with each band offering up to 15 dB of boost/cut. The on-board 18-bit A/D converter allows program material to be output directly to digital in either AES/EBU or S/PDIF formats. The sampling rate is switch selectable between 44.1 or 48 kHz, while digital outputs may be used simultaneously if desired. For further information, contact DigiTech, 8760 South Sandy Parkway, Sandy, UT 84070. Tel: 801-566-8800. Circle EQ free lit. #108.



MIX WITH MACKIE

ackie recently introduced the MSI402-VLZ compact mixer. The MSI402-VLZ features six lownoise/high-headroom mic preamps, six balanced/unbalanced line inputs, four stereo inputs, 3band EQ (12 kHz, 2.5 kHz, 80 Hz),



and PFL/AFL (Solo-In-Place) on every channel (globally switchable). The compact mixer also features Alt 3-4 for an extra stereo bus on each channel, a low-cut filter on channels 1–6, balanced XLR main outputs, 60 dB mic gain on channels 1–6, a very low impedance (VLZ) architecture, EFX to monitor switch, and a global Aux 1 pre/post switch. For more information, contact Mackie, 16220 Wood-Red Road, Woodinville, WA 98072. Tel: 800-898-3211. Circle EQ free lit. #109

ADD SOME VERB

eavey Electronics recently introduced the AddverbTM III multi-effects processor, which can be used in band PA systems, fixed PA systems, home and professional studios, and in musical instrument effects racks. The Addverb III's user-friendly interface is designed to simplify the process of programming patches. Features include: true stereo inputs/outputs,



Unmatched

Sound

fully user-definable algorithms, two independent chorus effects, two independent pitch shifters, a total of 35 different effect types, 128 user-definable patches, 128 factory programs, and eight different special ultra-reverb effects. The Addverb III also boasts 24-bit stereo processing, a 12 Hz–20 kHz bandwidth, and dual data knob programming. Suggested retail price is \$349.99. For more details, contact Peavey Electronics Corporation, 711 A Street, Meridian, MS 39301. Tel: 601-483-5365. Circle EQ free lit. #110.

Useful

Features

Used to be, you had to spend a lot of money to get great sound and features such as userdefinable algorithms, ultra-reverb effects and 24-bit processing in a multi-effects processor. Not anymore! The Addverb IIITM comes with those features and much more. Read on.



The MONITOR Magazine is a publication filled with the latest information musicians want to know. To receive 4 issues for only \$5 (price good in U. S. only) send check or money order to Monitor Magazine, Peavey Electronics, 711 A Street, Meridian, MS 39301



SAFE POWER

ulizzi Engineering, Inc. has introduced the Z-LINE® TPC 12/MTD power distribution system with up to 12 automatically sequenced time delays on power up. Systems can power up in a predetermined sequence to cut high in-rush current, and prevent momentary power sags from equipment, such as amplifiers, drives, and motors. Along with the patented Multiple Time Delay™ feature, the TPC 12/MTD also provides EMI/RFI high-frequency noise filtering, spike/surge suppression, and emergency power off. For power output, you have a choice of either twelve NEMA-style receptacles, or twelve three-pin AMP-style connectors for Form C switching. The TPC 12/MTD is available in 120 V or 240 V, and 15 amps, 20 amps, or 30 amps. For more details, contact Pulizzi Engineering Inc., 3260 S. Susan Street, Santa Ana, CA 92704-6865. Tel: 714-540-4229. Circle EQ free lit. #111.

LET'S ROK!

RK has introduced the RoK•Bottom subwoofer that utilizes the same enclosure shape as the K•RoK. The 2.1 cubic foot enclosure features a 12-inch polyglass longstroke woofer and is powered via a proprietary 250-watt K•Amp amplifier. The 8-ohm RoK•Bottom offers a maximum SPL of 110 dB and a sensitivity of 91 dB. For more information, contact KRK Systems, Inc., Group One Ltd., 80 Sea Lane, Farmingdale, NY 11735. Tel: 516-249-1399. Circle EQ free lit. #112.



- Stereo inputs and outputs
- Fully user-definable algorithms
- Automatic parallel and series effects linking
- Two independent chorus effects
- Two independent pitch shifters
- 35 different effect types
- Up to 8 simultaneous effects
- 128 user-definable patches, 128 factory programs
- Special Ultra Reverb effects
- 24-bit digital stereo processing
- Full MIDI implementation
- 8 parameters can be continuously controlled per patch
- LCD display (no hard-to-use LED display)
- 🧖 Intuitive programming utilizing dual data knobs



TECHNOLOGY LEADER



(601) 483-5365 • Fax (601) 486-1278 • http://www.peavey.com • AOL Keyword: Peavey • © 1995 Check out the new Addverb III at your Peavey dealer today! * U.S.A. Suggested Retail Price

CIRCLE 56 ON FREE INFO CARD

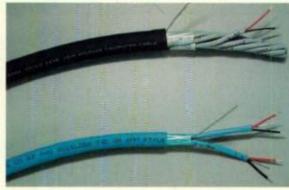


BOO!

Soundcraft's Ghost console offers a four-band EQ with two fully parametric mids, ten aux busses (two for stereo), a MIDI muting system, timecode synchronization, machine transport control, and four MIDI data faders. The Soundcraft Ghost console allows for control of digital multitrack tape recorders via MIDI machine control and video recorders via the Sony 9-pin professional standard. The Ghost also features a new low noise mic amp with phantom power and phase switch on every channel, four stereo returns, a MIDI-controlled snapshot-based mute recall system, timecode reader/generator, and LED meterbridge and 24-channel expander options. For more details, contact Harman Pro North America, 8500 Balboa Boulevard, Northridge, CA 91329. Tel: 818-830-8278. Circle EQ free lit. #113.

QUEST FOR TUBES

hirlwind has introduced the PM line of Precision Manufactured power and preamp tubes for music applications. Whirlwind's PM series tubes are custom designed, manufactured, and matched specifically for music applications. PM preamp tubes are designed to minimize microphonics and noise over the life of the tube and are selected to provide classic European and American overdrive characteristics. The Whirlwind PM tube range includes both precision manufactured classic and proprietary advanced designs. For more information, contact Whirlwind, 99 Ling Road, Rochester, NY 14612. Tel: 716-663-8820. Circle EQ free lit. #114.



WIRE WRAP

epco International, Inc. has created GEP-FLEX, a new flexible outer jacket compound for the 618 (22 gauge) and 724 (24 gauge) series multi-pair audio cable. Designed for both indoor and remote use, GEP-FLEX has an increased temperature range with cold weather

properties to minus 60 degrees centigrade. With its aesthetic matte finish, GEP-FLEX provides customers with increased flexibility, while still retaining the UL listed-type CM rating required for permanent installation. GEPCO will stock GEP-FLEX multi-pair cable in bulk lengths and will cut to customer length specifications. GEPCO will also terminate GEP-FLEX cable to connectors, patchbays, etc., as requested. For more details, contact GEPCO International, Inc., 2225 W. Hubbard, Chicago, IL 60612-1613. Tel: 312-733-9555. Circle EQ free lit. #115.



EIGHT GATE

reSonus recently unleashed its third product onto the market, the ACP-8. The ACP-8 is an eight channel compressor/ gate, housed in a steel, two-rackspace case that features eight sepa-



rate compressor/gates. Each compressor features control of threshold, ratio (1:1–20:1), attack, release, and gain. Each gate features control of threshold, release, and variable attenuation range. The ACP-8 accepts balanced or unbalanced inputs and balanced or unbalanced outputs. Each channel features a sidechain jack for spectral processing and a separate jack for gate keying, as well as full gain reduction metering and compression and gate threshold indication. For more information, contact PreSonus Audio Electronics, P.O. Box 84008, Baton Rouge, LA, 70884. Tel: 504-344-7887. Circle EQ free lit. #116.

"WOW, THIS BOX DOES A LOT OF AMAZING @#\$%."



266 Compressor Limiter

- Add tons of sustain to your screamin' guilar solos
- Punch up your bass rig both on the gig and in the studio
- Gate out unwanted noise and fix leaky drum tracks
- Tighten up kick drum tracks
- Even out background tracks
 - Tighten up that rhythm guitar tone
 - Protect your PA system from unwanted transients

dbx PROJECT and base rugs around dbx ugnal processors are the standard by which all others are nudged. Now you can get dbx quality for a price you can afford

I westerday

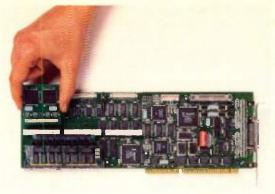
Whether in major touring systems, recording

For your guitar rig, vocals, bass rig, recording or live the 266 really does some amazing @#\$ So head on down to your local dox dealer and see and hear the 266 compressor/limiter today!

H A Harman International Company

dbx is a registered trademark of the Harman International email: customer@dbxpro.com • dbx 8760 South Sandy Parkway. Sandy Utah USA 84070 • (801) 568-7660 Fax (801) 568-7662





HAVE A V8

igital Audio Labs announced the release of its V8 DAW system for the PC. This modular hardware platform features a main board with an upgradeable DSP architecture and a wide variety of input/output options, including interfaces for ADAT and DA-88 machines. In its base configuration, the V8

is capable of recording and playing up to 16 discrete tracks. Application software for the V8 system will be available from third party audio, MIDI, and DSP plug-in developers. For more information, contact Digital Audio Labs, Inc., 13705 26th Avenue North, Suite 102, Plymouth, MN 55441. Tel: 612-559-9098. Circle EQ free lit. #117.



WHO'S THE (MIDI)MAN?

IDIMAN's Macman PT is fully GEO port compatible and works with any older 8-pin serial port Mac. Macman PT is self powered and also has a built in power jack for use with an optional 9-volt wall wart - perfect if you have a PowerBook 500. When the Macman PT is not functioning as a Mac MIDI interface it acts as a 1 In/2 Out MIDI Thru box so you don't have to replug the system when the computer is off. Macman PT is guaranteed for life and retails for \$59.95. For further information, contact MIDIMAN, 236 West Mountain St., Suite 108, Pasadena, CA 91103. Tel: 818-449-8838. Circle EQ free lit. #119.

SPECIAL FX

ower Technology combines proprietary DSP hardware and software in its DSP•FX. The technology is available either as a self-contained turnkey system called the DSP•FX PowerPack™, or as a cost-effective package of cards and software for any Windows-equipped PC, known simply as DSP•FX. Audio quality is main-



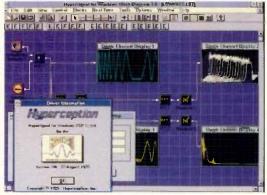
tained through Power Technology's 32-bit audio processing scheme, which utilizes floating point calculations. Visual Pro Audio's™ Display provides the on-screen centerpiece of each DSP•FX algorithm. Multicolored



icons offer real-time indications of the degree, type and spatial location of processing effects. The control screen sports familiar console-like knobs and faders. Parameters can be controlled in real time by a mouse, by any MIDI fader system, or by the DSP•FX Controller. For further details, contact Power Technology, 100 Northill Drive, Bldg. #24, Brisbane, CA 94005. Tel: 415-467-7886. Circle EQ free lit. #121.

WINNING TEAM

yperception, Inc. has integrated the Bridgenorth Signal Processing BN4000 DSP system board with Hypersignal for Windows Block Diagram 3.0. This combination of DSP hardware and visual programming software allows engineers to develop real-time signal processing functions on a floating point DSP processor using a graphical interface. Hardware features include dual access



SRAM, hardware address generation, a bi-directional data FIFO, and bi-directional control registers. For more details, contact Hyperception, 9550 Skillman, LB 125, Dallas, TX 75243. Tel: 214-343-8525. Circle EQ free lit. #118.

EASY AS QSC

SControl2 from QSC Audio Products is the control program for QSC's new Ethernet-based audio control network line. The control system provides the full programming capabilities for Microsoft Visual Basic. A complete control system is composed of a user interface system controller (a PC running QSControl2) for operation of the sound system, a network (Ethernet) for distribution of monitor and control data, and devices such as QSC's MultiSignal Processors that can be remotely controlled from the system controller. For more information, contact QSC Audio Products, Inc., 1675 MacArthur Blvd., Costa Mesa, CA 92626. Tel: 714-754-6175. Circle EQ free lit. #120.

meet your next rack. MR Rack.

64 Voices

1 Meg Wave ROM Expandable to 84 Meg (3 expansion card slots)

> 2nd -Generation TranswavesTM (with digital resynthesis)

> > 44.1 kHz Output

Four Outputs (four 18-bit DACs

All-new ESP-2 Effects (26 MIPS of 24-bit DSP power)

SoundFinder

GM **Compatible**

The MR-rack cuts no corners to give you great sound – with 44.1 kHz playback through hardware that measures in at an awesome 102 dB signal-tonoise level. (Four 18-bit DACs sure help with that.)

Our all-new waveform ROM has 12 meg of the best sounds you've ever heard, including bigger and better piano, strings, brass, and voices. Plus the second generation of our unique Transwaves[™], using digital resynthesis to give you sounds that move and transform under your control.

Great waves become great sounds thanks to MR-rack's all-new synth engine. Each sound is crafted with up to 16 sonic layers, creating the most realistic acoustic and the wildest synthetic sounds imaginable.

Our next-generation effects chip (*ESP-2*) provides 6 stereo busses for total mixing flexibility. Studioquality reverb and chorus are always available, plus your choice from a broad selection of insert effects. And our thoughtful design lets you change the insert effect while the MR-rack is playing, without affecting the chorus and reverb.

Forget you ever heard the term "MIDI mode". Working with the MR-rack is easy, because it's always ready to receive on 16 MIDI channels. And it's just as easy to set up splits and layers on a single channel.

We've made selecting sounds a breeze with our proven SoundFinder™ interface – one knob selects

totally new totally awesome

the sound type; the other knob picks the sound itself. And with hundreds of great sounds to pick from, you might never need to tweak one. But if you do, we made the most important sound parameters easily available for editing.

Want more? We thought so. The MR-rack has the best expansion options of any synth, with 3 card slots for up to 84 meg of wave data. And we give you tons of memory for your money. (Three times what anyone else offers!) Like our upcoming 8 meg Drums at only \$250, the 20 meg Dance/Hip-Hop card at \$425, and the 24 meg World and Piano cards at \$500 each.

Need another incentive? OK, buy an MR-rack and we'll give you a rebate coupon worth \$50 off your first expansion card. (*This is a limited time* offer good until we come to our senses!)

Whoops - ran out of room! So if you want to hear all the other totally cool things about the MR-rack (like the PCMCIA card slot, the 4 outputs, or the Unisyn MR editor we'll give you) just call us, write us, or visit your ENSONIQ dealer and say

World Radio History

"I'd like to meet MR Rack!"

You can reach us dt 800-257-1439 documents to your fat http://www.ensistig.com on the We GO MIENBONIQ on CompuServe S00-553-8151 phone 610-647-8905 fat

ENSONIQ

☐ Yes! I want to meet MR Rack! I'd also like info on:

Zip

s about TS Series synths ASR Series samplers KT Series weighted action synths KS-32 weighted action synth SQ Series synths DP Series parallel effects processors 1682-fx mixer

> Mail to: ENSONIQ Corp. Dept. Q-59, 155 Great Valley Plewy P.O. Box 3035, Malvern, PA 19355-2735

State

MS1402-VLZ

NE MIXEI

6 mono mic/line chs. 4 stereo line chs.

3-band equalization Low Cut Filter (chs.1-6) 2 aex sends per ch.

Mute/Alt 3-4 Bus

PFL-in place/AFL Solo

60mm Ing-taper faders Aux Send 1 master control

EFX Return to Aux Send 2 stereo aux returns loudness

Control Room/Phones matrix

Control Room output

Tape inputs/outputs

Balanced XLR & 1/4" outputs

Built-in power supply. No outlet-eating, hum-inducing wall warts!

RNEWMS



0

Studio grade mic

preamps (chs. 1-6) with high headroom, low noise (-129.5dBm E.I.N.) and phantom power. These are the same discrete mic preamps used to track and mix several gold & platinum albums on our

acclaimed 8.Bus console series. **Balanced** line

inputs. Channels. 1-6 are mono; channels. 7-10 are stereo. Low Cut Filter (chs.

1-6) cuts mic handling thumps, pops, room rumble and wind noise. Also lets you safely use Low Shelving EQ on vocals. Two aux sends per

channel with 15dB extra gain above Unity.

High shelf EQ. ±15dB at 12kHz.

Peaking midrange with wide, musical bandwidth centered at 2.5KHz. ±15dB.

Low shelf EQ. ±15dB at 80Hz.= Four buses on a 2-bus board! Mute switch routes channel output to extra ALT 3-4 stereo bus. Use it for feeding

multitrack recorder

channels, creating



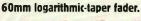
MACKIE.

1402-VLZ

a subgroup via Control Room/ Phones matrix (more info at right), monitoring a signal before bringing it

into the main mix or creating a "mix minus." Solo. AFL (for studio use) or

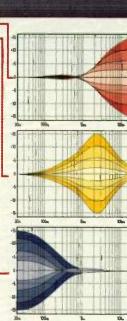
PFL (for live sound mixing) via global switching. Individual channel metering via Solo makes level setting easy.



Balanced XLR outputs with

switchable mic/line

output level.



CAUTION

outputs.

Control Room & Alt 3-4 Bus

Mackie's signature 3-way

Channel

(chs. 1-6).

Inserts

4

INTRODUCING FOR \$599*.



NEW MS1402-VLZ 14x2

More mixer in

less space. The

just 1.3

sq. ft. of

MS1402-VLZ takes up

work

space.

Balanced inputs & outputs (except RCA tape in/ out & ch. inserts). **Beefy head**phone amp can drive any headphones to shock volumes. 12-LED meter display with Level Set Indicator. Separate 60mm L/R **Main Mix** faders.

MAIN

ALT

TAP

ASSI IN

C IOC MIDINE

MARCAL / PHONES

48 10---

52

20 -

30 -

Control Room/Phones matrix



globally selects mono pre-fader (PFL) or stereo inplace (AFL).



60mm logtaper faders are accurate along their whole length of travel for smooth fades. They employ a

new long-wearing contact material for longer fader life & super resistance to dust, smoke, etc.



D/LINE MIXER NEW

NOW SHIPPING

THE NEW MICROSERIES 1402-VLZ. YOU CAN'T GET A BETTER COMPACT MIXER.

VLZ (Very Low Impedance) circuitry at critical signal path points. Developed for our acclaimed

8.Bus console series, VLZ effectively reduces thermal noise & minimizes crosstalk by raising current & decreasing resistance

his one really drove our **Engineering Department** nuts.





gazillion studio and live sound features into an extremely compact mixer. He wanted the same nice long 60mm

to pack



3-band EQ. AFL/PFL and deluxe tape monitor/Control Room features. In short, he wanted it to be the best under-\$600 mixer value ever offered. Many months and countless

faders we developed for

Diet Cokes[®] later, it's here. The MicroSeries 1402-VI7

We invite you to shop around and compare it to other mixers costing up to \$1000. When it comes to low noise. high headroom, impeccable mic preamps, sweetsounding EQ and sheer density of Cool Stuff, we think that you'll agree... Greg and his fanatical Engineering **Department** have done it again.





CALL TOLL FREE

FOR OUR NEW

40-PAGE

"IN YOUR FACE

BROCHURE &

APPLICATIONS

GUIDE

CIRCLE 50 ON FREE INFO CARD



Takes a stomping and keeps on mixing. Our unique

pressure-absorbing knob design, thick, horizontal fiberglass circuit boards and brass stand-offs resists downward impacts that would trash other mixer brands.



Why do we make such a big deal about our Low Cut Filter? First. it's a

real low cut filter with a sharp, 18dB per octave slope

beginning at 75Hz (some so-called "low cut" filters' shallow 6 or 12dB/oct. slope lops off audible bass). Our filter cuts stage

rumble, room noise, P-pops and microphone handling thumps - all of which muddy your

^{*}Su recet d retall price. Higher In Canada.

Solid steel chassis instead of aluminum or plastic. RCA-type tape inputs &

outputs. Balanced 1/4" TRS main

mix robs amp power and can

actually blow woofers. It has

benefit: without the filter, if

you try to add Low Shewing EQ

to vocal channels, low-frequency

garbage gets

boosted to

potentially

disastrous

levels. Engage

the Low Cut

filter and you

another equally valuable

can safely use Low EQ

to enhance

vocals.

outputs. Balanced stereo line inputs (chs. 7-14) are switchabie from +1dBu pro level to -10dBV semipro levels. Can be used to boost any weak input signal. Fast, accurate level

setting via Channel Solo. As the Rude Solo Light blazes forth, a soloed channel's level is displayed on the LED meters. Set the channel fader to Unity, adjust input level to the Level Set marker and you've optimized the MS1402-VLZ UnityPlus gain structure for maxiumum headroom & minimum noise.

EFX to Monitor switch. lets you send aux return signals (delay monitors through Aux Send 1.



16220 Wood-Red Road NE • Woodinynie • WA • 98072 • \$ 810/898 3211 • \$ 206/487-4337 mackie.com • Outside the US, > 206/487-4333 • Represented in Canada by S.F. Marketing \$ 800/363-8855

Master Pre/Post switch. for Aux Send 1. 20dB gain above

World Radio History



Two stereo aux returns with Unity



From Mangione T

STUDIO NAME: Barking Doctor LOCATION: Mount Kisco, NY

KEY CREW: Mick Guzauski; Marnie Riley PROJECTS RECORDED: Shawn Stockman of Boyz II Men, "Vision of the Sunset"; portions of Mariah Carey's televised Thanksgiving Special.

CONSOLE: Solid State Logic SL 4064 G Series with AT&T DISO Digital Mixer Core.

MONITORS: Tannoy SRM-10B 10-inch dual concentric and SGM-10B 10-inch dual concentric reference studio monitor with Mastering Labs crossover network; Magneplanar MG-IIIC's with JBL subwoofers; DynAudio.

AMPS: Hafler H500's; Yamaha P2700's. COMPUTERS & SOFTWARE: Macintosh Quadra 840 with Digidesign ProTools and Studio Vision.

RECORDERS: Sony PCM-3348 48-track digital, APR-24 24-track analog, and PCM-800's; Alesis ADAT.

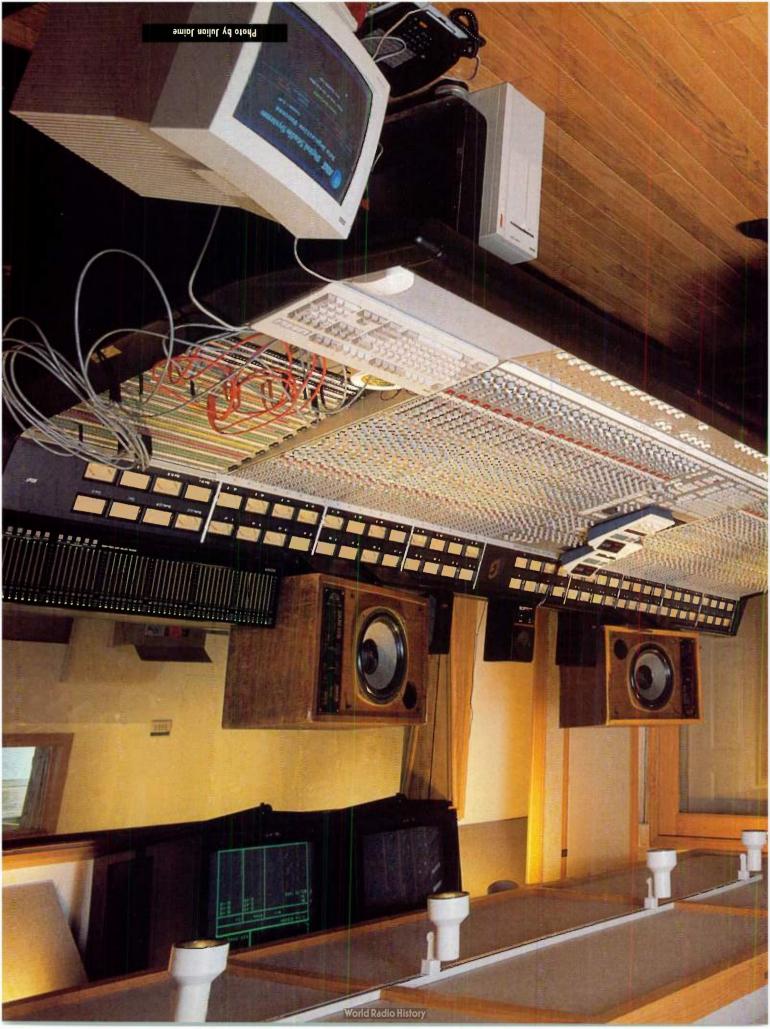
DAT MACHINES: Sony 7030 timecode DAT and 2700A; Panasonic SV-3700.

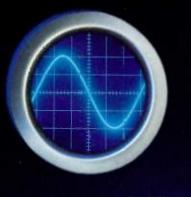
OUTBOARD GEAR: Lexicon Model 300; Eventide DSP4000, H-3000SE Ultra Harmonizer, and SP2016; Sony DPS-V77 multieffects processor; Yamaha SPX-990; Ensoniq DP/4; GML 8200 dual-channel 5-band EQ and Series 3 limiter/compressor; Lang EQ; dbx 900 rack; UREI LA-22's; Calrec preamps; Neve preamps; API preamps.

MICS: Sony C800G; AKG C24; Neumann TLM170 condenser; Milab DC96; Shure SM57's and SM58's.

STUDIO NOTES: Guzauski says: The important thing to teach about recording and I don't know if anyone does — is attitude. Not technology, not music, just attitude and how to deal with the people and the session. You really can't teach that. It's part of your personality and it's experience. I also think it's good to have a sense of humor about things and keep it light in the session.

EQUIPMENT NOTES: Guzauski continues: 1 like the Sony 48-track. It's extremely convenient, sounds great, and its sampling memory allows you to move stuff around and quickly locate to any part of the song. Also, you don't have to wait for several machines to sync up. For acoustic music, I like to use two analog 24-tracks with Dolby SR because of the incredibly large dynamic range and detail at low level. —Andrea Rotondo ENGINEER MICK GUZAUSKI SETS OUT HIS SHINGLE AT HIS MT. KISCO HOMESTEAD





AUDIO INTEGRITY

You are busy trying to create the perfect mix,

so we'll get right to the point. The new

Sony DPS-V77 is a single rack powerhouse -

a digital multi-effects

signal processor that

combines the best of

our DPS Series, for an impressive array of effects including reverb, delay, modulation, dynamic filtering and more. The V77 also offers balanced and unbalanced analog and digital l/Os, with high resolution 24-bit A to D, 20-bit D to A converters and Sony's proprietary 32-bit digital signal processing. Result: great sound in, better sound out.

DUAL EFFECTS ARCHITECTURE

With 50 effects per block, the dual block architec-

ture of the DPS-V77 makes it extremely flexible,

since it allows for various serial and parallel configu-

rations. Each block is equipped with a switchable

IT'S A SIGNA



SONY

POWER

AON GOF





IT'S A GARBA

pre or post effect EQ. You'll find 198 user presets in

addition to 198 factory presets. You'll also discover

several new, ear-opening effects, from intelligent

pitch shifting to irregular delays, to mono/stereo

conversion, and three-dimensional spatial place-

ment. The most important feature, however, may be

what this unit doesn't come with.

 $\ensuremath{\mathbb{C}}$ 1995 Sony Electronics Inc. All rights reserved. Reproduction in whole or

MORPHING

USER FRIENDLY

Nietzsche once said, 'get rid of the junk you don't

need.' Or something like that. Anyway, sound philos-

ophy for life is sound philosophy for signal proces-

sors. So, that annoying drop-out you get switching

PROCESSOR.

MULTI-EFFECT PROCESSOR DPS-V77

11 SAVE

SYSTEM

ENTER

Take a good look: nice, big LCD display and

"jog/shuttle" knob. Numeric key pad, descrip-

tive icons. Assignable direct access keys, located

where they ought to be. Call up a preset and

up to 6 of your most frequently used parameters

are there instantly. All this, plus full MIDI

control. So work flows

quickly, smoothly, easily.

As easy as calling

GE DISPOSAL.

3

between effects? It's history. Eliminated. Toast.

Seamless transition - say, between the tail end of

a reverb into a chorus - is now reality. With this

'morphing' function of the DPS-V77, a new effect

(like a chorus or flange) can begin while a current

effect (like a reverb or delay) is decaying, giving you

from 0 to 10 seconds transition time.

Imagine. No more fumbling, grumbling, mumbling, mumbling, the

1-800-635-SONY, ext. DPS, for more information

a whole rack of trouble. No more extraneous

SONY

stor hone beoplews

in part without Sony's written permission is prohibited. Sony is a trademark of Sony.

Dar Williams "Mortal City"

Steven Miller explains how he recorded Williams's newest release

BY ROBERT GRANGER

What was the basic "studio" setup you used to record Dar's *Mortal City* disc? Basically, with the budget that was there, the only way to have it come out the way we wanted was to make the record at Dar's house. I had been trying to figure out what kind of portable system to get and I decided to go with the Roland DM-800. So I brought up the Roland unit, some mic pres (including a few APIs and a John Hardy), and a few mics. I did all the stuff on headphones, but I brought up a pair of Genelec speakers. In reality, those weren't even really necessary because we ended up listening to most of the stuff on the headphones anyway.

What kind of mics did you end up using for her guitar and vocals?

I used two Schoeps with the switchable cardioid and omni capsule on the guitar. The other mics were pretty basic things, including a Neumann U87 and an AKG C414, which are not microphones that I particularly like, but they were stuff that I had lying around and we wanted to avoid big rental charges. We also used an Electro-Voice RE20.

How did you handle your tracks with the DM-800?

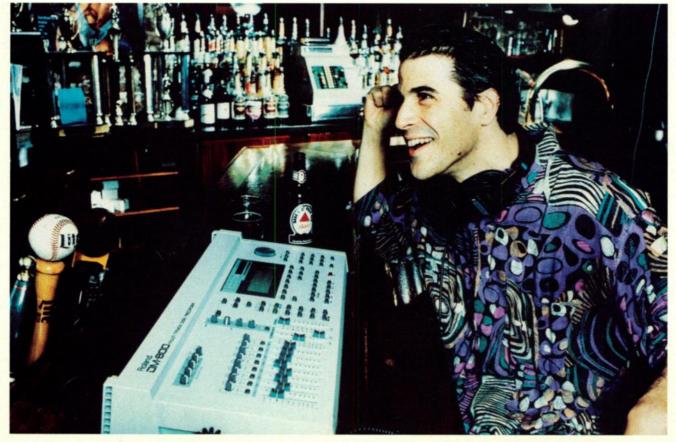
I wanted Dar to explore different ways of approaching the song, so I let her do a couple of takes, I'd compile them on the empty tracks, and we'd keep the one that we liked the best. I didn't do any "punching in"; in fact, I've rarely used that technique — even with my Windham Hill recordings. I like to let them sing takes and then go through and grade them for pitch, delivery, and diction, then compile a master track from there.

How did you find the editing process using the DM-800?

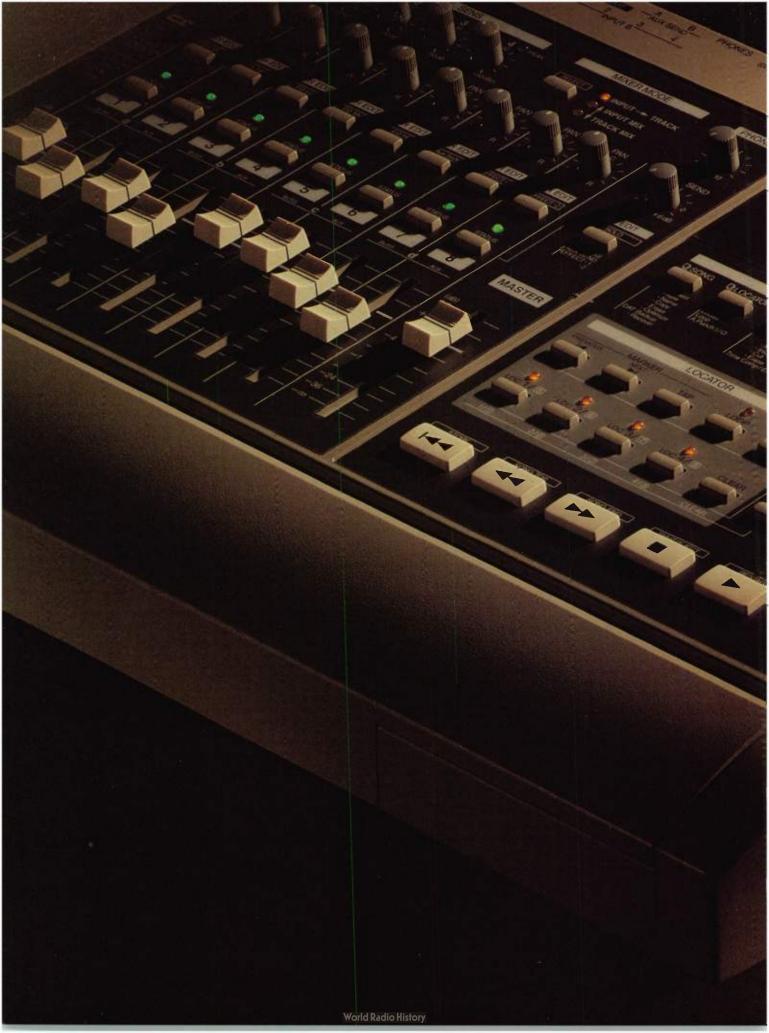
I really put this thing to the test. The tightest increment you can get on the DM-800 is milliseconds, which is good if you've got my mentality with the early days of tape editing. There are an incredible amount of edits.

How did you end up editing in a sports bar during the NBA playoffs?

I'm an enormous basketball fan and I thought, "I take this thing everywhere else with me, why not take it to watch the Knicks game?" The Riviera Sports Bar is right up the street from me, so I got the DM-800, brought it to the bar, and just asked if I could plug it in. So it was me, the regular bartender, and four other guys and I was working and watching and drinking, and it worked out really well.



What do you say to the world's first affordable 64-track Digital Studio Workstation?



Welcome home.

Say hello to the first fully integrated digital recording workstation practically anyone can afford. The VS-880 is for anyone who ever wanted more and more out of their home or studio recording environment.

It's a 14-channell digital mixer. It's an 8x8/64 virtual track digital recorder. It's two completely independent stereo multi-effects processors'. And it's up to 500 minutes or more of random access recording time.

It's time to turn your home or workplace into the studio you've always wanted. Turn on to the VS-880 Digital Studio Workstation and discover just how easy recording, editing, mixing and mastering can be to help you create your best work.



*With optional VS8F-1 Effect Expansion Board.

Who did you think would put it all together?

Only Roland could combine all the key components of studio recording

in a fully digital environment.

The VS-880 is the first digital workstation of its kind.

A 14-Channel Digital Mixer

The VS-880 comes with a 14-channel digital mixer with digital EQ. Fader, pan and mixer parameters can be automated by MIDI control change messages. Internal snapshot automation is included.

An 8x8/64 Virtual Track Digital Recorder

Eight primary tracks with eight levels of virtual tracks give you the ultimate in recording flexibility. Record multiple takes, edit or undo edits and compile the best parts of various takes to create the perfect track without compromise. You can accumulate up to 64 instantly accessible tracks and choose eight for final mixdown.

Digital Editing

With the VS-880, you get fully non-destructive editing capabilities. You can copy, move, exchange, insert, cut and erase tracks. Because it functions like a word processor, you can select the perfect chorus of your song, copy it or move it somewhere else without re-recording it. Or cut out another section and have the remaining material slide over and fill the open space automatically. And if you're not satisfied with a particular edit, simply return to like previous performance. Instant locate points make editing on the VS-880 as user-friendly as it gets. For advanced applications, functions like Scrub Preview or Time Compression/Expansion are available.

Digital Mastering

The VS-880 is fully digital and has digital in and out. This makes it compatible with all your existing digital equipment, both linear tape-based and non-linear hard disk format.

Two Digital Effects Processors

A VS8F-1 Effect Expansion Board can be easily userinstalled, giving you two totally independent multieffects processors. Reverb, stereo delay, chorus, flanging, vocoder as well as distortion/overdrive and guitar amp simulation are a few of the effects that can be added during recording or during final mix in realtime. Three dimensional effects based on Roland's proprietary RSS system are also included.

Digital Memory

You can choose the type of memory which best suits your needs. Select a 540 MB internal drive or an internal 1 Gigabyte removable lomega JAZ drive capable of 500 minutes of recording time. A standard SCSI port allows for easy connection to external drives such as lomega ZIP, magnetic optical or other storage media.

Synchronization

The VS-880 is MIDI compatible for synchronization with MIDI Time Code (MTC) as both a master or slave. MIDI Machine Control (MMC) also allows automated transport control, putting playback, fast forward, rewind and more at your fingertips.

The VS-880 Digital Studio Workstation is ready and waiting for your valuable input at your nearest Roland dealer. Or call (213) 685-5141, ext. 798 to order a free demonstration video.



Roland Corporation U.S., 7200 Dominion Circle, Los Angeles, CA 90040 (213) 685-5141 Roland Canada Music Ltd., 5480 Parkwood Way. Richmond. B.C. V6V 2M4 (604) 270-6626 COMPUSERVE®: GO ROLAND Fax-Back Information: (213) 685-5141, ext. 271 http://www.rolandUS.com All trademarks are registered by their respective companies.

TL Audio Indigo Series

A new line of processors guaranteed not to make you blue by steve LA CERRA

hese days recording studios have more tubes than a tire shop. That trend has only been on the increase in the last year since manufacturers have started introducing tube-based processors that mere mortals can afford. TL Audio is a name familiar to engineers as the producer of high-quality, tube-based compressors, mic preamps, and equalizers. In an effort to bring its technology to a broader market, TL Audio has introduced the Indigo series, a new range of more affordable and easy-to-use units that are likely to make a project studio owner's eyes light up.

The Indigo line consists of five new product introductions, with additional processors scheduled for later in the year. Available now are a 4channel mic preamp, a fully parametric stereo equalizer, a shelving/peaking characteristic stereo equalizer, a stereo compressor, and a stereo overdrive unit. All of the devices come in a compact one-rack space chassis. Four of the five units have line-level inputs and outputs on both 1/4-inch unbalanced and XLR balanced connectors. The exception, of course, is the microphone preamp, which has balanced mic-level XLR inputs and the same XLR balanced and 1/4-inch unbalanced outputs as the other units.

The IN-PA-2001 is the Indigo Series' 4-channel tube microphone preamp, and each of the four channels has identical front-panel controls. Two rotary pots (with knobs of different sizes) control input gain and output gain, allowing you to overdrive the input while still controlling the level to tape via output control. Push-button switches enable phantom power (thanks to TL Audio design engineers for putting them on the front, not the rear) and a 90 Hz lowcut filter is available for keeping the rumble out of your tracks. A peak LED shows overload at the input stage of the preamp and a phase reverse switch lets you check relative phase of multiple microphones. As mentioned above, the rear-panel microphone inputs are on balanced XLR connectors. The IN-PA-2001 is expected to retail at \$1295.

TL Audio's IN-EQ-2011 is a peaking/shelving type of equalizer with a continuously variable input control, in case you might like to dial in some "crunch" to your EQ'd sound. There are four bands of EQ available for each of the stereo channels, two of which are shelving and two of which are peaking-type circuits. Each band has a maximum boost and cut level of 12 dB. The low- and high-frequency shelving controls are switchable between two frequencies: 80 Hz and 120 Hz for the LF and 8 kHz and 12 kHz for the HF. The midrange controls are of the peaking type; the low-mid band is switchable between 250 Hz, 500 Hz, 1 kHz, and 2.2 kHz while the high-mid EQ band can center at 1.5 kHz, 2.2 kHz, 3.6 kHz, or 5 kHz, thus allowing a small area of overlap in the midrange. Signal status through the 2011 is monitored with two LEDs per channel: one showing peak signal level and the second indicating that the EQ circuitry is switched into the audio path (each channel may be individually bypassed). As an added bonus, TL Audio gives you front-panel auxiliary 1/4-



TRUE BLUE: TL Audio's Indigo line is priced more affordably than the company's other processors.

Don't be a Tape-based googlehumpher.

n 1856, Charles Darwin determined that the Tapebased googlehumpher species died off because of its failure to evolve with the times. 150 years later, E-mu's DARWIN 8-track Digital Hard Disk Recorder is proving to be the salvation of the digital recording species, with its highly evolved functionality and streamlined cost. It's true: DARWIN

GOOGLEHUMPHERUS TAPEZOIC PERIOD xtinct due to its diet of magnetic tape limited brain power and incredibly slow speed

delivers the power of digital audio workstation-style recording and editing at a price every *Homo erectus* can afford.

Play music? Doing some recording? You won't believe what DARWIN will do for your creative process! That's because DARWIN's heart pumps with the power of non-destructive random access recording technology. Consider the benefits of that over taped-based beasts:

- Effortlessly cut and paste that perfect chorus throughout a tune
- Completely rearrange a composition without screwing up the original
- Instantly resurrect that insane guitar riff you accidently wiped out
- Create a virtually unlimited number of tracks from a single unit
- Avoid time wasted waiting for tape rewind
- Operation so easy and evolved, even a dinosaur can run it

Any question about who will be picking who's teeth with who's bones in the recording food chain? Face it, you need this thing.

Don't make the mistake of the googlehumphers before you. Spit that tape out of your mouth and sail to your nearest E-mu dealer to learn about the DARWIN Digital Audio Disk Recorder in the flesh. After all, DARWIN is... the natural selection.

E-mu Systems

To find your local E-mu dealer, contact: PO Box 660015, Scotts Valley, CA 95067-0015 • 408.438.1921 UK Office: Suite 6, Adam Ferguson House, Eskmills Industrial Park, Musselburgh, EH21 7PG • 44.131.653.6556 CIRCLE 32 ON FREE INFC CARD

World Radio History

THIT

FIRST LOOK

inch inputs for each channel, allowing the 2011 to be used "direct box-style" with equalization for keyboards and guitars. The 2011 has a suggested list price of \$1295.

The IN-EQ-2011's brother is the IN-EQ-2012 stereo parametric equalizer, which features one high- and one low-frequency band per channel: channel A has band 1, continuously variable from 30 Hz to 1.2 kHz and band 2, continuously variable from 1 kHz to 18 kHz. Channel B has band 3, continuously variable from 30 Hz to 1.2 kHz and band 4, which is continuously variable from 1 kHz to 18 kHz. Bands two and three each have an additional switch labeled "divide by ten" and "multiply by ten," respectively.

Although TL Audio is giving you only two bands per channel, don't be fooled into thinking that this EQ unit lacks flexibility. First of all, each channel can operate over a very wide range: 30 Hz to 18 kHz. And TL Audio has a little trick up its sleeve that tracking engineers will love. A frontpanel switch labeled "mono mode" sends the output of channel A directly to the input of channel B. Engage this switch along with the "divide by ten" and "multiply by ten" functions and now you will have a single-channel, 4band parametric unit with EQ available at the following frequencies: 30 Hz to 1.2 kHz (band 1), 100 to 1.8 kHz (band 2), 300 to 12 kHz (band 3), and 1 kHz to 18 kHz (band 4). Boost and cut range is 15 dB and the IN-EO-2012 offers the same LED peak indicators, auxiliary inputs and bypass switches as its brother. Very versatile for a processor that lists at only \$1395.

The IN-C-2021 is the Indigo stereo compressor and features continuously variable controls for input gain, threshold, ratio, and gain make-up (you might note that while the photo shows a two-position switch for threshold adjustment, TL Audio has wisely changed the threshold control to a variable pot). The company simplifies the operation of this unit for less-experienced engineers by providing push-button switches labeled "fast" or "slow" for the attack and release times. A rear-panel sidechain insert allows you to connect an external EQ unit for "frequency-conscious" compression such as de-essing and a link switch straps the two channels continued on page 130

Two Studio Veterans Team Up. Ed Cherney and the AT4033

Grammy-awarc winning recording engineer and producer Ed Cherney has worked with some of the most talented people in the business. Bonnie Raitt, Eric Clapton, Little Feat, Elton John, and The Rolling Stones just to name a few. So it was inevitable that he would eventually work with one of the finest m crophones. The AT4033.

Here's what Ed had to say about it:

"When I first used the 4033, I was working on a ba lad with singer Jann Arden. But I'd always had trouble finding the right microphone to handle the level sne sings at in choruses as opposed to lower volumes in the verses. Until I tried the 4033."

"I just put it up with no compression, no EQ, and had her sing into it. And I mean, *it was right there.* The mic handled everything, stayed clear and open all the way through, and ended up sounding great."

For Ed and the AT4033, it sounds like the beginning of a long, successful partnership. But what about you? Just call, write, or fax for details on how you can team up with the AT4033.

(A) audio technica

Audio-Technica U.S., Inc. 1221 Commerce Drive, Stow, Ohio 44224 Tel: 216/686-2600 Fax: 216/686-0719 E-mail: pro@atus.com

Audio-Technica Limited Old Lane, Leeds LS 11 8AG England Te : 0113 277 1441 Fax: 0113 270 4836 CIRCLE 09 ON FREE INFO CARD

6

GET A PROCESSORS



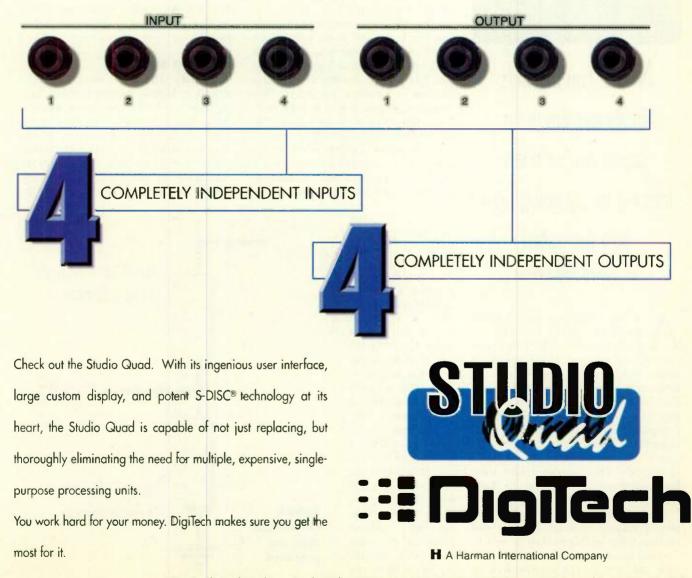
COMPLETELY INDEPENDENT EFFECTS

That's less than \$125 ea. Do the math. It's your money.

*Buying one Studio Quad is like buying 4 independent processors. Inside the Studio Quad is a multi-effects processor that is multi-talented and while most of us can't do more than one thing at a time, the Studio Quad can do 4. Why did we put the power of 4 multi-effects processors into one unit? So you, too, would be able to do more than one thing at a time. The Studio Quad features 4 completely independent inputs and outputs that give you the power of four independent mono Visit Digitech on the World Wide Web at http://www.digitech.com processors or two true stereo processors. Imagine the power to process two true stereo sources simultaneously without sacrificing control. But true stereo is only the beginning. Imagine having 4 independent reverbs for vocals, guitar, keyboards, and drums, where each reverb can be optimized specifically to achieve the best audio performance for each application. And imagine a processor that gives you the power to create any combination of effects, in any order. And then multiply by four.





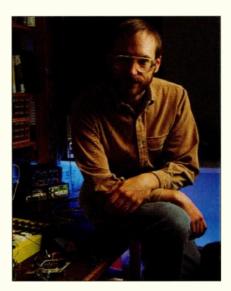


8760 South Sandy Parkway, Sandy Utah 84070 • (801) 566-8919 • Fax (801) 566-7005 • Int'l Fax (603) 672-4246

CIRCLE 28 ON FREE INFO CARD

World Radio History

The Ultimate Direct Guitar Sound



Now it can be told! Finally get the guitar sound you've been looking for by going right into the board

A re the neighbors complaining about your stack o' Marshalls going to 11 at 3 AM? Does your guitar lack clarity? Does the bass sound like someone shot holes in the frequency response at various random frequencies? Are you tired of getting a completely different sound live compared to the studio? Do you want something — anything — in your life that's actually predictable and repeatable? Then you're a candidate for recording your guitar or bass direct.

Going direct eliminates the hassles of miking an amp, leakage, and various other problems. But in return, you get an exciting new crop of hassles. This article will tell you how to overcome them and get a great direct sound. But first, some theory.

THE IMPEDANCE IMP

Guitar pickups, unless they're active (i.e., have a built-in preamp), exhibit a high *output impedance* that increases at higher frequencies. Output impedance is like shoving a resistor in series with your guitar output. Meanwhile, whatever input the guitar feeds has an *input impedance*, which is the equivalent of placing a resistor from the amp (or mixer, or recorder) input to ground. Get the picture? We now have an involuntary, laws-of-physics-mandated volume control (fig. 1).

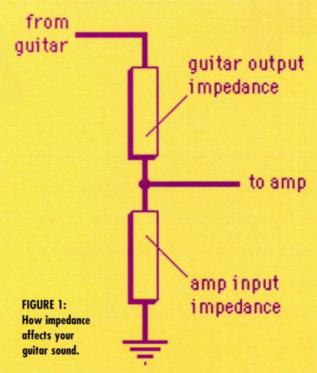
The higher the output impedance, and the lower the input impedance, the more signal loss your guitar will suffer — especially at higher frequencies. Conversely, to keep the sound as pristine as possible, you want a low output impedance and a very highinput impedance. Tubes and FETs have inherently high-input impedances, whereas transistors are (generally) of the low-input impedance persuasion — perhaps one explanation for why people like the sound of tube gear.

This is why plugging your guitar directly into a mixing console or recorder dulls the sound so much: the console will usually have a low-input impedance that sucks the life right out of your signal. A direct box converts the guitar's high-impedance output into a lowimpedance out, thus providing a better match.

However, not any direct box will do. Older types often use transformers whose impedance may load down the guitar more than the mixer input would. A direct box for guitar should have an input impedance greater than 250k (1 Meg is even better). You also probably want it to be able to generate balanced as well as unbalanced outputs. Any other features — tone controls, phase switches, etc. - can be helpful, but aren't essential. Many of today's mic preamps are also designed to work as a DI (direct injection) box for guitar; if you're handy with a soldering iron, check out the DIY DI (EQ, June 1991 issue), a full-function direct box optimized for guitar and bass. If you don't need something quite that elaborate but still need killer specs, the IGGY preamp elsewhere in this issue is the ticket.

DOWN AND DIRTY

Guitarists who have never experienced a high-impedance input are often astonished at the clarity, brightness, and definition they now hear from their guitar. However, while that's great for clean parts, what if you want a down and dirty amp sound? In fact, for some guitarists, an amp with a low-impedance input that robs high



Vintage Circa 1995

PVM T9000 CONDENSER TUBE MICROPHONE

he new Peavey PVM T9000 tube microphone features a very special, self-polarized condenser capsule coupled with a vacuum-tube preamplifier to provide all of the mellow warmth for which tube microphones are revered.

With smooth, extended-range frequency response and a very uniform cardioid directional pattern, the PVM T9000 is ideally suited for studio vocals and a host of critical applications.

This microphone will easily handle SPLs of up to 137 dB and includes a 10 dB attenuator and 80-Hz low-cut filter switches. The unique shock suspension incorporates a finned heat-sink to help dissipate tube filament heat! Get all of the vintage sound, state-of-the-art functionality and classic styling from a company dedicated to quality and performance in professional sound productions. Get the Peavey PVM T9000.

The MONITOR Magazine is a publication filled with the latest information musicians want to know. To receive 4 issues for only \$5 (price good in U.S.A. only) send check or money order to: Monitor Magazine, Peavey Electronics, 711 A Street, Meridian, MS 39301 • (601) 483-5365 • Fax: 486-1278 • ©1994

World Radio History

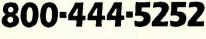
Research

Audio

Media







Price Quotes & Information 505-292-0341 Fax 505-293-6184

> 800 S-T Juan Tabo Blvd NE Albuquerque, NM 87123

MI INSIDER

frequencies is part of their "sound" since taking off the high end, then adding distortion, gives a smooth, creamy sound. Fortunately, there's a way to simulate various amp qualities when going direct.

THE MULTIEFFECTS CONNECTION

An alternative to just patching in a direct box is to use a multieffects designed for guitar, or a series of "stomp boxes" patched together to shape the sound. These generally have the high-input impedance required to maintain your guitar's signal strength. Most of today's multieffects let you string a bunch of effects together in any order you want, edit the sound of each effect, and save the whole shebang in memory as a program. However, the programming choices you make can drastically influence the

sound, particularly the order in which you place effects. Here's what works for me (see fig. 2).

COMPRESSOR

This goes first in the chain. If you're using a stomp box, it should have a high-input impedance to interface the

guitar with the rest of the system. Adding some compression increases sustain, gives a smoother sound, and makes the best use of the limited dynamic range of many signal processors. Use a fast attack (10 ms or less) and medium decay (300–700 ms). Adjust the threshold and ratio according to whether you want a natural sound (e.g., 6 dB of gain reduction with a 2:1 ratio) or a squeezed, squashed sound (12 dB of gain reduction, 8:1 ratio).

PARAMETRIC EQ

Patching EQ before the distortion customizes the distortion sound. For example, if you boost the midrange at about 700 Hz with a one octave bandwidth, single note leads higher up on the neck will be boosted more, thereby going into distortion earlier, and giving more sustain. Boosting the low end gives a chunky, meatier sound. Rolling off the extreme highs prevents that brittle, metallic kind of breakup that happens with some distortion boxes (and simulates the effect of loading down the guitar with a low-input impedance). Note that the pickup settings are important as well; the bass pickup gives a smooth sound (especially with the tone control rolled back), whereas the treble pickup gives more "bite."

DISTORTION

This is where the crunching occurs. Adjust to taste.

POSTDISTORTION EQ

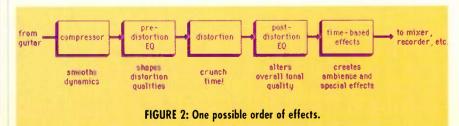
Parametric or graphic EQ works well

Guitarists who have never experienced a high-impedance input are often astonished at the clarity, brightness, and definition they now hear from their guitar. here to shape the sound's overall tonal quality. The predistortion EQ has less of an effect on the overall tonality because any subtle differences are wiped out by distortioninduced clipping. For example, even if the predistortion EQ rolls off the highs,

the distortion process itself will generate some additional highs. The postdistortion EQ can take care of this, as well as add some midrange peakiness so the signal cuts more, or roll off some of the bass to minimize interference with low-end instruments such as bass and low toms/kick.

One popular setting is to boost the midrange around 1 kHz to give more "honk" to the guitar for solos. For rhythm guitar, some engineers put a gentle midrange notch around 500–1000 Hz to create a sound that leaves some space for other midrange instruments, yet retains a bright top end and authoritative bottom.

Perhaps the best postdistortion EQ option is a speaker emulator. The







For more than three-quarters of a century, Tannoy has been designing and producing loudspeaker systems and components to meet the demands of the world's most demanding user. A philosophy of constant research and investment in state of the art materials, technology and processes enables Tannoy to ensure that every monitoring system we produce will re-produce

absolute fidelity to the source, true dynamic capability, and most importantly, real world accuracy. This is why Tannoy systems are used in more of the world's professional facilities than any other brand.

In the North American marketplace, Tannoy has been the number one monitor of choice for several years according to the Billboard's international recording and equipment statistics. This clearly illustrates why Tannoy enjoys its reputation as the world's leading manufacturer of reference loudspeakers. In fact, leading the market is what Tannoy is all about. While other multi-faceted manufacturers, not dedicated solely to the art of reference monitor loudspeakers, scurry to produce products to compete with Tannoy's original highly acclaimed and awardwinning PBM series, **Tannoy moves on**.

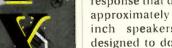


The new PBM II series, once again, is setting new standards in the industry. Pioneering new technologies such as variable thickness, injection molded cones with nitrile rubber surrounds are but one fine example of our dedication to perfection. The new molded cones are stiffer than conventional cones' producing more linear extended low frequency. They are better damped for reduced distortion and exhibit more naturally open and detailed midrange. They are immaculately consistent and durable for years of faithful trouble free use. From the high power polypropylene capacitors to the hardwired minimilist crossover, every component has been carefully selected for the new

breed of PBM II series. When leading edge technology is so affordable, Get Real. Don't settle for second best.

CIRCLE 69 ON FREE INFO CARD





You've just recorded a masterpiece on your cassette multitrack and you're ready for mixdown. But what can you do to eliminate that annoying hiss; while maintaining the punch. clarity and sparkle of your original tracks?

Combining our patented Sonic Maximizer and single-ended noise reduction in one unit allows you to cut hiss, AND, restore high-frequency detail and bass "punch".



Audition one today! 5381 Production Drive • Huntington Beach, CA 92649 FAX 714-896-0736 • 714-897-6766

CIRCLE 19 ON FREE INFO CARD



response that drops off like a cliff after approximately 5 kHz (12-inch and 10inch speakers were never really designed to do a good job on highs). Some "speaker emulator" modules found in multieffects are simply graphic EQs or other simple tone controls that give you the option to roll off highs; others are digitally generated response curves of different types of amps that even let you specify whether you want the characteristics of four 10-inch speakers, two 12-inch speakers, open back, closed back, and so on. These can make a huge difference in terms of getting a "true-tolife" sound.

average amp cabinet has a frequency

TIME-ALTERING EFFECTS

MI INSIDER

(As in chorus, delay, reverb, etc.). If you're using a multieffects, you might want to ignore the internal time-based effects and go through your studio's big-bucks, high-end reverb. In fact, any of the above stages might be served better with outboard gear - for example, use some nifty tube compressor instead of the one in a multieffects (however, make sure it has a high enough input impedance if it's the stage right after guitar).

One final caution involves stereo. Some stereo multieffects can cause problems when heard in mono because of phase cancellations. Always monitor a stereo guitar sound in mono before assuming it's okay.

After you've nailed your sound, there's a side benefit: because your sound is not dependent on an amp, your sound won't change as tubes age, and you'll have much more flexibility - if you want a brighter sound, a darker sound, or a crunchier sound, just tweak a few parameters. Furthermore, when playing live, you can just plug your output into the PA or other flat response system and it will sound the same as it did in the studio. Ain't (electronic) life grand? EQ

Crady An veriencis a card-carrying humanoid hiped. He still thinks guitars are cool, and, in fact, is a consulting editor to Guitar Player magazine in addition to his technology editor duties at EQ magazine. Visit his "Sound, Studio, and Stage" area on America Online.

<text>

The real passion behind a hit song lies in the performance. You want all the emotion and the expression of the performance to burst onto the track and shine its way through the mix. Whether it's a truly great performance or just a happy accident, you want that moment captured in all of its glory. When that inspired moment occurs, you need to have the right mic...the **RØDE NT2.**

The RODE[®] has all the special stuff that makes those vintage mics so desirable:

- Dual Gradient Transducers with Large Diaphragm and Gold Plated Membrane to warmly receive and translate the full body of the performance
- Internal **Gold Plated Connections**, insuring warmth and precise signal transitions
- Transformer-less low-noise circuitry
- Switchable Filter enabling you to change the low frequency response
- Switchable Pad enabling you to handle high gain.
- Switchable Pattern providing cardioid or omni patterns
- High SPL (sound pressure level) allowing it to work cleanly for screaming vocals or instruments, or even soft spoken word
- It even comes with a Shock Mount, Flight Case, and Pop Filter.

The RØDE[™] large Diaphragm Condenser Mic is hand-assembled in Australia. In the tradition of the great microphones, it's designed to furnish outstanding, vintage-flavored vocal or instrumentation micing—at a truly untraditional, breakthrough price.

Step on the road to recording success by seeking out your nearest EVENT Electronics dealer. Listen to the NT2 and hear the difference a great mic can make.

RØDE in is trudemilik of Freedman Electronic CIRCLE 30 ON FREE INFO CARD



P.O. Box 4189 Santa Barbara CA 93140-4189 voice 805-962-6926 fax 805-962-3830

Chas Chandler: Guiding Hendrix

The noted producer discusses the "Experience" of his lifetime

Bryan "Chas" Chandler began his musical career in the late '50s as bass guitarist with the Newcastle group The Contours. Keyboardist Alan Price, who along with Chandler and a fresh-faced Eric Burdon went on to form The Animals, was also a member. The band's distinctive R&B style quickly gained them a strong following in the North of England, and before long the group was brought down to London where they recorded their first single with then unknown producer, Mickie Most.

Chandler, however, was dissatisfied with Most's choice of song ("Let Me Take You Home"), and insisted that the next release should be "House Of The Rising Sun." It was an unpopular move that turned both the band's record company and management against them.

"They made life as difficult as possible for us," recalls Chandler. "They booked us into a London studio for eight o'clock in the morning after we'd been playing a gig in Liverpool (the night before). We got into the studio at 7:30 after driving all night, by 8:00 we were recording, and by twenty past we were packing up the gear for a gig in Bournemouth."

Despite all this, the record became a huge international hit and marked the start of an exhausting period of touring and recording that was to last for two years. Although the Animals were extremely successful, the band members saw little financial return, and by 1966 Chandler and Burdon were the only remaining founding members. Feeling disenchanted, Chandler decided to do one final tour of America and call it a day — it was just before the start of this that he came across Jimi Hendrix.

"We'd organized the final tour to try and put some money in the bank, because we'd never seen a penny. I'd gone to New York a few days before the rest of the guys to see this young lady I was enamored with at the time, and she played me this track by Tim Rose called 'Hey Joe.' 1 thought, 'I know what I'm going to do when I get back to England, I'm going to find an artist to record this with.' I literally decided then and there that I was going to be a record producer.

"The next day I ran into Linda Keith, who was Keith Richards's girlfriend,

and we got to talking about production. She said, "If you're getting into production, you've got to come and see this guy I'm working with." So we drove to Greenwich Village where I met this 24year-old guy called Jimi James (Hendrix's stage name at the time). We got along very well, so I went down to The Cafe Wha? to see him play.

"The band was just a pick-up four piece — the other guitarist turned out to be a 14-year-old runaway named Randy California. The very first number they played was 'Hey Joe.' I just sat there and couldn't believe it, the whole thing was far too much of a coincidence.



PRE-JIMI: Chas Chandler in the studio with The Animals.

"Jimi was very impressive, and he really carried the band with his guitar and vocals. I offered to take him to England once the Animals tour had finished, and it was all agreed in 20 minutes — the only worry he had was what kind of amplifiers they had in England. I said, 'Believe it or not Jimi, you can buy American amplifiers over there if you want to,' and he was happy."

THE BIRTH OF THE EXPERIENCE

How did you go about finding a band for Hendrix?

I'd been helping Eric Burdon put his new band together, which had just left

a product comes along that is so unique, so powerful, that it changes the way we look at things.

ICC in a while

Such a product is the Aphex 661 Compressor Limiter- creating a new standard by combining four Aphex inventions. A skillfully engineered instrument of unprecedented flexibility, ease of use and sonic excellence.

3

Tubessence[®] - true vacuum tube technology and warmth; High Frequency Expander (HFX)[™] for automatically retaining the high frequencies lost during compression; Easyrider[®] circuitry for an Auto mode that really works; and the world's best VCA - the Aphex 1001, the fastest, most accurate and transparent available.

The Aphex Model 661 - another revolutionary step toward improving the way the world sounds.



Improving the way the world sounds[™]

11068 Randall Street, Sun Valley, CA 91352 • Tel: 818-767-2929, Fax: 818-767-2641 CIRCLE 08 ON FREE INFO CARD to do some recording in New York. I was in the office and this guy knocks on the door and asks if he could audition for the new Animals as a guitarist. He had his hair cut like Jimi's, it was Noel Redding, and I thought here's coincidence number two. So I asked him if he could play bass. He'd never played bass before, but I lent him mine and we went down to a club called Birdland in St. James where we were going to audition some drummers.

He began playing and Jimi just said, "If he's prepared to keep it simple like that, he'll be fine." So we settled on Noel but we just couldn't make up our minds about drummers; in the end, we had to toss a coin to choose between Aynsley Dunbar and Georgie Fame's drummer, Mitch Mitchell.

Mitch got the job. He was only 17, but he was an incredible drummer. He used to drive you mad though; he was the most consistently late person you've ever known. He'd always disappear minutes before the band was due on stage, and there would be this last minute panic to find him. I used to think, "Do I really need to put up with this sort of crap," and then as soon as he started hitting the drums I'd think, "Yes I do!"

How long before you got the band into the studio?

Oh, very quickly. We went to Kingsway where the Animals used to record, and put down the backing track to "Hey Joe" on four track with engineer Dave Siddle. There was then a gap when the band went to France to play a series of gigs with Johnny Halliday. When we got back we went into Pye to record backing vocals with The Breakaways, because Kingsway was booked-up. The B-side and mono mixes were then done at Kingsway.

Did you then record the rest of Are You Experienced at Kingsway?

About half the album was done there, and the rest, apart from "Third Stone From The Sun," was done at Olympic. "Third Stone" was actually recorded at CBS Studios, and the engineer there gave us some dead time without the studio knowing about it - we went in at about midnight to record. The reason we went to Olympic was because Kingsway was about to close, and also, we'd had problems because it was under a bank — Jimi played pretty loud, and they complained that the vibrations interfered with their computers. It was Mick Jagger that recommended Olympic, saying you can play as loud as you like there and nobody will complain.

At Olympic we met up with engineers Eddie Kramer and George Chkiantz who were both very important to us. George was the tape-op, but he was much more than that, he was a bit of a boffin who talked electronics all the time, and he came up with a number of things for us, including phasing.

Along with chief engineer Keith Grant, he modified this old Revox — it was a bit hit-or-miss, but when it worked it sounded fantastic. The Small Faces were in the studio next door to us and they came in to have a listen and were so impressed that they rushed off and put it on *Itchycoo Park*, which came out before our album."

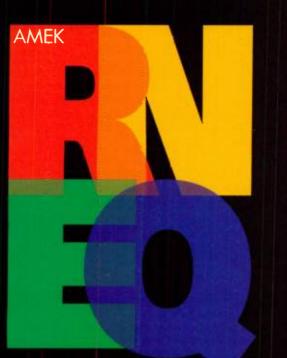
Was there much experimentation before that?

Yeah, it started with "Purple Haze" at Kingsway. Siddle was a more conventional engineer than Kramer, but we used to do some strange things like feed stuff back from tape to headphones and re-record it while moving them in and out over the microphone — it produced a kind of "phasey" sound. An awful lot of the effects were based around tape — echoes, delays, spin-ins, and so on. We did a lot of spinning-in of voices, silly noises, slowed up drums, and things.

We also used a lot of backwards effects, and Jimi was great at that you'd put a tape on back to front and he'd just play along, he'd know exactly where he was and what to play to get



PLAYING THE CROWD: Hendrix listens in the studio amidst many onlookers. Chandler felt Jimi was playing for the crowd, not for the studio.



The EQ The tradition The pedigree

System 9098 Equalizer by Rupert Neve the Designer



"The SYSTEM 9098 EQ is a high performance Equatizer and Preamplifier designed to originate microphone signals of the highest quality and to process signals generally in terms of frequency response. The circuitry is based on the research I put into the 9098 <u>console</u> and the approach bears

many similarities to that used in the 9098. Paramount importance has been given to the sonic quality of the audio path, taking great care to retain the highly-prized musical character of the famous old designs of this pedigree.

The SYSTEM 9098 EQ embodies the original curve shapes now enhanced by improved circuitry which provides swept frequency bands in place of the discrete switched steps of the past. Thus the EQ bas become even more powerful yet remains a subtle and creative tool, using the same basic circuit configurations which have been successful over many years. However, new amplifying devices and better quality components have resulted in lower noise, lower distortion and the ability to handle higher frequencies.

The result is an equalizer which has the solidity and sound of Class A without the cost, heat and weight penalties and thus provides the 'best of both worlds'. We have also left behind cumbersome and expensive hand cabling, noisy connectors, heavy separate power supplies and outdated assembly techniques which contribute nothing but nostalgia. Apart from the robustness, repeatability and reliability, we have now made one of my designs more affordable than ever before."



CONTACT CLEANERS ARE <u>NOT</u> CREATED EQUAL



- Improves Conductivity
- Reduces Noise & Distortion
- Deoxidizes, Cleans & Protects
- Reduces Intermittents

▶ Reduces RFI, Wear & Abrasion

Even the finest equipment cannot guarantee noise-free operation. One "dirty" connection anywhere in the signal path can cause unwanted noise, distortion and signal loss. Considering the hundreds (if not thousands) of connections in electronic equipment today, it's only a matter of time before they here to fail.



Available in Environmentally-Safe Spray, Wipes, Pen, Precision Dispensers & Bulk Containers

Some film deposits are effectively removed with washtype cleaners such as contact/tuner cleaners, degreasers, acohols and other solvents. Oxides and sulfides, however, become an integral part of the contact surface and <u>cannot</u> be removed by ordinary contact cleaners.

DeoxIT dissolves oxides and sulfides that form on metal surfaces, removing these sources of resistance. This restores the contact's integrity and leaves a thin (organic) layer that coats and protects the metal



DeoxIT's advanced formula contains deoxidizers, preservatives, conductivity enhancers, arcing and RFiinhibitors and antitarnishing compounds that significantly increase the performance and reliability of electrical components and equipment.

DeoxIT provides long-lasting (1-10 years) protection, reducing the expense of repeated cleaning with expensive and aggressive ozone-depleting solvents.

DeoxIT outperforms & outlasts all other contact cleaners. Non-flammable, Safe on Plastics & Environmentally-safe.

CAIG PRODUCTS USED BY THOSE WHO DEMAND THE BEST!					
Ampex	Federal Express	Honeywell	Switchcraft		
Boeng	General Electric	McIntosh Labs	Tektronix		
Diebold Inc.	John Fluke Mfg.	Motorola	Texas inst		
Doiby Lab	Hewlett Packard	Nakamichi	Xerox Corp.		
	LABORATORIES,	TEL: (, CA 92127-1904 619) 451-1799, 619) 451-2799 23		
CIRCLE 21 ON FREE INFO CARD					

TECHNIQUES PRODUCING

the effect he was after. At Olympic there were a huge number of echo plates, Keith Grant was obsessed with them, and that was a massive advantage in those days. They also had this mock-up of a Leslie cabinet that we used, but it was amazing just how much tape we got through, and the Revox seemed to be the answer to just about every echo we wanted. Quite often Eddie and George would come up with an effect and we'd go home that night and Jimi would start writing something based around it.

Did Hendrix always use the same equipment to record with?

He always used a Strat, I only ever remember him trying a Gibson once

and it was put aside very quickly. He'd mainly use a 4x12 Marshall setup that was driven to its limit; there weren't many effects pedals to begin with — there was a thing called a Fuzz Face built by Rodger Mayer who was a friend of Noel Redding's. It was one of the first fuzz boxes, but in actual fact limi didn't use much of the fuzz on it, it was used more to produce extra sustain, he used to get



POST-JIMI: Chas Chandler today.

fuzz automatically with his hands. Later on there was a wah-wah and an octave splitter called an Octavia also built by Mayer.

What was he like to work with in the studio?

Great, the only disagreement we had in two and a half years was a silly thing about an amplifier distorting — Jimi thought it sounded OK and I didn't. We couldn't agree, so Jimi said he might as well go back to America if he had to record like this. I'd just come back from the immigration department with his passport and return ticket, so I handed them to him and said, "OK, piss off then." He just looked at me and burst out laughing. We never fought after like that again until the *Electric Ladyland* album. Jimi was actually a very funny guy and a lot of fun to be with, he was also extremely intelligent. We shared various flats (apartments) together; the first Ringo lent us, but that caused all kinds of aggravation because the lease said no colored people. So rather than give Ringo any further grief we moved out and John Lennon moved in. He was only there two days before he got busted — the police had come to do Hendrix, but they got him instead. He was furious.

Did much material get composed in the studio?

Very little. It was all generally rehearsed before we went in. "The Wind Cries Mary" was an exception —

Mitch and Noel had never heard the song before, but it completed was along with three guitar overdubs in twenty minutes. The Are You Experienced album was done in six days and Axis Bold As Love in ten. We'd generally go into the studio at about eleven in the morning and finish up around ten at night.

We had to spend an extra day on Axis Bold As Love because Jimi lost one side of the

master. We'd brought the masters back to the flat and he'd taken them over to play for someone, but he left side one in a cab. The next night we had to go in and remix the whole first side again. The only problem we had was with "If 6 Was 9," we just couldn't get the same feel back into it, whatever we did it seemed totally dead. Noel said that he had a 7-1/2 ips copy of it at home and went off and got it. The tape he brought back was in a terrible state, we practically had to iron it to get it to play, but we ended up copying it and that's what we put on the record. It still amazes me how good it sounds.

Did the band get very involved with mixing?

Everyone helped out - Noel and

GUIET CONTROL.

E62 DUAL 31-BAND 1/3 OCTAVE GRAPHIC EQUALIZER

Never before could you get this much quiet performance and sonic control in a graphic EQ that sets new standards for quality.

High-quality, low-noise NJM 2068 ICs and a smart parallel filter design let you operate noise-free with minimal phase distortion.

Super-low Total Harmonic Distortion of less than 0.0035% (20 Hz-20 kHz) adds up to transparent audio.

Balanced XLR and 1/4" inputs and outputs for more effective interfacing.

±15 dB of boost and cut over 2 x 31 carefully selected frequency bands.

Extra-wide frequency response (10 Hz-50 kHz) retains the full harmonic integrity of your program material for a more "musical" sound.

Internal bipolar power supply gives you plenty of headroom (+28 dBu output), especially important in sound reinforcement.

Time-delay circuitry eliminates speaker "thumps" when powering on. Hard-wired bypass feature even lets you pass signal if you lose power! Peak level LED indicators for each channel light 5 dB prior to clipping so you can prevent distortion before it happens.

Bypass switch coupled with level control lets you quickly compare the direct and equalized signal.

All-steel chassis provides maximum shielding and longterm durability.

Optional security cover kit eliminates accidental tampering—a critical feature in fixed installation use.

Suggested retail list price \$349.

Stereo and dual mono operation for either stereo or individual channel processing.

If you want to get up to speed on the fastest growing line in Audio, please call (516) 364-2244, fax (516) 364-3888 or write Samson Technologies Corp., P.O. Box 9031, Syosset, NY 11791-9031.

WHAT'S NEW?

2 manual of

The E30 Dual 15-Band 1/2 Octave Graphic EQ. Same comprehensive control, super-quiet performance and crucially important features as the E62. \$239.



World Radio History



of experimentation to wreck a sound rather than just wrecking it right off the bat. Favorite quote: "Don't worry, we'll fix it in the shrink wrap."

Maestro: This engineer doesn't believe you might actually be able to create your own sound, so wants the band to sound like his/her production, instead of the production sounding like the band. Fave quote: "Whadayuh pluggin' in there?! We'll add that after."

Road Kill: This engineer has mixed live sound for the past 10 years, and can't hear anything over 8 kHz. Favorite quote: "What?"

Mr. Thin: This engineer refuses to believe that such a thing as phase cancellation exists with guitar cabinets. Can't understand why the guitar sound is so thin and lifeless. Fave quote: "Yeah, you're right, it does sound a little fatter at the amp."

Trust Me I'm A Professional: Similar to Master Control. Seldom (if ever) permits the guitar player to work the console's pan, volume, and EQ to get a guitar sound.



Mud Slide Sound: Consistently gets acoustic guitars to sound ultra muddy and smeared due to poor mic placement.

Telepathic Soul: Kind of short on social skills. Biggest danger is that this engineer invariably fails to communicate clear punch-in and punch-out points.

Mr./Ms. Thimk : Like our Clueless friend mentioned earlier, can't keep the tracks straight and "loses" your best take.

The Tunnel: Would never conceive of taking a suggestion or idea from a guitarist as to mic placement, use of outboard gear or anything that alters what's been done in the past. Favorite quote: "This is how I always mic an amp." El Cool Dude: The big-buck engineer the band chipped in for is constantly on the phone lining up the next project.

Journey To Forever: You came in for overdubs and the engineer takes hours to get the mics and the cue mix together.

HEY, IT'S JUST A JOKE

Q: What do guitarists and recording engineers use for birth control? A: Their personalities.

Bennet Spielwogel, AKA, The East Side Flash, owns Flashpoint Recording Studio in Austin, TX. He has been known to frequent both sides of the glass.



An integral tool in many of the world's top recording studios, KRK^{*} close-field monitors have gained acceptunce as the reference speaker of choice. Their extremely smooth frequency response, low distortion and superb imaging is without peer. This incredible performance can be attributed to a unique combination of exotic driver materials and fanatical engineering. Furthermore, each driver is extensively tested and monitor pairs are matched to within one-tenth of a decibel for unequaled accuracy. With a complete family of monitors to choose from, KRK has a monitor for every application, including video-shielded models. KRK, funatical by design.



Distributed by Group One East Coast Office: 916: 249-1199 • EAX (316) 753-1020 West Coast Office: 010: 006-8633 • FAX (316) 577-6407 IRIX Muniforing Systems 10460 Cothard Sc. Unit D Hammagron Beach. CA 92047 714: 541-1000 • FAX (714) 375-6426

CIRCLE 47 ON FREE INFO CARD World Radio History



ASIA & FAR EAST VW MARCETING TEL: +44 (0)1379 798481 FAX: +44 (0)1379 794009

AUSTRIA TON ESCHENGER TEL +431 4865 165 FAX +431 4865 165

BELGIUM AMPTEC TEL + 39 11 98 14 58 FAX +39 11 28 14 59

CANADA SASCOM MARKETING GROUP TEL +1 905 469 8080 FAX +1 905 469 1129

CYPRUS EMPIRE MUSIC TEL +357 2 490472 FAX: +357 2 490863

DENMARK DANISH AUDIO DISTRIBUTION TEL +45 3168 9811 FAX +45 3165 8449

FINLAND SOUND MEDIA LTO TEL +358 0 510 9355 FAX +358 0 510 2257

FRANCE WUSE BUSINESS TEL +33 1 43 38 15 95 FAX +33 1 43 38 70 79

GERMANY GROOVE MUSIC (S.E.A.) TEL +49 5903 93 88 FAX +49 5903 6141

GREECE ECE SOUND CONTROL S & AUDIO & VIDEO SYSTEM TEL +30 1 8837 69930 FAX +30 1 8836 377

HOLLAND AUDIO ELECTRONIES MATTUSEN TEL +31 90 699 04 80 FAX +31 20 699 36 41

KELAND R.F. PA SYSTEMS TEL +354 1 191 44 FAX +354 1 612 144

NDONESIA MULTI AUDIO PERICASA TEL +62 21 629 6009 FAX +62 21 629 8453

ITALY PRODUX SRL TEL +39 2 393 11571 FAX +39 2 393 12609

JAPAN HOOK UP INC TEL +81 33 643 5735 FAX +81 33 367 4778

PORTUGAL NDO MUSIC TEL +351 9 900 4616 FAX +351 9 908 4949

RUSSIA A & T TRADE TEL +7 095 978 9016 FAX +7 503 956 1105

SINGAPORE/MALAYSIA SWEE LEE COMPANY TEL +65 336 7886 FAX +65 339 7085

SPAIN REFLEXION ARTS TEL +34 86 48 11 55 FAX + 34 86 48 90 65

SWEDEN ESTRAD MUSIC TEL: +46 8 643 20 07 OR: +46 8 640 12 60 FAX: +46 8 702 20 16

SWITZERLAND ZAP AUDIO TEL: +41 92 340 05 70 FAX: +41 92 340 05 75

SWITZERLAND NAGRA BROADCAST TEL +41 21 732 01 01 FAX +41 21 732 01 00

TAWAN TILLE SOUND TRADING CO LTD. TEL +886 2 595 8512 FAX +886 2 594 0032

SASCOM MARKETING GROUP TEL: +1 905 469 8080 FAX: +1 905 469 1129



ADD THE WARMTH FEEL THE GLOW

VII 8 CM

14 O.

n

"I first tested the TL Audio VIin conjunction with a Yamaha ADRX A-D convertor and a Sony 3324 digital multitrack - I was so impressed with the warmth and oth bottom end that the unit smooth bottom end that the unit added to the rewording that I have ordered a VI-1 for our in-house studio and for our hire stock." Ian Silvester Digital Audio Technology, London

TL Augro

Give your recordings the rich, warm glow of Tube Technology, a series of tube based mixers & signal processors. Take for example the new VI-1 interface, which offers 8 channels of tube warmth making it the ideal partner for digital multitracks, hard disk systems, stereo mastering & copying. Then there's the superb new EQ2 - a sophisticated, twin channel, parametric 4 band Tube Equaliser. Both these new additions offer the smooth sound, very low noise floor, quality construction & everyday affordability that only TL Audio can deliver.

5.

\$795 VI-1 8 Channel Tube Interface Line amp, balanced ins & outs, unbalanced compatible, level matching for -10dB & +4dB equipment. . 3 1 6 \$5995 8:2 Tube Mixer 99.0---- Or ... \$595 Dual Tube Mic Pre-amp/DI 20 -... \$1695 Dual Pentode Tube Pre-amp

NEW!

THE TL AUDIO

..............

\$2395

......

EQ2 Stereo Tube Parametric EQ

4 band parametric, variable filters,

mic-amps, +48y phantom power.

dual mono or stereo linked modes

TUBE RANGE

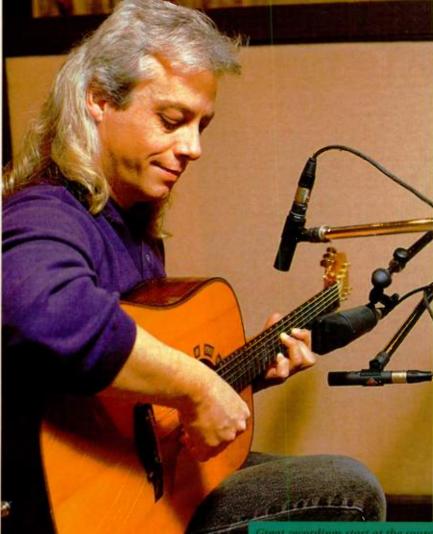
10000 0000000 \$1595 **Dual Tube EQ** ------\$1795 Dual Tube Compressor SASCOM GROUP For US & Canada Sales Tel: +1 905 - 469 8080 Fax: +1 905 - 469 1129

Worldwide Distribution Tony Larking Professional Sales Ltd. Letchworth, SG6 1AN (UK): Tel: Ha4 (0)1462 490600 Fax: +44 (0)1462 490700 CINCLE ON AN EREE INEA CARR

To feel the glow contact your nearest authorised dealer.

Acoustic Guitar Miking Techniques

How to best capture the natural sounds of an unplugged six string BY EDWARD GERHARD The guitar that I played on Luna was built by Ervin Somogyi (Berkeley, CA), and it is a very unusual-sounding guitar. I think that's one of the things that contributes to the different sound of Luna. A lot of guitar records you hear are very crisp, crystalline, and bright. This guitar has some of those characteristics but it also is very round and thick-sounding, almost like a nylon-string guitar. And



that is the texture that I wanted to capture on these records.

I recorded the pieces for Luna in a couple of different rooms. One room was fairly large with a high ceiling. The other is a relatively small, rectangular shaped room — basically an extra bedroom in my apartment. Both rooms were rather dead acoustically and that, along with close-miking, helped to keep the guitar sound consistent.

Ninety-nine percent of the basic guitar tracks were recorded with a pair of Neumann KM 140 cardioid condenser microphones through a Mackie MS1202 mixer. This was before I got my Alesis ADAT, so I just used the Mackie mic pres and recorded the guitar in stereo straight to a Panasonic SV-3700 DAT machine. On a couple of songs those stereo guitar tracks were bounced over to an ADAT so that we could add overdubs. While a lot of songs on the CD were just solo guitar, there was some overdubbing. On a few tunes I played a second guitar and on other tunes there were trumpet, upright bass, or drum overdubs. For those purposes we transferred straight into an ADAT using the analog input and I didn't feel like there was any loss in the transfer. If I had heard any problems or differences in the audio I would have tried something different. I tried to be as meticulous as I possibly could where the audio was concerned - I listened very carefully to the transfers and they didn't pick up any noise and I didn't hear any generation loss.

MIC PLACEMENT

I used two Neumann KM 140's to mic the instrument and I placed the first mic on the neck at about the 18th fret. Sometimes it was pointing straight in at the neck and sometimes I would angle it a few degrees. If I was getting a lot of string noise I would face it slightly away from the neck and point it more towards the sound hole. Or if I could hear myself breathing, I would

Great recordings start at the source — any experienced engineer knows that. If you take a fine instrument and put if in the hands of a competent musician you are more than half way to a great sound. Just ask Edward Gerhard, whose acoustic guitar sound on his third CD Luna has gotten rave reviews. Acting as artist, producer and engineer while recording Luna (as well as his previous release Christmas and his soon-to-be-released Counting The Ways) Gerhard has honed in on a wonderful technique for recording his guitar.



Wally Traugott Capitol Records



Pat Scholes Ardent Studios



Mark Egan Wavetone Records



Wayman Tisdale Recording Artist



Erick Labson MCA Music Med a



Brilliant Award Winning Amplifiers

Musicians • Broadcasters • Engineers • Producers... It's not a coincidence! Thank you from your friends at Hafler!



 P5000 Winner Technical Excellence and Creativity Award
 P1500 Winner Radio World Cool Stuff Award National Association of Broadcasters



P7000 Trans•nova D.I.A.B.L.O.



9303/9505 Trans•nova D.I.A.B.L.O.



Tom Jung DMP Records



Ronnie Montrose Recording Artist



Pat Weber Engineering



Chaton Recordings



Rusty Milner Marshall Tucker Band



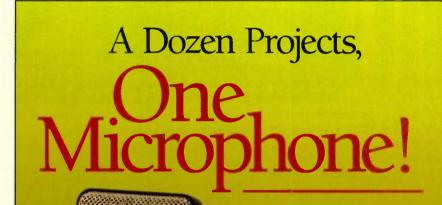
HAFLER PROFESSIONAL, A DIVISION OF ROCKFORD CORFORATION • TEMPE, ARIZONA 85281 USA 1-800-366-1619 • INTERNATIONAL FAX 602-567-8132 • IN CANADA, CONTACT CABLE TEK 604-942-1001 CIRCLE 3/2001/Perise/Infeg/CARD

TECHNIQUES RECORDING

angle the mic down a little bit towards the guitar. But it was pretty much onaxis and no more than six inches away from the neck. The second microphone was in a sweet spot just behind the bridge, pointing basically straight in at the soundboard, and, ocassionally, I would move it slightly off-axis.

I wanted a really big guitar sound, but I have heard many guitar records that have an almost exaggerated degree of intimacy to them, which I didn't want. I wanted to go for the "cinematic" approach to the guitar sound where it would be filling the entire frame but it wouldn't be in your face. Each of these mics was recorded to a separate track (panned hard left and hard right) and they stayed panned in exactly the same position through the mixdown.

The microphone near the bridge area was also about six inches away, and I think it's important to keep the mics pretty close to the same distance away from the guitar. Otherwise, you will start having phase problems



OVER HI-PASS PAD



Vocals, acoustic and electric guitar, piano, tom toms, snare, cymbals and oh yes... even kick drums. This uniquely designed condenser microphone features a servo head amplifier delivering unparalleled transparency and sonic performance, allowing you a level of flexibility not available from any other microphone.

... and all this flexible performance can be yours at a price making owning a dozen a wonderful possibility.

Also available in matched pain



World Radio History

The incredible Equitek E-100 Microphone from CAD.....Who Else!



For an authorized CAD dealer near you call! 1 800 762•9266

Invision of C11 Audio, Inc. 341 Flattor St. Compean, OFL 1405

CIRCLE 22 ON FREE INFO CARD

because the amplitude in different areas of the guitar is going to be different. I listened really closely for phase cancellation and if there were any problems with phase, they were very minor. My way of thinking is that unless the sound is totally collapsing in mono — for an acoustic guitar record — you are generally pretty safe. Acoustic guitar records are not getting played on hit radio stations so you don't have to worry that much about what it will sound like on somebody's clock radio. I would sacrifice a little bit of phase coherence for a killer tone.

A HINT OF EFFECTS

The only time I processed the guitar was during mixdown where I would add a touch of reverb. I used a bit of Lexicon PCM 70 on most of the tunes - sometimes for a little ambience, sometimes for a wetter reverb. On pieces like "Luna" and "Howl," I knew I'd be using a bit more reverb than on the other songs, so I increased the distance between the mics and guitar when I recorded them. To me, a tightly miked acoustic guitar swimming in a big spacious reverb is an audio oxymoron. The presence is fighting with all the space you're trying to create with the reverb. Back off from the mics a little (even two inches can make a big difference) and the reverb sounds more real.

During the mastering stage we did not normalize or compress, although there were a couple of peaks I cut back so that the overall level of audio on the master tape could be raised. There was one brief bass note on "Red Mountain Serenade" that we trimmed with a little EQ. The guitar got a little bit thumpy during one particular passage so basically we punched the EQ button in, knocked the note back a couple of dBs, and then punched the EQ out for the rest of the song. Except for that one note there was absolutely no EQ and no compression. The sound of the Neumanns through the Mackie mic pres was exactly what I wanted.

NEW WAYS

My new CD will be called *Counting The Ways* (an album of all love songs) and I'm recording the guitar with a slightly different mic setup. I am using the KM 140's again, but I have added a Neumann TLM 193. One of the KM 140's is pointing straight towards the neck at around the 18th fret, but closer to the treble side of the fingerboard. The other KM 140 is raised up above the

NOU DON'T NEED BBE <u>Until you hear it!</u> Then your point of reference is forever changed.

"There is an added fullness and clarity. Stereo program becomes more three dimensional, more spacious. Dense textures become more pleasurable, with a more palpable space around each discrete instrument. This is powerful stuff. Maybe its voodoo?"

"As close as we've seen to a magic black box, the BBE 462 is probably the most cost-effective improvement you can add to your rig."

> Listening on a variety of loudspeakers – JBL, KRK, Meyer, Tannoy, Yamaha and a few lesser-known names – I discovered that the BBE Process provided a nice punch to older recordings without being brittle or shrill. The LF contour emphasized the bottom end, without undue boominess."

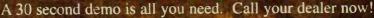
"All of the mixes we played gained extra sparkle and clarity when processed with the BBE 462."

erboard Magazine

HAT YOUR EARS HAVE BEEN

PROCESS

0-E-9-8





5381 Production Drive Hun

Huntington Beach, CA 92649 (714) 897-6766 CIRCLE 9 WordFREE-INFO (ARD Fax (714) 896-0736

R

Edward Van Halen, the master of of Van Halen's tenth album and the

By Steve La Cerra

World Radio History

fretboard magic, discusses the recording part his project studio played in it

Q: How did the songs for *Balance* come about? Did you write them while recording or during preproduction?

Edward Van Halen: I guess it really started when our manager passed away from cancer in October of 1993. My brother and I were so shocked that we gravitated to something we know, which is making music in the studio. And since I have my project studio, Alex and I just locked ourselves up and lost ourselves in making music. I let a cassette roll as I was writing. At the time we were also looking for a producer and we met with a few guys like Mike Klink, Bob Rock, and Andy Johns. Then we met with Bruce Fairbairn and he was a great guy. He walked in and said, "Play me some stuff." We had about 20 songs in rough form — some completed, some basic ideas on cassette, and some we just played live for him. Bruce asked when we wanted to start and we said, "Whenever you want to." So he went home to tie up some things — I think he was just finishing mixing a project. We started around June 1st. Out of the 20 songs, we cut about 16, 11 went on the record, and we have four or five left over. We always do that.

World Radio History

F BALANCE

Were those extra songs actually finished?

Everything except for the lyrics. When I get on a roll or focus on writing I come up with a ton of music. I don't write the lyrics - Sammy [Hagar] writes the lyrics. He actually said, "Hey Ed - slow down. I can't keep up." So we have some completed tracks without vocals on them. 'Baluchitherium" was one

of the ones that wasn't completely finished lyrically, and we decided to just leave it off.

That's how it ended up being an instrumental. It wasn't intentionally written as an instrumental. Bruce said, "Let's see how it sounds without anything." And we all thought it held up pretty well so we said, "What the f-?"

What about that drum machine in there?

That was actually a little ditty that I had written years ago, and it was a kind of cool way to end, so we just tagged it onto the end of "Baluchitherium." It was a really old Roland drum machine.

Do you find that you eventually go back to the music you have finished musically but not lyrically?

A lot of times we do, but generally we get more excited about the new stuff every time out. On every record there is something [older]. Like on the previous album, For Unlawful Carnal Knowledge, "Right Now" - I had written it years ago and no one really seemed to be too interested. There is a time and place for everything. Sometimes I'm into something that the other guys don't really care for at the time. I'll play it again for them a couple of years later and they'll go, "Wow, what's that?" And I'll say, "That's the same thing I played you five years ago," and now they'll think it's really cool. So sometimes we're not all in tune at the same time on the same type of music. But eventually everything surfaces. Every record we make is about 80 percent brand-spanking new material written for that record and sometimes we'll have something old laying around and we'll try it.

Do you record the rhythm tracks live? Oh yeah; Mike, Alex, and I play everything live, and a lot of times, Sammy will scat along. Most of the time when we are tracking, Sammy is not done with the lyrics yet. But on songs like "Don't Tell Me" I'll play the solo on the rhythm track, and then if it needs any overdubs, I'll do a rhythm part underneath it later. That's the way we have always done it. I have walked in on some other band's sessions and they're overdubbing a kick drum or they start out with the drum track alone, then overdub guitar and then bass and I say, "Don't you guys ever play together?" And they say, "No you guys do?" Well isn't that what a band is supposed to do? And that is why our music breathes.

And I'll assume that you don't have a click track going either.

No. It's really funny, because with "Big Fat Money" that was the running joke throughout the album. I guess a lot of bands that Bruce works with use a click, and so Bruce asked Al, "Do you want a click on this one?" Al said, "I don't need a stinking click!"

So the band is just going for it.

Well, yeah. Our stuff isn't perfect but that is how you lose the energy and the



WE'VE DISCOVERED THE MISSING LINK IN THE SEARCH FOR NATURAL-SOUNDING COMPRESSION.

The New AUTOCOM MDX 1200 with Dynamic Enhancement

Compression, particularly when applied heavily, can "dull" the sound of your signal. That's because the lower end of the frequency spectrum dominates the action of any broadband compressor, causing high frequencies to "disappear." The new AUTOCOM stereo Compressor/Limiter/Expander/Enhancer has been designed to solve this problem. Our exclusive new dynamic enhancer circuitry restores the perceived loss of higher frequencies that occurs during compression, ensuring your mix always retains its original brightness and energy. Exceptionally versatile by design, the AUTOCOM now also features a unique IRC (Interactive Ratio Control) Downward Expander/Gate which reacts intuitively to the program material to produce an exceptionally natural sound.

Built-in Auto and Manual Compression modes give you more options—from "set and forget" to a wide range of fully adjustable Attack and Release times for creative effects. Our unique Wave Adaptive compression circuitry delivers "inaudible" compression every time you use it.

AUTOCOM from Behringer. The dynamic new way to put life back into your music.



Your Ear Is Our Judge

For a copy of Behringer's dynamic full-line brochure, please call (516) 364-2244, fax (516) 364-3888 or write Samson Technologies Corp., Box 9031, Syosset, NY 11791-9031. Behringer is exclusively distributed in the U.S. by Samson Technologies Corp. @1996 SAMSON

CIRCLE 10 ON FREE INFO CARD

World Radio History

LEARN THE ART OF RECORDING You can get the practical, real-world

skills needed to successfully start your career as a recording engineer, producer or studio musician. •Hands-on approach, learning by doing • Previous experience not required • Complete 300 hours of training in less than 2 months •6 studios filled with the latest equipment •Small classes, excellent personal attention •Job placement assistance • Financial aid available Low cost, on-campus housing

For free brochure, call or write today 1-800-848-9900 1-614-663-2544



THE RECORDING WORKSHOP Recording 455-Q Massieville Rd WorkShop Chillicothe, Ohio 45601 his Staty Mard of Proprietary School Registration #80-07-0696T

CIRCLE 77 ON FREE INFO CARD



Vocal Booths Amplifier Isolation Enclosures Practice Rooms Tel: 423-585-5827 Fax: 423-585-5831 116 S. Sugar Hollow Rd. Morristown, TN 37813 USA

CIRCLE 93 ON FREE INFO CARD

F BALANCE

feel of it - by making it so sterile that it's perfect. I don't think that is what music is all about. What we try to do is to capture a moment and a magic. At the end of the day, if we feel like we have done that and there's a mistake in it or it speeds up or slows down a little, who cares as long as it moves you.

Since you do have 5150 [the name of Edward's project studio), if you are having a bad day on a particular song, do you come back to it the next day?

We never do more than two or three takes of a song. If we don't capture it, we move on to something else and then come back to try and nail that one the next day or whenever. A couple of times, I remember Alex and I saying, "We can do that one better," and Bruce would say, "Leave it - it's

fine." It is all about your emotions and what you are feeling at the time. If we went into the studio right this moment and re-recorded the album, every song would end up sounding different. That's the beauty of performing live. It's why playing songs off the first record on stage is exciting because every time we play them they're a bit different. It's really not just all about the notes. It's about the feeling behind the notes and what's coming from the heart.

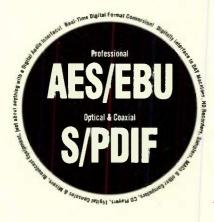
On "Can't Stop Loving You," what is happening with the guitar sound. It seems like there are a lot of layers. It's one guitar with two sounds. The guitar has a piezo pickup in the bridge and it also has the regular humbucking guitar pickup. So it's an acoustic pickup and an electric guitar sound coming from the same guitar.

What's the story with "Strung Out?"



World Radio History

Professional Digital I/O for Windows



Presenting Multi!Wav Digital PRO: The ultimate card for professional direct digital transfer in Windows.

Use Multi!Wav Digital PRO to make digital masters, digital mixdowns, sample edits, and direct transfer of audio files with NO GENERATION LOSS!

Use Multi!Wav Digital PRO for real-time digital format conversion!

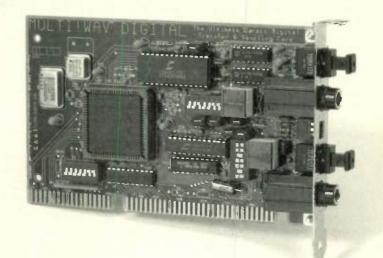
Multi¹Wav Digital PRO is future upgradable! Add NEW HARDWARE features, such as 24 bit audio, from software updates!

Control Multi! Wav Digital PRO with your favorite Windows WAV editor, such as SAW, Sound Forge, Wave for Windows, ...

Multi!Wav Digital PRO installs easily into your EXISTING setup. It works side by side with the hardware you already own!

Multi!Wav Digital PRO: It's the ultimate card for prafessional digital I/O in Windows.

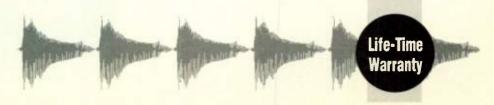




Compare the Superior Features of Multi!Wav PRO

MultilWav Digital PRO Features	What you can do with Multi!Way Digital PRO:	Multi!Wav*	CardD*
Professional AES/EBU Digital I/O Capability	Direct Digital Sampling and Transfer with any AES/EBU connection. Interface to Samplers, HD Recorders, Digital Consoles, Broadcast Equipment	YES	No
 Optical & Coaxial S/PDIF Digital I/O Capability 	Direct Digital Sampling and Transfer with any S/PDIF connection. Interface to DAT Machines, CD Players, Samplers, Macs & other computers, Digital Mixers	YES	Just Coaxial
 Real-Time Digital Format Conversion 	Convert Optical to Coaxial, S/PDIF to AES/EBU, or vice versa, in REAL-TIME!	YES	No
 Software Upgradable Hardware Design 	Add NEW HARDWARE features from software updates, such as 24 bit audio & more!	YES	No
 I/4" Phone Jacks for Coaxial S/PDIF 	MuitilWav uses pressure-contact 1/4" phone jacks. CardD uses single-ended RCA jacks.	PHONE	RCA
 Shielded Digital Audio Transformers 	Reduce common-mode noise and clock phase jitter.	YES	No
 Flexible Ground Strapping for Pros & Audiophiles 	Achieve optimal ground configuration. Reduce common-mode noise and clock phase jitter.	YES	No
I/O Overload Protection to 50 VDC	Protect against accidental misconnection of inputs and outputs up to 50 Volts DC.	YES	No
 256/512 Bit Upgradable RAM FIFO Buffer Option 	Improve performance by reducing sensitivity to hardware underruns & overruns.	YES	No
LC Filtered Analog Supply	Improve receiver PLL performance by reducing noise.	YES	No
Multi-Layer PCB	Improve performance by reducing "digital" noise.	YES	No
24mA Bus Drivers	Meet all IBM PC Bus requirements.	YES	No
► 64 Selectable Addresses	Eliminate hardware installation conflicts.	64	Just 2
Professional Sample Rates	48kHz, 44.1kHz, 32kHz (software selectable)	YES	Yes
► Life-Time Warranty	Protection against manufacturing defects.	LIFE-TIME	1 Year

* Comparison performed May 1995 using MultilWay Digital PRO (rev A) from AdB International and Bigital Only CardD (s/n 00009159 rev B) from Digital Audio Labs MultilWay Digital PRO requires one 16-bit ISA bus slot; a computer that meets the hardware requirimints specified by your Windows WAV editing informational and divital I/O cables. AdB and MultilWay are brademarks of AdB International. All other trindem rick are the property of their respective owners. Made in USA.



For more information please contact AdB International • 2180 Pleasant Hill Rd. Suite A-5100 • Duluth, GA 30136 • Voice: 770.623.1410 • Fax: 770.623.1429 • E-Mail: Adbdigital@Aol.com

CIRCLE O4 ON FREE INFO CARD

F BALANCE Back in 1983. Valerie and I had rented Marvin Hamlisch's beach house for the summer. We wanted to see if we might actually buy a place out in Malibu, so we thought we'd rent one to see if we even liked it out there. Marvin had this beautiful white Yamaha grand piano there. I used to drink a lot, and I got pretty wasted and started dismantling his piano with forks and knives and plucking on the strings. I bounced ping pong balls off the strings and used D-cell batteries to play

slide and I recorded about six hours of cassettes. I put a bunch of cigarette burns in the piano. I basically had to restring and repaint the piano before we left. I don't think he ever knew that anything had happened because I pretty much put everything back together. But if he ever reads any of these interviews he'll know.

So that was recorded at his place. Yes. When I had written "Not Enough," Bruce asked me if I had any weird stuff that I could use for an intro. I asked him, "Well, how weird are you talking about?" And he said, "Well, I don't know, what have you got?" So I handed him six cassettes and he picked 60 seconds from one of the cassettes and that became the intro to "Not Enough."

Do you normally record the piano tracks at 5150?

I have two houses up here: one right

World Radio History

by the studio and our main house, which is away from the studio. When we built the house, I ran cables underground to the studio so I can record in any room of my main house. I was wearing headphones, playing piano in the living room of my main house while Al and Mike were up here in the studio. We played it live.

So you obviously had the foresight to have all the tie lines run.

5150: WELL EQUIPPED

For the complete dope on the gear at 5150, EQ spoke to studio manager Ken Deane who had this to say:

"We're using two Studer A820-2-24-track, 2-inch analog machines. Although these machines are equipped with Dolby SR, we normally don't use it. The console is a 48-input, 24-monitor, custom design by Brent Averill based on API op amps and equipped with GML automation. The control room monitors were designed by George Augspurger and are powered by H & H 800 amplifiers. We have the usual complement of outboard gear, including UREI 1176 limiters, GML stereo EQs, and AMS delay lines. We also have an assortment of microphones including Shure SM57's, Neumann U87's, and some vintage AKG C-12's. This studio is for private use only, so don't call us to try and book time!"



Anyone who has heard it knows ...



Vacuum Tube Microphone Preamplifier

D.W. FEARN

610-793-2520

P.O. Box 57 · Pocopson, PA · 19366 · Fax 610-793-1479

ART GIVES YOU A MUCH BETTER WAY TO WARM THE SOUND OF YOUR MICS!

If you've been getting a less-than-great, cold, flat sterile sound from your mics or instruments, it's time to warm them up. Get ART's new Tube MP Microphone Preamplifier. You'll suddenly find yourself with lush, rich, and yes, WARM sounding mics. The Tube MP utilizes a hand-selected 12AX7 to impart all of the warmth, and richness you want. Plus, it's got 70dB of low-noise gain, built-in phantom power, and a phase-reverse switch. Use it for digital (or analog) recording or live shows. It's great when used as a direct box (particularly on bass) or with transducer-style or piezo pickups. Best of all, it's inexpensive enough to put one on every mic you own.

PROFESSIONAL TUBE MIC PREAM

70DB OF GAIN HAND-SELECTED 12AX7 TUBE BALANCED XLR & UNBALANCED 1/4" INPUTS BALANCED XLR & UNBALANCED 1/4" OUTPUTS

TUBE-MP

EQUIVALENT INPUT NOISE XLR-IN SPECS: -129DB DYNAMIC RANGE: >100 DB PHANTOM POWER TO 48V 20DB GAIN SWITCH PHASE-REVERSE SWITCH

APPLIED RESEARCH AND TECHNOLOGY

215 Tremont Street, Rochester, New York 14608 USA Phone 716-436-2720 • Fax 716-436-3942 • email: artroch@aol.com For A.R.T.'s full color catalog (and to be put on our mailing list) please send \$2.00, along with your name and address to: A.R.T. Catalog Dept., 215 Tremont Street, Rochester, NY 14608. Be sure to tell us what magazine you saw this ad In, and its cover date.

World Radio History

Clean. Clear. Right.





t's the new Mic Eliminator[®] from Whirlwind, and it sounds just like an SM57 in front of a Twin Reverb.

Plug in from your speaker output, preamp output, or right out of your guitar, and get a clean, open, classic sound, balanced so you can run it directly to your console without a mic. It's at dealers everywhere.



CIRCLE 83 ON FREE INFO CARD

F BALANCE

Yes, and I have a racquetball court in my house that we use for echo.

And you had ties run to the racquetball court as well?

Yes. We use a combination of the real room and digital reverb depending on the music. We can't change the racquetball court — it is what it is. So if we want a shorter decay time, we use a digital unit. But we use the court for mixing and it's pretty cool. We have a couple of speakers and mics in there. Of course, it

drove Valerie crazy cause we'd be working late at night and she'd be going, "Turn that crap down," and I'd say, "Honey we're almost done — only another week." "Another week?"

Do you have any preference for the mic you use on your cabinets?

I use Shure SM57's in the studio. Live I use the Groove Tubes MD-1 to get a slightly warmer tone, but live I use a different set up. In the studio I just use one cabinet and add the effects afterwards. I use two 57's, one directly off of center and the other angled in from the side. I pan those to the left. And to get the signal to the right side of the stereo spectrum I use a Harmonizer and then add a little delay to both sides. But it's just one signal coming from one speaker cabinet, whereas live I use a dry cabinet in the middle and two effect cabinets on the outside, to kind of duplicate the record.

Did you print the guitar tracks dry and then add the effects while mixing?

We always print the dry signal and the Harmonizer. The delay is added in the mix, but I do hear it in the studio off the monitor.

In "Baluchitherium" is there a synth doubling the bass part?

I overdubbed a Danelectro six-string bass, doubling Mike's bass and scraping the strings like I do in the beginning of "Don't Tell Me."

We started talking about Bruce Fairbairn before. What did he add to the project?

I prefer to work with a producer because they tend to have you look at your own music from a different perspective. Bruce is a very musical guy, and for some songs he would just say, "Great," and for other songs he would have arrangement suggestions. We would try them, and if they worked better, we would go that way. If not, then we always had the way we originally started.

If you ask Bruce what he did, he'll say, "Well I watched Eddie plug in, I watched Al grab his sticks, and I had fun watching them play." But being a producer is everything from being a babysitter to being an amateur psychologist to you-name-it. I think a lot of times that if Sammy was going somewhere lyrically that the rest of us didn't like, we would go through Bruce to make suggestions to Sammy. A lot of times if I would say it to Sammy he'd say, "Hey I don't tell you what to play!" So Bruce was kind of a mediator.

I guess the main thing that he brought was organization, because if we were left on our own, I'd want to work on one thing on one particular day and Alex and Sammy will want to work something else. Bruce would say, "No, this is what we're doing today guys." He'd have it all mapped out and say we're doing these overdubs today and the backgrounds tomorrow. He wouldn't let us get away with, "I don't want to do this today." So it was one of the fastest recordings we have ever done - it took us four months for writing, recording, and mixing - and it's because Bruce is very organized.

I like someone who lets me express my own ideas and get everything out that I want to before they make changes or suggestions. Bruce is very good about that. There are some producers — I won't name any names — who want to feel like they have contributed somehow, and before I have said what I want to say with my music they are already tearing it apart and saying, "Why don't you do this or try that?" And I haven't finished my thought. I think the producer's job is to get the best out of the artist. To me, if the band doesn't have it together on their own yet, then what's the point?

Bruce actually made me very nervous in the beginning. I was in the process of trying to stop drinking, and drinking has always been an easy way out for me. I got nervous a few times and he is such a great guy that he could tell and would just leave the room to let me be on my own. It was a great working relationship because he is very sen-

DIGITAL 6 TRACK RECORDER

Digital mixer with 3-band digital EQ 4 Aux Sends and Returns built in Full editing functions—move, copy & delete Synchronize any number of HDR-6 units MIDI Sync, mixer control and Auto Mix MIDI Machine Control Auto Punch In/Out, with undo Auto location for instant access Real time pitch control 2nd internal flard Disk capability Direct Out mode for external mixer Easy back-up and loading 18 bit A/D conversion 20 bit D/A conversion Option slots for expansion

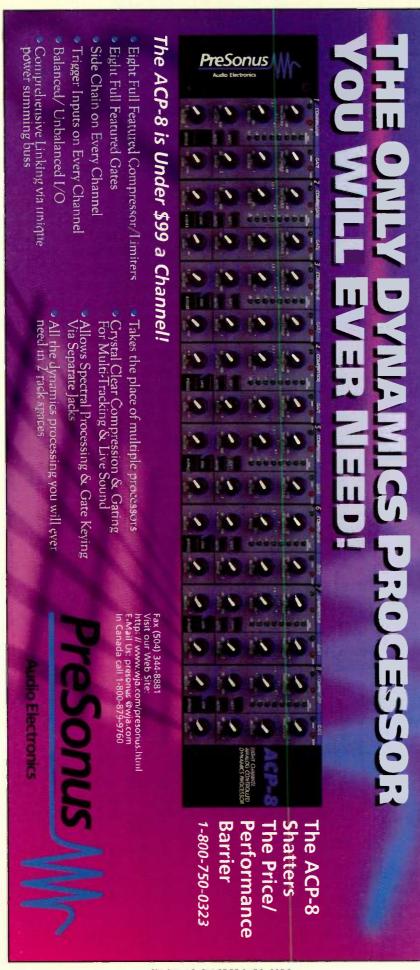
World Radio History



VESTAX CORPORATION 2870 Cordelia Road, Ste. 100 Fairfield. CA 94585 Phone (707) 427-1920 Fax: (707) 427-2023

Cubase is a trademark of Steinberg CIRCLE 81 ON FREE INFO CARD

MANUFACTURED IN THE US



F BALAR sitive to the artist. If there were any guitar overdubs to

were any guitar overdubs to be done, I did them on my own at night after everybody had left. All of the recording was done here except for a couple of the vocals which were done at Bryan Adams's house in Vancouver. Bruce lives up there and we told him we wouldn't mind going there so he could spend some time with his family instead of being down here every week. I think we did three lead vocals up there and the rest of the recording was done at 5150.

We mixed at Record Plant.

Why mix at the Record Plant?

That is where (remix engineer) Mike Fraser was comfortable. If Erwin [Musper, tracking engineer] would have mixed it, he probably would have done it here because he is familiar with my studio. Mike Fraser didn't do any of the recording, so he wasn't familiar at all with my studio. He said he would rather mix it where he was comfortable. The beauty of my studio is that basically everything you get out of here is true to what you are hearing, so you can take it anywhere and it sounds the same.

How involved were you in the mixing of *Balance*?

Mike Fraser and Bruce were mixing it down at Record Plant, and Al and I would go down and listen to what they were doing. On the very first day I think they had mixed "The Seventh Seal" and "Don't Tell Me," and it sounded completely different than it did up here. I said, "What the hell did you guys do?" They added a bunch of effects that just totally got away from the basic sound and I said, "Turn that s--- off." To me the more raw, the better. I always wondered why, if I take a rough mix it sounds great, but after it's mixed the punch and the power is gone. The more effects you add the more you cover things. So we took all the effects off the drums. There are no effects at all. That is the way Al's kit sounds in the room.

Can you tell me about your studio? It has an API console built by Brent Averill and it has GML automation. I have two self-aligning Studer 24-track

78 FEBRUARY EQ

CE

machines. We ended up only using one because we recorded real basic for this album.

Have you used two machines on past projects?

We did on a couple of songs for the previous record and I think we went a little bit overboard. I hate locking up two machines and making slaves because there are always problems.

What was the deciding factor for you to get your own studio?

I don't want to get into negative stuff, but it stemmed from a lot of aggravation with Ted Templeman and Dave [Lee Roth) at the time. They were so into redoing old hits, and our album Diver Down was the last straw for me. It was half cover tunes and I was sick of that. I'd already written "Jump" at the time and these guys wanted nothing to do with it. So I thought, "F--- you. I'll build my own studio and the next record is going my way [laughs]." My own philosophy is that I'd rather bomb with my own music than make it with someone else's. Then all of a sudden "Jump" became a number one hit and the biggest record we've ever had. Then Roth quit because he didn't dig the fact that I was taking control, I to do things my way.

How has it helped the band?

In every aspect. We have a place where everybody is comfortable. Even Bruce loves this place. He just got done producing Chicago and asked if he could use the studio for some of their overdubs. So I said, "Sure, knock yourself out." The place is open to my friends for free so go ahead and use it. I don't charge them any money because I'm not allowed to. They would close me down if I take the business away from anyone else.

For you what makes a great record?

Vibe. As long as it feels real and at the end of the day when we listen back it kicks our ass and we're excited about it. The most we can hope is that it moves somebody else and we really don't let anything out that doesn't move us. We stay true to ourselves and make music that we enjoy making. We've never followed any trends, and that's what I plan on doing until I don't feel like doing it anymore. And I don't see that coming in the near future.



—All Major Brands—



- Reel to Reel Decks (2-4-8-16-24 Track)
- Cassette Decks
 Mixing Boards
- Mikes
 Studio Monitor Speakers
- Special Effects Units Accessories
- Digital Recording



CIRCLE 65 ON FREE INFO CARD



Have an axe to grind about recording guitars? Here's some new gear that'll change your mind.

1

c

World Radio History

0

				SETUP		ENGINE		ENGINE 2		COMBINED	+2 54	APSHOTS	CONTROL		ADJUST
ON		OVERLOAD	THYPE DECRY	100	VO I		RECALL	1 and 1	RECALL	And in case of the local division of the loc	RECALL	and a lateral second		TAP	
OFF	-) =		NEU-I-IZ.SH	HUP		WIZARD		ALC: NO.	1	WIZARD		TAKE I	CANCIL		and the second se
	■ 4 ■ 4		ANY AND Hall	10000	ROUTING	the state of the s	STORE	Terr and	STORE	-	STORE	2	land of the land	1 9HFT	A CONTRACTOR OF
L.C. electronic M2000		HIDLIN	ALL PROPERTY AND A LOCAL DIST.	NAME:		490,000		ani moni		COLUMN TWO IS NOT	_	Hat 2	-		and the state of the
	— 34 —	CAND	CHERTER AD SHOT 1 241		LEVELS	the second second	EDIT		EDIT.	Margaret and	EDIT	3	the second second	A .	and the second se
	-40 🔜	TEHPO EDITED	Mide Stereo Chorus	AUTO LEVEL					-	-		TAKE J	NGE UP	CURSOR 4 P	
	- 4 🖸	EDITED		and it is a lot of the	UTL/HIDI	and the second	BYPASS	100	ETTA	-	ETTRASS	4	-		
				HER HONTOR								LAKE 4	INGE DOWN		

ART + SCIENCE = MAGIC

The Engineering Group at TC Electronic were given carte blanche to create the optimal studio signal processor. Being musicians and studio engineers themselves, they know what is important and have a feel for what you are looking for. Working on the borderline between art and science, they created the M2000.

Designed for the artist within you, you will create magic when using the M2000:

Superior Sound Dynamic Morphing #1 in Digital Ultimate User Interface

UNCOMPROMISING EFFECTS

The library of proprietary effects includes reverb, pitch shift, delay, chorus, ambience, equalization, de-essing, phasing, compression, gates, expansion, limiting, and stereo enhancement.

Among the reverbs is the unique Co-efficient Optimized Room Emulator - or C.O.R.E. Reverb[®] - that is built on a whole new concept for reverberation; the result of years of intensive research. The smoothness, intensity, and density must be experienced to be believed.

The array of enhanced pitch shift (up to 8 voices), chorus, and delay effects are characterized by their precision and versatility. Everything from

the fine and subtle to the wide and spectacular is handled with equal superiority. The algorithms in the dynamics section (compressor, limiter, expander, gate, and de-esser) are unique as stand-alone effects, but are particularly useful in combination with other effects. Those might be de-esser/room, gated hall, or compressed pitch. The possibilities are endless.

Using the Tempo Tap function, you can match the effect to the beat. The tempo can be adjusted in beats-per-minute and subdivided any way you like, for example in triplets. The tempo can also be read from MIDI.

A preset 'Gliding' function (called 'Morphing' by others) has been added to ensure seamless transition between effects. It is a useful function in live and mixing situations.

But another unique feature in the M2000 is the *Dynamic Morphing*^{®4}. By applying this function, you will be able to apply, say, Chorus to a vocal at low volume levels, while the effect is being morphed into Flanging at high levels. Or any other combination of your liking. This will give the signal a totally new dynamic dimension.

The powerful true dual engine configuration means that you can run two full-blown effects simultaneously, i.e., on two individual effects sends. There is 250 factory presets, and another 250 of your own settings can be stored in the unit. You will find that creating new presets is a breeze. The effects have been developed for optimal sonic quality, with no compromises made. This has only been possible because our engineers succeeded in combining the latest technology with TC's powerful DARC[™] chip.

#1 IN DIGITAL

As the only signal processor in its category, the M2000 is fully prepared for the digital age. Not only does it have the digital I/O connectors S/PDIF and the professional format AES/EBU, it also has dithering. Dithering ensures that the dynamic range in your M2000 is kept when you digitally connect it to a device with a lower bit resolution, such as a 16 bit DAT. Proper dithering guarantees a clean, undistorted signal even at extremely low volume levels.

COMES WITH A FREE EXPERT

The M2000 comes with a free expert: You!

Naturally, a manual is included with the unit;

however, you will not be referring to it much.

Great care has been taken to create the ultimate

user interface. With the Recall Wizard you can

instantly find the preset that suits your need in

any situation. You simply enter your preferences, such as 'gentle reverb for vocal' or 'extra pitch for guitar', and the Wizard serves you a short list of presets ready to compare and use.

The single-layer interface ensures that all parameters are always only one menu-level away. No more paging through endless menu-levels. And much, much more...



THE SOUND, THE SOUND

As you can see from the back of this brochure, the M2000 has impressive technical specifications.

However, numbers don't tell the whole story. What really matters is how your final mix sounds. This is where TC's long experience in high-end studio signal processing (with industry standards like the TC2290 and the M5000) becomes important. Our feel and experience have been fully utilized to your benefit in the M2000.

You deserve the sensation of the M2000.

Call 1 800 798-4546 for information about your nearest Authorized TC Dealer.



TECHNICAL SPECIFICATIONS

Analog Input

Connectors:XLR balance	xed (pin 2 hot)
mpedance:	15 KOhm
Vax. Input Level:	+22 dBu
Sensitivity:	@ 12 dB headroom: -22dBu - 10 dB
A to D Conversion:	20 bit (4 bit, 64 times oversampling)
Dynamic Range:	>105 dB
THD:	0.003% @ 1 KHz, +10dB
Frequency Response:	10 Hz - 20 KHz: +0, - 0.2 dB
Crosstalk:	-60 dB max, 10 Hz - 20 Khz

Analog Output

Connectors:XLR balanced (pin 2 hot)				
Impedance:	100 Ohm (active transformer)			
Max Output Level:	+22 dBu			
Output Gain Range:	0 to -32 dB			
D to A Conversion:	20 bit (4 bit 64 times oversampling)			
Dynamic Range:	>96 dB			
THD:	0.003% @ 1 KHz, +10 dB			
Frequency Response:	10 Hz - 20 KHz: +0, - 0.2 dB			
Crosstalk:	-60 dB max, 10 Hz - 20 KHz			

Digital Inputs and Outputs AES/EBU In/Out: XLR

AES/EBU In/Out:	XLR
S/PDIF In/Out:	Coaxial, RCA Type
Formats:	EIAJ CP-340, IEC 958, S/PDIF (20 bit)
	AES/EBU (24 bit)
Sample Rates:	32 KHz, 44.1 KHz, 48 KHz

PCMCiA Interface

Connector:	PCMCIA Type 1 cards
Standards:	PCMCIA 2.0, JEIDA 4.0
Card Format:	Supports up to 2 MB SRAM

Control Interface

In/Out/Thru: 5 Pin DIN 1/4 inch phone jack

Pedal: General

MIDI:

 Finish:
 Black anodized aluminum face plate Painted and plated steel chassis

 Dimensions:
 19" x 1.75" x 8.2"

 Weight:
 5.2 lbs (2.35 kgs)

 AC Power:
 90 - 240 VAC without switch settings 3 Pin IEC power connector

Radio Frequency Immunity/Interference

RFI/ESD: Conforms to FCC Class B, EN55022 Class B (CE), IEC 801-2, IEC 801-3

Environment

 Operating Temperature:
 32 °F to 122 °F (0 °C to 50 °C)

 Storage Temperature:
 -22 °F 167 °F (-30 °C to 70 °C)

 Humidity:
 Max. 95% non-condensing

Note: All specifications are subject to change without notice

t.c. electronic

you and TC

The Wizard M2000 signal processor will enable you to create Magical Sound. Uncompromising effects processed by the TC DARC[™] chip offer you a clarity, density, and feel beyond anything you have ever experienced before.

You deserve the sensation.

Wizard | M2000 Art + Science = Magic



Show us an engineer who doesn't have to worry about guitars and we'll show you an engineer who doesn't work much. EQ presents reviews of several new tools that will make that job easier and the guitar sound better.



Avalon U5 Direct Box



MANUFACTUFER: Avalon Design, PO Box 5976, San Clemente, CA 92673. Tel: 714-492-2000.

APPLICATION: Interfaces low-level signals and amp outputs to consoles and recording devices.

SUMMARY: It's expensive, but it's really clean. The U5 is the top-of-theline way to get a real-world and og signal into quality digital or analog recording systems.

STRENGTHS: Class A amplification; solidly built; "bonus" tone-shaping options; exemplary specs.

WEAKNESSES: Inflexible EQ; lacl dedicated 1/4-inch output on rear panel; no on-off switch.

PRICE: \$679

EQ FREE LIT. #: 122

A DIRECT BOX is, as they say, a good thing. It can match an instrument with a weak output, such as guitar or bass, directly to a console or mixer by providing proper level and impedance matching. But a direct box for \$679? I was pretty skeptical, and expected to see gold-plated knobs or something on its 2U, half-rack chassis

After opening up the package, though, I must admit I was pretty impressed. This unit isn't for everyone, but it justifies its price tag in several different ways.

SPEC-O-RAMA

The first thing I look for is a highimpedance input (at least 1 Meg) for minimum loading of guitars with nonactive pickups; the U5's input Z is 3 Megs. Next thing I look for is pin 2 hot on XLR connectors (the U5 has two XLR outs, one mic and one line). I figure if a company doesn't know there's a standard for XLR connectors, what else don't they know? Well, pin 2 is hot. There's also extended frequency response (5 Hz to 100 kHz, down 3 dB at 500 kHz), very good noise figures (quoted as -98 dBu line out, -104 dB mic out, unweighted), and +28 dBu of output juice.

Okay, so the specs are real good. But what justifies the extra bucks are the extras. Some of these are subtle, like in/out switches that glow orange when "in," DC-coupled outputs and discrete (not IC) Class A circuitry. (This is the most distortion-free amplifier class, although also the most current hungry—which is why it's used for preamp and small power amps, but not large power amps. This also explains the massive transformer and ventilation slots in the box.) Some of the goodies are much more obvious, such as...

THE GOODIES

If you're trying to get a direct sound from an amp, a front-panel switch selects between the front-panel, instrument-level input jack and a rearpanel, 1/4-inch phone input that accepts speaker inputs up to 400 watts.

Gain is variable from +2 dB to +32 dB in 3 dB steps. The stepped feature is helpful for repeatability.

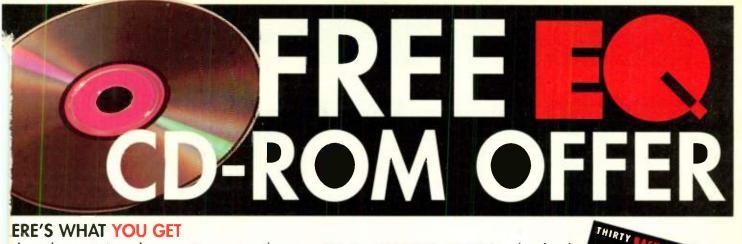
At first I was concerned that you can't interface the U5 with non-XLR devices, but a front-panel, 1/4-inch thru jack can switch between either paralleling the input jack, or providing an amplified and tone shaped output capable of feeding unbalanced, "project studio"-type devices. Speaking of tone shaping, in addition to a high-cut "hiss killer" switch, there are six preset tone curves using passive circuitry (like the tone controls on classic guitar amps). Passive circuitry generally gives kinder, gentler curves than active filters since there's no gain available to sharpen up the response. This kind of filter technology has a reputation for musicality, whether you're talking Pultec or Fender Twin; if you like those kind of gentle EQ effects, you'll love having them built into a box. The tone circuitry can be switched out of circuit if desired.

The manual shows the six curves, as well as suggested instruments that work well with the included curves (e.g., curve 2, which has a fairly steep notch at 1 kHz, is recommended for bass). I agree with most of what they think is appropriate tone-shaping (I often roll off the low end of acoustic guitar, which is one of the available curves). I'm real picky about EQ, and the U5 has very natural-sounding, albeit not very versatile, EQ.

I'm surprised, however, that there's no typical guitar speaker cabinet curve. I think this would help guitarists feel more at home when going direct if they don't want a crystal-clear high end. Also, there is no ultra-sharp notch filter for 50/60 Hz hum, but that's to be expected given the passive circuitry. Overall, the tone shaping does not replace a dedicated equalizer, but it provides some solid, unobtrusive voicings that may obviate the *continued on page 130*



World Radio History



ON A CD-R

ERE'S WHAT YOU GET

ubscribe to EQ right now & receive the new EQ BACK-ISSUE CD-ROM absolutely REE! Order a one or two year sub and receive your complete back issue refernce library on a single dual-platform CD-ROM. This innovative reference tool rganizes all the tips, techniques, products, reviews and opinions interactively - indexed, searchable and instantly available at a click of your mouse.

UT WAIT, THERE'S MORE ...

Ve're talking about EQ's entire past and its future for one single order. And es, you get your very own copy of EQ in your mailbox every month. You're ading this month's issue, so you know how much EQ helps you get the most from our project studio.

O HERE'S HOW IT ADDS UP

ne year subscription gets you 12 print issues and 30 electronic back issues (a total of 42 issues) for nly \$24.95 or (here's some auick math) less than 60 cents per copy. This is the deal of the century. 5 act now. Supplies of EQ BACK ISSUE CD-ROM are limited and offer expires 3/1/96.

- YES	I would like to receive E	of EQ BACK ISSUE CD-ROM with my EQ subscription. (Q 12 times per year (in print) at a rate of Two Years (\$39.95)				
REMASTERING THE BEAT	PAYMENT METHOD: Bill Me (CD will be sent on payment) Check/Money Order Enclosed					
A CHIERROW AND A CONTRACT OF THE REAL OF T		AMEX				
	Name Address	Studio/Company				
Price. all add	s good in US only. Canada add 3 \$30 per year for international a	State Zip				

EQ MAGAZINE • P.O. BOX 0532, BALDWIN, NY 11510-9830

ART MP1 Tube Mic Preamp



MANUFACTURER: Applied Research and Technology Inc., 215 Tremont Street, Rochester, NY 14608. Tel: 716-436-2720.

APPLICATION: Single-channel microphone and instrument preamplifier with tube-based circuitry.

SUMMARY: A tube microphone preamp and direct box affordable by even the most modest project studio.

STRENGTHS: Can be used to interface just about any audio signal to a tape machine; separate input gain control allows creative addition of distortion to audio signal; phase reverse switch.

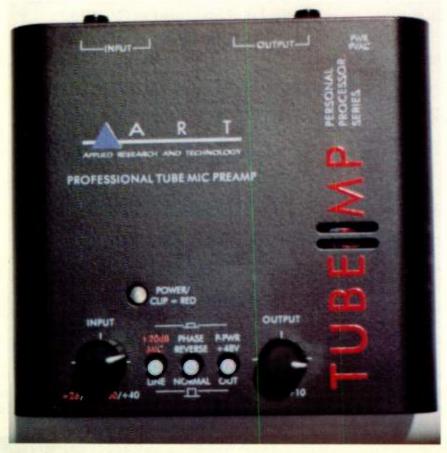
WEAKNESSES: Wall-wart power supply; non-rack mount design adds to console clutter; no front-panel marking for 0 dB output level.

PRICE: \$149

EQ FREE LIT. #: 123

THE TUBE MP1 from Applied Research and Technology is a singlechannel outboard microphone preamp in a compact chassis slightly bigger than a direct box. Similar in profile to ART's MR-1 reverb unit, the Tube MP is designed for table- or consoletop operation. On the rear panel are jacks for a 9-volt AC "wall wart" adapter and both 1/4-inch and XLR I/O connectors. While both outputs can be used simultaneously, ART does not recommend connecting more than one input at a time.

The front panel has rotary pot controls for input gain and output level and push-button switches for +20 dB input gain, phase reversal and



phantom power. There is no on/off switch, so the Tube MP remains powered as long as the AC adapter is connected. ART has provided a bicolor LED that indicates when power is on (green) and when the signal level at the output is 6 dB below clipping (red).

The input stage of the unit employs matched transistor pairs and a 12AX7 tube for amplification, and the gain of the Tube MP normally varies between +6 and +40 dB. When the input gain switch is depressed, that range is increased by 20 dB. At its maximum clockwise position, the output level control can provide an additional 10 dB of gain over that from the input stage, allowing for a maximum gain of 70 dB. I'd like to see a front-panel indication for 0 dB gain position of the output level control. This would make it easier to set the unit for minimum noise and maximum headroom.

Inside the Tube MP are three circuit boards arranged in a flattened Ushape. One of these holds all of the push-button controls, (sealed) pots, and a ceramic tube socket. This board is mounted perpendicular to the main circuit board and is secured to the front panel via nuts that thread onto the shafts of the pots. The main PC board (which holds almost all of the MP's circuitry) is neatly laid out and connects to the smaller boards via ribbon cables. Resistors are of the carbon-type and capacitors are electrolytics. The XLR and 1/4-inch input and output connectors are mounted onto the third board. I am generally not too crazy about PC-mounted connectors, but this circuit board is so small and lightweight that it is solidly held in place by the connectors. All semiconductors are clearly labeled including the TLO72 ICs for the output stage.

In the studio, the Tube MP made a great combination with a Neumann KM-84 condenser mic for recording acoustic guitar. The MP has a slight elevation in the lower midrange, which made the guitar sound really sweet without changing the character of the instrument. I have found that some consoles don't have enough gain

84 FEBRUARY EQ



ART TUBE MP1 MIC PRE SPECS

Input connections: XLR balanced, ¼-inch TS unbalanced Output connections: XLR balanced, ¼-inch TS unbalanced Input impedance: XLR: 2 k ohms; ¼-inch: 840 kohms Output impedance: XLR: 600 ohms; ¼-inch: 300 ohms Max. input level: XLR: +14 dBu; ¼-inch: +22 dBu Max. output level: XLR: +28 dBu; ¼-inch: +22 dBu CMRR: Greater than 75 dB typical Frequency response: 10 Hz–20 kHz (+0, –1 dB) Dynamic range: Greater than 100 dB (20 Hz–20 kHz) Max. gain: XLR to XLR, 70 dB

Equivalent input noise: -129 dBu (A weighted), XLR in to XLR out; 105 dBu (A weighted), ¼-inch to ¼-inch

Power requirements: 9 volts AC @ 700 mA Dimensions (inches): 5.0L x 5.5W x 2.0H Weight: 1.5 lbs

to get the 84's to line level without adding a lot of noise, but the MP could have given me another 10–15 dB of gain. When used with a Neumann U87 for vocals, the MP exhibited that same lower midrange characteristic, which helped fatten up a thin male vocal.

The MP's phase reverse switch is an important feature since many consoles in the project studio price range don't offer this feature, limiting the possibilities for multimiking of instruments. I noticed that the MP's phantom power ramps off very slowly, and it took about ten seconds for the U87 to actually shut down, so a bit of caution should be used when disconnecting phantom-powered microphones.

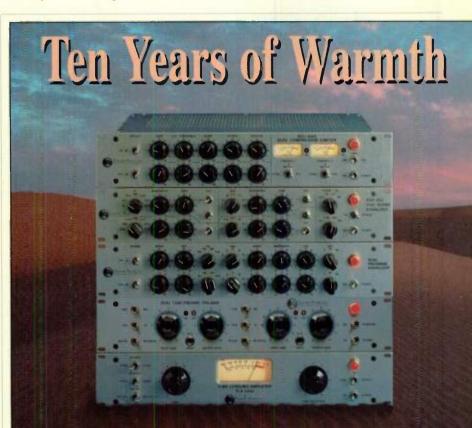
I also used the Tube MP as a direct box/preamp for bass, guitar, and electronic keyboards. Compared to a Countryman Type 85 FET direct box, the Tube MP was just as quiet. On bass, the Type 85 had a more aggressive top end and reproduced more of the pick sound. The MP definitely gave the bass a more "retro" tone with no hint of harshness and a slight coloration in the lower midrange that would be great for R&B sessions. This coloration was also evident on guitar and gave me the idea of using the MP as a front end for recording synths and samplers.

As a keyboard interface, the Tube MP was great at reproducing transients from bell and marimba-type sounds. It sounded faster and seemed to capture the attack more accurately than the console preamp. The only time that the MP's self-noise became noticeable was when we were (intentionally) overdriving the input stage with an electric guitar. Used in this fashion, the MP actually did a credible job of simulating an overdriven guitar amplifier. It gave the guitar a growl that reminded me of Pete Townshend's guitar sound from The Who circa Who's Next.

ART has created a cool little tube

preamp that can be used in quite a variety of recording situations. And at

a list price of \$149, the MP can fit into any studio's budget. —Steve La Cerra



It all started with the Summit TLA-100A tube leveling amplifier. If you bought one back in 1985, it's probably still working. And, worth more now than what it cost new.

Summit's growing family now includes the TPA-200B mic preamp, EQP-200 program equalizer, EQF-100 four-band equalizer, and the DCL-200 dual compressor-limiter. All share Summit's distinctive sound, and, of course, quality throughout.

HEAR THE WARMTH".



Summit Audio Inc. • P.O. Box 1678 • Los Gatos • California • 95031 (408) 464-2448 • Fax (408) 464-7659



CIRCLE 36 ON FREE INFO CARD 86 FEBRUARY EQ

Korg Pandora

PANDORA PX1 is a new type of "personal" processor - it looks like a garage door remote with an LCD, and fits comfortably in a shirt pocket. Yet it's actually a reasonably full-featured signal processor, headphone amp, metronome, and tuner. Connections include a 1/4-inch input jack for guitar, stereo mini-jack for line-level instruments (e.g., stereo tape deck for practicing), on/bypass/mute button, and stereo 1/4-inch headphone jack that doubles as a stereo line out. There's an AC adapter jack, or you can power the unit from two AA batteries for about 8-9 hours.

TAP TOYS

The Pandora's front panel is simple: two sets of Up/Down and one set of Left/Right cursor buttons, Enter button, Metronome button, Write button, and a nonbacklit LCD (if you need to see it in the dark, expose it to really bright light for a bit and the background will glow for a while). There's also an Off/Edit/Play switch on the side — which I found a little too easy to turn on accidentally.

There are 20 editable programs, each with a master output level parameter (there is no input control). Within those programs, you have parameters for:

• Drive stage (12 distortion options or compression)

• Tone (individually adjustable bass and treble)

• Modulation {one of the following, each of which has four options: chorus, flanging, vibrato, phasing, autowah, tremolo, and random (the last is a truly wild, bubbly sound)]. You



can also choose triangle- or square wave-based panning, or a fixed wah sound.

• Ambience (six single delays, six stereo echoes, room, hall, and eight pitch-shift options; however, if you use room, hall, or pitch shift, you can't use modulation).

· Cabinet simulator (four options)

Noise reduction

Each option (except cabinet simulator) has one adjustable parameter. For example, this parameter sets the drive input amount for the drive stage, the threshold for noise reduction, the processed/dry balance for delay or reverb, rate for modulation, pitch shift interval for pitch shift, and so on.

Although your programming options are limited — I kept wanting to turn down the wah resonance (designed to emulate the Vox Wah) or change the mix on the pitch shift, yet these parameters aren't available it's surprising how much variety you can add with a little tweaking. There *continued on page 130*



MANUFACTURER: Korg, 89 Frost St., Westbury, NY 11590. Tel: 516-333-9100.

APPLICATIONS: Recording direct in the studio with processing, as well as private practicing and guitar tuning.

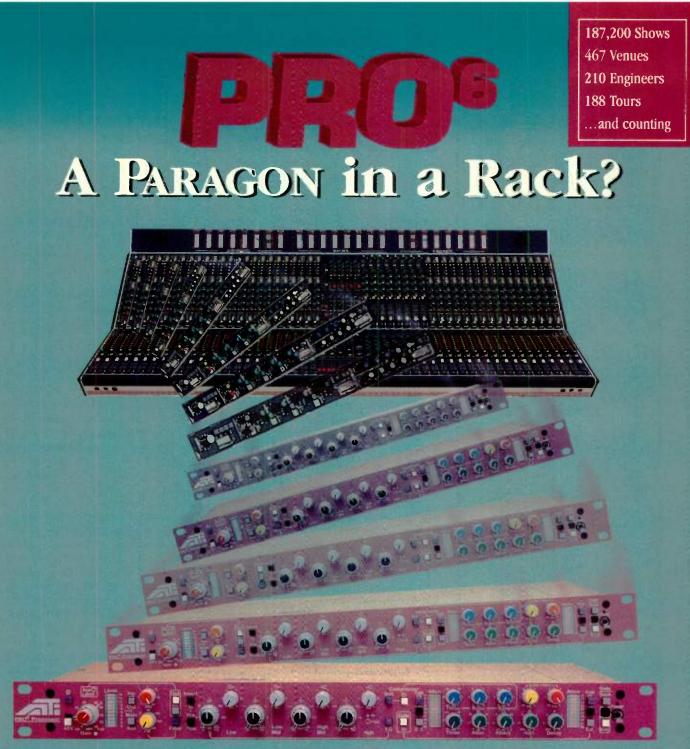
SUMMARY: Does more than you'd expect, but be ready to program it yourself to get the most out of it.

STRENGTHS: Inexpensive; easy to learn; compact; looks cool; more options, sonic flexibility, and programmability than you'd expect at this price.

WEAKNESSES: Factory presets don't show off unit's versatility; wah resonance is fixed at a very high value; tendency towards hiss with gobs of distortion.

PRICE: \$295

EQ FREE LIT. #: 124



The ATI **PRO**⁶ uses the same road-proven technology that has passed the rigorous testing of the most demanding engineers in pro audio. ATI's PARAGON mixing console is "the most highly-specified touring desk on the market."* Now you can literally have a piece of the PARAGON in *your rack*. Loaded with features, the **PRO**⁶ includes:

- high voltage mic preamp
- 24db/oct hi/lo pass filter
- four band parametric EQ

- parametric noise gate
- compressor/limiter
- ducker

Get a PRO⁶ today and listen to it become your favorite audio processing tool!



Audio Toys, Inc. • 9017-C Mendenhall Court • Columbia, MD 21045 (410) 381-7879 • Fax (410) 381-5025





H&K TubeMan



MANUFACTURER: Hughes & Kettner Inc., 1848 South Elmhurst Rd., Mt. Prospect, IL 60056. Tel: 800-452-6771.

APPLICATION: Guitar-specific direct box (unbalanced) and signal processor for the studio; signal processor/buffer for live use.

SUMMARY: Simple, uncomplicated, **cost**-effective way to get traditional tube sounds with minimal effort.

STRENGTHS: Smooth tonal quality; can obtain good "crunch" as well as full-blown distortion; funk, blues, and rock emulations very usable; gentle, but useable tone controls.

WEAKNESSES: No memory to store patches; difficult to replace tube; limited tonal variations.

PRICE: \$349

EQ FREE LIT. #: 125

THE H&K TubeMan is neither a barebones tube processor nor a do-all rack box, but falls somewhere in between the two. Designed as a floor box with accompanying wall-wart-type AC adapter, TubeMan has six knobs and two switches. A four-position rock/ blues/funk/jazz knob provides general voicing: bass, mid, and treble controls provide postdistortion tone shaping, while a predistortion midrange boost switch alters the sound's "fatness."

There are two gain controls, one for preamp and one for master. The remaining switch is an active/bypass footswitch with status LED. The rear panel has an input jack and three output jacks (all 1/4-inch, unbalanced): to power amp, to guitar amp, and to mixer. You can use one or all, depending on your application. The power amp output is the brightest, while the guitar out has some high-frequency rolloff; the mixer connection incorporates Hughes & Kettner's Red Box cabinet simulator circuitry (which, essentially, rolls off some high frequencies and provides a mild midrange boost). Incidentally, tube purists will be happy to know that Tube-Man puts a full 290 volts on the plate.

Speaking of sound, the Rock, Blues, and Funk voicings are excellent, although I found the Jazz setting to be somewhat bland. In addition to a creamy shred tone, the Rock setting can deliver gentle, smooth breakups if you pull the gain back a little bit. This sound is important to me, and is often difficult for solid-state devices to emulate. Blues gives a strong, stinging tone — sort of '60s Buddy Guy-ish — and Funk sounds really great if you pull the gain back to avoid obvious distortion, and crank up the treble a bit. These are solid sounds — just a bit of aggression combined with the characteristic smoothness of tubes.

The low operating voltage means the tube should have a pretty long life, which is a good thing considering that replacing the tube is a fairly major hassle: you have to remove not only the casing, but the knobs, control washers, jack washers, and internal circuit board.

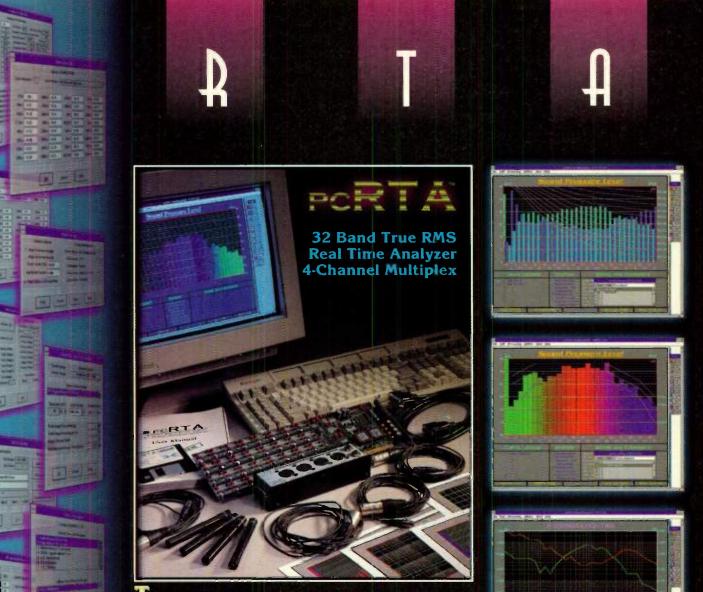
TubeMan is not programmable. It's not difficult to find good sounds, but the settings can be quite different. For live use, TubeMan probably works best as a set-and-forget device. Given the small number of controls, programmability in the studio isn't much of an issue — you can pretty much tweak up a sound in a minute or two. However, if you get a particularly groovacious sound, it's a good idea to write down the control settings so you can re-create them later.

Although TubeMan doesn't provide a lot of variations, overall it's a friendly, cost-effective, easy-to-use box that makes it easy to get quality guitar sounds with a minimum of hassle. —Craig Anderton



88 FEBRUARY EQ

World Radio History



doday's ever improving audio standards, and increasingly technical and creative acoustic applications, require precision realtime acoustic measurements and documentation more than ever before.

Whether your application is home theater, cinema, church, or stadium installations, the pcRTA has powerful features and solutions for you. The pcRTA gives you affordable laboratory preci-



PAC3 - The Portable Solution If your measurement requirements demand portability, we've got the answer. The PAC3 provides a serial interface to laptop or other computers, and more...

Call Today for a free Demo Disk! Tel: (503) 620-3044

computers, and more... LinearX Systems Inc 7556 SW Bridgeport Rd Portland, OR 97224 (ISA TEL: (503) 620-3044 / FAX: (503) 598-9258 BBS: (503) 598-9326 / Internet: info@linearx.com ©1995 Product and Trademark names are the property of their respective owners.



sion and quality with all the power and features of a PC-based system. Offering 4-mic multiplexing, the pcRTA makes surround sound alignment and spatial averaging a snap. Moreover it provides built-in alignment curves for NC, SMPTE X/N, and more. Complete with ANSI filters A,B,C,D,E and a host of

other features to numerous to mention!

International Dealers: Agentina:Interface SRL (54-1)741-1389/ Australia:/ME Technologies 61(0)65-50-2254/ Austria:Audiomax 49(0)71-31-162225/ Belglum:Beram 32(0)2-736-50-00/ Brazil:SDS Equipamentos 55(0)11-887-7597/ Canada:Gerraudio 416-696-2779/ China:HiVi Research (852)2-556-9619/ Denmark,Finland:A&T Ludproduktion 46(0)8623-08-70/ France:Belram 32(0)2-736-50-00/ Germany:Audiomax 49.71-31-162225/ Indonesia:Ken's Audio 62(0)21-639-5806/ Italy:Outline snc 39-36 3581341/ Korea: Sammi Sound:82(02)463-0394/ Luxembourg:Belram 32(0)2-736-50-00/ Malaysia:AUVI 65-283-2544/ New Zealand:ME Technologies 61-65-50-2254/ Norway:A&T Ljudproduktion 46(0)8-623-08-70/ Phillipines:Dai-Ichi Denshi (63)2631-6980/ Poland:inter-Americom 48(22)43-23:34/ Singapore: AUVI 65-283-2544 / Spain:Audio Stage Design 34(94)620-27-47/ Sweden:A&T Ljudproduction 46(0)8-623-08-70/ Switzerland:Good Vibes Ltd. 41-56-82-5020/ Taiwan:Gestion Taycan Irtl 886-2:562-3883/ Thalland:AUVI 65-283-2544/ The Netherlands:Duran Audio 31-41-80-15583/ UK(England):Munro Assoc 44(0)171-403-3808.







Tech 21 MIDI Mouse



MANUFACTURER: Tech 21, 1600 Broadway, New York, NY 10019. Tel. 212-315-1116.

APPLICATION: Remote program switching of MIDI devices, live or in the studio.

SUMMARY: A simple, foolproof way to change programs.

STRENGTHS: Sturdy construction; very easy to figure out and use; multiple power options.

WEAKNESSES: No "trap door" for easy battery replacement; no provision for "bank" or 0-127 program readout, only 1–128; loses channel setting unless there's an internal battery.

PRICE: \$125

EQ FREE LIT. #: 126

IF YOU'D LIKE a remote control that can send MIDI program changes to a device like a signal processor or "snapshot" mixer, there's a bewildering array of choices — from computer programs to footswitches designed specifically for guitarists. Tech 21's MIDI Mouse is one of the simplest, and most foolproof, options. The black, 4.5" X 3.5" cast aluminum box features:

• A three-digit, seven-segment LED display that shows the MIDI program number (1-128).

• Three footswitches: Up, Down, and Active/Search. In Active mode, the display remains lit as you press Up to increment to the next higher-numbered program, or Down to decrement to the next lower-numbered program. In either case, the program change message is sent out instantly upon selection. In Search mode, the display blinks; you can select a program without sending a program change message until you press the Active/Search button again, which transmits the message via the MIDI Out jack. (Note that in Search tery juice (about 15 mA average), but turns off after 10 seconds to save power. Tech 21 claims you can turn on the device and switch programs 2000 times from one battery. Environmentally speaking, an AC adapter is preferable, although using the battery does allow for an appealingly cable-free setup. You can set the MIDI Mouse to transmit on any of the 16 MIDI channels, but it will default to 1 when you remove power unless there's an internal battery.

The MIDI Mouse seems designed for live use, but it also makes a nifty little remote in the studio. The small footprint lets you place it on top of a keyboard or console (or wear it on a belt, for that matter) for quick program changes. A little Velcro on the back is a good addition in this context.

It ain't fancy, and bells & whistles are at a minimum, but you can run over this sucker with a car or drop it from your hotel room window and it will keep on working. If you want to change programs without having to become a rocket scientist, this is your box. —*Craig Anderton*

mode, pressing and holding the Up or Down button scrolls through the programs.) Either mode can wrap around from program 128 to 1 or vice-versa.

• DC power jack that works with most 9 V adapters.

There are three power options: DC power jack; phantom powering through the MIDI cable using the two "outside" pins of the connector; or battery. The LED sucks beaucoup bat-



LOOK AT WHAT THEY'RE SAYING.... LOOK AT WHAT THEY'RE SEEING....

"I grabbed Plus a few hours ago and have been "wow-ing" and "oooh-ing" all evening <g>. This is going to knock them out I think!"- Dave Tosti-Lane

Elle Lati Bentone Mattillert Process William projection and

5

165

11

17

11

"Very impressive. Never seen anything like it on a PC, that's for sure! Quite a piece of work."- Lauren Weinstein

"WOW! Of course, you've heard that before..... Incredibly awesome product, if it was a dessert it would be chocolate."- Chris Smith

"SAW Plus is just the greatest"- Rick Peeples

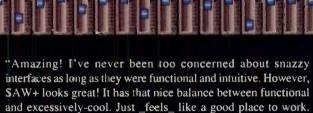
· 1 . -25 . +2 .





THE PART OF THE

A 16 Track (Mono or Stereo) Non-Linear Hard Disk Editing and Mixing Solution For The PC!



Compliments to the chefs!"- John Ciccone

H/S TL

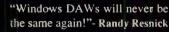
15.2 20.2 AU.2 AS.2

"This software is the finest I have ever had the opportunity to work with"- Elann Mandler



INNOVATIVE QUALITY SOFTWARE

2955 East Russell Road Las Vegas, NV 89120-2428 Tel: (702) 435-9077 Fax: (702) 435-9106 BBS: (702) 435-7186





"Get ready all, a new era in PC digital audio multitrack production is coming soon!!!!!!"- John Reid

> Visit us on the web: http://www.iqsoft.com/ Email:info@iqsoft.com



Time Remaining

Stop SRP Rec RecRdy





MANUFACTURER: Blue Chip, distributed by Music Industries Corp., 99 Tulip Ave., Floral Park, NY 11001. Tel. 516-352-4110.

APPLICATION: Convert guitar playing into MIDI data capable of triggering MIDI instruments.

SUMMARY: It's not perfect, but it does bring a few new tweaks to the party and triggers faster than the average.

GUITAR TOYS

STRENGTHS: Fast response; clever split options; tracks well; flexible (almost to a fault).

WEAKNESSES: Still glitches if you're not careful; user-hostile interface; manual assumes much prior knowledge and is unclear in parts.

PRICE: \$1195

EQ FREE LIT. #: 127

MIDI GUITAR refuses to go away. Once touted as the next big thing, it has settled down into a comfortable niche existence, with Roland's GR-1 and GR-09 (reviewed in the December 1994 issue of *EQ*) as the undisputed leaders. Into this scenario steps the Axon NGC 66, which not only has to overcome Roland's reputation and clout, but the perception built over the years that MIDI guitar is pretty much for propeller heads with perfect technique.

First, though, the standard disclaimer: playing MIDI guitar has very little to do with playing guitar. You're playing synth, but with a more familiar interface. The type of synthesizer you play is important as well (personal favorites: Yamaha TX81Z and TX802, Ensonig TS-10, Oberheim Xpander) since optimally, you should be able to play over multiple channels, and have each channel respond to only one note at a time (just like a real guitar). You also need to know about synth programming - many times tracking problems are caused by the synth programming, not the MIDI guitar.

However, the rewards of MIDI guitar can be considerable: a wide

palette of timbres, with a degree of expressiveness that is very difficult to achieve on a keyboard. Wind and string patches come alive, and human finger-based pitch bending beats an LFO any time. For those willing to slog through the MIDI guitar jungle, there is definitely the opportunity to find hidden treasure. What you need is a guitar-to-MIDI converter that's fast, accurate, and inexpensive. The perfect converter doesn't exist, but the Axon sure tries hard. Let's check it out.

The Axon NGC 66 is a half-rack unit designed to work with the Roland GK-2 guitar interface. The rear panel has MIDI in, out, and thru, bass/guitar switch (yes, it works with both), two jacks for footswitches (hold and patch advance), guitar audio out, and a jack for the GK-2 compatible multipin connector. The unit runs pretty hot, so make sure it gets a reasonable amount of ventilation.

The front panel has power on/off, four main buttons, six programming buttons. 32-character LCD, and a normal guitar audio level control. The Axon is a very deep and flexible unit; that's the good news. The bad news is that trying to program something this deep through this limited a set of interface tools is a major pain — the amount of button presses required to do even the simplest tasks seems excessive. Although, in Axon's defense, this is designed to be a "setand-forget" type of unit where you sort out all your assignments and patches beforehand, you might get frustrated enough to want to forget it before you finish setting it.

SPLITTING HAIRS (AND NOTES)

The Axon works well with General MIDI modules, and much of its flexibility comes from being able to set up different types of program/modulation splits. These can be set according to fret or string (e.g., notes played on or above a particular fret play a different sound from notes below the specified fret, or different strings can play different sounds).

More surprisingly, and most welcome, is the option to set a zone between the bridge and highest fret where picking sends out a controller such as modulation. For example, playing in the specified zone could add vibrato. What's best about this is that you can set the two points that specify the zone to different controller values. Suppose one point generates the maximum value controller, while the second point generates the minimum value. As you pick closer to the second point, the controller value that's sent decreases rather than just cuts off. Cool.

But wait — there's more. These options can be applied simultaneously, so that playing a note below a particular fret on a certain string, and picking in a particular part of the



World Radio History



string, makes one sound; playing the same way but on a different string or note range could make an entirely different sound.

However, I must admit that due to Axon's awkward programming, I found it easier to use a MIDI guitarfriendly synth (Ensonig TS-10) and leave the Axon set up to generate six channels of signal over six strings. Then I just set splits, different string sounds, and so on at the synth.

A little bit of time spent tweaking the TS-10 (e.g., glide mode set to mono, glide time to 0, system pitch bend to 12 semitones, MIDI mode set to Mono A) made it an ideal companion for the Axon. As I've said before, MIDI is like a combination lock, with the tumblers spread out over the guitar, MIDI interface, and synth. When everything is set correctly, great - otherwise you'll get glitching. Note that any problems are most obvious with very short, percussive patches such as kalimba. If they don't work, keep tweaking.

BUT WAIT, THERE'S STILL MORE ...

The Axon has several other useful, if not exactly ground-breaking, features: note pitch quantization, minimal double-triggering, decay threshold (keeps notes from overstaying their welcome), velocity sensitivity and offset for setting various velocity curves, transposition, and the ability to program a variation sound that appears when you press on the hold footswitch (supplied with the unit).



CD-ROMS: PLAY BLUES GUITAR & JAZZ TUTOR

Pop quiz: you're an recording engineer and you're trying to communicate to the guitar player where you're going to punch in. Do you say: "Okay, we're going to punch in ... ":

- a. At this here part coming up next
- b. At SMPTE time 00:01:50:12
- c. When I make this gesture with my hand
- d. Just before the second measure of the turnaround

If you said a, b, or c, you might consider learning a bit about music theory. Although there's no CD-ROM entitled "music theory for recording engineers," there are some decent discs out there for musicians...like Play Blues Guitar, from Kudosoft. Keith Wyatt, the instructor, analyzes the structures of various songs as well as the playing techniques used in those songs, but of equal importance is the terms and techniques section. This provides background from the basic (how strings and fingers are numbered, guitar terms, etc.) to what makes up chords. This section is handy for when you're trying to decipher something you don't understand about a song, but it's also great for browsing and picking up some musical basics.

As one example of what happens when you find a topic that piques your interest, if you click on Rhythm Phrasing, you can then check out playing bass lines, blues turnarounds, strumming, combining chords, and using horn voicings. If you then click on blues turnarounds, you're presented with two video demos of standard and improvised riffs, and two MIDI playback videos that show fingerings and play notes. There's also a section on phrasing — bends, slides, slurs, vibrato, etc. All of these help make you more aware of the guitarist's vocabulary, and can come in really handy when trying to help break creative blocks ("Why not try using some horn voicings behind the vocals rather than strumming?").

From a musician's standpoint, Play Blues Guitar works because it's aesthetically very well done so you want to use it, navigation is simple, the instructor seems into it, and there's a lot of really solid information. This disc makes good use of the medium because it encourages you to float around to whatever interests you; there's no real road map through the CD, although the contents page provides a reasonable substitute. Then again, this isn't really a course in the traditional sense — indeed, this is one of the factors that makes it more suited to engineers — but more like a teacher who answers whatever question you have at any given moment.

However, Play Blues Guitar is blues only; if you already have one foot in the musical world, or are a musician seeking to really expand your theory chops, Masterclass Productions' Jazz Tutor is the most industrial-strength music theory CD-ROM I've seen. Musicians will appreciate the fact that it's a complete, paced course involving slide shows, charts, and MIDI examples based around five new songs by Phil Woods; those who are more interested in theory will be interested in the reference section (featuring a glossary of jazz terms and chord/scale glossary). If your brain starts to hurt after absorbing all that theory, you can chill out with a slide show with highlights of Phil Woods' career or listen to the five songs in all their audio CD glory.

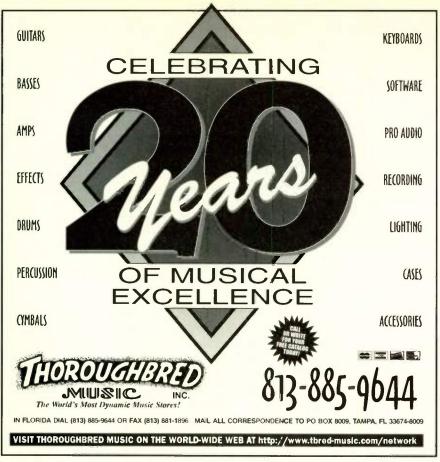
While its complexity may not make Jazz Tutor as applicable to recording engineers who want to know a bit about music compared to the simpler (and more user-friendly) Play Blues Guitar, it's a real find for musicians who want the equivalent of a music school-level jazz theory -Craig Anderton course without having to leaving their PCs.

Play Blues Guitar (\$59.95); Kudosoft, 11320 Chandler Blvd., Suite G, N. Hollywood, CA 91601. Tel: 800-887-7529 or 818-766-2958.System requirements: MPC-compatible 486 or greater, 256 colors, 150k/sec CD-ROM, 4 MB RAM, MPC sound card, mouse, Windows 3.1 or 95 and MS-DOS 5.0 or greater.

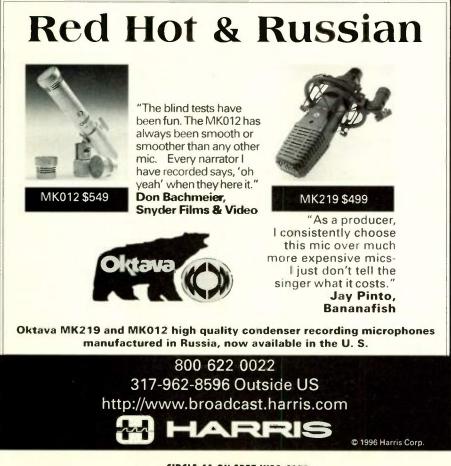
Jazz Tutor (\$99.95); Masterclass Productions, PO Box 304, Ridgewood, NJ 07451. Tel: 800-627-0833. System requirements: Same as above except 8 MB of RAM and 30 MB of hard disk space.

GET THE BLUES: A scene from Play Blues Guitar.

adio History



CIRCLE 79 ON FREE IMFO CARD





There are 64 "arrange" programs, which contain all available parameters. An additional 128 "scratch" programs lack the fancy split features mentioned above (although you can still send out a controller depending on pick position) and are a bit easier to set up. It is not possible to access them randomly; you have to scroll.

Finally, a chain mode lets you string up to 64 programs together through which you can cycle. This is intended mostly to simplify live performance, but of course, it has its uses in the studio as well.

So is the Axon the Holy Grail of pitch-to-MIDI? Once you get the synth and unit matched, the tracking is extremely good (and very fast; apparently the Axon need recognize only the initial picking transient in order to determine the note, and this really speeds things up). Compared to the Roland GR-09, it does seem peppier and tighter — but bear in mind it's an incremental improvement, not a mind-boggling quantum leap.

The Axon is also flexible, and the pick control feature is a real winner. If only there was a Galaxy or Unisyn editor/librarian, I'd be ecstatic. A unit this good shouldn't have to be saddled with such a frustrating interface.

Meanwhile, I'm glad that MIDI guitar continues to improve. The Axon is indeed a step forward, and deserves a close look from guitarists who have been waiting for the "next step" in MIDI guitar. If you want to join the wide world of high-tech guitar, you couldn't go wrong (although you might go broke!) with a Roland VG-8 for great guitar sounds, and an Axon NGC 66 to handle the MIDI end of things.

One caution: always try a guitar synth before you buy, with the sound generator you're going to use. My experience indicates it takes at least an hour or two, as well as a fair degree of MIDI literacy, to really figure out how to set up the guitar and synth properly. Some will try the Axon and decide it "doesn't work" because the other elements of the system are not set up correctly. But if you persevere, you can really get this box to sing. Blue Chip couldn't have picked a tougher product category with which to make their US debut, but it's a credible box with a lot to offer those who have both patience and savvy. -Craig Anderton

CIRCLE 41 ON FREE INFO CARD



VG-8 UPDATE

Regular readers of this magazine may recall my enthusiasm about Roland's VG-8, which to me was the product of 1995 for high-tech guitar players. Now Roland has introduced several new expansion options that take the VG-8 even further.

First is a major, user-installable software update that provides several new features:

• Four new instrument models: hollow body (electroacoustic) guitar; hollow body plus amp; dual amp with two signal chains; and "Vio" guitar, which adds comb filtering.

• Seven new pickup options: Tele; Les Paul with P-90; Les Paul with PAF; Danelectro "lipstick tube" pickup; Rickenbacker; Gretsch; and single-coil active pickup.

• Three new amps: Soldano; Marshall with master volume; and Vox/Matchless.

• Three new speaker cabinets: Fender Super Reverb; Celestion stack; and Vox 2 X 12.

There's also a wah function (pedal or automatic sweep), pedal control over pitch shift amount for any number of strings, "intelligent" polyphonic pitch shifting, ability to assign several HRM parameters to pedal control, and 64 new presets that take advantage of the above features.

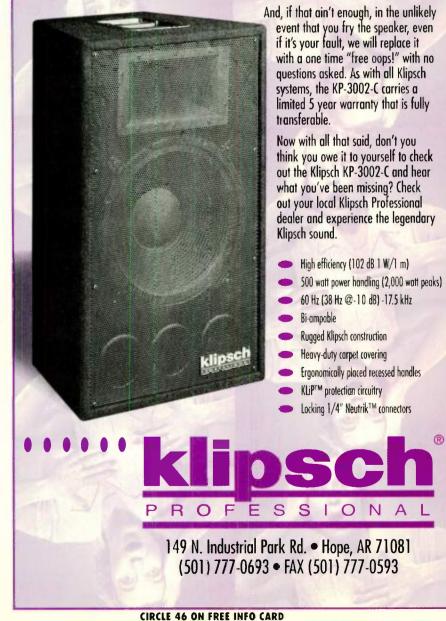
In addition to the software update, there are two new ROM cards. The VG8D-1 "Traditional 1" contains 64 patches designed for country, folk, blues, jazz, and classic rock. The VG8D-2 "Modern 1" offers heavier patches intended for rock, metal, and fusion.

For more information, contact Roland at 7200 Dominion Circle, Los Angeles, CA 90040; 213/685-5141. — Craig Anderton

The Trap With One Free Oops!

The all new Klipsch KP-3002-C could be just the ticket, especially if you're looking for a high output PA box that's expandable and won't break your pocket-book or your back.

The enclosure's trapezoidal design allows it to be used individually or in multiple system arrays without fear of phase cancellation. The pro components include a die-cast 15" aluminum frame woofer for gutsy lows and a two-inch titanium voice coil driver for sizzling highs, coupled to a 60°x40° Tractrix Wave™ horn for smooth and even coverage.



If You Need More Power Than This...

ALEHIY

A HLY

ABHLY

ASHLY

m, R

ASHLY

Maybe You Should Run For President

1.0

Our new MFA Series of amplifiers deliver the power that digital audio sources demand. Whether used for motion picture sound tracks or live contemporary music, these models faithfully reproduce the extended dynamic range of today's technology. High power, great sound, and a full Five Year Worry Free Warranty. Why buy anything else?

* 24,000 watts as illustrated: Eight MFA-8000 amplifiers combined, each putting out 1500 watts per channel into a 2 ohm load. Also available: the MFA-6000, with up to 900 watts per channel.

POWER AMPLIFIERS BY ASHLY

Ashly Audio Inc., 847 Holt Road, Webster, NY 14580-9103 • Toll Free: 800-828-6308 • TEL: 716-872-0010 • FAX: 716-872-0739 • Canada: Gerraudio Dist. Inc. 416-696-2779 International Agent: E and E Exports Inc. 714-440-0760 • Internet: http://www.ashly.com

CIRCLE TS ON FREE ONFO CARD

SIX STRINGS ON STAGE





World Radio History

SIX-STRINGS ON STAGE

Pete Anderson, Dwight Yoakam's favorite guitarist, shares the secrets of getting good guitar sound live

By LIANA JONAS

PETE ANDERSON is more than a guitarist. He's got his own project studio, his own record label, Little Dog Records, an album called Working Class, and has been country superstar Dwight Yoakam's guitarist and righthand man for over ten years now. He is a multiplatinum Grammy award-winning producer, and is presently up for Country Album of the Year/Producer for Dwight Live [Reprise]. Not bad for this one-time factory worker.

Ironically, despite Pete Anderson's fame as a country guitarist, his roots are deeply seeded in blues and rock music. Anderson didn't really play country guitar until he moved to Los Angeles in the early '70s. He found that he could play a wider variety of music in the local honky tonks, and really learned his instrument in that environment. This is where he and the then struggling Dwight Yoakam met in the early '80s.

Anderson maintains a basic rule of thumb in building a good guitar sound: "Use a great sounding amplifier and a great-sounding guitar. Once you've accomplished that, you have a strong foundation upon which to build a complete guitar rig." His personal brand of choice is Fender. "Ninety-nine percent of the time, I use a Fender guitar with a Fender amplifier," notes Anderson. "Once I've got that settled, I can choose my speakers. I like to use Electro-Voice speakers because I don't like distortion coming from anywhere but my amplifier. I find EV speakers to be very clean sounding. Some people may use softer types of speakers to get a distortion-like sound, but it's not

my preference. I like my distortion to come from the tubes and the preamp section of my amp."

As Anderson's gigging roster moved onward and upward, his stage rig has evolved in accordance to meet the growing needs. His first initial rig began with a Fender Blackface Deluxe with an EV speaker. The effects loop consisted of an old grey tube Echoplex delay box and a modified Boss Chorus. This was the rig Anderson used from clubs to the early Dwight gigs. Later on Anderson expanded his rig with two identical Deluxes and the addition of a Goodrich volume pedal. "The Goodrich had a battery unit, which created a low Z signal that went back into the amplifiers. Because it was a low Z signal, I didn't lose any high end or tone no matter how much cable I used. My guitar went directly into the Goodrich and then to one side of the

amplifiers, which was clean, and drove the other side that had the Echoplex and Boss Chorus. Both sides had onboard reverb. As the stages got bigger, I had to fill more space. I started jacking up the amps with power transformers and went as far as I could with a two-tube amplifier."

The next level in Anderson's evolution of stage rigs included a Fender Twin, "which is like two Fender Deluxes." But Anderson felt that the Twin was lacking something, and the bright switch did not yield the sound he was looking for. "No matter what I did, I couldn't get the brightness on the Twin the way I wanted to. So I ended up having the Twin modified where the down position held the same brightness value as a Fender Deluxe, and in the up position it had the same as a Fender Super Reverb. The changes were done, I tweaked the now-modified Twin to my

Greatness runs in the family

Great sound, features, reliability, and good looks, all shared by three families of S.R. loudspeakers from Fender Pro Audio.

From the largest Tour Series concert array, to an SPL Series stack in a club, to a pair of Artist Series speakers on stands, the family resemblance runs deep.

By intention, all are designed to give your audience the power, emotion, and subtlety of your performance, through an open sonic window that doesn't color your sound and won't wimp out when the heat is on.

Hear for yourself what mere pictures and words can't begin to tell. Listen to them today at your nearest Fender Pro Audio dealer - but be forewarned: some of that greatness just might follow you home!

Fender

ELECTRONICS

Shown, front to rear: Artist Series 110-ELC, SPL Series 1225 Mk. II, and Tour Series 2912's and 118s subwoofers.

•

Tour Series concert rig courtesy of Spectrum Audio.

For more information, send \$2.00 postage and handling to Loudspeakers, c/o Fender Pro Audio, 7975 N. Hayden Rd., Scottsdale, AZ 85258. ©1995, F.M.I.C.

> CIRCLE 39 ON FREE INFO CARD World Radio History



liking, plugged in and played, and it sounded like a big Deluxe. I was so happy.'

At this point, Anderson had made the move to wireless guitar (Nady 701) and added on a Korg SDD2000 digital delay, which he had modified to sound like his retired Echoplex. "Everyone thought I was nuts wanting to modify the Korg delay. I actually requested that it be

degenerated from 16 kHz to 12 kHz just to get it to sound the same. It worked out well." MIDI control entered Anderson's rig by way of an ADA floor controller. Using the controller Anderson stored all of his effects settings.

The third and most current incarnation of Anderson's stage rig came in the form of a Fender Tonemaster and a Dual Professional, which are both

four-tube heads. Bob Bradshaw designed for Anderson the 4x (read: four by), a headswitching system that switches between up to four heads. The system has a feature that sends a "speaker dummy load" into the heads as speakers are switched in and out of usage. This is so the heads don't blow out from pumping power into nothing. Anderson notes, "On stage I have two cabinets that

NY 11010

100% (1700 Fark No., Flanklin Sq., NY 1 (516) 328-7500 • FAX (516) 328-7506

147 New

Wellst. in Canada by Erikson Pro Audio

look exactly like Fender Twins. except that they have no heads. They get the heads switched into them by my 4x. One cabinet is dedicated for clean signal and the other is for effects. My wireless guitar is switched between heads by way of a rack-mounted Patchmate MIDI control device. The 4x and the Patchmate are the two devices that make my rig work."

An entire show is programmed this way on a per song basis, and it is all operated by a guitar tech off stage. The only objects present on stage are Anderson, his guitar, the two cabinets, and a volume pedal. His effects rack, which is also off stage, is kept simple, consisting of a Peavey Valverb, a Roland SE-50, and a Korg SDD2000 digital delay. "I try to get my effects from on-board the amps as much as possible," Anderson comments.

HIGH FIVE

According to Anderson, the following top-five ingredients will yield a good live guitar sound:

1. Knowledge of the room type: "You want to adjust your delay and reverb according to the room. Is it a dead room, is it really live sounding? Adjust accordingly. Also, you want to be heard. The best way for a guitarist to be heard is to play mono and dry. I know that might not be very fun, but it will give you impact you're looking for."

2. No compressors: "I never use compressors on stage because it takes away the high end and impact of your sound. If your rig is dialed right, you'll get natural compression."

3. Build your bass: "Dial in the bass according to the room. It's easy to create a bass trap in clubs/bars because you are often positioned in a corner. Build your bass from the lowest level up."

4. Simplicity: "Try to keep your signal chain as simple as possible in order to get the



CIRCLE 16 ON FREE INFO CARD



MIXDOWN CLASSIC 8 - THE LATEST GENERATION OF PROFESSIONAL AUDIO CONSOLES FOR MULTITRACK RECORDING OR LIVE SOUND REINFORCEMENT

Launched in early '88 the Mixdown was followed two years later by the Mixdown Gold with extra features and upgraded spec. Both were critically acclaimed everywhere they landed. That was then. We've now smashed through to a new dimension with the Mixdown Classic 8 - the latest generation of professional audio consoles designed to face the future. The MIXDOWN CLASSIC 8 offers unsurpassed quality, features and specifications that a few years ago would have been unthinkable at the price! With three models in the range 16, 24 & 32 channel, and with the full width meter bridge, MIDI muting and rackmount EP5 power supply - you're entering into a whole new world of music making.

METER BRIDGE - The adjustable meter bridge has a two colour, 12 segment peak reading LED display for every channel and moving coil VU reading meters for the left and right outputs. INPUT CHANNELS - All input channels are crammed full of features including balanced mic and line inputs, inserts, direct outs, +48V. 20dB mic pad, three band EQ with sweep Mid sweep Low & EQ cut, six aux busses, MIDI mute, PFL and smooth 100mm fader.

TAKE CONTROL TAKE CONTROL AND DISCOVER

OUTPUT SECTION - The sixteen tape monitors can be used as extra inputs bringing the total number on a 16 channel up to 34. The upper row of inputs even feature two band EQ, PFL, a couple of aux sends and fader reverse. All output groups have insert points. 12 segment bargraphs and 100mm faders. A line up oscillator, stereo return and a built-in talkback mic.



WHEN YOU NEED TO GET SERIOUS



K E E Studiomaster Inc. 3941 Miraloma Avenue, Anaheim CA92807 USA Tel 714 524 2227 Fax 714 524 5096 World Radio HistStudiomaster House, Chaul End Lane, Luton, Beds, LU4 8EZ UK Tel 01582 570370 Fax 01582 494343

MIXD

AC Hum

Gone!

Hum Eliminato

The Hum Eliminator[™] from Ebtech stops the 60Hz buzz that's caused by groundloops, quick and easy. Just plug it into the signal lines between offending pieces of equipment and you're done.

And since it's not a filter or gate, it won't change your sound.



- Works with balanced or unbalanced signals at either end.
- Breaks groundloops safely while leaving all equipment grounds intact.
- Also available in Line Level Shifter[™] version for converting back and forth between -10dBV and +4dBu.
- Two channels for \$69.95 retail. Eight channels for \$279.95 retail.

For Sales, Call On The Road Marketing: Tel: (201) 389-1718; Fax: (201) 389-1917

CIRCLE 94 ON FREE INFO CARD



CIRCLE 95 ON FREE INFO CARD

cleanest signal to your amp's front end. Turn the effects on, but keep your mix knob turned all the way down, and bring it up only as necessary. You want unity gain in your loop."

5. Active Volume Pedals: "If you use a volume pedal, I recommend low-Z, batteryoperated ones that go in between the guitar and amplifier. I find that most nonactive volume pedals dull your tone out."

OTHER TIPS

Anderson cites single-coil pickups as one of the main causes of a noisy guitar rig. "Single coil pickups pick up (no pun intended) the hum often generated from light dimmers. I recommend having copper shielding installed in the cavity of your guitar and make sure that it is grounded really well. Another way to avoid noise is to put your delay at the end of the chain because, for example, you don't want a delay signal hitting the chorus signal."

When using a wireless system, Anderson recommends walking the stage during soundcheck. This is the way guitarists can identify drop-out spots, and mark the spot(s) on stage where not to step. When gigging, bring your own bag of gear and don't rely so much on rental systems. "Often times these rentals are very beat up and break down. You don't want to be caught in a bind. Also learn how to operate a Fender Twin Reverb because they are the standard in most rental situations."

The final tip Anderson offers is to get off monitors. "There are plenty of monitors on stage, particularly around the singer, and you can listen to those. Learn to hear what you need to hear, learn to hear yourself. I don't want to hear the rest of the band except for the snare drum and the hi-hat. Guitar players should be concerned with these two instruments because they provide the pattern that guitarists play to."

102 FEBRUARY EQ

World Radio History

SPIRIT 8 THE EXTRAS COME AS STANDARD THE PRICE COMES AS A SURPRISE

Spirit 8 is a quiet, compact and rugged 8-bus mixing console, bardly surprising from a company with over 22 years of knowbow in audio design.

What may be surprising is that it comes with every feature you could wish for, as standard, at a price* that will make the competition wonder how we did it.

MONO

METER BRIDGE BUILT IN UP TO 40 BULLET-PROOF MIC INPUTS 4 MUTE GROUPS INDIVIDUAL CHANNEL BOARDS 10 X 2 MATRIX SECTION

ULTRAMIC⁺ PREAMPS on mono channels. An incredible 66dB of gain range provides 28dB of headroom - enough to accomodate virtually any input signal, from the hottest line level electronics to the lowest level condenser mics. **48V PHANTOM POWER** individually selectable on each mono channel **HIGH PASS FILTER** at 100Hz with steep 18dB/octave slope, essential for combatting stage rumble and mic popping (mono channels)

+BAND EQ with 2 swept mids on every mono input. 2-BAND EQ on stereo inputs

EQ In/Out bypass switch on every input

6 AUX buses with 6 controls, 2 of which are selectable pre- or postfader at the master. Internal solderless jumpers can be accessed for pre- or post-EQ configuring

8 GROUP BUSES

4 MUTE GROUPS for scene setting

Multi-tap PEAK/PFL

DIRECT OUT on every mono channel allows multitrack recording or individual effects sends (internally selectable pre- or post-fader for live recording)

GROUP

DUAL

INPUT

14 C

81

.2: 🔿

П

STEREO

12:0

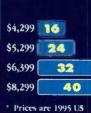
.

....

SECTION

 8 STEREO RETURNS for additional stereo sound sources or effects units: 4 have EQ

10 X 2 MATRIX section for additional mixes



suggested retails

Also available: Spirit Live 3² 3-bus Spirit Live 4² 4-bus

CIRCLE 72 ON FREE INFO CARD

- 4 FRAME SIZES: 16, 24, 32 and 40 channels
- UP TO 64 INPUTS AVAILABLE (40 channel version)
- 4 STEREO INPUTS with 2-band EQ and full access to Auxs
- METER BRIDGE AS STANDARD, showing mono input, group and master levels
- INDIVIDUAL CHANNEL PCBs for case of maintenance and replacement
- Every pot INDIVIDUALLY FASTENED to the steel facia, the professional way, to cope with knocks
- RUGGED "NO TWIST" STEEL CHASSIS built to survive heatic tour schedules
- Side checks can be removed for FLIGHT CASE INSTALLATION if desired

APPLICATIONS

Front of House Mixing

- Theatre/Venue
- Installations

Touring

Recording direct to multitrack

Spirit by Soundcraft¹⁶⁸ Inc., 11820 Kemper Road, Auburn, CA. 95603 Tel: (916) 888 0488 Fax: (916) 888 0480







I HAVE BEEN working with Blues Traveler in preproduction and live mixing for eight years now, and have watched the band grow both musically and dynamically. Other artists that I have worked with include Soul Hat, Chris Whitley, Colonel Bruce Hampton and the Aquarium Rescue Unit, and the Screaming Headless Torsos.

Blues Traveler has become more fluid and relaxed on stage over the years. There was a time when they were a bit trapped in "over-professionalism," and almost came across as rigid in their performance. But now that the band has achieved a level of technical proficiency, they are more relaxed on stage. As a result, we're doing a lot of live recording for a live album due out in May; as a matter of fact, we've begun the mixing process.

For recording, my partner Dave Swanson is using a Mackie 32•8 with four TAS-CAM DA-88's, splitting off of the monitor or FOH console, depending on which is more discrete. We're using

four channels of Drawmer 1960's for some tube mic pre warmth and smooth limiting, four channels of Dirk Schubert (more on these awesome units later on) parametric EQs, and three sets of ambience mics. Two pairs of the ambience mics are placed facing the audience (one from up in the truss and a broad stereo stage L/R from the side); for these we're using your basic condensers, AKG 460's and Audio Technica 4051's. The third pair is a set of AKG 414's with one set to cardioid and the other set in a figure eight pattern, which is run out of phase in mixdown. The 414 pair gives an out-of-phase, super-wide stereo image from the figure eight, and then we dial the cardioid up the middle for a three-way room perspective.

ROAD ARSENAL

In my FOH racks, I have BSS graphic EQs; Drawmer and Aphex gates (which are in the racks for the opening act); a t.c. electronics M5000 mainframe (which I love); Yamaha REV5, SPX900, and an SPX90II for an old pitch change program I use on Popper's vocal; some older dbx 165's with discrete circuitry; and Dirk Schubert, very highlysecret, 6-band parametric EQs. You just try and find one of these things! These EQs have been compared to Massenburg's; they've got a super-tight bandwidth. I keep them around for inserts or for the downfills that I use across the front of the stage. The Schubert EQs are my secret weapons, as are the older dbx 165's, and I bring them to all the Blues Traveler's shows.

Speaking of secret weapons, there is one other that I bring along with me - the dbx 120XP. [The unit is designed to produce an additional octave of bass. Ed.] I can't say enough about the dbx 120XP Subharmonic Synthesizer, which is made, believe it or not, for discos and DIs. It costs around \$150, but it gives you 50 Hz from here to kingdom come! On the hollow stages, I use less actual acoustical rumble in my kick drum and put the 120XP to use. This is another one of my toys that I load into my rack for the shows.

IN THE VINK: Engineer and author Richard Vink.

RIDING A WAVE OF POPULARITY, BLUES TRAVELER HITS THE ROAD IN SUPPORT OF THEIR NEW ALBUM *FOUR*

By RICHARD VINK

SEIZE THE MEANS OF PRODUCTION!



THE NEW K2500 SERIES: THE REVOLUTION HAS BEGUN.

THE PRODUCTION STATIONS Kurzweil most powerful sampling synthesizers ever: the **K2500 Series.** So powerful and flexible, you can take a project from conception to completion without leaving the instrument – and interface with a myriad of analog and digital formats. The **K2500** instruments, available in rack-mount, 76-note semi-weighted and 88-note fully-weighted keyboard versions.

STUNNING ON-BOARD SOUNDS – A stunning new soundset of 200 programs and powerful performance setups, plus a separate disk containing 1,000 useful programs. Load samples in Akai*, Roland*, Ensoniq*, AIFF and WAVE formats from disk, via SCSI or MIDI, and process them with VA.S.T.*

V.A.S.T. POWER – Variable Architecture Synthesis Technology offers 60 DSP functions arrangeable in 31 algorithms. Use different methods of synthesis within the same program, including the unique $KB-3^*$ harmonic synthesis mode, as simple as the drawbars of a

tonewheel organ. **SAMPLE WHILE PLAY** – The SMP-2 Sampling Option has digital and optical



I/O, use the sample while play facility to record samples while continuing other operations, or sample the K2500's audio outputs or sequencer.

MULTI-TRACK SEQUENCER – The K2500 Series' on-board sequencer offers state-of-the-art, computerbased sequencer features plus flexible live performance interaction. Trigger sequences from keys or controllers via the Interactive Sequence Arranger (ISA).

UPGRADABILITY – Optional upgrades for the **K2500** include the KDFX Digital Effects Option, providing four stereo buses of state-of the art DSP the Orchestral ROM, Contemporary ROM and Stereo Grand Piano SoundBlocks, for a total of 28 MB of Kurzweil's

finest sounds; and the *Digital MultiTrack interface (DMTi)*, allowing sample rate/format conversions and direct connection to DA-88 or ADAT*.

SOFTWARE UPGRADES IN A FLASH – With Flash ROM, upgrade the operating system from floppy disk or via SCSI.

SEIZE THE MEANS OF PRODUCTION – Play the K2500 Series at your authorized Kurzweil dealer today.

%B 3 Harmonic Synthesis Mode available 1996



I is a product line of Young Chang

PLAYERS OF THE WORLD...

Visit our Web site – get a FREE T-Shirt http://www.youngchang.com/kurzweil/ Get a FREE Poster at your Authorized Kurzweil Dealer."

Kurzwell Music Systems, 13336 Alondra Bivd., Cerritos, CA 90703 Tel: +1 310/926 3200 Fax: +1 310/404-0748. Young Chang (Piano) Canada Corp., 395 Cochrane Drive, Markham, Ont L3R 9R5 Tel: +1 905/513-6240 Fax: +1 905/513-9445 E-mail: kurzwell@aol.com Trademarks and registered trademarks are the property of their respective owners. "Otters subject to availability.

PURE INSPIRATION

CIRCLE 48 ON FREE INFO CARD



PICK A SPOT, ANY SPOT

I like to mix from as close to a crowd perspective as possible. And I don't mind mixing from the worst-sounding spot in the room because you just get used to that in theaters. (I'm talking about under the balcony!) The only place that I really avoid mixing from is all the way off to one side of the room. The reason why I don't mind being in a tough spot is because if I can make the mix sound good from there, it's an achievement. If I'm in a sweet spot, like on a riser in the middle of the floor, the sound is unrealistic. I try to never be on a riser because by the time you get all of your gear out in front of you, you've created a little bit of a bass trap. Being positioned up in the horns, higher than the rest of the crowd, and blocking the sight lines is not for me.

My ideal mix position is slightly off-center and in the crowd at their level. This will give me a nice, realistic blend of two perspectives: the crowd's and the room's.

POPPING SENSATION

Having a lead singer who



also plays harmonica also offers some challenges. For John Popper's vocals we're using an Audix OM5. We chose this mic because John is a big guy and sings very loud. The Audix handles his high SPL superbly and has such strong presence. This has worked out very well for us in the live recordings. Additionally, John's vocal has a lot of top end, so using the board EQ I'll create a dip at around 3k for a presence curve. I use the dbx 165 compressor set for hard limiting in a protective manner. After that I use a limiter soft-knee with a slowed down attack. This helps gather him up from 500 Hz and down.

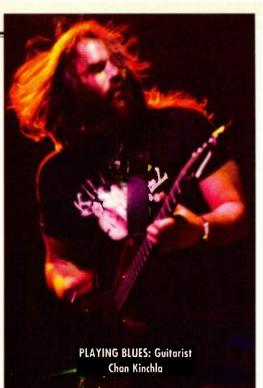
For the harmonica, a Hohner Special 20, we're using a Shure SM58. So John has the Shure in one breath and the Audix in the other. The SM58 has two toggle switches about half way down the mic's shaft, and it's duct taped from here to next month. The right side toggle switch is an amp channel (lead and overdrive) changer switch for his Soldano 100 watt guitar head, and the left side is a speed switch for

his Leslie cabinet. Just on top of the two switches is a volume pot that controls how much of his harmonica is going to the Leslie. (The pot controls the slave out on his Soldano, allowing John to turn on his Leslie cabinet as he wishes.) He will use his two thumbs to control the toggle switches, and his index fingers to maneuver the volume pot. It's quite a talent he's got there. Popper's SM58 was customized by the techs at Eastern Stage Productions.

John cups the mic a lot and uses that to get a very thick low-end feel, and we usually set up his gain structure on stage to work off of that. Popper has got his lead channel turned up very loud and rides the distortion. He uses both channels on the Soldano and switches back and forth between overdrive and a

clean sound. We carry a DI for the harmonica for extreme emergencies. Popper runs an Eventide H3000 and a DigiTech multieffects unit on a coffee table full of tricks that are controlled by a MIDI foot controller — he'll use some pan delays and harmonization. It's funny, a lot of people think that I'm doing all of this kooky stuff on the harmonica when it's all Popper. I simply bring up the amount of effects that he's playing and blend it with the house mix.

My harmonica FOH channels are two Shure SM57's, a clean DL two effects channels, and two Leslie channels that are panned wide and in stereo. We only mic the top horn part of the Leslie and spread it very wide in stereo to create more tremolo, using two AKG 460's. For the Soldano, we use two SM57's. One is perpendicular to the cabinet on a 90degree angle, and the other is next to it and out of phase by 45 degrees. This allows me to leave the EQs on the inputs flat in almost all scenarios, and just use the 57's in a presence-oriented manner. It gives me more edge, and I just balance between the two



faders, and in a way, it kind of turns into a 2-band EQ on the faders. We do this on both Popper's harmonica and the guitar rigs. It works really well with 12-inch speakers.

HOLLOW STAGES

I've found that the Crown 5000's [amps] we're using on the EAWs give a lot of punch to my low end, but as the crowd fills in, the low end starts nondirectionally rolling backwards under the stage. This kind of tickles the band on stage and bounces them around like chess players. When this happens, it is not good because they start misperceiving the feel of their music.

If the situation permits, I like to build a solid barricade going around the back of the speakers to prevent the bass from rolling back underneath the hollow stage. Unfortunately, Blues Traveler plays on six-foot high hollow stages quite often, and this is the absolute worst possible scenario! You've got to battle 60 Hz just going berserk on stage, but not in FOH. This situation gives me an unre-

continued on page 130



INTRODUCING A LINE OF RECORDABLE MEDIA FOR PERFECTIONISTS. THE TDK PRO LINE. IT'S **Pure Precision**. Because every format, every length, every grade is 100% specification guaranteed. For your Finest work. It's more innovation from the people thinking about what you do, and the tools you need to get ahead. From the company that's taking professional recording fast forward.





For more information on our complete line of recording products, please call I-800-TDK-TAPE or check out our site on the World Wide Web.

World Radio History



http://www.tdk.com



NEW GEAR FOR YOUR NEXT GIG

FEELING SXY

Samson has introduced the SX6 Mixer Amplifier. The SX6's integrated mixer design combines XLR/line

inputs with a power amp that delivers 150 watts into 4 ohms. It comes with a threespring reverb, independent two-band EQ for each channel, and a five-band graphic EQ on the main output. Other features include an aux/reverb send on each channel, separate reverb and aux/effects returns on each channel, 48 V phantom power, protection relay circuitry with front-panel LED. front-panel peak overload LED, and a rear-panel amplifier input and preamplifier output. The SX6 is housed in an all-steel chassis and features an anodized aluminum heat sink. For further information, contact Samson Technologies Corp., P.O. Box 9031, Syosset, NY 11791-9031. Tel: 516-364-2244. Circle EQ free lit. #128.



transmission via standard

able gain preamps accept

Each input also provides

switchable phantom

power. A unique fea-

into multiple optical

outputs to different

locations. For further

information, contact

tems, Inc., 102 Grove

Street, Worcester, MA

ELEX-CELLENT

Inc. recently introduced its

professional UHF wireless

guitar FMR-450G and bass guitar FMR-450B systems.

wireless WT-450G transmit-

ter for guitars and the WT-

450B transmitter for bass

guitars. The 450 system is

compact, lightweight and

self-contained. The trans-

pager and clips onto your

belt or guitar/bass strap.

mitter is about the size of a

The guitar cord jack for the WT-450G/B is designed to

accommodate guitar pick-

ups with impedances up to

50K ohms. A toggle switch is

provided to allow the user to

mute the guitar if desired.

The FMR-450G/B receivers

are powered by an external

AC supply via the power

The systems include the

Telex Communications.

01605. Tel: 508-754-

4858. Circle EQ free

lit. #129.

Telecast Fiber Sys-

ture splits stage inputs

microphone or line inputs.

ST optical connectors. Vari-

iack on the rear, or with any 13 VAC RMS/12-14VDC source. The true receivers feature 0 to 180 degree lights to show diversity operation. The

unit's high performance RF front end includes a high Q input coupled with a GaAs-Fet amplifier. For more information, contact Telex.



9600 Aldrich Avenue South, Minneapolis, MN 55420. Tel: 612-884-4051. Circle EQ free lit. #130.

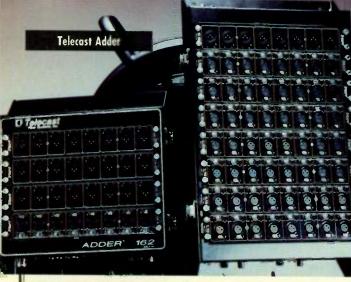
COOL COMBO

HI-MU recently expanded its amplifier line with the inclusion of the Model PPEL84 18 combo. The PPEL84 incorporates dual EL84's in a push-pull configuration yielding a power output of 18 watts. HI-MU's version of the EL84 is a recreation of the classic. late 50's Mullard circuit. modified to eliminate the negative feedback used in the original. The PPEL84 runs in Class A operation. which is standard fare on all HI-MU models. For more details, contact HI-MU, 52 Wheeler Avenue, Cranston, RI 02905. Tel: 401-781-7314. Circle EQ free lit. #131.

ADD IT UP

Telecast Fiber Systems announces the Adder, a family of bi-directional snake systems. Adder snakes simultaneously transmit and optically split up to 64 audio channels plus intercoms and data/control channels both ways over a single optical fiber. Telecast's new Adder family consists of the Adder 162 and Adder 322, capable of carrying 32 and 64 channels, respectively, over a optical fiber cable The Adder is available as portable "shells" or 19-inch rack enclosures. Digital modulation and multiplexing combine all signals onto a single optical fiber for two-way

Telecast Adder I CARE ADDER' 162





PLAY WITH ZOOM

Zoom raises the ante in guitar effects with the introduction of the new Player 3030 multieffects system. The Player 3030 features 32 of Zoom's newest analog distortions and digital modulation effects, with up to seven that can be used simultaneously. Also included is an assignable expression/volume pedal. Specific volume and effect parameters on this pedal can be changed in real time and when used with the optional FP01 foot pedal, volume and expression can be manipulated simultaneously. Equipped with 28 user and 28 Zoom presets, the 3030 features three types of amp simulation that can be used in tandem with eight different analog distortions for a total of 24 sounds from crunch to ultra-high gain. The 3030 also boasts Zoom Noise Reduction (ZNR), and an "intelligent" harmonized pitch shifter (HPS). Additionally, the 3030 has an auto-chromatic guitar tuner for tuning on stage. For more information, contact Samson Technologies Corp., P.O. Box 9031, Syosset, NY 11791-9031. Tel: 516-364-2244. Circle EQ free lit. #132.

SCRATCH THAT

Denon recently introduced the the SMX-2000, the world's first Digi-Scratch DJ Mixer. The SMX-2000 Digi-Scratch function emulates the sound of scratching on vinyl records. The effect may be actuated by turning the Digi-Scratch turntable and is available in three different playback modes: forward, backward, and both. The unit also features a sampler which has a maxi-



Zoom's Player 3030 multieffects system

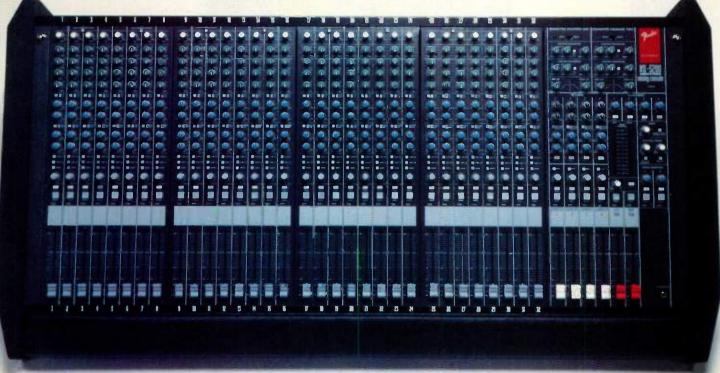
mum record time of 16 seconds and a frequency response of 20 Hz to 20 kHz nels with six inputs, stereo

@ ±50% pitch. The mixer section features three chanoutputs and a sub out. Channels one and two can each be switched between line/phono and line input, while channel three switches between mic and line input. For more information, contact Denon Electronics, 222 New Road, Parsippany, NJ 07054. Tel: 201-575-7810. Circle EQ free lit. #133. ER

TIM Get up and check out Pyramid. We'll give you what you want in pro-audio equipment and what you need in service. And that's something you won't have to lose sleep over. 708-339-8014 708-339-8024 Fax Phone RAMID CALL TO GET ON OUR HIEROGLYPHS MAILING LIST



FENDER MX-5232 MIXING CONSOLE



WHAT'S IN A NAME? Well, ouite a lot really, because some names are more evocative than others. Take Fender - you immediately think of contoured bodies, machineheads all on one side, and single-coil pickups that made rock 'n' roll. Fender may be guitars and screaming backline to most people, but it now also makes a rather good range of sound reinforcement products, including speakers, power amps, and SR mixers. The Fender MX-5200 range of consoles are the sort of no-nonsense. down-to-earth desks there really ought to be more of. This is good gear.

Available in 16-, 24-, or 32-channel versions with separate power supplies, the 5200 series is extremely well built. It provides four groups with 3-band EQ, four auxes, and two monitor sends on the channels, four stereo effects returns, and (unusu-

al) two 2-band parametric EQs with high- and low-pass filters that can be assigned to the two monitor sends or can be patched physically into any insert on the desk. [A spare power supply and rack-mount adapter for mounting up to two power supplies in a 3-space rack are available as accessories. The 5200's power supply includes a built-in automatic changeover and protection provision for redundant operation, meaning that if the supplies are hooked up together and one should fail. the remaining supply will automatically switch over and operate the console. Spare supplies retail for \$399.99; rack carrier tray to hold two supplies: \$74.99.]

Each channel has balanced inputs plus a TRS jack insert and a mic input with switchable phantom power in four channel blocks. Channels have a gain pot with 48 dB of boost, access to the four aux sends from two pots (1 and 3 are post, 2 and 4 are pre, but may be jumper modified) and two monitor send pots switchable pre/post. EQ offers ±15 dB on fixed shelves at 12 kHz and 80 Hz and a sweepable mid peak covering 100 Hz to 7 kHz. There's routing to the main mix; four subs via a pan pot; a solo (switchable globally for Stereo-in-Place); a nice big Mute switch; and a smooth longthrow fader with associated

ROAD TEST

MANUFACTURER: Fender Musical Instruments Corporation, 7975 North Hayden Road, Scottsdale, AZ 85258. Tel 602-596-9690.

APPLICATION: Live sound.

SUMMARY: Four-bus console with 3-band EQ, four auxes, two monitor sends, and four stereo returns, plus two "assignable" parametric EQs.

STRENGTHS: Dead easy to use; well built and laid out; excellent sonic performance.

WEAKNESSES: Lack of metering; no EQ bypass; aux access somewhat restricted; no master Solo indicator.

PRICE: \$3259.99 (32-channel as reviewed). EQ FREE LIT. #: 134

BY ZENON SCHOEPE

Make Your Studio World Class

For Under \$2,500

Thousands of project studios equipped with ADAT or DA-88 digital recorders are discovering that recording direct, through on of the Focusrite Reds, enables recording quality the equal of the world's finest studios.

Each of the Reds featured here include the premier outboard microphone preamplifier, the classic Rupert Neve – designed Focusrite Mic-Pre. Transformer coupled for optimum impedance matching, this classic delivers all the performance of your chosen microphone direct to the recorder (RED 8) or via a subsequent processing stage.

Focusrite

Θ

0

Focusrite

focusrite

Recording direct, with optimised signal level, is clearly superior to tracking through a mixer, the performance of which is compromised by the design budget.

Each of these Reds enables you to match the level to the recorder, getting the best performance from the A-D converter, without compromise. RED 6 combines the classic mic-pre with the equally classic Focusrite EQ: High and low pass filters, high and low frequency shelving controls and two parametric mid bands. Also featured are balanced line input, VU metering and output fader. \$2,495*

RED 7 is similar, but features dynamic control instead of EQ. The mic-pre is followed by a compressor, (identical to the TEC Award winning RED 3) and a de-esser /exciter stage. This is the best vocal front end in the business (in our humble opinion!). \$2,495* RED 8) simply provides two channels of the Focusrite mic-pre. If you've never had the chance to use one, you'll be amazed when you do! Your microphone will reveal hidden tonal & dynamic qualities whether it's a condenser, ribbon or dynamic. \$2,295*

 \bigcirc

0

The Red Range - World Class.



Made in the United Kingdom by: Focusrite Audio Engineering Ltd. 2 Bourne End Business Centre, Bourne End, Bucks, SL8 5AS England. Tel: +44 1628 819456 Fax: +44 1628 819443 Distributed by: Group One Ltd. 201 Wilshire Blvd. # A18 Santa Monica CA. 90401 Tel: 310 656 2521 Fax: 310 656 2524 In Canada: Sonotechnique Tel: 416 947 9112 Ontario Tel: 514 739 8731 Ouebec

CIRCLE 86 ON FREE INFO CARD



the center of the recording industry. Learn recording engineering at the

INSTITUTE OF AUDIO RESEARCH

Our 9 month Recording Engineering and Production Program includes: Multitrack Recording, Mixdown Techniques, MIDI, Digital Audio & more. + a diploma from a world renowned school.

Call 800-544-2501 212-777-8550

Your focus is music, our focus is you.



INSTITUTE OF AUDIO RESEARCH 64 University Place New York, NY 10003

CIRCLE 42 ON FREE INFO CARD



peak LED. Mutes are also available with peak indication on the four subs and the two main output faders.

Only two meters which follow the main stereo mix - are provided, but they also register Solo levels. The lack of more extensive metering is to an extent overcome by the presence of peak LEDs, which make setting up optimum gain structures through the desk very easy and inspire confidence. Fender offers an optional, analog VU meter bridge for \$599.99. The bridge has eight illuminated VU meters one for each submaster, the stereo master, the mono sum out, and the PFL/AFL solo.

The four stereo effects returns can be routed to the main mix or to a sub in mono and can also be fed to the two monitor busses. Subs can be switched individually to the main mix via a pan. The four aux masters have Solos and the two monitor masters additionally have mutes and peak LEDs. A summed stereo mono output is also supported on its own level pot for feeding center clusters.

The inclusion of two "floating" 2-band parametric EQ sections is a peculiar, but extremely useful, addition. A three-position switch bypasses these sections individually or hard wires them to the two monitor busses for EQing stage monitors. The third position routes the EO into whatever insert you have physically patched them into. As the main output, the four subs, and each channel all have TRS inserts, it means you can patch an EQ section anywhere you like on the desk as well as using it on external sources.

The two parametric EQs have high- and lowpass 12-dB/octave filters sweeping 10–300 Hz and 3.3–50 kHz, while the two fully parametric bands offer ±15 dB and cover total ranges via a x10 multiplier switch of 80 Hz to 22 kHz and 8 Hz to 2.2 kHz with fully variable bandwidth from 0.4 to 2 octaves.

IMPRESSIONS

The above-described EQ really is wonderful, and while it would be nice to have this level of control on each channel, the basic channel EQ is not bad. The HF and LF are superbly responsive and the midrange is broad enough to be useful but not so broad as to be unmanageable. Patch a parametric into a channel when processing a really critical source and you'll have no need for outboard equalizers (more frequently, it could be used to eliminate feedback in the on-stage monitoring feeds). The filters are a real luxury and the mid bands are sweet as a nut on their broadest bandwidth. This is quality stuff.

Indeed, the whole desk is quality — it feels rugged and dependable. The sound quality is really quite exceptional and it's clean through the board with predictable overload characteristics on the inputs, good crosstalk, and excellent mute isolation. Even the headphone amp is hi-fi, with stacks of level on tap — something you can't say about all desks.

On the downside, there is no master Solo indicator LED; the auxes are a bit fixed unless you choose to jumper them to suit yourself; and there is no EQ bypass. There are, however, LED indicators beside each input and output Solo switch, plus LED indicators on each mute switch. Still, I'm very

impressed. This live desk is fast and easy to use, which is what you want from a live desk.

Fender guitars? Mixers as well now. Good gear.

the e r an c o f power of N/DYM. the 4 P.M. MS2500 Wirelen Systems -A GLOBAL SUCCESS ...

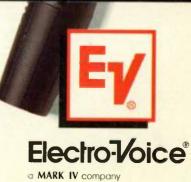


After topping the charts in Japan,

Australia and Canada with their single "Sukiyaki," 4 P.M. found today's live audiences expect CD quality sound. 4 P.M. lead singer kay Pena says, "During our first world tour, we were using a different wireless system every night. Only the EV MS2500 gave us the strong,

instant response we learned to expect from wired microphones."

Check out 4 P.M.'s latest album, Years from Here, available from London/Island Records. Check out the Electro-Voice MS2500 wireless systems, featuring the N/DYM N/D757B microphone, at your local EV dealer.



... A NEW WORLD WIRELESS.

THE MS2500.

Electro-Voice, Inc. a MAR& IV company 600 Cecil St. Buchanan, HI 49107 616/695-6831 800/234-6831 In Canada: 613/382-2141

World Radio History CIRCLE 87 OM FREE INFO CARD



AUDIO TOYS PRO⁶ AUDIO PROCESSOR

WHEN ATI introduced the Paragon console, it included every feature anyone had ever thought to be useful on a live mixing console. There was no need for a rack of dynamic processors and all those patch cables because every input channel included a complete compressor and noise gate. The Paragon console defined the highend features for live-sound mixing, but it was financially out of reach for most touring companies and venues. For many users this level of processing on every microphone input may be unnecessary, so ATI (now known as Audio Toys, Inc.) presently offers the complete input section of the Paragon console in a single-rack-space processing unit, the PRO[®]. [Dimensions: 1U, 1%" H x 19" W x 11" D; weight: 7 pounds.] You can now have a Paragon mixer's input for the voices and instruments that most need it.

As a stand-alone processor, the PRO⁶ remains a high-end processor with all the power you would expect from an individual mic preamp, parametric equalizer, compressor, or ducker/noise gate. The single-channel PRO⁶ can not only produce the great sound of a high-quality

processor, it can also produce the kind of special effects that only professional-strength processing is capable of achieving. This is not a device with a control range that remains within the limits of what will always sound good - dramatic effects or drastic repairs are also possible. After all, this is the input section from what was designed as the ultimate live mixing console, and in live sound there is no Take 2, so last minute drastic action is not unknown in processing signals.

The PRO⁶ can accept mic or line-level inputs and, via a detented control, apply as much as 65 dB of gain. The input polarity can be reversed and 48 V phantom power can be switched from the front panel. Mounted concentrically with the input gain is an auxiliary output level control. This auxiliary output can be fed from the signal immediately after the mic preamp section, following the EQ section or following the dynamics sections. With the very flexible signal routing within the PRO⁶, this offers users an output for recording or as a monitor send that can share or ignore

ROAD TEST

MANUFACTURER: Audio Toys, Inc. (ATI), 9017-C Mendenhall Court, Columbia, MD 21045. Tel: 410-381-7879.

APPLICATIONS: No compromise analog signal processing for live sound or recording.

SUMMARY: From the microphone preamp to the main output, the PRO° offers complete control over the signal and all the flexibility of discrete processors.

STRENGTHS: Powerful fully-parametric 4-band EQ; access to the sidechains of both the compressor and ducker/noise gate; major amounts of headroom.

WEAKNESSES: Powerful processing can be badly misadjusted by the uninitiated; cooling fan may be audible in some situations.

PRICE: \$2295

EQ FREE LIT. #: 135

BY WADE MCGREGOR

any or all of the primary processing required for the main signal. A total of 80 dB of gain is available to the auxiliary output signal, an additional 15 dB more than the main output.

The equalization of the PRO[®] is more flexible than many stand-alone parametric EQs. First, there are continuously variable high-pass (10 Hz to 320 Hz) and lowpass (1.2 kHz to 40 kHz) filters with a bypass switch. Then there are four bands of parametric EQ, which offer generous overlap and can be switched to peak or shelf filter types. Add to this a variable Q of between 0.3 and 3 octaves [see fig. 1], and you should be able to find a filter shape that suits any signal you will encounter. All the EQ controls serve double functions: concentric controls for EQ amount (dB) and frequency center, while the continuous control for Q is also a push-switch to alternate between shelf and peak. I found that some of these switches would occasionally need to be overpressed to activate, because the control's shaft was binding slightly on the front panel.





SUBSCRIBE



DON'T MISS THE NEXT ISSUE! SUBSCRIBE TO EQ & SAVE

Yes! I can't afford to miss the next issue of EQ. Sign me up to subscribe for ... \$24.95 for 1 year (12 issues) \$39.95 for 2 years (24 issues) Prices good in US only: US: 1 yr. \$24.95, 2 yrs \$39.95; Canada add \$10.00 per yr for surface; other countries add \$15.00 per yr for surface; All add \$30.00 per year for Airmail VISA, MasterCard, American Express, or International Money Order only. 4-6 weeks for delivery of first issue. Payment in US\$ drawn on US Bank only. Method of payment: Bill me Check/Money Order Visa Master Card AMEX Card # Exp. Date Signature Phone (Name Address City State/Prov. Zip/PC

BUSINESS REPLY MAIL

FIRST CLASS

PERMIT NO.680

BALDWIN, NY

POSTAGE WILL BE PAID BY ADDRESSEE



P.O BOX 0532 BALDWIN, NY 11510-9938

հուկուկվերուկկիսերիներուկիսերին

NO POSTAGE NECESSARY IF MAILED IN THE UNITED STATES



SUBSCRIBE



PO BOX 0532 BALDWIN, NY 11510

	REE PROD	υC	Т		IN	1 F	0	RI	MA	AΤ	Ι	0	V	
	this FREE Literature Ca products and services													
adver	tisements or editorial	and	you	wil	l re	ceiv	e b	roct	iure	s, b	uyin	g ti	ips	
and literature from EQ	manufacturers absolute	ly FR	EE.	Cir	<u>cle</u>	as n	nany	nur	nber	s as	you	ne	ed.	
FOR FREE INFORMATION AND BR										or must	ic, rea	ordir	s &	
 Print your name and address Circle all the numbers you 		sour a.		upmen er \$5	nt dur .000					.00,000				
Name		b	_\$5,0	00-\$1	0,000		E	\$100,	000-\$	250,000				
Street	StateZip	c			25,000 50,000	9	J•	_Over	\$250	,000				
	7ax					0 77				— ••		Ξ.		
1. Check one category that	cSerious hobby		IS YC		n e	Q 80	JBSC	KTRI	SKT		88 999)		NO 888)	
describes your primary	dStudent	01	02	03	04	05	06	07	08	09	10 :	11 1	2 13	3
involvement in recording & sound;	eOther (Specify)	14	15	16	17	18	19	20	21	22			5 26	
01 Musician	section in EQ that is most	27 40	28 41	29 42	30 43	31 44	32 45	33 46	34 47	35 48			8 39 1 52	
02Producer 03Studio_owner	important to you:	53	54	55	56	57	58	59	60	61			4 63	
04Engineer	1New products 2. Product reviews	66	6 7	68 01	69 22	70	71	72 05	73 00	74			7 78	
05 <u>Songwriter</u> 06 Educator	3Techniques/Workshops	92	80 93	81 94	82 95	83 96	84 97	85 98	86 99	87 100	88	89 9	0 91	L
07Sound reinforcement	4Columns	ED.	TOR 4											
08Videographer/Editor	5People profiles 4. Check the one that best	101	102	103	104	105	106	107	108		110	111	112	-
09Technician/Consultant 10 MIDI Prog/Multimedia	describes your current	113 125	114 126	115 127	116 128	117 129	118 130	119 131	120 132		122 134	123 135	124	
11Record Company/A&R	investment in music, record-	137	138	139	140	141	142	143	132		146	147	14	
12Pro Audio/Video Dealer	ing & sound equipment:	149	150	151	152	153	154	155	156	157	158	159	160)
13 Manufacturer/Rep/Agency 14 Other (Specify)	aUnder \$5,000 b\$5,000~\$10,000	161	162	163	164	165	166	167	168		170	171	172	
2. Describe your level of	c\$10,000-\$25,000	173 185	174 186	175 187	176 188	177 189	178 190	179 191	180 192		182 194	183 195	184 19(
involvement in music,	d\$25,000-\$50,000	197	198	199	200	201	202	203	204		206	207	208	
recording and sound: a. Full-time occupation	e\$50,000\$100,000 f\$100,000\$250,000	209	210											
bPart-time occupation	gOver \$250,000	Valid		Anni	1 30, 1	1996					ebruary	/ 1004	Teen	-
FOR. FREE. INFORMATION. AND. BR					to des nt dur					or musi	ic, rec	ordin	g &	
Circle all the numbers you		a	_Und	er \$5	,000		e			.00,000				
Name		b	_\$5,0	00\$ 1	0,000		E			250,000				
Street City	State Zip	c			25,000 50,000	9	5 •	_Over	J250	,000				
	?ax					0 ~	-					<u> </u>		
1. Check one category that	cSerious hobby				in e	N 2(198C	RIBI	sr?		88 999)		NO 888)	
describes your primary	dStudent	01	02	03	5 04	05	06	07	08	09	10 :	u 1	2 13	3
involvement in recording & sound:	eOther (Specify)	14	15	16	17	18	19	20	21	22	23 :		5 26	5
01 Musician	3. Check the one editorial section in EQ that is most	27	28 41	29 42	30 43	31	32	33	34	35			8 39	
02Producer 03 Studio owner	important to you:	40 53	41 54	42 55	43 56	44 57	45 58	46 59	47 60	48 61			1 52 4 65	
04Engineer	1New products 2Product reviews	66	67	68	69	70	71	72	73	74			7 78	
05Songwriter	2Product reviews 3Techniques/Workshops	79	80 93	81 94	82 95	83 96	84	85 00	86 00		88 4	39 9	0 91	L
06Educator 07 Sound reinforcement	4Columns	92	93 TOP 4	94	20	96	97	98	99	100				
08Videographer/Editor	5People profiles	E D. 101	102	103	104	105	106	107	108	109	110	111	112	2
09Technician/Consultant 10MIDI Prog./Multimedia	4. Check the one that best describes your current	113	114	115	116	117	118	119	120		122	123	124	
10	investment in music, record-	125	126	127	128	129	130	131	132		134	135	136	
12Pro Audio/Video Dealer	ing & sound equipment:	137 149	138 150	139 151	140 152	141 153	142 154	143 155	144 156	_	146 158	147 159	148	
13Manufacturer/Rep/Agency 14Other (Specify)	aUnder \$5,000	161	162	163	164	165	166	167	168		170	171	172	
2. Describe your level of	b\$5,000-\$10,000 c\$10,000-\$25,000	173	174	175	176	177	178	179	180		182	183	184	
involvement in music,	đ\$25,000-\$50,000	185 197	186 198	187 199	188 200	189 201	190 202	191 203	192 204		194 205	195	196	
recording and sound:	e\$50,000-\$100,000	209	210	222	200	201	202	203	204	200	206	207	208	,
aFull-time occupation bPart-time occupation	£\$100,000-\$250,000 gOver \$250,000 World R	adio 1950	er											-
	9 () Ver (9200,000	USHA		Ameri	1 20 1	inne							÷	

ŗ



հվեսինումիսինունեններիունիստենությի

AD INDEX

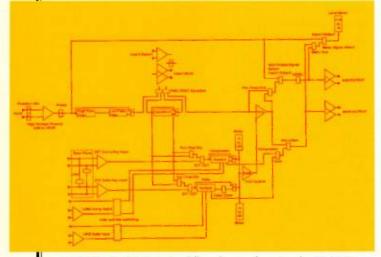
PAGE	BRAND	INFO#	PHONE#	PAGE	BRAND	INFO#	PHONE#
67	3M	01	800-752-0732	95	Klipsch	46	501-777-0693
46	Acoustic Science Corp.	03	503-343-9727	60	KRK	47	714-841-1600
73	AdB International	04	404-623-1410	105	Kurzweil	48	310 926-3200
2-3, 13, 57	Alesis	02, 05, 06	800-5-ALESIS	56	Leigh's Computer	49	215-658-2323
53	Amek Systems & Controls	07	818-508-9788	89	Linear-X	64	503 620-3044
9	Ampex	13	415-367-3809	26-27	Mackie Designs	50	206 487-4333
51	Aphex	08	818-767-2929	128	Martin Audio Duplication	51	800-553-8906
75	Applied Research & Technology	11	716-436-2720	144	Musicator, Inc.	52	800-551-4050
96	Ashly Audio	15	716-544-5191	66	New Frontier Electronics	53	215-862-9344
87	ATI	14	410-381-7879	129	On Point Productions	55	800-331-8841
39 100	Audio Technica	09	216-686-2600	20-21, 43, 48-49, 119		56, 58, 35, 88	601-483-5365
132-135	Azden Corporation B&H Photo	16 18	516-328-7500 212-807-7474	121	Phonic Hi-Tech	90	714-253-4000
46, 65	BBE			14	Polyline	60	708-390-7744
40, 03	Behringer	19, 91 10	714-897-6766 516-364-2244	78	PreSonus	61	504-767-7887
128	Bell Electronics	20	903-784-3215	74			
5	beyerdynamic	12	800-293-4463		Pro Sound & Stage Lighting	62	800-672-4128
54	Caig Laboratories	21	714-897-6766	109	Pyramid Audio	63	708-339-8014
64	Conneaut Audio Devices	21	800-762-9266	15	QSC	54	714-754-6175
66	Countryman Associates	22	415-364-9988	79	Rhythm City	65	404-320-7253
59	Crown	17	219-294-8200	128	Ridge Runner Products	66	800-FRET-PRO
129	Crutchfield	24	800-955-9009	63	Rockford/Hafler	57	800-366-1619
74	D.W. Fearn Company	92	801-568-7660	33-36	Roland	67	213-685-5141
23	dbx	26	801-568-7660	55	Samson Pro Audio	68	516-364-2244
40-41	DigiTech	28	801-566-8800	147	Sennheiser	70	203-434-9190
120	Disc Makers	29	215-232-4140	30-31	Sony	XX	800-635-SONY
56	Discount Distributors	31	516-563-8326	14	Sound Deals	71	800-822-6434
102	Ebtech	94	619-679-6510	103	Spirit by Soundcraft	72	818-909-4500
38	E-mu Systems	32	408-438-1921	102	Stedman	95	616-629-5930
113	Electro-Voice	87	616-695-6831	10	Støder	73	615-391-3399
128	Empire Records	33	716-871-3475	101	Studiomaster	59	714-524-2227
25	Ensoniq	25	215-647-3930	85	Summit Audio	74	408 464-2448
47	Event Electronics	30	805-962-6926	127	Sweetwater Sound	89	219-432-8176
99	Fender	39	602-596-9690	11	t.c. electronic	75	805-373-1828
111	Focusrite	86	516-249-3662	45	Tannoy America	69	519-745-1158
44	Forge Recording Studios	34	610-935-1422	16-17	TASCAM/TEAC America	76	213-726-0303
86, 88, 90	Full Compass	36	800-356-5844	107	TDK	78	516-625 0100
128	Geoffrey Daking & Co.	38	302-658-7003	72	The Recording Workshop		
44	Grandma's Music & Sound	40	800-444-5252	94		77	614-663-2544
94	Harris Allied	41	217-222-8200		Thoroughbred Music	79	813-238-6485
91, 117	Innovative Quality Software	27, 37	702-435-9077	61	Th Audio	80	905-420-3946
112	Institute of Audio Research	42	212-777-8550	77	Vestax Corporation	81	707-427-1920
128	International Audio	43	708-734-1695	79	West L.A. Music	82	310-477-1945
148	JBL Professional	XX	818-895-8190	76	Wnirlwind	83	716-663-8820
129	Jim's Music Center	44	714-552-4280	72	Whisper Room	93	615-585-5827
129	Klarity Kassette	45	800-458-6405	7	Yamoha Pro Audio	84	714-522-9011



The compressor and noise gate are completely independent processors. The compressor includes controls for threshold, ratio, and make-up gain, plus switchable soft-knee mode and high-frequency enhancement. The combination of soft-knee compression with ratios up to 10:1 and the high-frequency enhancement offer remarkable degrees of compression without the usual loss of definition. [Manufacturer's Note: The PRO compressor circuitry is a patent of ATL The high-frequency enhancement is always present in the compressor circuitry, but acting on the sidechain of the VCA to reduce the amount of high-frequency content that

ing where you want some instant help in overlaying two sounds.

The combination of mic preamp and compressor is especially useful in live sound applications to reduce the large dynamic range of some performers and to help when ringing out feedback from the monitor or FOH system during setup. Set a relatively low threshold on the compressor, and then as you bring up the microphone gain to the level of feedback (with an active sound source at the mic), the howl is held at a level that prevents damage to ears or loudspeakers. As you notch out each feedback ring, the compressor will bring up the gain to the next feedback



GET IN THE FLOW: A signal flow diagram for using the ATI PRO.

would be attenuated under high compression ratios.] The noise gate is complete with on-board key filters to add frequency as well as level discrimination to the key source. Attenuation can be adjusted from unity up to -60 dB: Attack from 10 ms to 100 ms; and Hold and Release adjustable between 4 ms and 4 seconds. This is enough range to transparently remove leakage from a vocalist's mic or deliberately add transients to a dull sounding kick drum. The gate can also be switched to perform duckring. Quick and painless feedback equalization.

Metering is complete and concisely grouped on three very bright 10-segment LED ladders. Input and Output, both main and auxiliary, can be selected with pushbuttons on a meter adjacent to the Input/Aux level controls. Compression gain reduction is displayed beside the compression-mode switches, and gate attenuation is displayed beside the gate switches. It would be my preference to have these

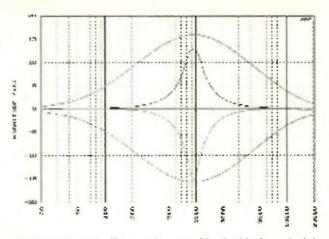


FIGURE 1: The PRO° offers a wide range of bandwidths for each of the four EQ bands. Shown above is EQ band two set to 950 Hz with full boost and cut (±15 dB) and Q settings of 0.3 and 3 octaves. Each band can also be switched between peaking (shown above) and shelving (not shown) filter types.

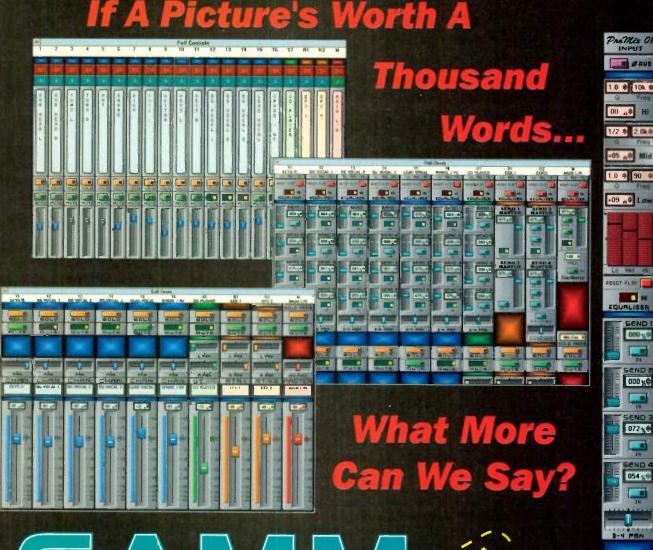
meters active even in bypass so that I am aware of the level of gating or compression I can anticipate when engaging this processing.

Front-panel labeling is dense, and some labels for pushbuttons do not clearly indicate which mode is active when the button is depressed. Fortunately, the labeling is consistent, and with use this confusion should be short-lived. However, it is not immediately apparent from the front panel that the Q controls for the parametric are also the shelf/peak switch. Any confusion can be quickly dispelled by the user's manual, where concise descriptions of each function and clear drawings of the related controls or connections appear.

In keeping with the high-end heritage of the PRO⁶, all audio connections are fully balanced. In fact, the rear panel is covered in them. Nonetheless, there is still room for ground-lift switches (input, outputs and inserts) and clear labeling. The input connection and main and auxiliary output connectors are XLRtype. All the other audio and DC control connections are 1/4-inch phone jacks. Both the compressor and

gate include separate link connections for In and Out (DC control of the VCA for stereo operation) in addition to external audio inputs for triggering the thresholds. All four of these connections can be selected individually from the front panel, thus making permanent connections to a patchbay very practical. There are also insert send and return connections that can be switched in/out of the signal path and routed pre/post EQ. These are more connections than many dedicated dynamic processors provide.

The ATI PRO⁶ takes high-end mixing console technology and places it within reach of the project studio, regional/local SR company, and club performer. This is a great combination of an excellent preamp and processing that will allow them to make adjustments that can range from the tiniest tweak to a sledgehammer blow. The PRO® qualifies as both an esoteric mic preamplifier and as an excellent signal processor, and is worth the investment even if you are only after one of its many capabilities. After all, there are bypass buttons on every section. EQ





Experience the joy and power of virtual mixing technology today!

KICK DBU

Multiple zoom levels provide simultaneous views of the entire console or control details.
 Create groups to control all functions simultaneously, even across multiple consoles.
 Control Up To 16 ProMix 01 consoles at once. (Additional licenses required)
 Create Cue sequences to allow complex mix changes to be automated.
 Generate or chase-lock to SMPTE for complete mix automation.
 Control All-Mixer MIDI-controllable Functions on-screen
 Manually step or cross-fade through Cue sequences.
 Fast, smooth scrolling of entire graphic image.

INNOVATIVE QUALITY SOFTWARE. Designed by Audio Engineers for Audio Engineers!

TECH INFO 702-435-9077 TO ORDER 800-844-1554 FAX 702-435-9106 BBS 702-435-7186

CIRCLE 37 ON FREE INFO CARD World Radio History

Steroids for Electric Guitar

How to build your own IGGY and get better sound from guitars BY CRAIG ANDERTON

ne of the surest ways to kill a guitar's tone is to load it down with a low-impedance input (like most mixing consoles and proquality effects have). As mentioned in this month's MI Insider, an impedance mismatch can result in a dull, lowerlevel sound — definitely not the way to have your axe cut through the average wall of noise.

Throw in the nasty effects of cable capacitance, hum pickup, and other

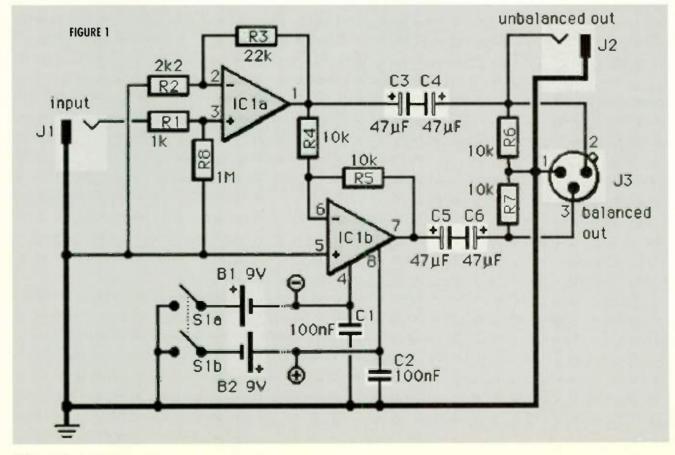
gremlins, and it's amazing guitars sound as good as they do. One solution to all these problems is to immediately follow the guitar output with a preamp that presents the guitar with a very high-input impedance, and transforms the signal to one with a very low impedance. Not only does this preserve the level, it also counteracts a cord's cable capacitance with long cable runs.

Although there are lots of preamps in the world, IGGY (Interface Gizmo for Guitarists) is designed specifically to make guitars and basses sound better in the studio. For a little work and about \$30 in parts, this box will let you patch a guitar directly to a mixing console, drive effects with XLR balanced inputs, and allow for extremely long cable runs without sound degradation. And if you're a real purist, you can also record right into the balanced inputs of tape or harddisk recorders and DATs, thus bypassing all the mixer electronics (IGGY's signal-to-noise ratio is better than 90 dB, and the frequency response extends past 100 kHz). The improvement in sound quality can be significant.

WIRE FOR HIRE

Fig. 1 shows the schematic; the main ground lines are drawn a little thicker. IC1 is an NE5532 dual low-noise op amp (this chip is used a lot in pro gear because of its excellent specs), however, you can substitute a TL082 or TL072 (slightly lower noise than the 082) if the NE5532 is hard to find. To find good parts sources, go to your newsstand and buy a copy of *Electronics Now* or *Popular Electronics*. A variety of parts houses advertise in these magazines (and you'll get much better prices than Radio Shack).

IC1a converts the guitar's output from high-impedance to low-impedance, while adding approximately 20



PEAVEY AND EDWARD VAN HALEN ANNOUNCE THE WOLFGANG GUITAR

Todd Cowden

ANAHEIM, CA - At the 1996 NAMM show, Peavey Electronics and Edward Van Halen announced the release of the Wolfgang guitar. -

the design of the guitar was a joint effort by Hartley Peavey and Edward Van Halen

Peavey Electronics

711A Street

Mendran, Missessippi 39301

Lix Armstong H

For set answers and no tricks Grab the phone and stat call La-

Qator

ion: Anaheim, California

fgang Guitar

wolfgang

(601) 483-5365 • Fax (601) 486-1278 • 1995 State Meridian, M

PARTS LIST

Resistors (5% tolerance, metal film
preferred for fixed resistors)
R11k
R2 2.2k (2k2)
R3
R4-R710k
R8
Compresidence 125

Other Parts

IC1-IC3..... NE5532 or TL082 dual op amp

J1, J2 Mono, open circuit,
½" phone jack
J3 XLR plug or chassis-mounting
male XLR jack (see text)
S1a+b DPST or DPDT switch
B1, B2 9-Volt battery (see text)
Misc IC socket, perf board, case,
wire, power supply, etc.

Specs

Freq Response: ±0.1 dB, 10 Hz-100 kHz

S/N Ratio: Greater than 90 dB Input Impedance: Greater than 500 kohms

Output Impedance: < 600 ohms Headroom (+15 V supply): Greater than 26 V peak to peak Gain: 20 dB



dB of gain. IC1b provides the additional output needed for balanced line operation. J2 sends the signal to unbalanced inputs, and J3 patches to balanced inputs. Since you can use both outputs simultaneously, IGGY is also an active splitter.

BUILDING IT

The simplest approach is to mount the parts on a perf board or printed-circuit board, which then mounts inside a metal case with a cable trailing out to an XLR plug. Install the 1/4-inch jacks in the box itself. Use metal-film resistors for lowest noise, and a socket for IC1 to prevent heat damage while soldering.

Power IGGY with two 9 V batteries as shown, or any bipolar power supply between +5 and +15 V. If you're not going to use batteries, remove them and S1 (which turns the battery power on and off); feed in the external supply at the points marked (+) and (-) on the schematic.

To use IGGY, plug your guitar into J1 using as short a cable as possible, and patch J2/J3 into the console input, processor input, or long cable run. Play away — you might be surprised at just how clean and crisp a guitar can sound when it's not being loaded down.

MOD SQUAD

You can alter IGGY's gain by changing R3's value. The amount of gain equals (R3 + 2.2)/2.2, with R3 in kohms. For example, R3 = 22k gives a gain of 11; 100k gives a gain of approximately 45.

J3 is wired with pin 2 "hot," in accordance with the IEC standard for XLR connectors. However, some maverick gear uses pin 3 as "hot." You can accommodate this gear by reversing the wires going to J3 pins 2 and 3, but this isn't really necessary unless you're using IGGY as a splitter.

And that's all there is to it. Aren't you glad you know how to solder?

(Note: this article is adapted from the book "Do-It-Yourself Projects for Guitarists," published by Miller-Freeman books.)

Craig Anderton is technology editor to EQ magazine and a consulting editor to Guitar Player magazine. Visit his "Sound, Studio, and Stage" area on America Online.

120 FEBRUARY EQ CIRCLE 29 ON FREE INFO CARD



MM122 Retail \$349

PMC-B Series Starting at \$599

ARE YOU TIRED OF PAYING TO MUCH?



M-Series If you have to ask...



CALL FOR FREE LITERATURE 1-800-524-3050

Y

IN REVIEW

Valley Audio 730 Dynamap



MANUFACTURER: Valley Audio, a division of Galaxy Audio, Inc., 601 E. Pawnee Avenue, Wichita, KS 67211. Tel: 316-265-9500.

APPLICATIONS: Audio processor for the recording studio.

SUMMARY: Lots of versatile features makes this unit stand out among its contemporaries.

STRENGTHS: Very versatile; plenty of features; good A/D-D/A converters. WEAKNESSES: Won't get my morning paper; interface takes some getting used to. PRICE: 730DD (digital), \$2798; 730ADD (w/analog), \$3548 EQ FREE LIT. #: 136

IF YOU remember the Kepex, the Gain Brain, the Headroom Horseman, and the Dynamite, then you have been around audio for a while and are familiar with Valley Audio. Valley Audio has been around for 25 years producing solutions for audio dynamic control. Not a company to rest on its laurels, 1 offer a hardy (1 couldn't resist) congratulations on their entry into digital dynamic control.

The Valley Audio 730 Dynamap provides an entirely fresh approach to dynamic control. Besides doing an exemplary job at the normal dynamic parameters, the 730 has added features that are virtually impossible with analog dynamic processors. The expected parameters, such as Attack, Release, Threshold, Gain and Ratio, have been joined by Range, Shape, Mix and Stereo Width. The only thing I haven't been able to make the 730 Dynamap do is get my morning paper.

The 730 Dynamap is available in two versions: 1) The 730DD (not to be confused with any measurements on the 50-foot woman) is fully digital in and out; 2) The 730ADD has all of the digital capabilities plus stereo 18-bit delta sigma A/D and D/A converters. The analog inputs can be balanced, unbalanced, pin 2 or pin 3 hot, and ground lifted. The digital inputs available are AES, S/PDIF, and optical. Any two of the three inputs can be operational at the same time. The second input can be routed to the Key/Mix input. If Mix is selected from the Key/Mix menu, then both inputs can be mixed together digitally.

FEATURES

Peak Limiter: Because the 730 is a digital device, it can look ahead to see what is going to happen before you actually hear it. LSB truncation in conjunction with the look ahead feature provides distortion-free peak limiting with the elimination of pesky "overs."

High-Frequency Compressor (De-esser): Digital domain de-essing is one of the best applications of the technology. High Q filters with phase correction allows processing sibilance without turning "S"s into "F"s. With the resurgence of analog mastering, sibilance processing has become more important than it was for CD-only mastering.

Stereo Dimension Enhancement: You can reduce the stereo separation all the way down to mono, or you can enhance the stereo separation by up to 6 dB. 1 tried it on some old 2-track tapes with excellent results.

Digital Level Control: You can raise or lower the level digitally to get the most use out of those 16 bits.

Mix: This feature lets you do some amazing things with the compressor. You basically have two threshold levels. As the level exceeds the first threshold, compression starts compressing at the set ratio. As the level gets higher, some of the source audio mixes in and starts canceling the compression. Loud peaks get through uncompressed-compressed, while the meat of the audio can be made to sound loud without pumping. [The Mix parameter is available when using the Standard, Enhanced or Map compressors simultaneously with the Expander. When the Mix value is set to 99 percent or less, some unprocessed signal will be present regardless of whether the threshold has been reached. As the amount of unprocessed signal increases, the knee softens, that is, the effective ratio changes. The amount of effective make-up gain decreases at the same time.]

Memory: There are 495 user-programmable storage registers for settings and parameters.

OPERATION AND SOUND QUALITY

The 730 Dynamap has more parameters to set than you can shake a stick at. It takes a while to get used to surfing around the menu tree, but once you get the hang of it you get from



anywhere to anywhere else with just a few keystrokes.

I spent about a month cruising the compressor and de-esser. With the extra parameters added to the compression settings, the combinations are almost endless. For me, this is a good thing. There have been many times when I have inserted a compressor on an entire mix to see if I could get the mix to sound a little louder. It would sound fine for a while, and then when the drum fill came, SUCK, SUCK, duhSUCK. With the 730 controls, you can set up the compressor so that things like drum fills are let through the compressor, instead of driving down the level of the rest of the track.

This brings us to the namesake processing option of the 730, the Dynamap. The characteristics of most compressors can be defined using only a single ratio, threshold, and gain setting. With the 730 Dynamap, you can construct a dynamic map made up of eight segments. The segment slopes and locations are defined by selecting Threshold/Gain endpoints. After you select the Threshold and Gain of each endpoint, the 730 calculates the ratio for each segment to connect the endpoints. The ratio in each segment may be greater than 1:1 (compression) or less than 1:1 (expansion) depending on the endpoints you select.

The High-Frequency Compressor is used mostly for de-essing. There are three listening modes to help you set up the de-esser for optimum operation: Normal mode allows you to listen to the broad band product with deessing active; Tune mode allows you to listen only to the compressed high-frequency band to help you optimize your settings; and Invert mode suppresses the broadband signal 24 dB and the compressors operation is inverted. This causes the level of the effected band to increase as it crosses the threshold, making the problem area jump out so you can further fine tune the parameters.

The 730 Dynamap also has a mode called Static. No, it doesn't add clicks and pops to your audio. In this mode there is no dynamic processing. This is the mode that allows you to adjust the stereo field width, input and output levels. The Key/Mix input is also adjustable in this mode, allowing you to mix in another digital signal if needed. The Main In, Key/Mix In, and Output levels can be changed in 0.5 dB steps from 0 dB to mute. The Stereo Width can be changed from +6 dB to -71.5 dB to Mono in 0.5 dB steps. I haven't figured out exactly how they do it yet, but the +6 dB position actually does widen the stereo image. On the oscilloscope it does not look like the phase relationship is getting out of hand, so I would not be against trying this on some old mixes I want to spiff up.

SUMMATION

The 730 Dynamap is a great sounding piece of equipment. The converters sound good enough to use as your main A/D-D/A for critical listening. Nice and smooth. If you have any doubts about using a digital compressor on analog material because of the extra conversions, forget about it. These conversions won't get in your way. Don't forget that you can use an analog source to key the dynamics on a digital signal, or vice-versa.

If you have ever sat in front of a compressor saying to yourself, "I wish this thing would...," well, this is just the puppy that can.

If you don't have room in your rack for the 730, don't worry. With the polished stainless-steel chassis, you will want to put it out on top of your rack where everyone can see it. Good work, Valley Audio. I can't wait to see what's next. —*Roger Nichols*





CONSOLE ANNUAL. Once a year our editors get their hands on the coolest consoles around, putting the latest desk configurations to the test. Here's what's new from the major purveyors of console technology.

EQ LIVE. Steve La Cerra takes the Lexicon PCM 90 out on the road. And the first in a series of new club reports takes us up I-95 to Toad's Place in New Haven, CT, to find out what the house system has to offer the band-in-a-van.

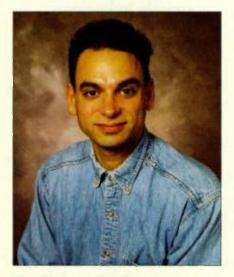
GREAT GEAR. This months' product section is devoted to the mega-intros at NAMM. We're also the first to review the new lomega Jaz drive, and take the very first look at the Roland VS-880. And software heads get a taste of four new noise reduction programs for Windows.

TO BE A PART OF THIS ISSUE, CONTACT:

Kathleen A. Mackay; Associate Publisher (ext. 152) Matt Charles (ext. 147) Daniel A. Hernandez (ext. 150) Andrea Berrie (ext. 148) Christine Cali (ext. 155) Advertising Sales TEL: 212-213-3444 FAX: 212-213-3484

Tube Amplifier Tips

Don't assume the worst (i.e., most expensive) thing when your tube gear breaks down BY EDDIE CILETTI



Vacuum tube (VT) guitar amp designs are anything but conservative. Simple, perhaps, but not conservative. In fact, so much performance is squeezed out of each tube that the end result can be mostly inconsistent — and, occasionally, magical. This explains why enthusiasts are particularly obsessed about subtleties like make, model, and vintage of both the gear and its component parts. Purists will insist only on original replacement parts and zealously argue against replacing a VT rectifier with a solid-state (SS) device.

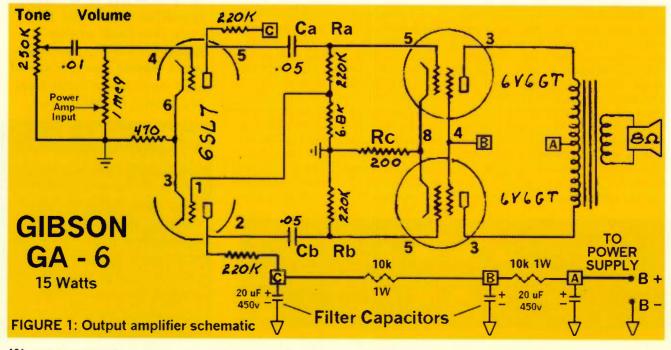
In contrast, some hi-fi gear and most pro equipment is of such conservative design that even marginal tubes will allow them to meet spec. For this — and many other artistically related reasons experience has taught me to treat guitar amps differently from hi-fi/pro gear. For example, distortion in excess of 1 percent is plainly visible on a 'scope. Guitar amps, however, have typical distortion figures of 10 percent — *before* going into overdrive. They never look great on a 'scope, and present a temptation to "tweakheads" like me who must resist the urge to "improve via modification."

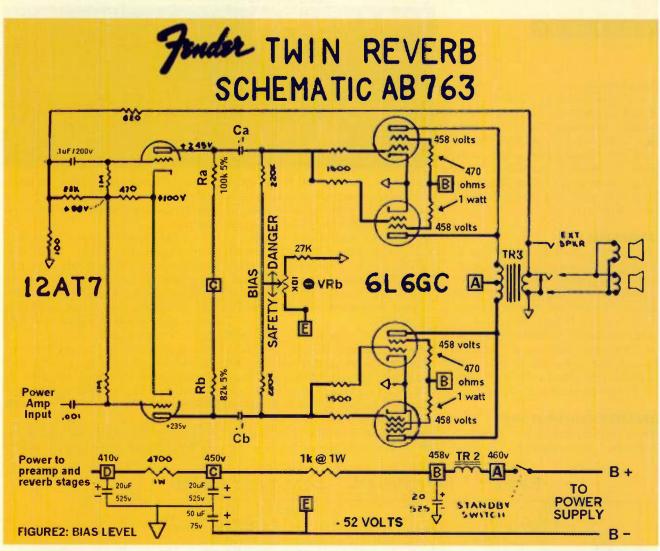
In all cases, failure-prone parts should be replaced even when they *appear* to be good. This approach will save considerable time. Don't, however, assume that the cause of an intermittent problem is the most expensive, hard-tofind part, such as an output transformer. Look instead for the simple, affordable solution: components such as high value resistors (100 kohm and up) and interstage capacitors, both of which often become unstable when they're critically needed.

Here are five, common, easy-tosolve, tube-amplifier maladies:

Hum: Power supply filter or smoothing capacitors lose their ability to hold a charge, resulting in a saw-toothed hum that will vary in frequency depending on the type of rectification. The conversion from AC to DC will yield either a 60 Hz, "half wave" hum when a single diode (VT or SS) is used or a 120 Hz "full wave" hum when two or four diodes are used. The solution is to change the capacitors, which have capacitance specified in micro-Farads (abbreviated as "µF" or "mF") and operating voltage specified in either "volts" or "working volts" (V or WV). Typical values are 20 uF @ 525 WV. Replace solid-state, plug-in rectifiers either with a new model or a vacuum tube. The older type are failure prone and can short-circuit a power transformer designed for VT rectification; a sad and most unhappy smell.

Noticeable Loss of Output Power or





Preamp Gain: After years of burning the midnight oil, the cathode's electron emission will be reduced. This can be verified with a tube tester. The mystery ailment, however, is "gas," and a tube suffering from this disease will be particularly nonlinear — a.k.a. distorted — even when not overdriven.

Noise: While tubes can be a source of microphonics and noise, recurrent problems point to carbon resistors. A significant improvement can be made by replacing them with the metal film variety.

Intermittent Problems: Since guitar amps are subject to road abuse, look for loose components or tube sockets that have lost their grip. Suspect a socket that produces a crunchy noise when the tube is wiggled. Loose sockets will not only cause bias-related, output tube problems such as the dreaded "cherry-red plate" syndrome, but also interrupt the filament power. (I solved one mystery simply by close, visual inspection of the miniature, dual-triode (12A"7) preamp/driver tubes.) Confirm that the filaments in both halves are lit. Tube Matching & Bias: For any amplifier using two or more power output tubes, each tube should be by the same manufacturer and each pair tor quartet, if applicable) should be matched If one tube fails or is damaged, always replace both! Mismatched tubes will make proper adjustment impossible. Not only do guitar amps subject tubes to great amounts of stress (especially if you are keen on distortion), but the relative amount of bias will affect sound and performance.

BIAS: GOOD FOR YOUR TUBES

Bias optimizes the degree to which both tubes and tape are "turned-on." (No bias = the tube will overheat; too much bias = the tube will barely let a signal through.) The primary difference is that biasing tape requires a high frequency (AC) while biasing vacuum tubes requires no frequency (DC). (Transistor amps have bias adjustments too.) Insufficient or disappearing bias, either through poor adjustment or component failure, will shorten tube life, possibly damage related circuitry, and trash the output trans-

World Radio History

former (the most expensive component)

VISUAL INSPECTION

Not all amplifiers provide or require a bias adjustment. If you suspect a problem, one quick and simple user test requires only a close look. Power output tubes have filaments that are supposed to glow red, but they also have plates that should not glow at all.

If you have changed the output tubes with no improvement, then it's a good idea to change the grid resistors (Ra and Rb) on each tube. This resistor is connected to pin 5 of a 6L6, 6V6, 6CA7/EL34, 5881, 6550, and/or pin 6 of a 7027 and pin 2 of a 6BQ5/EL84. The DC blocking capacitors Ca and Cb between the driver and the output stages — should also be replaced. This sort of information can be found in...

THE TREASURE MAP

Three bias circuitry variations are depicted in figs. 1, 2, and 3. These can also be found on America Online in Craig Anderton's SSS under "EQ On-line." Just look for the button with my name! Fig. 1



is a schematic of my most prized possession, a Gibson GA-6. I was lucky enough to find the Master Service Book for this no-frills, tweed-era 15-watt amplifier. Figs. 2 and 3 have been simplified to show two Fender power output variations. [Incidentally, enthusiasts could not go wrong acquiring any of R. Aspen Pittman's tube amp books (Groove Tubes, 818-361-4500).]

CATHODE BIAS

Fig. 1 shows how bias voltage can be developed across a cathode resistor, Rc. If the tubes attempt to draw excessive current, more voltage will appear across Rc (typical values are between 12 and 20 volts), raising it even higher above ground potential. To look at it another way, the grid (which is referenced to ground and essentially at "0" volts potential) will appear to be more negative with respect to the cathode.

SEPARATE NEGATIVE BIAS SUPPLY

When an independent bias voltage is desirable, it is developed by a "tap" on

BEAM A LITTLE BEAM

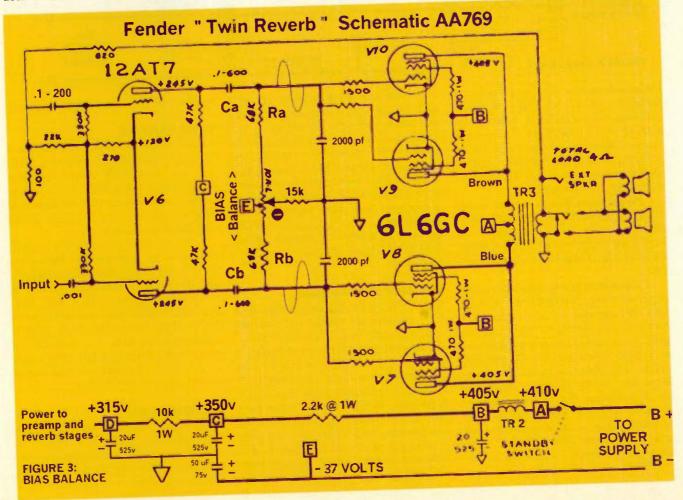
For a detailed look at a beam power tube, check out the June '95 issue of $\mathcal{G}Q$. On page 84, the illustration to the left shows each component clearly abeled. (The sidebar above it also lists sources for parts.) The outermost metal housing is called the Plate, which is made positively attractive to electrons (which are regatively charged) by applying hundreds of volts. Too little bias will accelerate the electrons to ecstasy, bombarding the plate at such force as to cause it to glow red. Eliminate all other sources of illumination to make sure that each plate is not even faintly glowing.

the power transformer to which a reverse-biased semiconductor diode is connected in a half-wave configuration. The voltage ranges between -60 and -35 volts (with respect to ground) when measured before the bias resistors, Ra and Rb. If the juice is missing or not within spec, check the diode, replace the capacitor, then spray and exercise the pot (replacing the latter. if necessary).

The Bias adjustment, VRb, will either vary the amount of voltage to both tubes (fig. 2) or change the balance of the voltages between the tubes (fig. 3). In all cases, if either of the plates in the output stage is glowing, try another set of matched tubes. If the problem persists or is intermittent, replace the bias resistors, the coupling capacitors, and, if loose, the socket(s). While the output transformer may be suspect, all of the other solutions are far more affordable steps worth trying first.

Note: Measuring the julce in a high-impedance circuit requires a voltmeter capable of 20,000 ohms-pervolt (less is not better). Unless an oscilloscope or a really high-impedance DC voltmeter is used, attempting to measure the grid voltage will yield little information of value. In fact, it could momentarily "fix" the problem. In addition, be especially careful when working on high-voltage circuitry. Do not attempt to steady yourself by placing one hand on the chassis.

Contact Mr. Ciletti at: eddieaudio@aol.com.



WHY IS SWEETWATER SOUND THE "RIGHT CALL" FOR YOUR MUSIC?

Anyone can sell you boxes. But what's best for you and your music? There's a good reason why Sweetwater has been America's fastest growing music retailer for three straight years. We understand that when you're investing your hard-earned dollars, you want to get exactly what you need, when you need it and within your budget. Why is Sweetwater best equipped to do what's right for you?



KNOWLEDGEABLE SALES STAFF Our sales engineers are professionals with serious, real world knowledge of the products we represent. Most have their own studios. Many have spent years as engineers at major studios. We even have a Grammy winner on board. We make it our business to be absolutely sure the product you order is exactly the product you need.



HUGE IN-STOCK INVENTORY

We stock products from over 100 of the best names in the business like Alesis, Digidesign, Kurzweil, Mackie, Roland and TASCAM, just to name a few. The "other guys" may tell you they can get it fast, but how long will it *really* take? Our giant warehouse is packed with the latest gear, ready to ship right to your door — in most cases the very same day you order!

BRICES THAT ARE AFFORDABLE Not only do we have the best selection of the finest products in the entire music industry, but cur prices are so affordable, you won't have to take a second job just to start making music!

OUTSTANDING TECHNICAL SUPPORT

Got a question about a product you've purchased or are thinking about purchasing? Just give us a call! Whether you're a novice or a seasoned pro, our technical support team has the knowledge and experience to answer any question you might have. And no other dealer has anything like our huge, online product database that gives our staff instant access to technical information about thousands of products.

BUY WITH CONFIDENCE Because if you're not totally satisfied with any product you purchase from us, we'll work with you to make it right! If you're not happy, we're not happy!

DEPENDABLE OR-TIME DELIVERY Whether you need it in a week or a day, we have a shipping department that's up to the task. We can have your equipment packaged and shipped out within hours of receiving an order.

> We ship via all major carriers. Just tell us when you need it and it'll be there!



LATEST UP-TO-THE-MINUTE INDUSTRY NEWS There are new products being released row about them You

every day, and you want to know about them. You also want technical information and tips on making the most of your equipment purchase. That's why we're the first major retailer to maintain a full-time presence on the Internet. And why we publish our exclusive, free full color *Sweet Notes* newsletter. STATE-OF-THE-ART SERVICE When your gear isn't working, you want it fixed right the first time. Sweetwater is a factory authorized service center for all the products we sell, with over a dozen highly skilled technicians trained right down to component level repairs. What's more, all the work is done right here on the premises, so your equipment is returned to you the minute it's ready with no delays and no excuses.



CUSTOM FINANCING PROGRAMS

We accept cash, checks and all major credit cards, but we also offer our own Sweetwater Technology Card, as well as our exclusive new line of credit accounts and leasing plans.

ADD UP THE VALUE!

Close to 100,000 customers world wide already call us. Shouldn't you be one of them? Call right now and experience the Sweetwater difference!

> THE ULTIMATE MUSIC RETAILER!



CALL NOW! YOU MAY BE **PRE-APPROVED** FOR A \$10,000 LINE OF CREDIT ACCOUNT!

CIRCLE 89 ON FREE INFO CARD

EQ SHOPPER



EQ SHOPPER



FIRST LOOK

continued from page 38

together for stereo operation. The IN-C-2021 has two eight-segment LED meters that can either show output level or gain reduction — very handy for monitoring the status of your signal. And like the two EQ units, the compressor has auxiliary instrumentlevel inputs for taking keyboards or guitars direct to tape. The IN-C-2021 lists at \$1395.

The last member of the Indigo series is the IN-O-2031, a descendanton-steroids of TL Audio's VI-1 vacuum tube line interface. But this unit lets you manipulate the signal with considerably more facility than the typical tube-based line driver. Each channel has continuously variable input and output controls with a boost control (defeatable) providing extra gain at the input to overdrive the tubes. A jack on the rear panel accepts a footswitch for turning the boost function on or off.

In addition to the expected I/O controls, there are some unexpected EQ functions in this unit. There are three rotary pots for fixed-frequency low, mid and high band EQ. While this single set of controls does affect both channels, there are independent EQ bypass switches per channel as well as a continuously variable low-pass filter (also with bypass). These extra features make the 2031 attractive to owners of digital multitracks who are looking to reduce the coldness of their tape machines and experiment in attaining new sounds.

But the really cool thing about the 2031 is that the instrument-level input, signal boost control with foot switching capability and EO enable the unit to function as a high-quality guitar or bass preamplifier. So studio owners get a high-quality guitar pre as a bonus when adding this unit into their rack. Signals passing through the unit are monitored by three-stage LED meters (one per channel) that indicate signal present, peak and clip levels (note that in the photo, the clip LED is labeled "boost" but production units will be labeled "clip"). The IN-O-2031 EQ lists at \$1295.

For more information, contact the Sascom Marketing Group, 34 Nelson St., Oakville, ONT., Canada, L6L 3H6. Tel: 905-469-8080. Circle EQ free lit. #137.

AVALON U5 REVIEW

continued from page 82

need for EQ further down the line.

The remaining features include a "signal present" LED, ground-lift switch, headphone jack (and wow, does it ever sound good when you listen to your guitar through this!), and 115/240 V operation. One feature that doesn't show up on a spec sheet is the design, which can best be described as the krells meet Hallicrafters (inside joke for those familiar with both the movie *Forbidden Planet* and communications transceivers). It's built like a tank. I didn't venture dropping it from my roof, but I bet it would survive if I did.

There is one caution: the input ground is isolated from chassis ground, which is good design practice. However, if you're listening through headphones and don't have the XLRs feeding something, there's no return path to ground from the input jack ground, which leads to beaucoup hums and buzzes. If you find yourself in this situation, clip the input cord ground to anywhere metal on the chassis.

DO BIG BUX BRING DELUXE?

Yes, but think carefully about your studio before you get too excited. This box is overkill for many applications, because unless the rest of your equipment is on a par with the U5, you're paying for performance you can't really hear. But with ultraquiet digital recording gear, the U5 provides a top-of-theline link between the analog world and all those digital audio A/D converters that can't handle low-level signals. The sound quality is beyond clean, and the construction is solid. This is a direct box that comes very, very close to the ideal of a straight wire with gain. For many people, that will be worth the price tag. -Craig Anderton

KORG REVIEW

continued from page 86

are a huge number of sounds lurking in here, because there are so many possible combinations of parameters.

So what's there not to like? Well, the factory presets favor aggressive alternative rock and thrash styles, and feature highly distorted effects. Some players will like the presets as is, but others working in different music styles may find these presets unsatisfying. That would be too bad because the factory patches do not show how versatile Pandora is.

Because I prefer smoother, more vintage sounds, I was ready to write off Pandora until I became intimate with the Editing mode. Playing with the parameters yielded smooth leads, some great rockabilly country sounds — even decent compression once I pulled back on the drive control. If you plug in and don't like what you hear, just remember that there's much more to Pandora than meets the ear initially.

Overall, Pandora gives you major bang for the buck, and the effects are surprisingly good once you figure out how to tweak them. Pandora is not going to blow away a digital multieffects costing twice as much, but the gap between the two is narrower than you might expect. The tuner function is also helpful. Plan to program Pandora if you buy it, because editing will get you beyond the presets and open up lots of more versatile, cool sounds that definitely belie the small size and low cost. —*Craig Anderton*

BLUES TRAVELER

continued from page 106

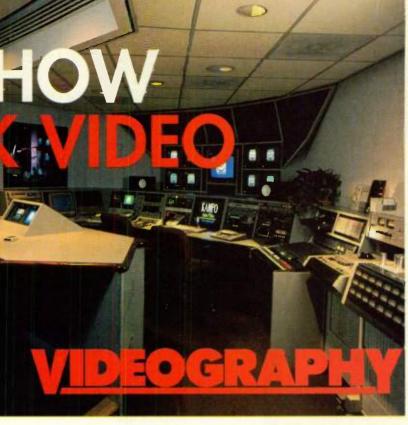
alistic impression of what's going on up there. To combat this, I try to get the subs uncoupled with the floor as much as possible and position them out to the sides. Or I'll form some creative speaker columns. I just try to find different ways for the bass to couple with the area of the room it's working in. When I go way out to the sides with the subs, I get the natural low end tendencies of the room. Then the cabinets are responding more to what's sensitive to me and not what's sensitive on stage. When the sensitivity point spills in towards the stage, it's really difficult to deal with. I'd rather have the sound a little more thick and low-end oriented in the FOH before it gets to the stage. The resonant coupling is twice as bad for the band as it is for the audience. Bass is nondirectional and I think techs sometimes overestimate the time-alignment aspects of it.

Blues Traveler is trying to communicate to the audience, and this makes my job more enjoyable. The most intense part of their show is that it's like a canon — they come on strong and finish strong. Those first 12 bars are a real adrenaline boost for me!

LEARN H TO TALK

THE VIDEO HOW-TO MAGAZINE

Find out everything you need to know about professional video production in the #1 video production magazine — VIDEOGRAPHY. It's the only video production magazine written by video professionals for video professionals.



FILLED WITH EXPERT TIPS

Every month VIDEOGRAPHY contains all the practical, how-to information you need to design, record, edit, and mix your audio for your video customers. It offers you all the tips you need to profit from the growing demand for audio-for-video.

YOUR ESSENTIAL TELEPRODUCTION HANDBOOK

Subscribe to VIDEOGRAPHY today and receive one free issue. If you don't agree that VIDEOGRAPHY is an essential handbook for today's dynamic video production scene, there is no obligation.

VIDEOG	RA	PHY	RECEIVE YOUR FREE ISSUE!
 Yes, I want to subscribe to VIDEOGRAPH One Year (\$30) Two Years (\$50) *Payment in US\$ drawn on US bank only. Enclosed is my check or money order. 	One Year S	Surface Foreign (\$45)	One Year Airmail Foreign (\$60)
Account#Signature			Exp. Date
Name Address			
City			Zip
VIDEOGRAPHY P.O.	BOX 051	3 BALDWIN	NY 11510-0513
	World Padio H	istory	



"THE PROFESSIONAL'S SOURCE

FOR ORDERS CALL: 800-947-5518 212-444-6688

OR FAX (24 HOURS): 800-947-9003

212-444-5001

OVERNIGHT AND RUSH SERVICE AVAILABLE

E-Mail at 76623.570@compuserve.com



Consistently the choice of professional per-formers around the world, the SM58 is a ruged undirectional dynamic mix with a high-by effective built-in wind and pop filter. It's per-formance charactenistics and compact size make it especially will-suited for music and speech applications – anytime the mic musit be used outdoors or close to the mouth. A genuine world standard and adulo legend, the SM58 makes all out vocalists sound their best.

 Self-contained soberical filter to control explosive breath Self-contained spherical filter to control explosive breath sounds ("poof) and wind noise in outdoor locations,
 Unusually effective cardioid pickup patiern to minimize back-inground noise and undersitable effects of studio and location acoustics. Rear and side rejection uniform to very low frequen-cies and completely symmetrical about microphane axis.
 Bright, clean sound, Especially good for vocal music, announc-tion activities and dividences.

Bright, clean Sound, Especially good for vocal music, announc-ing, narration and dialogue.
 Cartridge shock-mounted for protection and quiet operation.
 Wind and "pop" filter removable for cleaning or replacement.
 Versatifity – designed for both field and studie use – hand-held or on a stand (slip-in swire) adapter included).

SM57 Unidirectional **Dynamic Microphone**

The classic SM57 is the "workhorsa" mic of stages and studios worldwide. Its carefully contoured tre-quency response means class, well-refined instru-mental reproduction and rish, warm vocal pick-up on lectems or stages. The SM57 is the standard for drum, percussion and instrument amplifier miking

- drum, percussion and instrument amplifier mixing. Unusually effective cardioid pick-up pattern minimizes effects of studio or location acoustics and background noise. Rear and side rejection uniform to very low frequencies and completely symmetrical about microphone airs. Bright, clean sound. Especially effective for announcing, narra-tion and dialogue, vocal music, and rythym packages. Carridge shack mounted for guiet operation. Versatility for use in the hand or on a stand for use indoors or outdoors. Microphone supplied with adjustable swivel adapter pirmite titing through 90° from vertical to horizontal.

SENNHEISER'

RF CONDENSER MICROPHONES



response to audio transients over an extended frequency range

MKH 20 P48U3 Omnidirectional

Intern 20 F 4003 Official Control of the second sec

MKH 40 P48U3 Cardiold

MR.N.4U P48U3 CaPGIDIO Highly versatile, low distortion push-puil element, transformer-less RF condenser, high output level, transparent response, switchable proximity equalization (-4 dB at 50 Hz) and pre-attenuation of 10 dB to prevent overmodulation. In vocal appli-cations excellent results have been achieved with the use of a pop screen. Recommended for most situations, including dip-tal recording, overdubbing vocals, percussive sound, acoustic guitars, piano, brass and string instruments. Mid-Side (M-S) stere, and courventional X-Y stereo.

MKH 60 P48U3 Short Shotgun

Short interference tube RF condensor. Ilightwayfin metal alfoy, transformerless, low noise, symmetrical capsule design, smooth off-axis frequency resonas, switchable low cut filter (-5 d8 at 100 Hz), high frequency boost (-5 d8 at 10 KHz) and 10 d8 attenuation. Handles extremely high SPL (135 dB), ideal for broadcasting, film, wideo, sports recording, interviewing in crowded or noisy environments. Excellent for studio voiceovers.

MKH 70 P48U3 Shotgun

Extremely lightweight RF condenser, rugged, long shotgun, low distortion push-pull element, transformerfess, low moise, switchable presence (+5 dB at 10 KHz), low coll filter (+5 dB at 50 Hz), and 10 dB preatenuation. Handles 133 dB/SPL with excellent sensitivity and high output lavel. I deal for vitileofilm studios, lheater, sporting events, and nature recordings

TASCAM DA-88 Digital Multi-Track Recorder

The first thing you notice about the eight channel DA-88 is the size of the

The first thing you notice about the eight channel DA-88 is the size of the casetier – it's a small Hi-Brim wideo casetite. You'il also unotice the record ing time – up to 120 minutes. These are just two of the advantages of the DA-88 is innovative use of 8mm technology. The AFF system ensures that there will be no tracking errors or loss of synchronization. The DA-88 descrit even have (or need) a tracking adjustment. All eight tracks of audio are per-tectly synchronized. What's more, this system guarantees per-tectly tracks!) to 128 tracks!) Incoming audio is digitized by the on-board 16-bit D/A at either 44.1 or 48KHz (user selectable). The frequency response is flat As you would expect from a CD-quality recorder, the wow and flutter is Jumasurable.

flutter is unmeasurable



 One of the best features of the DA-88 is the ability to execute seamless Punch-ins and Punch-inits. This feature offers pro-grammable digital crossfades, as well as the ability to insert new material accurate, into tight spots. You can evin delay individ ual tracks, whether you want to g neare special effects or com-pensate for poor timing. All of this can be performed easily on a deck that is simple and intuitive to use

FOSTEX RD-8 Multi-Track Recorder

Foster has long been a leader in synchronization, and the FD-8 redeft that commitment, with its built in SMPTE / EBU reader/generator. Ith RD-8 can stripe, read and jam sync time code - even convert to MIDI time code. In a sync environment the RD-9 can be either Master or Stave. In a MIDI environment it will integrate semilessly mib the mos complex project studio, allowing you complete transport control from within your MMC (MIDI Machine Control) compatible sequencer.



Winnin your MMC (MID) Machine Control compatible sequencer: Full transport control is available via the unit is industry-standard RS-422 port, providing full control right from your video bay. The RD-8 records at either 44.1 or 48KHz and will perform Pull-Up and Pull-Down functions for fim/video transfirs. The Track Slip feature helps maintain perfect sound-to-picture sync and tae 8-channel Optical Digital Interface keeps you in the digital domain. All of this contributes to the super's sound quality of the RD-8. The audio itself is prive ased by 16-bit digital-ico-nain (QLAS) con-verters at either 44.1 or 48KHz (user selectable) sampling rates, with 64X oversampling. Plashat is accomptished with 18 bit ana-ting-to-digital (AOVs) and 64X oversampling, Thus delivering Co-quality audio. The 5-VHS transport in the RD-8 was selected because of its proven reliability, ruged ionstruction and ituger bandling capabilities Eight tracks on 5-VHS tape allow much wider tack widths than is possibil on other digital lage recording formats with its LCD and 10-digit display panel, the RD-8 is remarkably easy to control. You can readily access Table Incate joints, and cross-fade time is fully controlable in machine to machine editing. Table of Contenti- data can be recorded to incline, since the RD-8 is fully ADAT compliant, your machine can play tigges made on other complicities machines, and canible controlled by other manufacturers ADAT controllers. Your tapes will also be playable on any other AINAT deck.

Roland[®] DM-800

Digital Audio Workstation

The DM-800 is a compact stand-alone multi-track disk recorder that provides an amazing array of lea-tures at an unbelievably low price. Whether for music production, post production or broadcast, the DM-800 will make your work simpler, faster, more productive and more profitable. A full function workstation, the DM-800 performs all digital mixing operations from audio recording, to editing, to track-bouncing, to final mixidown. It fully supports SMPTE and MIDI time codes and also features a built-in Sample Rate Resolver to synchronously lock to any time code.

POWERFUL EDITING

Time Compression, Pitch Compression
 Completely Non-Destructive Cutting, Erasing, Copying
 Very Fast Looging for Music or Ambiance Editing
 Serub Preview and Preview to, from and thru
 Sic Levis of Waveform Zoom for Fast Editing
 Optional RS-4220 Interface (D10-800D) for 9-pin Control
 more Wite Sec.

from Video Edito

FLEXIBLE I/O STRUCTURE

 Full Digital Patch Bay
 Stereo Aux Send Bi
 Two Stereo Aux Returns
 Direct Channel Out
 Digital Stereo Input and Two Digital Stereo Outputs
 Four Balanced Analog Outputs with Gan Controls
 Four Balanced Analog Outputs with Option for 4 More Stereo Aux Send Buss
 Direct Channel Outs

FULL AUTOMATION Oynamic and Snapshot Automation of Level, Pan 2-Band EQ, netuding Frequency Select, Boost and Cut Microscope Editing of Automation Data
 Phase Level Editing of Level, Crosstade and Fade In/Out

TRIGGER FEATURES:

Trigger Mode to Play any Combination of 8 Tracks for Voce Fly Ins or Sound Effects Plazements Advanced Trigger Mode for Live Operation with Preset or Dial up Cue of Phrases to be Played One after Another

MIDI FEATURES:

- MIDI FCALORES. MIDI Machine Control Internal Tempo Maps Accurate Editing by Bars and Beats and Sub-Beats MIDI Clock and Song Position Pointer Output 8 MIDI Triggers for Instant Phrase Playback

Also...

🛋 Apple

audio accessories

MIDI Trigger of Record and Punch In/Out
 Tempo Maps from External Sequences, MIDI or Tap Input.



ACCURATE SYNCHRONIZATION

- Frame Accurate Structure of the Ender Locks to MTC
 Generates and Reads all Types of SMPTE, including 24 25, 29 73 0.0 Torophon-drop) "rames per Second Incoming SMPTE Reshaped to Durput Jack
- RECORDING OPTIONS

- Records to Standard SCSI Drives Up to 24 Hours Recording Time Possible Uses Magneto Optical or Syguest Brives for Fast Project December 2010
- Ontional Internet ? 5" Drives for Portative Operation

PROJECT CATALOGING

- Up to 150 Projects on Line at Once Easy Cataloging of Sound Effects and Projects Easy Transfer of Loundat from One Project to Another File Compatible with DM-80

HIGH QUALITY SOURD

Sampling Rates of 48, 44, 1, 32 kHz
 I8 Bit A/D and DX with 128 and 8 Tim is Oversampling
 24 Bit Internal Processing
 Superb Converter:

VIDEO OUT

- Composite, S-video, Digital RGB Output
 All Track Overslew with Infinite Level of Project Zoom
 Views of Phrase and Waveform Editing
- Very Accurate Leval Meters
 Track Status and Time Location



SEVEN DAY CUSTOMER SATISFACTION GUARANTEE CIRCLE 18 ON FREE INFO CARD

beyerdynamic) TG-X

Dynamic Mic Series

The entire TG-X family of microphones is based on neodymium technology coupled with rugged construction and designed for optimum performance in a stage environment

TG-X 5

- Clip-on gooseneck drum mic with high SPL capability. Acoustically coupled to capture shell sound as well as skin sound Internally shock mounted for isolation from mechanical noise. Rugged construction with convenient clothes peg mounting.

TG-X 30/35

- Ultra-slim flexible goosenack for easy positioning
 Broadast performance microphone with rugged contruction
 Lightweight, low profile adjustable neckband.
 Applications, Hands free vocals (drums/keyboards)
- Applications inality free vocals (drums/keyboards) and aerobics
 TGX 30 15 Omni neckworn with pre-amplifier
 TGX 30 16 Omni neckworn for wireless transmitter
- TGX 35 15 Cardioid neckworn with pre-amplifier
 TGX 30.16 Cardioid neckworn for wireless transmitter

TG-X 40

- Hypercardioid polar pattern with excellent isolation from unwanted sound. Flat, wide range frequency response for uncolored sound reproduction. Rugged construction. Applications, Vical piano, orchestral/recital provideor stream and sounding. recordings, strings and sampling

TG-X 50

 High quality hypercardioid vocal mic Full range frequency response Studio quality accuracy
 Excellent off-axis isolation

Hypercardioid kick-drum with high SPL capability
 Reduces shell ring in bass drum.
 Exceptional gain-before-leedback
 Superb transient response and extended low frequency range.
 Applications Kick drum, floor toms, congas and timpani.

TG-X 80

EV Bectro Voice

TASCAM

103 Mastering

Cassette Deck

tachines. With its three head design you can hear what is actually on the tape as it is recorded. Auto Monitor Function switches from playback to input automatically while in recordpause mode, allowing you to set record levels or match tape levels. Dolby NX PRO circuitry provides extended high frequency performance while keeping distortion and noise to a minimum. Tape type is automatically sensed and adjusted for by the Auto Tape Selection feature.

102 Cassette Deck

Designed primarily for mixdown, the 102 provides solid perfor-mance and sound quality with durability and reliability. Although in is a two head unit, the 102 closely matches the per-formance and features of Tascam's 103 Mastering Deck.

202MKII

Dual Recording Cassette Deck

Dual record cassette deck offers dubbing and copying capabili-ties at a reasonable price. The deck is capable of simultaneously making two identiliaal recordings from the same source, or mak-ing a single extended, recording first on one tape (both sides) then on the other (also both sides). Copying can be easily per-formed by using one side for playback and the other for record can also provide continuous background music, playing first both sides of one tape, then both of the other.

Cost effective three

cassette deck, appro priate for audio and video production facilities. With its

head steren mixd

 Shock absorbing rubber ring
 Applications Vocals, piano, strings, brass, percussion, woodwind, overheads sampling We are also full stocking dealers of (A) audio technica.

FOR PHOTO & VIDEO"

TO INQUIRE ABOUT YOUR ORDER: 800 221-5743 · 212 807-7479 OR FAX 24 HOURS: 212 366-3738

SONY TCD-D10 PRO II **Portable DAT Recorder**

Packed with features and functions, the eimpact TCD-D10PR0II is everything prefersionals need in a portable DAT recorder. Measuing only 0° 2 4 \times 7 \ll 11 provides high performance, reliability and Heixbility. Arong its many fea-tures are high speed 100x search. LCD molit-function display. 4-motor Direct Driver tansport mechanism and absolute time recording capability. With Astochule time deter by a source material for the CM-2002 series DAT edicing system Offening maximum performance and capabilities in a mini-mum package. Let CD-D10 PR0 II is the recorder of choice for any field application.



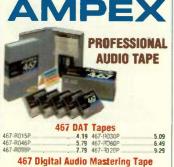
· Has balanced XLR input switchable mixrophone (-60dB) or the test of t

system is a sasting in also provides the convenience or unguar dubbing and editing withhut any degradation. Equipped with a compretensive self-diamostics function that constantly monitors the rotation of the head drum, capstan and reels. The tape transport mode and bad/unload time are continuously clacked as well. Upon detaxtion of truble, the tape is brought to a forced stop and un eaded automatically to protect the tape and the eccorder.

Tape is brought to a forced stop and un baded automatically to protect the tape and the ecorder. • Up to 99 start IUS can be recorded in the subcode area. When the record buttin is preseived, the start ID is recorded automati-cally for 9 seconds. During recording, it can also be added manually to any position at the tape. Second for these start IDs is performed in two modes at 100 timezenormal speed. Offers a maximum spooling time of 141x normal speed A two hour tape can be rewound or fast forwarded in under a minuta 20-segment oligital peak ere interies in clude overtcad indica-tors. Closely tracks input signal for accurate level indications During playback, the data and time of fracting is displayed. Has a 5-segment battery indicator. The last segment blinks on and off, nothing oy ou to share border with a tast attack time of 300ms. The microphone-intenuator prevents distortion by suppressing the signal level 20 dB • Immediate playback is playsible through a butti-in speaker. • A wred remuce controller is supplied the control the recorder. The top end in the controller i designed to accept a microphone noted. Two play stop, and pause functions of the recorder. The top end of the controller i designed to accept a microphone holder. Two

the controller is designe a to accept a microphone holder

 The supplied NP-22H re hargeable bat, ry pack provides 1.5 hours of continuous operation. The opional NPA-D10 battery adapter enables 1 hour of continuous operation on AA-size batteries With the use of the supplied ACP-88 AC power adapter, it can also be o erated on 10C-240 VAC, 50-60 Hz



467-173J1J 4 x 4600 101 NAB Reel 64.49 467-27GJ11 x 4600 101 Heavy Duty Precision Reel 79.95 467-57GJ11 1 x 4600 101 Heavy Duty Precision Reel 149.95 489 % Digital Mastering Tape Audio

DM810C0 * x 810		_
-----------------	--	---

DM810A0 X x 810' Gray Library Box	8.99
DM810C0 x 810 Clear Sleeve	8.09
400 04 47 14 4 4 1	
456 Studio Mastering Tape	
456-151111 1 x 1200 Plastic Reel	7.19
456-17311J x 2500 10 "NAB Reel	18.99
456-272111 x 2500 Plastic Hub	26.99
456-273111 # x 2500 10 NAB Reel	33.99
456-572111 1 x 2500 Plastic Hub	52.99
456-573111 * x 2500 fD NA8 Reel	58.49
456-97G111 2 x 2500 to PREC	125.99
400 Connel Marshau Cold	
499 Grand Master Gold	
Studio Mastering Tape	
499-174111 x 2500 to NAB Reel	20.49
499-274111 x 2500 to NAB Real	37.49
499-97M111 2"x 2500 JC "PREC	134.99

499-174111 x 2500 to NAB Reel 499-274111 x 2500 to NAB Reel 499-97M111 2"x 2500 JC." PREC

119 WEST 17TH STREET, NEW YORK, N.Y. 10011

Store & Mail Order Hours: Sun 10-4:45 • Mon & Tues 9-6 • Wed & Thurs 9-7 30 • Fri 9-2 • Sat Closed RUSH OR OVERNIGHT SERVICE AVAILABLE (extra charge)



SONY

PCM-2300/PCM-2700A DAT Recorders

The superb audio quality of digital recording and the convenience of DAT casetters is not restricted to broadcasting and post-production applica-tions. In fact, it is just as relevant to business and educational applica-tions. To meet the requirements of all these sectors Sony offees the PCM-2300 and PCM-2700A DAT recorders. While both are perfect for simple recording and playback at recording studios, the PCM-2700A is also well suited for simple program transmission at broadcasting sta-tions. Both feature exceptional sound quality, three sampling timplen-ces, absolute time recording, long record/playback time, alternative subcode recording and digital fader.

PROFESSIONAL PERFORMANCE PROFESSIONAL PERFORMANCE The serve system is complexely controlled by carefully pro-grammed software. This features sophisticated, smooth trans-port control, which optimizes tape handling and increases the search speed to approximately 150 times normal Absolute time can be recorded in the subcode area of the DA^T tape DAT tape with absolute time is ideal for editing with the Somy PCM-7000 Series Editing System, as it translates A-time into SMPTE time acide.

Sony PCM-7000 Series Editing System, as it translates A-t into SMPTE time code. DIGITAL FADER A 64-step digital fader is incorporated in both units. Both far and fade-out functions are available on the digital and analog inputs/outputs. The fade duration can be set over from 0.2 seconds, and can be controlled either locally or by the remo both units Both fade-in g 15 PCM-2300 AND PCM-2700A

Seconds, and controlled entry locally to by the remote. PCM-3200 has unbalanced digital input/output, while the PCM-2000 has unbalanced digital input/output, while the PCM-2700 has dual digital input/soutputs (balanced and unbalanced), allowing most digital audio equipment to be direct-ly connected. Both also have balanced analog fine lavel input/soutputs which can be adjusted over a range of -12 dBs to e8 dBs, for connection to many other types of audio equipment. - The PCM-2300 has a reliable tape transport system driven by three servo-controlled motors, two direct-drive types for head drum and capstan, and a single DC motor for the reels. - The PCM-2000 employs a four motor direct-drive tape trans-port with head drum, capstan and reels, all individually driven by servo-cont speed motors has a 20-segment peak level meter display with 0.5 dB step peak margin. The PCM-2700A has a 29-segment peak level meter with a 0.1dB step peak margin indication.



SUSCODES

They record various subcodes suparately from the audio data to provide tape search functions. The subcodes include the Start ID, Skip ID, End ID. Program Numburs, Absolute Time, and Date functions. Program numbers can be recorded sequentially as the Start IDs and are written (and can be renumbered) in the correct sequence whenever additional Start IDs are inserted.

OTHER FEATURES

They operate in the Loop Play LPJ mode, yielding twice the normal record/playback time. This also makes them compatitively the LP mode of consumer products. not P mode, a maximum of four hours of recording and playback its possible. Built-in crystal clock generates a date function, which automatically indexes ach recording with time and date info (mmute/hour/day of the week/monthyear). When tape is enabled to a standard tig-inch rack. Both units can be installed in a standard tig-inch rack. Both included a wireless remote control fair remote aperation
 Ped. The date of the date.

PCM-2700A ONLY FEATBRES

PCM-2700A.0HLY FEATURES Read After Write function allows you to novinor the recorded signal immediately after it has been stored un the tape. This saves you valuable time, as reviorded signal can be inheck moni-tored in real time. High-speed location, in either direction, to points or the tape with a resolution of 1 second by entering the assolute time address via keys on the front panel. Oroverinent remote operation is performed by the optional RM-D7100 Remote Coarticler. Connect via the 37-pin parallel remote connector, it allows basic tape transport control. It is particularly useful for On-Air applications.

 Add or pin ferriture interface to prive a rate of an additional to the pin ferriture interface of the pin ferriture is an enhanced rehearsal function which is available for accurate ID recording. While monitoring the ID recording, these can be relo cated forwards or backwards in 0.3 second steps.

Panasonic SV-3700/SV-4100

Professional DAT Player/Recorders

Panasonic's SV-3700 and SV-4100 are designed for professional applica-tions. They have highly accurate and reliable transport systems with search speeds up to 400 times normal pily speed. They also feature davanced, high-quality analog-to-digital (A-D) ard digital-to-analog (D-A) converters and input/output circuitry designed to interface with the widest variety of devices.

(D-A) converters and input/output circuitry designed to interface to SV-3700 Features:
When recording via the analog inputs, a front panel switch permits selection of the sampling rate (44 1kHz or 4kHz). This avoids the need for a conversion of the sampling frequencies of 20kHz, 44 1kHz or 4kHz). This inputs, it alumatically clocks to incoming frequencies of 20kHz, 44 1kHz or 4kHz). This descends fade-out provides automatic level changes at the start and out of a correcting up to 250x normal speed is possible orne the tape has been scanned in Pay, Fast-Forward or Reverse mode. This ensure sccess to any point on a two-hour DAT in approximately 27 secons.

FRAME ACCURATE INDEXING AND EDITING

FRAME ACCURATE INDEXING AND EDITING Using the trim and rehearsal functions, you can accurately determine points to write, start and skip IDs. These IDs can be written, rewritten or erased at any point in the recording and automatically renumbered. • With two SV-4100s rounceted via the 8-pin parallel remote ter-minal, synchronized frame-accurate editing can be performed Continuity of edit points can be checked by rehearsal playlack. By entering and editing end position in one of the Locate but-tons, you can determine a punch-out point as well.

SV-4100 Has All the Features of the SV-3700 PLUS:

Offers enhanced performance required for protestional production, broadcast and live sound systems. Feature: such as instant start, external sync capability, enhanced system diagnostics, additional digital interfaces and exceptional 20- in audio make the SV-4100 the DAT quality standard.
 Output
 Standard

 OUICK START WITH TRIM AND REHEARSAL

 • With 8M8 of memory holding live seconds of audio data, the Quick Start function provides sound almost instantly after a Data Start function provides sound almost instantly after a Data Start function provides sound almost instantly after a Data Start function provides sound almost instantly after a Data Start function provides sound almost instantly after a Data Start function provides sound almost instantly after a Data Start function provides and pack the Data Start function provides and pack the Data Start function almost function almost and the Data Start function almost function almost and the Data Start function almost and pack and p

FLEXIBLE SEARCH raddess jour A-Time You carl specify Easily and accurately address our A-Time. You can specify hour, minute, second and frame In most modes, the currently displayed A-Time can be assigned to one of the Locate buttons. Then from Stop, Pause or Play you can rapidly c.e b any of these four addresses by pressin its Locate key. In addition, Locate Last Likes you to the most recent Quick Start A-Time position • Search is also possible by Start ID or program number

5-MODE EXTERNAL SYNC 5-MODE EXTERNAL SYNC Has 5 external sync modes. External sync is essential for applica-tions such as video postproduction and size submix recording. It assures uniformity of timing between difficient equipment so the audio data consistently matches up with the tarret media - Select from 3 video external sync mode: (25, 29 ST and 30 frames per second) or use the word sync art Digital Data modes (which lock to the input sampling frequency). Explanetron Solution

ENHANCED SOUND The SV-4100 satisfies the highest professional expectations both in terms of sound and functionality. If decuises new 20-bit (equiv-alent resolution) digital-to-analog converses

MULTIPLE DIGITAL INTERFACES

MULTIPLE UNITAC INFERENCES Has XLR-balanced d gial inputs and output, plus unbalanced dig-rial coarial and optical inputs and outputs. Analogimputs/out-puts are XLR-balanced and cutput level is switchable between +4dB and -10dB, providing compatibility with other equipment. **3-WAY REMOTE CONTROL**

3:WAY REMOTE CONTROL OPI input allows simple triggering of Oulick-Start Play 8-pin parallel remote term hal connects to an inther DAT deck, com-puter or wired remote Inclumes wireless remote control



Portable DAT Recorder

With rotary two nead design and two direct dive motors the DA-P1 offers one of the

- the DA-P1 offers one of the best transport in its class 1000 XLR-balanced mic/line inputs (with phaetom power) accept a broad range of signal levels from -60d8 to +1dB

- Analog line inputs and outputs (unbalanced) pluis S/PDIF (RCA) digital inputs and outputs enables direct digital transfers. Uses next generation A/D and DIA converters to deliver amaz-ing sound quality.
- orts multiple sample rates (48, 44,1 and 32 kHz) and
- Supports multiple sample rates (48, 4.1 and 32 kHz) and SCMS-free recording.
 Included in its disrgin is a MIC limiter and 20dB pad to achieve the best possible sound without outside disturbances.
 Io monitor your sound there is a TRS Jack and level control for use with any headphones.
 Built tough, the DA-P1 is housed in a solid, well-constructed hard case. The BA-P1 includes a shoulder belt, AC adapter and one butter.
- and one battery

SONY TCD-D7 **DAT Walkman Player/Recorder**

 High-quality Standard Play (SP) mode provides up to two hours recording of 16-bit digital audio ona DT-120 DAT cassette. The SP mode is ideal for recording live music



Long Play (LP) mode allows up to 4 hours of record/play

Long ray (Lr) mode allows up to 4 hours of record/playback of 12-bit audio on a single DAT cassette. The LP mode is ideal for meetings, conferences or other voice recordings. Equipped with digital coaxial and cptical input connector. Equipped with rlightal coarial and cplical input connector. Maintains highest signal purity for recording and playback of digital sources with all information retained in digital format.
 Also has analog Mic and Line inputs for recording from analog sources with all information retained in digital format.
 High-speed Automatic Music Sensor (AMS) search function finds and playsitracks, skips forward or back up to 99 tracks, all at 100x normal speed.
 Has a Digital Velime Limiter System (DVLS) that increases listening comfert and sound quality by automatically adjust-ing for sudden evel changes of the recording It also helps prevent sound teaks through headphones.
 Two-speed cute-review lefs you hear sound while player is in fast-wind modes, up to 3X or 25x normal speed.
 Omgat and tortable, it has an arti-shock mechanism that permits accurate recording and playback even while in motion.

- permits accurate recording and playback even while in motion. LCD display with backlit windows clearly shows recording

• LCD display with backfill windows clearly shows recording level, track number, operating status and 4-segment battery indicator, even in low ambient light conditions. Optional RH-D3X System Adapter Kit for complete digital interface. The kit is equipped withithe input/output connectors for both the optical cable and the socaria cable. Therefore you can use it as a relay between the TCD-D7 and other digital enumbersh. Then includes a wireless remote control. equipment. Also includes a wireless remote control



An integrated puckage, the XD-P1 Pro combines a DA¹ recorder and a microphone with digital output in an unbiliev able light package · Records and plays at al three standard

Abk/2 for mastering if CD production or Abk/2 for mastering if CD production or Abk/2 for highest fidelity • Operates without the restrictions of

- 48kt z for highest fidelity Operates withmut the restrictions of SCMS (Serial Copy Management System), perriting on generation of digital to digital copy, using 44.1 [Hz] sampling freqency. You can digitally dub or copy avecording mode on the XD-P1 Pro as many times as necessary. The supplied letachable digital origont microphone provides two pick-up gat-terns, "Telescopic" and "Stereo". A collapsible microphone stand is also supplied, altwork up gat-terns, "Telescopic" and "Stereo". A collapsible microphone stand is also supplied, altwork up gat-terns, "Telescopic" and "Stereo". A collapsible microphone stand is also supplied, altwork up gat-terns, "Telescopic" and "Stereo". A collapsible microphone smale the XD-P1 Pro compact, lightweight and durable. The main module weighs under a pound (12.5 oz). With microphone and battery if weighs only 22 oz. Alt basic contols are on one side of the unit and readily access-ible. You can operate the recorder using one hand. Avanced power-saving design and low power-consumption cir-custs give the XD-P1 Pro long recording capability on a single battery charge. Onlineal rechargebibe batteries further extend "econding the".
- recording time Supplied AC adapter/charger works anywhere in the world regardless of voltage or frequency



• Built-in shuttle wheel has two variable speed ranges: 3 to 15x normal speed in Pay mode and 1/2 to 3x mormal speed in Pause mode – an ideal way to find tape locations. Comprehensive display includes program numbers, absolute time, program time, remaining time and Table of Contents which displays total recorded time and traile PNC count for commercial prerecorded DAT tapes – (digital autoenced and unbanced (phono) digital nputs and outputs. They provide direct interfacing with compared tise plays – (digital autoin workstations and other pomorents in a recording studio workstations and other components in a necording studio or production facility. Ais- has XLR-balanced plays. A support of the system with the systemation with compared table between +4db and -10db. The input level is +4db.



Samick乡

MM Series

MM842/1242/1642/2242

8/12/16/24 Channel Modular Mixers

..............

· Assignable ber graph level indicator for left and right or

dorrough

With today's audio systems stretching the limits of program dynamics if is biscene orthinal for engineers to obiain maximum loudness with the minimum of distortion comitionents. to fully utilize the dynamic range available. It is of equal importance that they have a method el inonitoring and establishing the maximum safe level at which a system can operate. That's why every, Donough Aud o Level Meter simultaneously shows 3 dimensions filterovan matterial content. That's

The model 40 A has scal allowing 140B of headroom in 10B steps A stand-a ose unit, it measures 8 " x 2 " x 6" and has an internal power supply. Model 4D AP has a geat-hold option

Loudness Meter Model 49-8

The Model 40:B provides metering of relative lauliness to peak modulation. The 44-6 is a socief differentiation of the 40-A and is calibrated in percent "9) modulation, with the lower scale in dB from +3 dB to -3 dB. The 40-Bit also has a yeak hold option.

Symetrix

Signal Processing Products

The The Area Control of the Source of Markowski and Source of Sourc

488 Dyna-Squeeze 8-Channel Compressor/Interface

We are a full stocking dealer

for the entire Symetrix line

601 Digital Voice Processor

An epts mic or one level anclog signals, con-versithem to digital (18 bit) and then signal processing Pr

shows 3 dimensions of program material content. Peak Average Power and Compression are displayed on a color-coded 40-segment LED so le. Meters are easily we well while providing precision indications of program energy content. Loudness Meter Model 40-A

Ileni all

animitation statements

· Electronically balanced XLR input Channel gam trim LED heatroom indicator 3-band acime EO rweep mid-bund Individual channel inserts for patching Direct output: on each channel

• XLR and 1/4" balanced outputs

"THE PROFESSIONAL'S SOURCE

FOR ORDERS CALL: 800-947-5518 212-444-6688

OR FAX (24 HOURS): 800-947-9003

FLEXIBLE FO SECTION

ADVANCED SIGNAL ROUTING OPTIONS

ERGONOMIC DESIGN The M-2600 has a big studio feel All buttons are tightly spring loaded, lock into place with confidence and are large enough to accomdate even the biggest fingers. The laders and knobs have a tight, smooth "expensive" feel and are easy to see, easy to reach and a pleasure to manipulate. Center deletist assure zero positions for EQ and PAN knobs. Smooth long throw 100mm

rocal as neadphone or cue mixer, level matching pro audio "tool kit", drum or effects sends submixer. 8-track monitor mixer phone or cue mixer, level match

INTELLIGENT EQ POINTS • Low frequency EQ is at 80 Hz where it has more depth and less hollow midbass "bonk" Midrange is centered at 2.5 KHz, pro-viding for more control of vocal and instrumental harmonics specially-shaped HF curve that shelves at 12 KHz creates more sizzle and less aural fatigue BEAL MC BECAUGE

sizzle and less aural fatigue REAL MIC PREAMPS The CR-1604 has genuine studio-grade phantom powered, bal-anced input mic preamps on channels 1 through 6 Ail CR-1604 (and XLR10) discrete input mic preamp stages incorporate four conjugate-pair, large-emittler geometry transistors just like the big mixers use So, when recording nature sound effects to heavy metal or mixing flutes or kick drums, you get the qui-dest, clapned receives possible

BUILT TO LAST

BUILT TO LAST The CR-1604 is designed for non-stop. 24-hours-a-day profes-sional duty – even for tours that log 100,000 miles in three months. It has sealed rotary potentiometers that are resistant to airborne contamination like dust, smoke, liquids, and even the oxidizing effects of air isself.

Optional Accessories

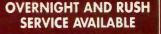
Uptional ACCESSIONES OTTO-1604 Add sophisticated computer controlled automation to your CR-1604. When connected to the MIDI port of your computer (PC, Mac, Amiga or Atari), each bane of the 16 input channels can be programmed to change gain or to mute, just as you would pro-gram a sequencer. Mister levels can be programmed as well, along with all buss channels

XLR10 While the standard CR-1604 comes with 6 high performance mic inputs, there are times when you need more. Enter the XLR10 This simple-to-install accessory adds 10 more (for a total of 16) mic inputs, with the same quality, performance and features as those in the CR-1604.

etest, cleanest results possible

SEVEN DAY CUSTOMER SATISFACTION GUARANTEE CIRCLE 18 ON FREE INFO CARD

212-444-5001



E-Mail at 76623.570@compuserve.com

SENNHEISER

Incorporating state-of-the-art technology and decades of experience, Sennheiser headphones offer outstanding design, superb audio quality and luxurious comfort. From the ear padding to the O.F.C (Oxygen Free Copper) cable, all materials are carefully selected and precision engineered to ensure incomparable quality

HD414 Classic

A re-issue of the world's first open-Aire dynamic headphone, the HD414 Classic dynamic headphone, the H0414 Classic offers maximum transparency, fidelity and comfort. A limited edition model , the HD414 Classic also features Professor Sennhieser's signature on the headband - Radial based diaphragoms for more accuarte reproduction - Field replaceable parts for forog-term engipment - Neodymium-ferrous magnets for broad trequency responce (16-21,000Hz) - 10ft. Keviar-reinforced oxygen-free copper signal cable with 1/8"-1/4" stereo phone plug.

HD25 SP **Studio Monitor Headphone**

Offering dramatic isolation from external sounds, the HD25SP is designed for pro-

sounds, the HU2SSP is designed for p fessionals who rely on studio monitor headphones for work and pleasure. High-efficiency drivers for portable and field use. • Modular parts for long term value and durability. • Neodymium-ferrous magnets for broad fragmency resonase.

- broad trequency respon
- Frequency Response: 16-22,000Hz Sensitivity 105dB • Impedance: 700



\$11995

Designed to meet the stringent demands of studio profes-sional and audiophile alike, the HD265 breaks new ground in the sealed-chamber headphone format. Sealed-chamber headphone format. Triple-wound aluminum voice coils for quick transients. Polycarbonate dome-damping minimizes distortion Dysgen-free copper signal cable with 1/8" to 1/4" stereo phone hun.

- \$199⁹⁵

HD535 **Dynamic Hi-Fi**

- systems offers excellent transient and dynamic response

- dynamic response Supported by the open structure, tonal quality develops high dimensional sound qualities. Circumaural, oval earpads for good wearing comfort. Can be connected to all digital and analogue deficiency of the open structure of th \$13995 Hi-Fi components

Digitally Compatible

- · Oxygen-free copper signal cable assures optimal \$169⁹⁵

Digitally Compatible

- Offering natural sound reproduction without foral distortion the choice for use with the most sophificated equipment. Sik dome-damping for brilliant highs and ultra-smooth midrange reproduction Copper-coated aluminum voice coils for high linearity. Bass-tube tuning for extended, powerful low frequencies. Velvet ear cushions provide lavish comfort. Daygen free copper signal cable with 1/8" to 1/4" stereo phone plug.
- - \$23995

- stereo phone plug Frequency Response, 10-25,000Hz

Stereo Headphone

The HD535's earcups surround your ears rather The most sectors surround your ears fait than resting on them, for a more natural listening experience
Open-back construction and classic design
Light aluminum coils in the transducer

HD545 **Circumaural Headphone**

The HD545 is the headphone to which all in its price

- Ine HUS45 is the headphone to writer all in its price category must be compared Polycarbonate dome-damping materials for clearer highs. Triple-wound aluminum voice coils mean powerful bass, high output and greater durability. Velvety soft, circumaural earpads and adjustable, padded headband for extendedcomfort and perfect fit. Owners, free concert cload is table servers on fitmal
- signal transfer Includes 1/8" to 1/4" stereo phone plug.
 - HD565

- **Circumaural Headphone** Offering natural sound reproduction without tonal distortion

TASCAM M-2600 Series 16/24/32 Channel Eight Channel Mixers LOW NOISE CIRCUITRY

 Combining completely redesigned, low noise circuitry with Absolute Sound Transparency™ the M-2600 delivers high-quality extremely Sound transparency - the M-cool derivers in gin-quarity entre clean sound. No matter how many times your signal goes through the M-2600, it wont be colored or altered. The signal remains as close to the original as possible. The only colorin you hear its what you add with creative EQ and your outboard

you hear to what you add whith clearing by and you builden signal processing gear. Double reinforced grounding system eliminates any hum World-class newer supply provides higher voltage output for befter headroom and higher S/N ratio.

PREMIUM QUALITY MIC PRE-AMPS PREMIUM QUALITY MIC PRE-AMPS 1 The M-2600 s mic pre ampn yeld an extremely low noise flictry, enormous headroom and an extremely list frequency resporse. This lowers distortion and widens dynamic range. It also increases gain control to an amazing 51dB. Pius you get phun-tom power on each channel. The M-2600 accepts balanced or unbalanced 1/4" inputs, and low-impedance XLR jacks. Better still, the TRIM controls oper-ate over a 51dB input range. For the hottest incoming signal-all it lakes is a press of the 20 dB PAD button atop each chan-nel stim to bring any signal down to manageable levels. Plug anything into in keyboards, guidars, basses, active or passing microphones, sampliers and more. No matter what you put into it, you can be confident that signal can be placed at optimum levels writter La list of fuss. THE EXT ALLY SECTION IN THE BUSINESS

THE BEST AUX SECTION IN THE BUSINESS

The most versaille AUX action in its class, rivaling expensive high-end consoles 8 sands total 2 in stereo. Send signal in stereo or mono, pre- or post-fader. Available all at once. Return signal through any of 6 stereo paths

PREMIUM QUALITY MIC PRE-AMPS

You'll find both shelving and split-EQ sections on some mid-level consoles. But that's where the similarities with the M-2600 end the M-2600's bi-directional split EQ means you can use either or both EQ sections in the Monitor or Channel path. or deleat the effect allogether with one bypass button. Most other compara-bly-procedmizers will lock the shelving mix into the Monitor path only, limiting your EQ application. Direct channel input switching. Assign to one of eight busses, or direct to tape or disk, or to the master stereo bus. Because the group and direct-out jacks are one and the same, you can sets either without repatching. You won't find this kind of speed or flexibility in one-site. (His all' board

faders glide nicely yet still confidently allow you to position the securely without fear of accidentally slipping to another position

MICRO SERIES 1202

12-Channel Ultra-Compact Mic/Line Mixer

Usually the performance and durability of smaller mixers drops in direct proportion their price. Fortunately, Mackie's fanafical approach to pro sound engineering has resulted in the Micro Series 1202, an affordable small mixer with studio specifica-tions and rugged construction. The 1202 is a no-compromise, professional quality ultra-compact mixer designed for professional duty in broadcast studios, permanent PA applications and editing suites where nothing must ever go-wrong

An applications and editing suites where nothing must ever gowrong.
 Sing Console FEATURES
 Working SN ratio of 9008, distortion below 0.025% across time
 and regramme single solution of solution below 0.025% across time
 and regramme solution of solution below 0.025% across time
 and regramme solution of solution below 0.025% across time
 and regramme solution of solution below 0.025% across time
 and regramme solution of solution below 0.025% across time
 and regramme solution of solution below 0.025% across time
 and regramme solution of solution below 0.025% across time
 and regramme solution of solution below 0.025% across time
 and regramme solution of solution below 0.025% across time
 and regramme solution of solution below 0.025% across time
 and regramme to the solution of solution of the solution

- Main outputs operate either balanced/unbalanced as required - Switchable three-way 12-LED peak meter displays

CR-1604 16-Channel Mic-LineMixer

The hands-down choice for major fouring groups and studio session players, as w as for broadcast, sound contracting and recording studio users, the Mackie CR-1604 offers leatures, spees, and day-in-day-out rehability that roral far targer bards! I features 24 usable line inputs with special heartoom/ ultra-low noise Unityplus circuity. seven AUX sends. 3-band etualization, constant power pan controls. Or segment LED output metering discrete front end phantom-powered mic inputs and much more.

LOWEST NOISE, HIGHEST HEADROOM

LOWEST NOISE, HIGHEST HEADROOM With the CR-1664. having the lowest noise and highest head-room 190 dB working SXM and 108 dB dynamic range) at the same time are net mutually exclusive. It is free of commonly encountered headroam restrictions, and is able to handle the occasional geoged input with lase. In lact, many drummers considient if the only mixer capable of handling the attack and transients of acaustic and referronic drums. Only with constant power pan pots will a source panned hard left or hard (ngh have the same toudness as when it is sitting dead center. While most small mixers pass simple balance cou-trols for pan poist, the CR-1604 scartelialy optimized constant power pan circuitry make it a professional tool with the kind of performance necessary for CD mastering, video posing and other critical audio production IN-PLACE STERED SOLD

other critical audio production IN-PLACE STERED SOLO Stereo "in place" solo allows not only the monitoring of level and ED, but also stereo perspective. Usually found in very expensive mixers, stereo solo allows you to critically scrutinize and carefully but d a mix using all the channels with their respective sends and AUX re

UNITYPLUS GAIN STRUCTURE

UNITYPLUS CAIN STRUCTURE Proper gain settings are facilitated by proper gain labeling, along with center-click detents on the faders, clearly under-standable input time controls and output meters that read chan-nel levels in solo mode. With property set levels you achieve very high beadroom and low noise at the same time. EFFECTS SEND WITH GAIN EFFECTS SEND WITH GAIN Unusual circuit design that provides two different "zones" that reflect real world use send from each channel can vary in lever from off to unity gain, which is the normal range of effects sends in other mixers. Since you also get another whole zone from the center detent to +15 dB of gain. the channel fader can be pulled down and the effects send can be boosted above unity when more effect is needed

FOR PHOTO & VIDEO"

TO INQUIRE ABOUT YOUR ORDER: 800 221-5743 · 212 807-7479 OR FAX 24 HOURS: 212 366-3738





- High finguency switch mode power supply fully charges 120,000 times per record (1000 times faster than conven-tional power supplies) inquiring far less capacitance for filter ing and storage
 High speed recharging also reduces power supply "sugging" that all filts of ther dus gris
 Incredibly efficient, 5 PA-1000 or PA-1400's (4 PA-1800's) Conventioned and the grission of the second second second second second second second second of the second second of the second second second second of the second s
- Increaning encient, 5 PAP 1000 of PAP 1400 s (4 PAP 1400 s) can be run on one standard 20 amp circuit. There is no need for staggered turm-on configurations or other preventive mea surse when using nultiple amp set-ups, as current drawn during turn-on is only 6 amp set only.
 They produce smooth and uncolored sound, while offering very full detailed to wan et response and tons of thersepower.
 They each carry a 5 year warranty on parts and labor
- PA-1000 weights 9 lbs us 15" deen and occupies the stundard

PA-1000 weighs 9 (Is, is 15) deep and occupies are standard rack space. Delivers 1000 waits into 42 when bridged to mone PA-1400 weighs 16 (bs, is 15) deep and tables 2 stundard rack spaces. Delivers 1400 waits into 442 when bridged to mone. PA-1800 weighs 17 (bs, is 17) deep and tables two rack spaces Delivers 1800 watts into 442 when bridged to mone.



Performance Series Amplifiers



Performance Series 1 300 Watt Power Amplifier

- Measuring only 3.5 inches high and weighing 24 pounds, the Series 1 delivers more than 150 watts per channel.
 Its welded steel charses is unbelevably strong while a puston heat simil extension provides exceptional thermal capanity
 An internal fan provides multi-background noise levels for cri-ical monitoring applications and when pushed hard the cool-hie sector inverse inollineum cool exceptions. ing system insures continuous cool operation even in the
- most demanding situations Most demanding situations • Active b lanced inputs with both XLR and 1/4" phone jacks • Supplied with qualit 5-way binding posts for highly reliable
- Supplied with quality 5-way obtaining posts for highly reliable speaker connection.
 Front panel handles are riversible for either rack mount insta
- lation or easy handling LEDs are provided for signal presence and clip indication; the
- detented gain controls have large knobs for easy front p adjustr

Performance Series 2 600-Watt Power Amplifier

Same as above except the Series 2 weighs 32 prounds and delivers more than 300 watts per channel

Performance Series 4 1200-Watt Power Amplifier

Same as above except the Series 4 weighs 53 priunds ind delivers more than ED0 watts per channel · Has a switch selectable clipping eliminator that greven's dam age to the speakers.

119 WEST 17TH STREET, NEW YORK, N.Y. 10011

Store & Mail Order Hours: Sun 10-4:45 • Mon & Tues 9-6 • Wed & Thurs 9-7:30 • Fri 9-2 • Sat Closed RUSH OR OVERNIGHT SERVICE AVAILABLE (extra charge)



TANNOY System 6 NFM I

A 6.5 inch Dual Concentric with Tulip HF wave guide forms the heart of the System 6 NFM II provid-ing a reference single point source monitor in a more compact evolosure than ever before. Every aspect of design fully complements the drive unit's capability. The right cabine with hearbluly con-toured battle and trim minimizes diffraction and the high quality minimalist DMT crossover and gold-plated BH-Wire terminal panel optimize the signal path. Pin-point arteries occuracy with wide frequency response, good power handling and sensitivity make this an ideal nearfield monitor.

PBM Series II Reference Monitors

The PBM II Series is the industry standard for reference monitors. They feature advarced technologies such as variable thick-ness, injection molded cones with mitrite rubber surrounds and the highest quality components including polypropylene capac-itors and carefully selected indicators. With a Tannoy monitor system you are assured of absolute fidelity to the source, true dynamic capability and most important, real world accuracy

PBM 5 II

· Custom 5" miect Custom 5" injection-molded Lass driver with a nitrite rubber surround for extended linearity and accurate low frequency reproduction. They are better damped for reduced distortion and exhibit more naturally open and

feduced insuring and cancer necessary of a polymide soft detailed midrange. • Wooler blerds seamlessly with the x^m polymide soft dome ferro-fluid cooled tweeter providing extended bandwidth for extremely precise sonically-balanced

Daliciwidin the extention process someany outprocess monitoring. • Designed for nearfield use, the PBM 5 II cabinets are produced from high density medite for minimal reso-nance and features an anti-diffraction radiused front balfle design

PBM 6.5 II

Transportable and extremely powerful, the PBM 6.5 II
 is the ideal monitor for almost any project production

avironment 6.5" lowfrequency driver and 3/4" tweeter are fed by a

• 5.5. Towfrequency driver and 3/4" tweeter are ted by completely redesigned hardwired hand selected crossover providing uncompromised detail, precise spectral resolution and flat response.
• Fully radiused and ported cabinet design reduces resonance and dirfraction while providing deep linear extended bass.

PBM 8 II High tech 1" soft dom: hweeter with unmatched pattern control and enormous dynamic sapability. 8" driver is capable of powerful bass extension under extrivme SPL demands. • High tech 1" soft dom

- demands. +lard wired crossover teatures twice bi-wire capability and utilizes the finest high power polypropylene capacitors and components available. Full cross-braced matrix medite structure virtually elimi-
- nates cabinet resonance as a factor
- Ensures caunter resonance as a factor.
 Ensures precise low fri quency tuning by incorporating a farge diameter port teacuring laminar air flow at higher port velocities.

TURTLE BEACH SYSTEMS

Turtle Beach audio cards bring true studio r Turtle Beach auto cards bring true studio performance to the PL planorm. Isaeb on the termology used in Their professional products, the performance of both hielt hardware and software has gameett acculades from users and press alike. Unlike other PC-bailed audio cards, Turtle Beach products are not designed for sound card game compatibility. Instead, the design philosophy is to give the best possible performance and quality, in the Windows operating environment, while bypassing traditional PC limitations.

MULTISOUND MONTEREY Multimedia Sound Card

The next generation of Turtle Beach's award winning Multisound board, Monterey offers high quality performance and features at a very reasonable price. Multisound's Hurricane technology provides very high speed audio data traisfer between the card and the hard drive. This approach allows for data throughput up to eight times faster than the more common DMA designs, while putting minimum burden on the CPU (critical Hyou're recording audio and video simultaneousty). The Motorolo DSP-5600 (Digital Signal Processor oper-ates at 20MIPs and performs all digitization functions with very low system methane all digitization functions with very low system methane.

Alls at 20 mr 3 and performs an orginization infinitions with very low system overhead. High S/M (-894B, A weighted), low distortion (<0.01% ThD, <0.01% IM - both A weighted), flat frequency response (DC-19KHz0.5dB).

On-board real-time effects processor for a variety of effects, including reverb and echo

playback (4MB of real instruments audio). Up to 4MB of standard SIMM-type memory can be added for simpling of ew sounds

True 16-bit recording, using 64x oversampling and sigma-delta conversion technology True 16-bit playback, with 64x oversampling and sigma delta conversion utilizing 18 bit DACs and an 8x interpolating filter

\$34900

WAVE 2.0 Professional Sound Editing for Windows 3.1 Compatible Sound Card

Wave is simply the best audio recording/editing/effects program available under Windows 3.1. With its logical layout, plethora of functions and wealth of effects, Wave will become an indispensable fool in your studio. Wave has the features (and interface) of a lage recorder. If supports stereo or mono recording/playback at 11.025, 22.05 or 44.1KHz sampling rates. Wave will work with any Windows 3.1 compatible audio board.

Supports cut & paste editing (just like editing in a word processor) and has a full Undo function to eliminate fatal mistakes.

Gain adjustment can be made to a whole file or just the section you highlight

the section you highlight. • Wave can import and export a large variety of file formats, including SMP, SFI, WAV, 16, B and VOC. In addition, you can change the sample rate of previously stored files. EFX Clips function provides real professional effects like dis-tortion, flange. digital delay, reverb, auto-stutter and others.

- A four band parametric equalizer function (WaveQ) gives you the ability to fouch up sections of your audio as well as the whole file. You can easily correct/flaws in the shund or go for the special effects.
- the special effects. Speed control leds you adjust the playback rate of your record-ed audio (up to 200% Hister or 51% slower). You can mix up to there isourd files into a new fourth one, with control of volume and starting time of each file. Clitches in the recorced sourch can be repaired by simply drawing the correction on the waveform with your mouse.

\$9900

QUAD STUDIO Professional 4-Track Recording System

Minimum Shipping USA (Except AK & H1) \$7.00, up to 3 lbs. Add 60¢ for each additional lb. For ins. add 40¢ per \$100. 1995 B&H Photo - Video. Not responsible for typographical errors. Prices vaild subject to supplier prices.

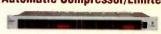
Quad Studio is the first real-time four track recorder for the PC. Now you can record two tracks directly to your hard drive (16ms or faster), then record two more tracks while playing back the first two. Whit is high performance and ease of use. Quad Studio is the perfect replacement for do hannel mixed proceed actions. 4 channel muter/cassette decks

 Graphics interface emulates all common mixer/tape deck controls
 Record, overdub, mix and bounce tracks in real time with no loss of quality.

· Previously recorded audio can be effects processed Includes Turbinal Wave program) and then mixed, in real time, with new audio.
 Includes Turtle Beach's high-performance Tahiti board
 Software is also separately available for Monterey cards. \$3990

a peet tota damage a tota

BEHRINGER MDX 1000 Autocom Automatic Compressor/Limiter



- Incorporates an interactive auto processor for intelligent program Incorporates an interactive auto processor for intelligent program detection. With the auto processor, the attack and release times are derived automatically from the respective program material – preventing common adjustment errors.
 The auto processor also allows you to compress the signal heavily and "musically" in dynamic range without any audible "pumping" freathing" or other side effects
 Provides both Attackand Release controls allowing for deliberate and variable equid processor.

- Provides both Attackand Release controls attorning for upmotrate and variable sound processing
 Switchable soft kneedhard knee characteristics. Soft knee is the basis of the "inaudible" and "musical" compression of the mater-al Hard knee is a prerequisite for creative and effective dynamics piocessing and for limiting signal peaks reliably and precisely.

MDX2000 Composer Interactive Dynamics Processor

E

- Powerful and versafile signal processing tool provides 4 most commonly dynamic control sections fully automatic compresser, manually controlled compressor, expander and peak limiter. Innovative IKA (Interactive Knee Adaptation) circuit combines the "musicality" of the "soft knee" function with the precision of the sound allows creative dynamics processing Auto processor provides fully automatic control of attack and release times. There is also manual control
 Interactive Ratio Control (IGC) expander eliminates "chatter" on or around the threshold point.
 Interactive Ratio Control (IGC) expander eliminates "chatter" on or around the threshold point.
 Interactive Ratio Control (IGC) Peak Limiter combines a clipper and program limiter. This allows for "zero" attack, distortion-free limitation of signal peaks.
 IGC is invaluable in the applications. Servobalanced inputs and outputs. Operating level switchable from -10dB to +4dB

PEQ305 Studio Parametric The Musical Equalizer

- Five independent, swächable bands. The quality of each of the five frequency bands can be modified gradually from notch to broad-band characteristics. This offers more flexibility than any graphic equalizer can provide.
 Samst 1 and 5 are switchable between shelf and peak. This is between shelf and peak. This is
- Excremely useful, since acoustic problems usually occur in the upper and lower frequencies
- Differs the "Consister (1° principle to eliminate interaction of the parametrics frequency, bandwidth and amplifude The same applies to interaction between the individual frequency bands
 Parallel arrangement of the individual filters reduces phase shift-
- ing and associated deays to a minimum
- Potentiometer response follows human hearing characteristics
 Relay-controlled hard bypass with auto-bypass function during nower failure

DEQ8000 Ultra-Curve 31-Band Digital Graphics Equalizer/Analyzer

The DE08000 is an innevative programmable graphic

The DEUBUDU is an innevative programmable graphic equiliber/Spectrum ana yere built with dightal technology. A two-clearnel unit, it features Burr-Brown 20-bit A/D and D/A convertes for input and output. It ischives the dynamics and audio quality of analog equipment while avoiding the drawbacks of analog filters such as telerances in components.

- digital basis. Filter settings are displayed either in the form of display slide con-
- rois or as a filter curve representing the actual requency response and taking the influence of several adjacent lifters into helving function makes the moving of groups of faders
- · Up to 3 additional notch filters can be used whose frequency and
- Up to 3 additional notch inters can be used whose frequency and bandwidh is freely selectable search and Destroy function automatic feedback suppressor, nuencies, thus acting as an automatic feedback suppressor.
 The integrated Real Trme Analyzer features both peak and RMS weighting, a noise generator with a separate output and a broad selection of auxiliary functions such as variable integration time, revealed in the second sec
- peak hold, etc. in Analyzer mode, a cursor is used to poll the amplitude of the
- in activation mode, a clinisor is used to poin the ampiritude of the simple bands with an accuracy of 25dB. The signal source (messuring microphone or equalizer input) fed into the analyzer input is freely selectable. Various analyzer messurements can be saved and recalled for reference purposes Au o EQ mode combines the analyzer with the equalizer section

Auto EU mode combines the analyzer with the equalizer section and thus allows for thir automatic equalization of acoustic envi-ronments. Within 0.5 reconds, the Ultra-Curve provides a linear frequency response based on the actual room acoustics
 MIDI interface allows for selecting memory locations from and external MIDI controller. Remote control of all Ultra-Curve para-meters can be relatized via system exclusive information. ALso, laveral units can be linked together via MIDI.





ACOUSTICAL FOAM



World Radio History

CREATIVE INPUT • 800.839.4678

EQUIPMENT DEALERS



DUPLICATION/REPLICATION SERVICES



DUPLICATION/REPLICATION SERVICES



140 FEBRUARY EQ

DUPLICATION/REPLICATION SERVICES



DUPLICATION/REPLICATION SERVICES



142 FEBRUARY EQ



ACROSS THE BOARD

continued from page 146

issues a serial number to any new form and keeps track of who is using it and what information was entered into it. This is the perfect setup for a studio with more than one room. Work orders can be checked out by the engineer who is working on the project and later by the accountant for entry into the billing system. I would rate this one four stars.

DIGITAL SOFTWARE

There are a couple of software packages that I have been playing with that you should know about. The first one is AD-1 Pro Audio Analyzer by Intelligent Devices. It is basically a digital meter/spectrum analyzer/ phase scope/waveform monitor for the Mac. It comes in two versions, one that runs under Digidesign's TDM bus, and another that is stand-alone. The stand-alone version requires either a Power PC Mac or a Digidesign DSP board.

The Supermeter section of the display shows digital audio levels on a 242-segment meter with a resolution of a tenth of a dB. There are actually five meter bars so that you can see stereo peak, average, and phase information all at the same time. The Spectrum Display is a 1024-point FFT display with 0.1 dB resolution and floating peak display for long-term spectral information viewing. The Phase Scope works like an old analog phase scope to display the phase relationship between the left and right channels. The Waveform Monitor displays a running history of the digital audio passing through the system. You have your choice of stereo, summed mono, or L/R difference. You can see past clipping events and evaluate compression and fade shapes with this display. The prices are \$349 for the stand-alone version and \$449 for the TDM version. I like it.

Apogee Electronics has been shipping Master Tools with UV22 for about a year now. Master Tools provides a unique metering environment for your digital audio as well as a way to preserve the high-resolution audio that is available on the Digidesign TDM bus. The metering front end of Master Tools shows a three-dimensional history of the audio levels and phase relationship as they go by. In addition to levels, Master Tools displays DC offset and allows you to perform DSP processing on the digital audio. You can remove DC; swap the left and right channels; reverse the phase of left, right, or both channels; mute any channel; and add UV22 processing to your digital audio before it comes out of Digidesign at 16 bits. If you care anything about the quality of your audio, you will get Master Tools. It is \$795 and you can check it out on Apogee's Web Site at http://www.apogeedigital.com.

These software packages are cheap when you compare them with the hardware to do the same thing. The hardware to swap channels, flip phase, mute tracks, and remove DC will cost much more than \$2500. So just think about how much money you will be saving by buying both of these packages.

Well I have successfully met another deadline (almost...only one day late) and am on my way to the NAMM show in Anaheim, CA. I have a few empty spaces in my new rack that need to be filled up. Don't tell my wife I was there buying equipment — she thinks I am just off somewhere having an affair.

Check out Roger's new web page at http://www.digital-atomics. com. It links to the Steely Dan Home Page and much more.

Leading the Digital Audio Revolution

Only Musicator Audio combines hard disk recording, MIDI sequencing, and music scoring in a single, elegantly-designed Windows[™] program. Record and playback up to 16 tracks of audio using a single sound card. View, edit, and mix audio and MIDI tracks side-by-side. Where else can you get recordable, real-time control of volume and pan for each audio track? Use multiple sound cards simultaneously and meter the output of each audio track. Import wave files. Drag and drop them at will. Why bother with inflexible multi-track tape decks or expensive hard disk recorders? Musicaton Audio does it all in a creative and easy-to-use environment.

A Sequencing and Scoring Powerhouse

Create giant scores and in seconds extract individual parts. Transcribe any tuplet and beam notes any way you want. Add chords, lyrics, and dynamics. Draw in your MIDI controllers and tweak notes on the Piano Roll. Sync MIDI to SMPTE. You can even export notation to PageMaker™ and other DTP programs. From the makers of *Musicator Win* and *Musicator Intro* comes a program that will provide you years of productivity and enjoyment.

Musicator products are available at finer music and computer stores everywhere. For a free brochure, music samples and \$5 demo disk, call:

800-551-4050

CIRCLE 52 ON FREE INFO CARD

Musicator USA P.O. Box 16026 Oakland, CA 94610 Tel: 510-251-2500, Fax: 510-251-0999 CompuServe: GO MCATOR E-Mail: 74431.40@compuserve.com

New! Musicator Audio \$39900 Musicator Win \$29900 New! Musicator Intro \$12900



- 16 Tracks of Hard Disk Recording
 MIDI Sequencing
 Music Notation
 - IIITI 51

Where Talent Meets Technology





Gand Music & Sound presents the Sixth Annual

Thur, Feb. 8, 5pm to 9pm Fri, Feb. 9, 10am to 6pm

See and hear the latest in:

- Digital Audio Workstations
- Digital multitrack recorders
- DAT recorders
 Digital effects
- Digital mixing
- CD recorders
- A/D-D/A Converters



Radisson Hotel & Suites Chicago 160 E. Huron • Chicago, IL 60611 For hotel room reservations, call 800-333-3333

Featuring

- Yamaha 02R Digital Console
- Roland VS880
- Alesis ADAT XT
- Digidesign ProControl
- Emu Darwin with Removable Jaz Drives
- Amek's New Digital Console
 Easter DMT 0 Digital Console
- Fostex DMT-8 Digital 8-Track
 Panasonic MDA 4 ADAT
- Panasonic MDA-1 ADAT
- Drawmer 1962 Digital Tube Processor

• Marantz CDR-620 Compact Disk Recorder Other exhibitors including: Akai, Digital Audio Labs, Emagic, Eventide, Genelec, KRK, Focusrite, JVC, Lexicon, Mackie, Opcode, Rorke Data, Sony, Spectral, Tactile Technology, and Tascam.

Seminars:

"Hit" man Mick Guzauski, engineer for Madonna, Mariah Carey, Kenny G, Boyz II Men, Michael Bolton, Quincy Jones, and many others.

Learn from this expert on Digital Recording Sponsored by

EC & ALESIS & DIGITAL AUDIO LABS

"Studio Acoustical Design" John Storyk

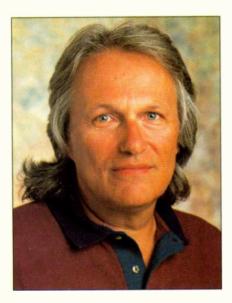
of Walters-Storyk Design Group Winner of Three TEC Awards for Studio Design Sponsored by Gand Pro Audio

Admission: \$10 in advance

For registration information, contact Bob or Kathy at Gand Music & Sound 780 Frontage Rd. • Northfield, IL 60093

Phone: 847-446-GAND (4263) Fax: 847-446-9082 On-line: http://www.gand.com Co-Sponsored by

Tales of a Blithering MIDlot



he ramblings of a technology junkie

I have mentioned before that technology takes advantage of its users. When there were only five or six television channels, there was nothing worth watching. When cable TV arrived on the scene there were 40 channels of nothing to watch. Now there are satellite dishes with hundreds of channels. There is still nothing worth watching; it just takes longer each day to surf through the choices to convince yourself that there is nothing to watch.

I have noticed the same thing with online services. They are now all offering connection to the Internet. They have lowered their prices to make you think that you are getting a great deal, but as more and more people connect, the response time gets worse and it takes you longer to do whatever it was you were doing than it did before when you were paying a higher rate; so it ends up costing you more in the long run.

Synthesizers now come with billions of presets so that you won't have to spend a lot of time programming in the sound that you want. Great, but it takes you longer to search through the presets than it would have taken to program your sound from scratch.

JBL & STUFF

As I write this column, I am sitting by the pool at the Ritz-Carlton Hotel in Palm Springs, CA. It is JBL's 50th anniversary. They are having a big bash for all of their international distributors and a few press people. I snuck in.

The theme for the meetings was "teamwork," pulling together to get the job done. There was a guest speaker who knows all about working together as a team, Jim Lovell, commander of Apollo XIII. Talk about teamwork! He told the whole story, and it was much worse than it seemed in the movie version. This guy was 200,000 miles away trying to change the course of his space craft so that he could get back to earth. I no longer have any reason to get mad when installing Digidesign software and it eats my hard disk. It is no big thing if my cat pulls the tape out of my ADAT. Lovell was in deep space doo-doo and pulled it off. I am impressed.

Speaking of Digidesign, I did get a copy of Session to try out. It uses the same copy protection scheme as Pro Tools 3.1. It does not like the Mac IIci with the Daystar 601 card. I tried installing the new software and it ate all of the installs currently on my hard disk for Pro Tools 3.0, Sound Designer 2.8, Masterlist CD 1.0, my TDM install, and probably a few other things that I haven't found yet.

I mentioned the problem with PT 3.1 last summer, but nothing got resolved. It is now January and it has not sent any fixes. I will try once more to try to talk Digidesign into changing its protection scheme to the dreaded *dongle*. My E-Magic sequencer uses the dongle method, and after a dozen hard disk crashes 1 can still use E-

Magic without any problems. I can even have the software installed on multiple hard disks without worrying about losing an install. Just plug in the dongle and you are in business.

INTERNET COMMENTS

I noticed on the Internet that some people are interested in computer-based track sheets. I have been using Fastforms by Power-Up Software. This is a pretty simple program for form generation and printing on the Mac. You basically draw the form on your computer screen and name each field where data will be entered. The data entered into each form is stored in a database. You can print a form with the data in the blanks, or you can print out blank forms for hand entering the data. I use Fastforms for DAT J-cards, CD labels, track sheets, cassette J-cards, and tape box labels.

Opcode used to make a Mac-based program called Track Chart, but I am not sure if it is still available or still supported. Track Chart lets you group tracks together in stereo pairs and enter icons denoting live tracks, drum tracks, or sequencer tracks. Track Charts lets you display your track information as a track sheet (with any number of tracks you choose) or as a timeline. If you have a Studio 3 MIDI interface, you can feed the audio from any track into the audio input of the Studio 3, and, along with timecode or MTC, Track Chart will detect where there is audio information and where there is blank tape. This information is displayed in the timeline so that you can see where you have spaces in your recording. It works great for those times when you need to find a few empty bars to stick in some percussion or guitar lick.

Last, but not least, InFORMed by Shana Corp. is the top end in form generation and distribution. A suite of programs called InFORMed Designer, InFORMed Manager, InFORMed Number Server, and InFORMed Revision Distributor provide all the functions you will ever need to maintain formbased information in your studio. InFORMed is designed to work on a Mac network. After the forms are laid out, any user on the network can "check out" a form from the form library where he can fill it out locally. InFORMed

continued on page 144



...the first new Neumann Tube microphone in more than 30 years.

Neumann understands tube microphones. We've been making them since 1928. The U 47, M 49, U 67, and KM 53 are legendary – coveted by engineers and producers worldwide. When asked to reproduce a microphone with the 'classic' characteristics of our older jems, we decided to go one better. Enter the new M 149 Tube. Utilizing the K 49 capsule and headgrill from the classic M 49, coupled with the transformerless FET 100 circuitry from our TLM 50, the M 149 Tube is a modern microphone in the classic Neumann tradition.

The M 149 Tube features 9 polar patterns and a 9-position high-pass filter. With a self noise of 11 dB-A, the M 149 Tabe is the quietest tube microphone in the world. The tube and associated circuitry for the M 149 Tube are mounted on modular 'circuit cards.' In the future, a variety of tube modules will be offered, allowing you to customize your M 149... different tubes for different recording situations.

Neumann has brought more than 50 years of design expertise to the M 149 Tube, and it shows in every detail of this truly outstanding microphone.

Neumann... the choice of those who can hear the difference.



THE HOLDER HOW



Neumann USA

6 Vista Drive • PO Box 987, Old Lyme, CT 06371 • Tel: 860.434.5220 • FAX: 860.434.3148 Neumann1USA West Coast • Tel: 818.845.8815 • FAX: 818.845.7140 In Canada: 221 Labrosse Ave., Pointe-Claire, PQ H9R 1A3 • Tel: 514-426-3013 • FAX: 514-426-3953 World Wide Web: http://www.neumann.com

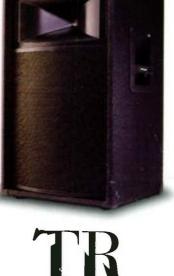
CIRCLE 70 ON FREE INFO CARD

The M 149 Tube

Made for the way you play.



Turn it UP



With the all new TR Series loudspeakers, you can crank it all the way up. Because with protective SonicGuard^m Circuitry, they are more bulletproof than any speakers in their class.

Fact is, we've already put the TR speakers through the most merciless torture test in the industry by cranking them at full throttle for 300 hours. So a 3-hour gig is just a warm up act.

And while you're playing hard, rest assured your sound is never compromised. With features found only in higherend speakers, the TR speakers always maintain a balanced output of crisp highs and killer lows.

At JBL our goal was to bring you a more affordable JBL speaker, not a cheap one. So turn it on. Turn it up. And give the new TR Series a listen. At this price, it's time to get a speaker that was made for the way you play.



H A Harman International Company

TR. MR. SR. The unbeatable line-up of sound reinforcement products from JBL Professional.