

LOW-COST DISITAL MULTITRACKS SPINAL TAP'S STASE MIKING TIPS AES NEW PRODUCT AWARDS

THE RECORDING & SOUND MAGAZINE

1.1

Recording HADDONNA BY SHEP PETTIBONE



EXCLUSIVE EQ REVIEWS Boss EH-50 Stereo Enhancer Lexicon LXP-15 Effects Processor Zoom 9030 Processor

TRI-PO

VOCALISTS-DON'T COMPROMISE.

Don't settle for second rate sound from your performance microphone. AKG's new Tri-Power™ performance microphones are revolutionizing live performance with a pure, powerful sound that gives vocals the punch and presence to cut through to your audience. AKG's Tri-Power design team worked for years with major performers and touring sound engineers to develop mics that were perfect for live performance musicians. The resulting Tri-Power series ombines the acoustic performance of our world-standard tudio microphones with new levels of ruggedness, cedback rejection, high output and mechanical noise uppression. 🦀 Hear and feel the power of Tri-Power woral and instrument mics at select musical instrument dealers near you. Tri-Power is what you've been waiting for-live performance microphones from AKG.

Tri-Power D3900 Vocal **Performance Microphone**



PowerGrip" housing

Rear ports for hypercardioid pickup patter

Acoustic chomber

Bass Rolloff switch

Reticulated open-cell foam inner shield

DuraShell[®] protective baskets

Paratex Filter

Makrofol " diophragm

Extra-heavy gauge wire mesh grill

Blast-diffusing, fine-meshed fabric

Positive oction, shockabsorbing, click lock bayonet mount

Hum-suppression coil



AKG Acoustics, Inc. 1525 Alvarado St., San Leandro, CA 94577 Tel: (510) 351-3500 Fax: (510) 351-0500 r is it trademark of AKG Acoustics, Inc mark of Akustische ur Kino-Gerare Gesim biH , Austria relacknowlenged. 1992 AKG Acoustics, Inc AKG is a regist

World Radio History

Treble Boost switch

CIRCLE 02 OH FREE INEC CARD

*CRANK IT UP

and compare the 3900's handling noise, feedback rejection and sonic performance against **any** other microphone.

When you set out to unearth some strange and startling sounds, start with a tape that's truly out of this world.

That's the philosophy of Mark Mangini who, as the

continually expand the boundaries of our digital audio tape technology. The proof is found in our new Pro DAT Plus. Ultra fine metal particles help deliver improved playback

head of Weddington **Productions, created** the sound effects for major science-fiction thrillers such as Star Trek, Gremlins and Poltergeist, just to name a few.

> Mark now relies on Sony

Pro DAT

SONY

É

The new Sony Pro DAT Plus,



Plus tape for recording and archiving his one-of-akind sounds and hauntingly beautiful melodies.

"The low error rate and consistent reliability of Sony Pro DAT Plus is critical in our search to capture and preserve the unusual sounds

used in fantasy and horror films," says Mangini. and unique abum cose. People like Mark put Sony's engineers in the right spirit to output in addition to a superior S/N ratio, while special binders and surface treatments help to achieve error free performance in any natural or supernatural environment.

You'll also discover a heat-resistant shell, anti-static lid and new

two-tape album case to protect your tapes from things that go bump in the night. And an erasure prevention tab to make sure you won't accidentally lose your "voices".

As Mark Mangini knows, a quality tape library is the lifeblood of any post sound company. He also knows that with new Sony Pro DAT Plus, life at Weddington Productions is a lot less scary.

SONY SONY RECORDING MEDIA



FEATURES

AES WRAPUP: SECOND ANNUAL BLUE RIBBON AWARDS

See	our editors' and contributors' picks for the best in show
AES I	WRAPUP: AES PRODUCT REGISTRY
Cou	Ildn't make the show? Afraid you missed something major? Our product dire

tory brings you all the hot gear introduced at the show
DIGITAL TAPE WARS: S-VHS VS. 8MM By Rick Schwartz
The gear is here! But which format is best for you?

THE MADONNA DIARIES By Shep Pettibone



BAND IN A VAN

STINKING UP THE GREAT INDOORS By Tony Savona	
MEET MY RACK: JASON SOUND	
THE PERFECT LIVE GIG COUNTDOWN By Mr. Christy	62
LIVE MIXING FOR \$3000 OR LESS By Wade McGregor with Hector LaTorre	

WORKSHOPS



2007/00-concers Step Pertilionis In Pener Managers Capite of Concentry, MSS-Dimensionsby, Mark Albander,

COLUMNS/DEPARTMENTS

WI INSIDER: THE COMPUTER QUAGMIRE B	y Craig And	erton	26
SYSTEMS: FIGURE 8 By J.D. Sharp			84
BASICS: NO NOISI! By Len Feldman			88
FAST FORWARD: MEMBERSHIP HAS ITS PRIVILEGES By Martin Polon			
MAINTENANCE WITH MOORE: MAKING THI	E MOST OF YO	UR MICS By Dr. Richie Moore, Ph.D	92
ACROSS THE BOARD MY FIDELITY By Rog	er Nichols		114
ETTERS TO EQ	8	IN REVIEW:	
Q TIPS	12	• ZOOM 9030 PROCESSOR	
A30	14	BOSS EH-50 STEREO ENHANCER	100
ROOM WITH A VU/ALAN MENKEN	24		
		& MIDI REMOTE CONTROL	103

World Radio History



......

CIRCLE 07 ON FREE INFO

2

EINSTEIN SUPER 'E'

THIS CONTROL SURFACE DOESN'T NEED A GENIUS

AMEK's large EINSTEIN expands the 64-input concept of the standard console, remaining, as ever, within the bounds of sensible pricing.

EINSTEIN SUPER 'E' delivers

 80 inputs all with 4-band AMEK equalization – more than enough for a small mountain of multitracks, synths, multi-output samplers and effects devices

 80 inputs all with Mute and VCA automation using AMEK's SUPERTRUE system – total control over all sources in the most demanding of mixdownes with SMPTE synchronized CUE LIST with MIDI triggers and compatability with AMEK MOZART and HEN-DRIX installations.

• Optionally, 40 or 80 channels of AMEK VIRTUAL DYNAMICS – allowing you to select a Virtual device for each channel from nine standard templates including 3 Compressors 3 Gates Expander Panner and Limiter

• Onboard jackfield with 576 break points and space for 384 Effects jacks which can be added in blocks of 24. All inputs and outputs terminated to EDAC for rapid installation.

 40 meters selectable o er the 80 paths allowing precise monitoring of levels throughout the signal chain

• Compact size - only 2 15 Metres (85 inches) mide

Head Office, Factory and Sales: AMEK Systems and Controls Ltd., New Islington Mill. Regent Trading Estate. Oldfield Road, Salford M5 4SX, England. Telephone: 061-834 6747. Telex: 668127, Fax: 061-834 0593.

AMEK TAC US Operations 10815 Burbank Blvd, North Hollywood, CA 91601. Telephone 818 508 9788 Fax 818 508 8619



Demos to masters. Creativity to tape. Dreams to reality. Magic phrases for those who want to make music that sounds as good as it feels.

-9 -12 -15 -18 -22 -27 -32 -38 -44 -51 -60

RECORD

POWER

The inspiration for these thoughts is the Alesis ADAT Professional Digital Audio Recorder, a technological revolution that tears down the walls to your creativity while delivering world class master recordings. Too good to be true?

Here's the concept. ADAT fuses a supersonic Alesisdesigned very large scale integrated chip set with the proven reliability of an industrial grade S-VHS* tape transport and a logical, sensible user interface. The result is a digital tape recording system that exceeds the most demanding requirements of professional audio and that can be used by literally anybody. Hard to believe?

Here's some specs. Bandwidth 20Hz to 20kHz ±0.5dB. Total Harmonic Distortion plus Noise 0.009%. Wow and flutter unmeasurable. ADAT uses the professional standard 48kHz sample rate and delivers better than 92dB dynamic range.

Here's some features. ADAT uses the familiar tape recorder controls that you already know how to use so recording is fast, intuitive, effortless. Connections are provided for balanced +4dBu levels on a single 56 pin ELCO** connector and unbalanced -10dBV signals on 1/4" jacks. And ADAT uses S-VHS tape because it's a proven, robust recording medium with wide 1/2" tape to solidly support ADAT's 8 recording tracks while delivering 40 minutes of recording time.

6

The best part. ADAT's Proprietary Synchronization Interface (Patent Pending) locks multiple ADAT's, independent of the audio tracks, to single sample accuracy $\pm 5\%$ of 1/48,000th of a second! In other sciences this is referred to as 'air tight'. So multiple ADAT's function in perfect mechanical and electronic unison: up to 16 ADAT's without an external controller. That's 128 tracks!

More best part. ADAT's Proprietary MultiChannel Optical Digital Interface (Patent Pending) simultaneously sends all 8 tracks of recorded information out the Digital I/O for perfect safety tapes and perfect track bounces.

Even more best part. The optional BRC Master Remote Control opens a whole other door to the ADAT miracle. With it you can control up to 16 ADATs (128 simultaneous tracks) with full transport functions, track offsets, machine offsets,



8 Tracks to Megatracks, Megatrack and ADAT Worldwide Network are trademarks of Alesis Corporation *VHS is a registered trademark of JVC **ELCO is a registered trademark of Elco Corporation-a Kyocera Group Company Also available: The AI-I ADAT to AES/EBU and S, PDIF Digital Interface with sample rate converter. RMB 32 Channel Remote Meter Bridge.

Call 1-800-5-ALESIS for information about the ADAT Worldwide Network



digital assembly editing via the Digital I/O, SMPTE and MIDI Time Code, Video Sync and more.

What does all this mean? Here's just a few benefits.

It's commonly known that many hours are wasted during expensive album projects while the artist, producer and engineer work in vain to reproduce the rhythmic feel and tonal nuance of demos. Demos that couldn't be used because they suffered from noise, limited bandwidth and overall sonic feebleness. Those days are over forever. ADAT's Sync and Digital I/O perpetually link your demos to your masters making them all part of the same creative process. Every track you record on ADAT is a master track that can be flown into any other ADAT recording, at any time. The best part is that ADAT can be there at any time to catch you at your best, flawlessly stored in the digital domain...forever.

Need more tracks? ADAT studios can be expanded at any time. The cost of a single ADAT is remarkably inexpensive and new ADATs can be added as budgets permit. Add the BRC at any time for more control and advanced editing. Producers please note: with ADAT. Megatrack™ recording is a reality. Your favorite sax player lives in Idaho? No problem. Send 'Supersax' a formative' tape with a guide track of your song. You'll get back 7 tracks of burning solos you can fly back into your production. All in perfect sync, all in the digital domain. All dripping with soul. Want more tracks? Just send more tapes.

ADAT is not only a new recording medium, it is the new recording standard. Imagine a network of ADAT users from bands, composers and project studios to professional studios, video editing suites and broadcast production studios. All recording master quality tracks with full compatibility and no barriers between their creative disciplines. In fact, we're launching the ADAT Worldwide NetworkTM multitrack recording group to facilitate communication between ADAT studios.

In time we'll all start taking these little miracles for granted. Before that inevitable event, unpack your first ADAT and track a minute of single notes and chords on your favorite instrument. Play loud, play soft. Play it back and listen really close. It's always a good feeling to have your mind completely blown.

See your Alesis ADAT Dealer today and start Megatracking on ADAT.



Alesis Corporation 3630 Holdrege Avenue Los Angeles CA 90016

World Radio History



A PSN Publication Vol. 3, No.5 December 1992

PAUL G. GALLO Publisher/Editor of Director

HECTOR LA TORRE Executive Director

MARTIN PORTER Executive Editor

ANTHONY SAVONA Manoging Editor

DAVID JACOBS Senior Editor

CRAIG ANDERTON West Coast Editor

DENISE GRAHAM, JON VARMAN Assistont Editors

DAVE BRODY, EDDIE CILETTI, ALAN DIPERNA, LEN FELDMAN, BOB LUDWIG, WADE MCGREGOR, DR. RICHIE MOORE, ROGER NICHOLS, JIM PAUL, MARTIN POLON, J.D. SHARP, TIM TULLY Contributing Editors

MP&A EDITORIAL

Editorial/Design Consultants

KATHLEEN MACKAY Director of Advertising Sales

DANIEL A. HERNANDEZ, MATT CHARLES Advertising Sales

RIVA DANZIG

Creative Director

MARK ALHADEFF Art Director

FRED VEGA Praduction 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, Fait: (212) 213-3484

EQ (ISSN 1050-7868) is published bi-monthly by F.S.N. Publications, 2 Park Avenue, Suite 1820, New York, NY 10016 BPA Consumer Magazine mempership capited for 11/92. Sev and class postage pending at New York, NY and additional mailing affree. PCISTMASTER. Sen4 adites changes to EQ, P.O. Box 0532, Baldwin, NY 11510-0532. SUBSCRIPTIONS: U.S. 1 yr. S19 97, 2 yrs 533.97. 3 yrs 349.97; CANADA add 552 per yr. for surface, Other countries add \$10 per yr. For surface, All add \$20 per yr. for Airmail. Back issues \$5. All product "nhumation is subject to dhange publisher assumes no responsibility for such change. All lited model numbers and product names are manufacturers' registered trademarks.



CATCH YOUR BREATH

I have been asked to clarify a point that I made in the August 1992 EQ&A, "Floored by Noise."

Briefly, the "breathing" effect that I referred to when using dbx generally stems from an improper user-set record level. I would like to point out that with proper monitoring of one's record levels, dbx is not likely to exhibit this phenomenon. While both dbx and Dolby will "breathe" when record levels are set too hot, it should be noted that, for the less experienced user, Dolby is generally tolerant of such occurrences.

As a matter of good practice, it should also be noted that, since timecode is a steady space signal and will never be present in a final mix, one should generally avoid noise reduction of any type on a sync track.

> Roger Maycock Senior Product Specialist Fostex

BREATHLESS

Some generalizations and misconceptions need to be addressed regarding Roger Maycock's comments about noise reduction in the August 1992 issue.

Roger is an employee of Fostex, and Fostex uses only Dolby noise reduction on its analog recorders.

dbx noise reduction offers significantly greater improvement in dynamic range than Dolby C, and it does so across the entire audible frequency range. In fact, dbx Type I Noise Reduction can effectively double the dynamic range of an analog open-reel recorder up to a maximum of 115 dB. This is nearly 20 dB greater than the dynamic range of a compact disc. dbx typically produces an improvement in the overall signal-to-noise ratio of 30 dB, while Dolby C typically offers only a 20 dB improvement and only at higher frequencies. The better signal-to-noise ratio provided by dbx processing allows more bouncing of tracks with less degradation from noise buildup.

dbx uses a 2:1 linear compression/ expansion ratio in combination with the carefully tailored high-frequency preemphasis/deemphasis circuitry in both the audio path and the control sidechain, in addition to providing greater dynamic range. This full-bandwidth, linear compansion design eliminates any possibility of multiband or "spectral" mistracking. It is also immune to mistracking caused by variations in record or playback level settings. dbx processing effectively maximizes available headroom. In narrow format analog recorder such as 1/4-inch 8-tracks, 1/2inch 16-tracks, etc., maintaining sufficient headroom is as important to a good-sounding result as is the prevention of excessive hiss buildup.

As with most "problem solvers," noise reduction can be used ineffectively. It can be made to "breathe" under certain circumstances, but these are easily avoided. Some dbx noise reduction users also ran into tracking errors in the early days of narrow-track formats when certain dbx-licensed manufacturers built their own systems using components inferior to those in dbx-manufactured units.

Another phenomenon observed by early users was the appearance of "noise tails." These turned out to be the modulation noise of the tape itself — no longer masked by the hiss that used to cover it up. With improved tape formulations and proper biasing, this also ceased to be a significant issue. Any "heavy breathing" you may experience today is likely to be your own, from the excitement of having your recording sound so good!

As to the timecode issue, most multitracks with onboard dbx offer a defeat for the last track (4, 8, 16, etc.) and the dbx outboard units offer bypass for each channel. (There are plenty of anecdotal stories about sessions where someone forgot to defeat the dbx on the timecode track and it worked just fine.)

Which noise reduction system is better? After all is said and done, it really boils down to personal preference. There are no absolutes. (Is a Les Paul better than a Stratocaster?) The questions really are: What is going to work best for your needs? What is the track format? Are you going to do lots of track bouncing? What about compatibility with other machines? What other key features does the machine offer?



EQ wants to dialogue with you. Write to: Letters to the Editor, EQ, 939 Port Washington Blyd., Port Washington, NY 11050. Letters must be signed, and may be edited for clarity and space.

Many people prefer the added punch and dynamic range that dbx gives their tapes and it is the overwhelmingly dominant system in narrow-format applications.

> Iawxillion Loeb Product Manager dbx Professional Products

FLASHBACK

In the AES article by J.D. Sharp in the October 1992 issue of your fine rag, the "Summer Of Love" was improperly attributed to the year 1968, when in fact, it was 1967.

> Linette Wilson Red Star Sound Seattle, WA

A HARP, BUT NO ANGELS

Regarding the article "I Had a Secret" in the August 1992 issue of EQ: It seems to me that project studio owners should be aware that this selective harassment by the HARP organization may begin with high-profile home studio owners like Chas Sanford, Hank Sanicola, Eddie Van Halen, Steve Vai, and Jay Graydon, but the message to the smaller project studio owners is clear. At some point when you acquire enough recording tracks, outboard gear, prominence, or whatever, you cross the line as far as HARP is concerned, and become "unfair competition."

Your magazine is the forum for home and project studio owners. I truly hope that this issue will continue to be addressed by you and your readers. A good start would be contact information for us Los Angeles readers so we can help put an end to this monopolistic harassment by the HARP organization.

Alan Wilcox Studio City, CA

NOT TO HARP, BUT...

Re: "I Had A Secret," FQ, August '92.

Despite the best efforts of HARP's studio membership, there still appears to be a great deal of misinformation regarding its attitude toward home studios. Anybody reading the article in EQ's August 1992 issue might be left with the impression that our association has targeted home studios, such as the one belonging to Chas Sanford. This is not the case.

HARP was set up several years ago

to address the changing requirements of our industry, the long- and short-term concerns of the Southern California studio community, and to specifically address the new generation of artistand producer-owned facilities and the confusion being caused by home studios "for hire."

At that time, the association was of the opinion that project studios would work in liaison with commercial studios on certain portions of their projects, and not rent out their facilities "for hire."

Let's be very clear on this very important distinction: A number of artists, producers, and engineers own recording equipment, that is installed in their home or garage, and intended for use on their personal projects. Such facilities are an essential part of the creative process of making music. They allow artists, bands, and their producers to develop ideas or to record portions prior to coming to a commercial facility for tracking and/or mixing.

What HARP does have a problem



LETTERS TO EQ

with, however, are the home studios that are hired out for projects as if they were *commercial* facilities. HARP members feel that it is important for all commercial studio businesses to compete on equal terms. We achieve this equality through city, state, and federal business law guidelines and local zoning laws. What HARP has *never* done, however, is to report owners of home studios to any authorities. Our style is based on persuasion rather than on confrontation.

As to the specifics of the Sanford case, the truth is far more boring than the fiction presented by Sanford in EQ. In 1989, having printed some very nicelooking, four-color brochures for his Secret Sound Studio, Sanford distributed them in various locations including a commercial recording studio. A brochure was eventually shown by an independent studio owner to a representative from the L.A. Building and Safety Dept., which enforces zoning laws. Chas was subsequently visited by a representative of that department, to whom he explained that he only used his 24-track facility for personal use.

When the time came to sell his property in early 1992, Chas Sanford apparently discovered that California's real-estate disclosure laws required that he reveal the presence of a business — a recording studio. He then contacted the zoning department to request a zoning variance to operate a commercial business on residential property (which, incidentally, would increase its sales value). The department held a hearing (which is a matter of public record) and his request was denied.

Because Chas Sanford had originally claimed that Secret Sound was for his personal use, then requested a zoning variance to reclassify the studio as a business, the Building and Safety Dept. revisited his property. They allegedly then issued a cease and desist order regarding his studio operation. HARP had nothing to do with these events.

Specifically, HARP's position on the role of home studios is simple: Our association has strong objections to home studios being offered for hire by their owners. City zoning laws are implemented to prevent residential areas from being inconvenienced by the noise, disturbances, and extra traffic that's normal for business areas. Also, commerciallyzoned facilities contribute to a city's dayto-day operating costs through levies, taxes, and various physical obligations imposed upon them.

It's time to play fair. Commercial studios, by their very nature, can have a substantial impact upon a residential area. Zoning laws were developed for a purpose: to prevent commercial activities within a residential setting. Despite the paranoid rumors being circulated by individuals such as Sanford, HARP simply wants the playing field to be leveled; we do not intend to put the competition out of business.

In fact, HARP has been very active in addressing the day-to-day needs of the California recording community. Lack of space precludes mentioning more than a few of our recent achievements:

• Developing a common ELCO connector standard, now being used by all major studios and rental companies.

Offering technical seminars and demonstrations from major equipment

The Ultra Effect Affect:



914 NIDIO SNEEDING SOCIETY CONVENTION

The SDR[™] 20/20 Digital Effects Processor

is your catalyst for creativity in both the performing and recording environment. There are 21 basic effects such as EQ, Reverb, Chorus, Delay, Distortion, etc., which are totally independent of each other and may be combined in any order. These effects represent the sound cornerstones of the SDR 20/20's amazing possibilities. The innovative capabilities of this processor call for visualization more than explanation. For example, up to eight effects can be strung together in an "effect block" and each may be independently mixed and its level separately adjusted, with the final mix assignable to the preset of your choice. 128 permanent factory presets and 128 duplicate presets which may be altered, changed completely, or saved, give you endless possibilities in creating your own custom sound effects.

In addition to these numerous effects possibilities, the SDR 20/20 offers the ability to focus all processing power to a single "Ultra" effect. An example of this Ultra

vendors and improving relationships between studios and manufacturers through special forums.

• Setting up a central tape-purchasing account for association members, offering substantial discounts.

• Conducting a review of Workman's Compensation with California's rating bureau, to establish a separate, lower-rated category for recording studios.

• Creating a set of standard guidelines for booking and client relations, plus documentation standards for sessions.

• Establishing dialogs with record companies to help redefine bookings and PO processes.

• Setting up a scholarship fund for student engineering education.

• Forging stronger links with SPARS and England's APRS (Association of Professional Recording Services), to serve the international interests of studio clients.

With my having taken the time to redress the balance of negative publicity directed against our association, I'm sure *EQ* readers will better appreciate the wide scope of HARP's interrelated activities. You can contact HARP at 8033 Sunset Blvd., Suite 89, Los Angeles, CA 90046. Tel: 213-655-2452.

> Teri Piro President HARP

UP IN THE SKY! IT'S A...

Regarding David Jacobs's "Flying Trapezoids!" article in the October issue of *EQ*, I was concerned about the impression that the article left me with. Although I do not think it was the author's intent, the article seemed to imply that all trapezoidal loudspeakers can be flown. Nothing could be more false.

A loudspeaker manufacturer must provide many safeguards — both internally and externally — to build a flyable loudspeaker enclosure, whether it is trapezoidal or rectangular. A complete flying loudspeaker system includes internal bracketing, external rigging components, and a structurally designed enclosure; all of which must meet at least a 5:1 design factor when tested. My concern regarding the article stems from the fact that the readers of EQ might not be schooled in rigging and the physical universe. If just one reader thinks that all trapezoids are flyable and attempts to fly a low-grade unflyable loudspeaker because of its trapezoidal shape, and that loudspeaker then falls and injures or kills someone, the professional rigging community would be in serious trouble.

Rigging is not an easy profession. Anything being placed over the heads of people should be done only by trained and qualified rigging personnel.

> Andrew T. Martin President ATM Fly-Ware Carson, CA

CORRECTION

On page 42 of our last issue (October 1992) we incorrectly listed the name of Lindsey Buckingham's console. It should have read "Neotek Elite."

Imagination Powered by Innovation



effect is the "**Uitra Reverb**" currently included in the palette of selectable effects. With software upgradeability the SDR 20/20 can be customized to meet future studio requirements.

Balanced ins and outs with selectable inputs and a 64x oversampling 16-bit Delta-Sigma A to D converter make the SDR 20/20 studio ready and ensure low noise and better sound quality. So, go ahead and power up your imagination through the innovations of Peavey with the SDR 20/20 Digital Effects Processor.

Made in the U.S.A.

FEATURES

- High quality Reverb and Ultra Reverb
- Multi-effects formed from individual effects
- Independent mix and level control where applicable
- Dynamic effect parameter control via MIDI
- Remote programming capabilities via MIDI
- Balanced ins and outs
- +4 dB to -10 dB selectable and programmable
- Four types of EQ
 - Software upgradeable



A Division of Peavey Electronics Corporation • 711 A Street • Mendian, MS 39302-2898 • Telephone (601)483-5365 • Telex 504115 • Fax (601)484 4278

MEMBER

CIRCLE 48 ON FREE INFO CARD

AUDIO MEDIA RESEARCH



CIRCLE 28 ON FREE INFO CARD



EDITED BY DAVE BRODY

RUSHIN'/DRAGGIN'

When you are doing drum replacement, the Russian Dragon timing accuracy meter can help you line up a triggered drum sample with the old drum on tape. You feed the old drum into Russian Dragon Channel One (the reference signal input) and the drum sample into Channel Two. The LED display tells you if the sample is "rushin'" or "draggin'" compared to the old drum. But there is a bit of confusion about whether or not the LED display of the Russian Dragon is calibrated - in other words, about whether the LEDs indicate a specific amount of timing error.

Well, here's the scoop: The Russian Dragon *is* calibrated, so the user can tell immediately how many milliseconds off the two sounds are. The confusion seems to have arisen from a knob on the older models, that is called the "WINDOW" control. This knob adjusts the time sensitivity of the LED display in milliseconds per LED. If the knob is set on "4" and the 7th LED on the draggin' side is lit, the Russian Dragon is indicating that the new sample is 28 milliseconds late compared with the old drum.

You then move the sample (by shifting in a sequencer, offsetting using SMPTE, using a digital delay, etc.) to get rid of the 28-millisecond error. Keep tweaking the position of the sample until the Russian Dragon indicates that it is in perfect sync with the old drum. The two sounds are now aligned to an accuracy of .1 millisecond.

> Marius Perron Chief Engineer Jeanius Electronics

ROCKIN' THE REELS

For those of us who haven't gotten our mitts on a Russian Dragon yet, there's still a highly accurate - albeit slower way to correct trigger-timing problems. Let's say you're doing a drum replacement (as in Marius Perron's tip, above) on a multitrack format. Record about 20 seconds or so of the replacement sample on a separate track. Rewind to the first hit. Now advance the tape by hand as if you are marking an edit point. Manually move the tape back and forth over any spot where the original drum and its new replacement live. (This time honored operation is called: "Rockin' the Reels," "Scrubbing," "Jogging," "Skwudging," or [in Milspec-speak] "Signal Acquisition and Zero-cross Locus Verification.") Watch your track meters and see who hits first. It may be helpful to solo these two tracks (or mute everything else) and pan, say, the old track left and the new track right. This allows you to use both eyes and ears to scope out who's rushin' and who's draggin'

You can now "guesstimate" what SMPTE offset, sequencer tick-shift, or delay product you need in order to bring your replacement sample into sync. You can do this by trial and error (it usually takes a surprisingly small number of whacks to get it right). Or, if you're working in reel-to-reel tape, you can get very close on your first try by actually measuring the misalignment. Mark the hit points of both original and replacement with a grease pencil as if you were about to perform a physical tape edit. Measure the distance between marks. Divide by the tape speed and convert to milliseconds, clockpulses or frames and subframes depending upon what box you're using to tweak your sync.

It is often a good idea (and, indeed, has become standard procedure in many studios) to record a quarter- or eighth-note click on a separate track as an absolute time reference even when working with carbon-based performers rather than silicon-based ones. Comparing to click by rockin' those reels can get you into the sync ballpark very quickly.

> Tap out those tips to: EQ Editorial Offices, 939 Port Washington Blvd., Port Washington, NY 11050

	ALESI		DRAVER	20 BANDWIDTH SMEALTANE ORS DIGITAL EFFECTS PROCESSON	and the second	and the second	Carl Carl			115
	CLIP 🚺 -6 dB ● -12 dB	1	e	PROGRAM 99 "CLASSIC REVEPB"		Fills	REVERB	DELAY 2	PITCH 1	
.)	-1.8 dB 🅤	INPUT	OUTPUT			STORE	CONFIG 6	MIX	MOD a	-

100 Reasons to Own a Quadraverb

- 1. It's the all-time classic reverb
- 2. It's fully programmable
- 3. Nobody tries to mix without it
- 4. I already own 3 and need 1 more
- 5 Everybody I know has one
- 6 Stereo in and Stereo out
- 7. Incredible MIDI implementation
- 8 Does up to 5 effects at the same time
- 9. Incredibly quiet
- 10. Great Dynamic range
- 11. Vocals sound incredible with Quadraverb
- 12. My guitar gently weeps
- 13. My girlfriend won't let me mix without it
- 14. My boyfriend won't let me mix without it
- 15. My producer won't let me mix without it
- 16. It works as well on stage as in the studio
- 17. 16 Bit Linear conversion
- 18. It's easy to program
- 19. It's easy to hook up
- 20. It works great in my MIDI system
- 21. It works great in my guitar rack
- 22. It works great with my bass rig
- 23. My flute sounds better through it
- 24. 20Hz to 20kHz Bandwidth
- 25. 127 MIDI sources can modulate any of 57 parameters
- 26. Works great in any country
- 27. You can take it on cruise ships
- 28. Everybody in Hollywood thinks it's cool
- 29. They use it back East
- 30. They use it out West
- 31. The North and South think it's great
- 32. Affordable enough for demos
- 33. Perfect for masters
- 34. It will be with you throughout your career
- 35. Those cute buttons with the little lights
- 36. Fressure sensitive programming buttons!!
- 37. The chorus is unreal
- 38. Reverb + chorus Programs are awesome
- 39. Pro engineers can': believe it's so inexpensive
- 40. They've got one in every room where I rehearse
- 41. I need one in my rack and one at home
- 42. The vocals stunk until I added the Quadraverb
- 43. Justant vocal doubling chorus + reverb + eq

Call 1-800-5-ALESIS for more information and more reasons to own a Quadraverb

Alesis Corporation 3630 Holdrege Avenue Los Angeles CA 90016

World Radio History

- 44. My guitar finally has a voice
- 45. 20 Bit processing
- 46. Incredible resolution for perfect reverb tails
- 47. It sings
- 48. Fills in the cracks of my keyboard
- 49. It's like a synthesizer for processing
- 50. There's no better reverb

- 51. The multitaps are unbelievable
- 52. You can delay up to 1.5 seconds
- 53. With everything on, there's still 800ms of delay
- 54. The price is unbel-evable
- 55. All Alesis reverbs are great
- 56. It makes my drum machine sound real
- 57. It was the finishing touch on my demo
- 58. ...and my demo got me a deal
- 59. It's great to borrow, but better to own
- 60. Late at night under headphones I leave the planet
- 61. My mixes sound like movies
- 62. My speakers sound bigger
- 63. Close your eyes and you're there
- 64. For a successful music career
- 65. For birthdays
- 66. For Christmas
- 67. The auto-panner is g cat
- 68. Michelangelo would have used one
- 69. It gives me new keyboard sounds for free
- 70. It's so musical sounding
- 71 It's as wet as you can get
- 72. Use as much as you like
- 73. The resonators are great for guitar
- 74 The ring modulators are great for sci-fi
- 75. The sampling is a cool convenience
- 76 You can automate parts of your mix
- 77. It's a necessity
- 78. The specs are great
- 79. It makes my ears feel better
- 80. It makes A & R departments hear better
- 81. I'd be lost without it
- 82. I don't need it at the beach...but I might
- 83. It's the first thing ON in my studio
- 84. It's the last thing OFF
- 85. Hove Alesis
- 86. Alesis loves me
- 87. It does 24 million instructions per second
- 88. It does -10
- 89. It does +4
- 90. It handles +20 peaks
- 91. The 32 character backlit LCD display
- 92. It only weighs 4.5 pounds
- 93. You can program advance with a footswitch

ALESIS

CIRCLE 04 ON FREE INFO CARD

- 94. The graphic eq
- 95. The parametric eq
- 96. Programmable panning

100. it's the all-time classic reverb

- 97. The flanger
- 98. The phase shifter 99. It looks great in a rack



Discover The Secret Of The Pyramids.

We've discovered a new acoustical foam that outperforms any we've ever seen (or ever sold).

ALPHA Acoustical Foam.

The secret is in its unique pyramid design.

- Superior Performance
- Uniform Pattern
- UPS Shippable

Call us, We'll gladly send you a brochure and a free sample.

Call 1-800-782-5742. Acoustical Solutions, Inc. License Alpha Audio

CIRCLE OI ON FREE INFO CARD

DISC MAKERS

Audio Manufacturing for the Independent Musician

500 CD'S AND 500 CASSETTES

FOR ONLY

two-color inserts

complete graphic design, film, and printing

Dolby HX Pro duplication, BASF Chrome Tape

clear shell and box

poly wrapped

ready to sell in stores and at gigs

Call Today For Our Free, Full Color Catalog

> 1328 N. 4th St., Phila., PA 19122, TEL: 215-232-4140 FAX: 215-236-7763





COPY CAT

I have a Tascam DA-30 DAT. It sounds great, the price is right, and it's been a great addition to my studio. I'm very disappointed, though, that as a pro deck, it includes the Serial Copy Management System (SCMS). Is there anything I can do about this, short of buying an outboard box?

> Eva Trautler Seattle, WA

If you're willing to void the warran-A ty, and have the necessary technical competence, you're covered. Unplug the DA-30, remove the top cover, and locate the wire jumper labeled W-402 (looking at the DA-30 from the front, this jumper is toward the left front of the large, green circuit board to the right of the transport). Snip the jumper with a small diagonal cutter, replace the top, and you're done.

The Tascam User Guide, issue #8, describes this modification in more detail on page 22. For back issue information, contact Tascam at 7733 Telegraph Rd., Montebello, CA 90640 or call (213) 726-0303, ext. 634.

> Craig Anderton West Coast Editor EO

MIDI THRU AND THRU

Q I'm making a MIDI thru box but can't find a good source for PC-900 opto-isolators. Jameco has them, but requires a \$30 minimum order. Is there an equivalent, more common part?

Also, I'd like to see the standard interface MIDI circuit described in the original MIDI 1.0 specification; where can I find this?

> I.D. Brown Afton, TN

The HP 6N138 opto-isolator can substitute for the PC-900. The 6N138 is available for \$1.45 from Digi-Kev; tel: 800-344-4539. They have no minimum order, but there is a \$5 handling fee for orders under \$25.

The IMA (International MIDI Association; tel: 310-649-6434) sells the detailed MIDI 1.0 spec (\$25 to members, \$40 for nonmembers) as well as supplements to the spec (Standard MIDI Files, MIDI Machine Control and General MIDI; prices range from \$7 to \$25). The original MIDI 1.0 spec is also reprinted in MIDI For Musicians (AMSCO Publications, a division of Music Sales).

> Craig Anderton West Coast Editor EO

NAMM-SKULL

In the article "Apocalypse NAMM X Revisited" [EQ, April 1992, p. 26], you mentioned Marion Systems in the "Annual NAMMINATIONS" sidebar. Could I get the location of this company, which offers a SCSI hard-disk interface for the AKAI MPC-60 and MPC60-11?

> Floyd Adams III Compusult Percussion Copenhagen, Denmark

Marion Systems is run by synthe-A sizer pioneer Tom Oberheim, and is located at 1078 Carol Lane, Suite 200, Lafayette, CA 94549; tel: 510-283-4631, fax: 510-283-6882. The SCSI hard-disk interface is compatible with both versions of the MPC60 and sells for about \$295.

> **Craig Anderton** West Coast Editor EO

GRABLE'S CABLE STABLE

I have been tying my cables in knots for years. Is there a better way to bundle them? I have tried a few of the cable-tying systems that I've seen adver-

> This is where your questions get answered. Send your query with your name and address to: EQ Editorial Offices. 939 Port Washington Blvd., Port Washington, NY 11050 Fax: 516-767-1745

World Radio History

tised and found each of them to be either a pain or too expensive to use on the large number of mic cables I now own. How can I tie up all these cables without taking out a loan?

> Bob Grable Wilkes-Barre, PA

A There are many ways to bind cables for storage between gigs. The problems of fastening mic or adapter cables for storage in the road box or on the cable peg require a solution that will create a neat bundle, bind quickly and snugly, stay with the cable without getting in the way, be reusable, and, lastly, be cheap to buy.

I also have tried many of the systems around and I didn't find any that solved all of these problems until I discovered spiral wrap. This stuff is a flat, fairly thick PVC that has been formed into a spiral tube (such as Alpha SW-13, which is available from Alpha Wire Corporation, Elizabeth, NJ; tel: 201-925-8000). It comes in a number of sizes and is intended to be used over the length of a set of cables to create a neat harness that remains flexible. Instead of using the stuff in long sections, I cut the 1/2-inch-diameter spiral wrap into two-inch pieces. They will then wrap two or three times around a coiled 25-foot mic cable. The longer the cable is, the longer the piece of spiral wrap for tying it up. Small diameter cables and short adaptor cables can be tied with smaller diameter spiral wrap tubing.

If you like to store cables in a box instead of on a peg, then it might be better to wrap the coiled mic cables across the coil, ending up with figure eights, as these are easier to separate than coils once the box has jiggled around in a truck for a few hundred miles.

When using a cable, the spiral wrap can be twisted back onto one end of the cable and you will then have it at hand when you go to pack up! You can also use spiral wrap to create instant mic snakes by binding a group of mic cables together, such as from the drum kit to the stage box for a three week tour. Label each cable connector, and the ends will come out to the right place every night. At the end of the tour, the mic snake breaks out to ordinary mic cables.

Once you are used to wrapping these onto your cables, it's as fast as gaff tape and over the course of a few months, considerably cheaper — all without leaving goop on those cables that haven't been used in six months. It will also avoid the sharp bends in your cables produced by tying them in knots and it will not catch on everything as velcro fasteners can. Plus it can go into your pocket, someone else's pocket or on the floor during set up like many plastic cable ties.

You can buy spiral wrap at most electronic wholesalers by the foot. Buy a few feet in a few different sizes and try it out. I prefer the black version, but it also comes in translucent white. It must be cheap to use because I haven't seen anyone with his or her sound company's logo printed on it, yet.

> Wade McGregor FQ





Westlake Audio makes it easy to hear every nuance of sound in your program background or foreground, dialog, music and effects.

Hear it all where you need it—right up front. Clear. Loud. Well-defined.

Audition a pair at your nearest pro audio dealer or call us.

> Westlake Manufacturing Group 2696 Lavery Court Unit 18 Newbury Park, Californ a 91320 18051 499 3686 • FAX (213) 498 2571

acoustic design to down beat...



MARK OF THE ADAT

Can the ADAT be synched to Mark of the Unicorn's Performer (or viceversa) using ADAT's internal SMPTE without recording a SMPTE track? Also, the ADAT has -10 inputs but the +4 inputs are replaced by a 50-plus pin connector. What's the connector for and how would I connect my console to the +4 inputs?

Michael Hammar Heart Music Productions Hsin Chu, Taiwan

To synchronize Performer to the internal SMPTE of the ADAT, you will also need Alesis' BRC remote control as this is the only place to get SMPTE in and out of the system. You can take the SMPTE in and out of the BRC to your MIDI Time Piece (or whatever SMPTE converter you are using to slave your computer).

The ADAT uses a straightforward multipin connector for its +4 I/O. To connect to the +4 I/O of the ADAT, you could build a snake with a 50-pin connector on one side running to a patchbay or mixing board with separate connectors for each channel. Alesis uses this

multipin to save space on the back panel and to keep the cost of production down. Daniel Rose

Mark of the Unicorn, Inc.

GETTING ON THE OT

I've heard a lot about the QSound process. How does it work? Peron Nilton Rarez Belo Horizonte, Brazil

QSound is a proprietary virtual A sound localization process for the production of stereo program material, e.g., audio-for-video games, film soundtracks or recorded music. QSound makes it possible for the listener to perceive discrete virtual sounds within a mix as being outside the normal stereo field.

At its core, QSound is a process that takes mono inputs and, from them, produces stereo signals encoded in such a way that the listener will perceive the selected placements (virtual images), when on axis, as being unrelated to the playback speakers. This enlarges the sound field to a 180° arc in front of the listener. As a bonus, encoding produces

enhanced separation and clarity for the off-axis listener as well. No special decoding equipment is required, and the encoded signal may be broadcast or reproduced in any stereo format.

Encoding involves manipulating amplitude, phase, and delay as functions of frequency. To achieve the required changes over very narrow frequency bands requires high-speed DSP technology.

The QSystem is the professional real-time QSound processor. It consists essentially of A/D and D/A converters linked to the QSound processing hardware, with an integral host computer providing the user interface.

Our new system features built-in QSound panning controlled by a remote joystick and internal SMPTE/MTC automation, internal software autopanners or external MIDI control. QSystem II provides eight channels of OSound processing and AES/EBU digital I/O at 48 kHz.

For further information contact: QSound Ltd., 7551 Sunset Blvd., #201, Los Angeles, CA 90046; tel: 213-876-6137.

> Scott Willing, CET Field Support Supervisor, OSound

DV777755HH Tommy Lee * Alan White * Jim Keltner

Dvzzzshh! Boowjhsh! Krzxsschh! That's about as close as we'll get to describing these sounds in print.

Q Up Arts now offers you the most explosive drum libraries ever presented—and you've gotta' hear them to believe them.

Not the same old same old, these libraries deliver the cleanest, most precision-engineered sounds, direct from the artists and engineers who make them happen.

Blxghwzhww!

However you spell it, Q Up Arts has it. To learn more, call us today.

Q UP ARTS PROUDLY PRESENTS **HEAVY HITTERS GREATEST HITS™** SAMPLE LIBRARY

Featuring three of the most respected drummers in the industry : The distinctively solid crisp sounds of renowned studio drummer Jim Keltner. The explosive sounds of Motley Crüe's Tommy Lee. The uniquely powerful sounds of Alan White of Yes. Digitally produced by Reek Havok for Sounds Amazing. All sounds recorded in the best studios in the world: A&M Studios, Power Station, Ocean Way Studios, & Music Grinders. Available in CD-Audio format now only \$149.00. CD-ROMs coming soon!

Q Up Arts also exclusively CARTIES OPTICAL MEDIA INTERNATIONAL'S entire catalog of CD-ROMS including the Sonic Images Library and the Master Studio Collection in the following formats: Akai S1000/S1100 E-Mu Systems EIII, EIIIXP, Emax II, Digidesign Sample Cell, Ensoniq 16 Plus.



рн: 408.688.9524 FAX: 408.662.8172 Q UP ARTS: PO BOX 1078 APTOS, CA 95001-1078

CIRCLE 51 ON FREE INFO CARD

THEY DO

MORE.

THEY DO IT

BETTER.

THEY DO IT

FOR LESS.

A simple message that is not often easy to back up. But Behringer's European reputation for excellence in signal processing gives these words new meaning.

Consider for a moment our latest interactive processors, conceived and engineered to deliver transparent performance for studio and live applications.

COMPOSER The Composer functions as a studio grade, automatic and manual controlled stereo Compressor, Expander, Gate and Peak Limiter. The automatic mode offers program-dependent attack/release times for inaudible processing,

Unique Interactive Knee Control circuitry combines the musicality of soft knee compression with comprehensive hard knee control. Special Interactive Gain Control Peak Limiter circuitry enables simultaneous Clipper and Program limiter functions while serving as a *zero* attack, distortion-free absolute gain threshold.

INTELLIGATE The Intelligate is an Expander/Gate/ Ducker with Ultra Transient Response circuitry for instantaneous attack. Proprietary high performance Class A VCAs provide exceptional audio quality. Precise key filters permit frequency-sensitive keying.

Like the Composer, the Intelligate includes Interactive Ratio Control Expander circuitry to eliminate "chatter" on or around threshold. Both units feature servo-balanced Neutrik 1/4" and XLR inputs and outputs, precise metering and a 5 year warranty.

Behringer. When you listen, the message becomes clear.





CIRCLE 17 ON FREE INFO CARD

THE SECOND ANNUA

The show is over, but far from done. Our editors and contributors tirelessly walked the floor in search of the most impressive new equipment upon which to bestow our coveted Blue Ribbons. We may have had a little fun with the award names, but each of these products proved to represent the future direction of the professional audio industry.

They have not been arranged in any particular order, although before we begin we would like to give the city of San Francisco an honorary "Great Place To Hold An AES Show" Award because its musical history and many scenic locales provided a very stimulating atmosphere.

THE UNPLUGGED AWARD TO THE RANE MAP 33 MIDI/PRO-GRAMMABLE INSTRUMENT PROCESSOR



The MAP 33 MIDI/Programmable Acoustic Instrument Processor from Rane accurately reproduces acoustic instrument timbres onstage with flexibility, power, and minimum feedback. The MAP 33 utilizes a multipickup approach (usually mic and piezo) with separate programmable 7-band equalization, and with notch filtering, band limiting, and level controls for each pickup. The MAP 33 is extensively outfitted with additional functions, such as separate Vocal Mic input with its own programmable EQ and level; a separate Monitor output with dedicated 7-band EQ and individual mix of vocal and pickups (all programmable); balanced stereo master outs with noise gates; stereo line inputs for tape or drum machine; several insert loops; foot pedal inputs; and much more. Retail price for the MAP 33 is \$1995. For more information, contact Rane Corporation, 10802-47th Avenue West, Mukilteo, WA 98275-5098. Tel: 206-355-6000. Circle EQ free lit. #101.



THE PAVAROTTI AWARD

DSP4000 ULTRA-HARMONIZER

Eventide's new DSP4000 Ultra-Harmonizer® fea-



tures AES/EBU digital I/O for all-digital signal manipulation, and a unique modular software architecture. You can build your own algorithms by linking modular effects "building blocks." The DSP4000 will generate up to eight simultaneous voices of pitch shifting plus simultaneous stereo reverb. A unique Effects Segue feature provides smooth transitions between presets. For more details, contact Eventide, Inc., One Alsan Way, Little Ferry, NJ 07643. Tel: 201-641-1200. Circle EQ free lit. #102.

THE NEW CONTENDER AWARD TO ATARI'S FALCONO30 COMPUTER SYSTEM

The Atari Falcon030 computer is the first personal integrated media (PIM) system. Features include: Atari's Graphic Environment Manager (GEM); standard-equipped MIDI input/output ports; 16-bit stereo sound at sample rates up to 50 kHz; and external video sync for high quality genlocking. You can record on two stereo tracks and add graphic equalization and audio special effects such as surround sound, harmonizing, reverb, and echo. Prices start at \$799 with 1 MB of RAM, and units are scheduled to ship the first of the year. For more information, contact Atari Computer Corporation, 1196 Borregas Avenue, Sunnyvale, CA 94089-1302. Tel: 408-745-2000. Circle EQ free lit. #103.



THE UNITED NATIONS AWARD TO THE ROLAND SRC-2 DUAL SAMPLE RATE CONVERTER

Roland Pro Audio/Video Group introduced the Dual Sample Rate Converter. The SRC-2 provides a variety of digital conversion and mixing features. It enables mixing two stereo digital signals at different sample rates into one stereo digital output at any selectable sample rate, with balance and overall level control. Features include AES3 (XLR), IEC 958 (coaxial), and EIAJ CP-340 (optical) inputs and outputs. The device allows for any incoming sample rate between 30 kHz and 52 kHz. Selectable output rates of 48.048 kHz, 47.952 kHz, 44.1 kHz, 44.056 kHz, and 32 kHz are provided. Sync sources include House Sync 30, House Sync 29.97, House Sync 25, Word Clock, Internal Clock, and Input A. In addition there is full 24-bit AES/EBU implementation for optimum real-time sample rate conversion. The two digital inputs and one digital output, each selectable between AES/EBU Professional Format Type I (XLR) and IEC 958 Consumer Format Type II (coaxial or optical), allow you to mix two different sample rates together. The SRC-2 will automatically deemphasize any incoming signal that has been preemphasized. In addition, there is metering for L and R channels and a digital "CLIP" indicator for inputs and outputs with two output modes, "Real Time" or "Peak Hold," with manual reset. The SRC-2 retails for \$2595. For the full story, contact Roland Pro Audio Video, 7200 Dominion Circle, Los Angeles, CA 90040-3696. Tel: 213-685-5141. Circle EQ free lit. #104.

THE 8 IS ENOUGH AWARD TO THE TASCAM EIGHT-TRACK DIGITAL RECORDER

Tascam's DA-88 is an 8-track digital recorder that records on Hi8 tape. It has the capability of recording up to 100 minutes on a standard 120 tape at either 44.41 kHz or 48 kHz, with a variable pitch of 6 percent up or down. Up to 16 DA-88 units can be locked together by connecting a simple 15-pin D-sub connector between units. Word sync I/O utilizes two BNC connectors and Digital I/O uses a 25-pin D-sub connector for easy copying of tracks from one machine to another. Options include the SY-88 Synchronization Board for SMPTE syncing as either master or slave without requiring the striping of an additional track, a remote controller, and an expanded, remote meter-bridge. To get on track, contact Tascam, 7733 Tele-graph Road, Montebello, CA 90640. Tel: 213-726-0303. Circle EQ free lit. #105.





THE ADD IT TO MY TAB AWARD TO FOSTEX'S DCM-100 DIGITALLY CONTROLLED MIXER AND MIXTABTM CONTROLLER

The MIXTAB controller and DCM-100 mixer from Fostex have been designed as two discrete units which fully interact to form a complete MIDI-controllable mixing system. The DCM-100 is a 1U rack-mountable mixer that is entirely controlled via MIDI messages. The MIXTAB is the human interface to the DCM-100. The MIXTAB can store up to 100 settings in a "snapshot" memory — all of which can be accessed by a simple MIDI Program Change Message. For the full story, contact Fostex, 15431 Blackburn Ave., Norwalk, CA 90650. Tel: 310-921-1112. Circle EQ free lit. #106.

THE SWISS ARMY DAT AWARD TO JVC'S XD-P1PRO DAT MACHINE

JVC's XD-P1PRO is the world's smallest DAT recorder. The main unit is about the size of a DAT cassette box and features JVC's servo driven "Compressed M" loading mechanism. Its modular design allows it to be used with either a snap-on A to D converter or digital output microphone module. Recordings can be made SCMS free through digital or analog sources at any sample frequency from mic, line or digital port. Flexible mini jack splitters are provided for unbalanced analog and digital inputs and outputs. For more information, contact JVC Company of America, Customer Relations Dept., 107 Little Falls Rd., Fairfield, NJ 07006. Tel: 800-252-5722. Circle EQ free lit. #107.



THE SECOND ANNUAL EQ BLUE RIBBON AWAI



THE TIME FOR A CHANGE AWARD TO THE ZOOM 9200 ADVANCED REVERBERATION PROCESSOR

The new ZOOM 9200 Advanced Reverberation Processor has a flexible architecture that is quickly and easily configurable to a variety of applications. The 9200 features 31 effect templates, with deep parameter adjustments that allow for virtually infinite effects possibilities. Up to 297 programs are available at any given time. Twelve quick control buttons and a large detented edit dial enable fast editing. The front panel's 99 program RAM/ROM card port makes archive and retrieval easy. The +4, XLR back panel is designed for professional interface. For more information, contact ZOOM Corporation, 385 Oyster Point Boulevard, Suite #7, South San Francisco, CA 94080. Tel: 415-873-5885. circle EQ free lit. #108.



THE ACE IN THE HOLE AWARD TO THE ALESIS BRC MASTER REMOTE CONTROL FOR THE ADAT

The BRC Master Remote Control for the Alesis ADAT 8-track digital recorder is an important peripheral because it delivers ADAT's full potential — digital multitrack recording, remote control, and editing. The BRC gives you the required comprehensive remote control of the tape transport and editing capabilities. The BRC provides precise access to the digital data stream of each track in an ADAT system. When connected, the BRC automatically becomes the master, assigns an ID number to each slave ADAT, checks if all units have a tape installed, checks if the tapes are formatted, and automatically locates all tape slaves to the position of the first ADAT. For the full story, contact Alesis Studio Electronics, 3630 Holdredge Avenue, Los Angeles, CA 90016. Tel: 310-558-4530. Circle EQ free lit. #109.





THE GARY BURGOFF AWARD

Anatek's RADAR (Random Access Digital Audio Recorder) is a standalone device that can record and play back a full 24 discrete tracks of digital audio directly to its onboard hard disk drives. No external computer is needed and it has been designed to work much like the standard tape recorder. For further information, contact Creation Technologies, 400 Brooksbank Ave., North Vancouver, B.C., Canada V7J-1G9. Tel: 800-665-4175. Circle EQ free lit. #110.



THE BEST SERIES SINCE STAR TREK AWARD TO CREST AUDIO'S CENTURY SERIES CONSOLES



The Crest Consoles division of Crest Audio has been launched with the release of the Century Series models. These consoles are offered in frame sizes from 16 to 52 inputs. Among the features offered by the Century Series are: an independent Mono bus; eight Auxiliary Mix busses; selectable Pre- or Post-Fader on the front panel; four bands of sweepable EQ; a sophisticated Scene Mute system with four presets controlling channel and EFX inputs; and two Matrix Mix outputs standard. Prices range from \$5000 to about \$30,000. For the full story, contact Crest Audio, 100 Eisenhower Drive, Paramus, NJ 07652. Tel: 201-909-8700. Circle EQ free lit. #111.

CALL 1-800-937-7171, EXT. 300 FOR THE DEALER NEAREST YOU. CIRCLE 74 ON FREE INFO CARD

World Radio History

THE BETTER MOUSETRAP AWARD



Sony's Pro Audio Products group introduced the K-1203 Super Bit Mapping Processor, which creates CDs with improved sound clarity. The K-1203 employs digital processing techniques, enabling mastering houses to use the capabilities of 16-bit recording generated from 20-bit source

TECHNOLOGIES INC.

signals. Standard CD-quality audio is quantized at 16 bits, Super Bit Mapping requires a higher number, such as 20 bits. The finished product consists of only the most critical 16 bits. Contact Sony, Corporate Communications Dept, Sony Drive, Park Ridge, NJ 07656. Circle EQ free lit. #112.

THE SECOND ANNUAL EQ BLUE RIBBON AWAI

THE GREAT COMMUNICATOR AWARD TO BEC TECHNOLOGIES' AUDIOPLEX

BEC Technologies' new AudioPlex Pro-Line/Pro-Series transmits "digitally clean" audio over both copper wire and fiber optic cables, yielding better than CD audio quality. Up to 16 digital channels can be placed on each twisted pair of copper cables, and fiber allows 64 channels of bidirectional communications. AudioPlex's self-contained modules are also specifically designed to guarantee fail-safe communications. BEC's Fault-Tolerant Redundant Communications provides triple redundancy with three separate outputs. Dynamic range weighs in at 90 dB, frequency response is 0-22.5 kHz, signal-to-noise ratio is 87 dB. For more information, contact BEC Technologies, Inc., 1101 N. Northlake Way, Suite 3, Seattle, WA 98103-0891. Tel: 206-632-2431. Circle EQ free lit. #113.

THE WAIT TILL A PRO GETS ONE AWARD TO THE SONY MINIDISC

Sony's MiniDisc (MD) is the first fully recordable and erasable optical disc format available to the consumer market. Up to 74 minutes of digital audio can be stored



on a 2 1/2-inch optical disc. It is portable and provides shock resistance through use of a 10-second memory buffer system. The model line-up consists of the MZ-1 recorder/player (\$750), the MZ-2P player (\$550), and the MDX-U1 car stereo player (\$980). Models offer auto track-numbering, title functions, blank disc space search, and random programmability. For the full story, contact Sony, 1 Sony Drive, Park Ridge, NJ 07656-8003. Circle EQ free lit. #114.



THE "WE'RE ALL CONNECTED" AWARD TO THE AT&T DISQ DIGITAL MIXER

AT&T's DISQ Digital Mixer is a fully digital system that works in conjunction with analog boards and it is software-based. Developed jointly by AT&T Bell Laboratories, Russ Hamm, and George Massenburg Labs, the DISQ Digital Mixer uses an AT&T parallel processor with supercomputing capabilities to provide digital signal processing for audio postproduction and mixing. The DISQ Digital Mixer is connected to a Neve VR or SSL G series console via an Ethernet cable. The DISQ system utilizes the console's control surface but totally bypasses its analog circuitry. All mixing processes are performed in the digital domain with the AT&T processor. Components of the system consist of the Parallel Processor, the AT&T Digital Audio Interface, and the GML Series 2000 Automation Environment. Prices start at \$325,000 for a 32-channel system. For more information, contact DISQ, Inc., 1790 Broadway, New York, NY 10019. Tel: 212-765-3417. Circle EQ free lit. #115.





THE GOLDEN GATE AWARD TO THE DBX SUPERGATE

The new 172 SuperGate® from dbx uses sophisticated VCFs (voltagecontrolled filters) to produce 24 dB/octave slopes that are twice as

selective as those found on most conventional gates. The VCFs are also faster to set up because their quasi-parametric design allows you to zoom in on exactly the part of the instrument or part of the attack that you want to catch. In addition to the filter section, each channel includes: independent Threshold, Ratio, Attack. Hold, Release, and Depth controls; Gate, Expand, and Duck buttons and status LEDs; 13-segment Gain Reduction meter with Open and Hold status LEDs; bypass; and more. For more information, contact AKG Acoustics, Inc., 1525 Alvarado Street, San Leandro, CA 94577. Tel: 510-351-3500. Circle EQ free lit. #116.

THE RICH LITTLE AWARD

TO THE SYMETRIX 601 DIGITAL VOICE PROCESSOR

The Symetrix 601 Digital Voice Processor acts as a "bridge" from the analog to the digital domain, the 601

accepts mic or line level analog signals, converts to digital (18 bits), and then performs 24 bit digital domain signal processing at a rate of over 50 million instructions per second (MIPS). For more information, contact Symetrix, 4211 24th Avenue West, Seattle, WA 98199. 800-288-8855. Circle EQ free lit. #117.



The Fishman Acoustic BLENDER. The power to be heard.

Bring your instrument to life! Capture its breath! Sense its pulse! Hear the voice you always knew was there!

Separate bass and treble controls for mic and pickup
Input gain and level controls for both mic and pickup
Master volume control
Three separate effects loops
Phase control for mic and pickup
Low level phantom power for the mic
Balanced XLR and 1/4" outs
Mute button
9-volt battery or AC power supply
Headphone output

The Fishman Acoustic BLENDER." Powerful! Affordable! And essential for all acoustic stringed instruments!



Benefit from the advantages of being able to combine and control both your pickup and an on-board microphone. The Fishman Acoustic BLENDER[™] has many professional features found only in much more

expensive equipment. Think of the possibilities you'll have at your fingertips! The Fishman Acoustic BLENDER." For guitar, dobro, banjo, mandolin, violin, cello, harp, dulcimer, autoharp and piano.



TRANSDUCERS Pickups & Accessories for Acoustic Instruments Five Green Street, Woburn, MA 01801 617/938-8850 Call us or write for free product literature

Beauty and The Beat

LOCATION: Studio A in Alan Menken's home in Upper Westchester County, New York. Studio B is under construction.

MAIN MEN: Alan Menken, composer; Rick Kunis, engineer, studio manager

CREDITS: Beauty and the Beast, The Little Mermaid, Home Alone II, Rocky V, Little Shop of Horrors, Aladdin, Newsies, ABC documentary Lincoln, several musical theater works that include Mr. Rosewater, Real Life Funnies, and The Apprenticeship of Duddy Kravitz, four Academy Awards, four Golden Globe Awards, two Grammy Awards, eight BMI Career Achievement Awards, the New York Drama Critics Award, the Drama Desk Award, the Outer Critics Circle Award and the London Evening Standard Award.

CONSOLE: Soundtrac Quartz 32-in, 24-out (discrete) with automation

MONITORS: KRK 9000s, small pair of Realistics

AMPS: Crown Microtech 1200, Hafler 600 for headphones

RECORDERS: Otari MX 5050 8-track, Tascam 238 Syncaset, Tascam DA-30 R-DAT, Sony DAT 75ES, Tascam 122MKII cassette, Onkyo Integra cassette, Sony TCW7ES hi-speed duping deck

KEYBOARDS: Kurzweil K1000, Yamaha DX7, Yamaha Disklavier

MICS: Neumann U87, AKG 414

VIDE0: 27-inch JVC S-VHS monitor, Hitachi 13-inch monitor, Mitsubishi BV-1000 S-VHS VCR, Sony MDP LaserDisc player

OUTBOARD GEAR: Kurzweil 1200 Pro, dbx NR, E-mu Proteus 1 & 2, Macintosh 950 Quadra, Mac IIci, Mac SE30, Powerbook 170, GCC BL II S laser printer, Yamaha SPX 900, Yamaha Rev 7, Roland AV controller, Roland A80, Korg O1RW, Korg M3R, Roland R8M drum machine, Yamaha FBO 1, KMX 16x15 patchbay, Mark of the Unicorn MIDI Time Piece, 20-inch Ikegami 24-bit computer monitor, Sony headphones, Furman Power Conditioners

EQUIPMENT NOTES: Menken states: "I do a lot of MIDI mixing direct to DAT. I tend to favor the real sounds. I want a true piano sound and that's why I love the Kurzweil. The Proteuses are my favorite modules. But I still get a lot of use out of the DX7. I use MOTU Performer software — I think it's the best choice. The MIDI Time Piece offers me a tremendous amount of flexibility. I can just call up instruments from storage — both channels and cables — so I have much more capacity than I could ever possibly use in terms of instruments. Using the MIDI Time Piece, I sync music to video through the Mitsubishi VCR. I use Composer notation software, but I will be switching over to Mosaic."

The Yamaha Disklavier is a central tool for Menken's composing and arranging. Says Kunis, "Alan does a lot of composing on it. It allows him to hear his work with objectivity and to instantly review his work. The sophisticated electronics are entirely internal so it fits in the house as well as the studio. Alan often works at it down in the sun room where he can overlook the terrace, the pool, watch the sunset."

PRODUCTION NOTES: Menken says, "My personal studio is used essentially for composing, scoring and demoing of my own work. It's my personal workspace. For larger group sessions I go out-of-house to larger studios in New York and L.A. I often use my old friend David Lucas's Warehouse in Manhattan. He has a company called 4/4 and it's a very friendly comfortable studio. We do our masters at BMG, but it's unfortunate that they plan to close it and sell it off for office space. In L.A., we frequently use the Todd-AO and Sony Studios."

ADVICE: "MIDI and computers are an incredible boon to the process of scoring," explains Menken. "They give you the ability to hear the music you've written against the picture and modify it as you wish. The technologies allow you to demonstrate the music to directors and get their comments, which saves a lot of artistic train wrecks. And being able to go to a studio with a click to a tenth of a beat per minute that you know will work exactly as you need it to work is a great advantage. But on the other hand, don't ever forget that even with all this equipment and all the technological wizardry, it's all about drama. It's all about human feelings and you've got to make sure that you don't get caught up in clicks or in the sound of a synthesizer or some great effect. You've got to remember that it has to rest in the music and I think my career's been an example of how much people still respond to basic musical values." EQ



11 11 11 11 11 11 11

N'S SCORES IEY IN HIS DIO

朝川

 MI INSIDER

The Computer Quagmire



Sure the election is over. There's still time to pick your favorite (computer) platform.

ow is not necessarily a good time to buy a computer. Not that it ever is — something faster, better, and cheaper will always come along six months later. But these days it might be particularly wise to hang on to the machine you have a little while longer.

The reason is that the second wave of personal computers is looking pretty old. The first wave of 8-bit machines (Apple II, Commodore-64, Atari 800, original IBM PC) peaked in the mid-1980s, at which time the second wave hit. The Mac, Atari ST, and Commodore Amiga brought a graphic interface, faster processors, and 16-bit operation to the party, while IBMmachine "clones" were developed that became the budget alternative for textoriented applications. Today's machines have more memory, greater speed, and better peripheral chips, but are still based on 10-year-old technology. In high-tech, that's an eternity.

This also locks companies into the past. Windows, the *hype du jour* for PCs, remains a slave to DOS with its 8-character file names and other oddities. Mac users, disenchanted with System 7's sluggishness (as well as with the cost of upgrading software to 7), are sticking with System 6 for as long as they can. The Amiga's oncebreathtaking graphics and sound capabilities are commonplace, and Atari has milked the 1040 ST for all it's worth. Flagship machines of yesteryear, such as the Amiga 1000 and Mac Plus, are becoming boat anchors.

Nonetheless, if you own a studio in 1992, you pretty much need a computer. So what are your options?

IBM: WINDOWS WASHER

IBM always seems the last to "get it," from the PCjr., to the Clamshell portable, to just now discovering MIDI (and getting upstaged by a Microsoft operating system). Yet the installed base is huge; developers love doing projections: "If we sell to just 5 percent of the PCs out there, we'll all be multimillionaires!". Music software continues to take a back seat to business, but the situation is improving rapidly.

Still, Windows — a slow, clunky memory hog — does not turn a PC into a Mac; it gives your machine a face-lift without really addressing some of the PC's problems. Clone prices are unbeatable, though. A \$2000 clone will scream, but a \$2000 Mac will only talk in a loud voice.

Windows NT will supposedly break with DOS completely and unchain the PC from the 1980s, but it's not here yet. You might want to wait long enough to see if Mac maintains its operating system superiority vis-a-vis Windows NT. Meanwhile, cheap clones do make cost-effective "appliances" to run particular pieces of software.

MACINTOSH: '90s YUPPIE COMPUTER?

Based on the phenomenal sales of the Mac Powerbook, the answer is yes. Apple's commitment to music has traditionally been soft; MIDI Manager is moribund, and no one at Apple seems to care. They could at least adopt Opcode's OMS, but I suspect the "not invented here" syndrome applies. Some say the Beatles's lawsuit prevented Apple from committing to music, while others blame Apple's internal priorities.

Apple has made some good moves lately: dropping prices, producing more models, and coming up with a fun laptop. But gone are the days of the closed architecture, onesize-fits-all philosophy, so the famed "ease of use" isn't what it once was. Also, some products have incredibly short lives — the Classic's paint seemed hardly dry before the Classic II came out. Still, the Mac family has the smoothest operating system, the price of admission is getting lower, and there's plenty of music software.

So why wait? Simple. There's a gnawing feeling that when the dust settles, Apple may not be able to hold its own against the clone tidal wave any more than IBM. Windows NT could pole-vault the clones over both companies unless Apple continues to increase its market share.

ATARI: DAVID VS. GOLIATH

Atari targets its products mainly at musicians, and with good reason: music is the one area where Atari has a significant market share, especially in Europe.

Atari is also in the trough between technology waves, but at least we have a preview of the next wave: the Falcon030. It's impressive — a 68030-based machine with onboard 56001 DSP chip, MIDI ports, and graphics coprocessor, all for \$1295 with a 65 MB internal hard drive and 4 MB of RAM. A \$200 program allows for stereo hard-disk recording. Sounds great, right? Well...

The reason for waiting is to see

Atari is also in the trough between technology waves, but at least we have a preview of the next wave: the Falcon030. what kind of software develops for the machine. Many developers had written off Atari, and though the Falcon030 is turning a lot of heads, it's hard to predict whether the amount of software will reach "critical mass." Atari's best hope is that, despite what Apple says, the Mac is no "computer for the rest of us," and perhaps the Falcon030 will grab that title. But only time will tell if this is the machine that puts Atari on the map big-time.

COMMODORE: HOLES IN THEIR SHOES

Several years ago, a film-scoring friend wanted to buy a computer and commissioned an expert to evaluate available computer hardware. His order of preference: Amiga, Atari, Mac, and, lastly, the PC.

Yet Commodore's marketing has been so seemingly dismal that, had it not been for NewTek's Video Toaster and the Amiga's inherent videotapefriendliness, it might have been a footnote in the history of computing. It's a shame, really. The Amiga was so ahead of its time when it came out that few people got the point, so it never really got the support needed to stay ahead of its time.

There aren't many choices in music programs; however, video enthusiasts swear by the Amiga, and when they need to do audio or music, the Amiga certainly has enough software to get by. So why wait on buying an Amiga if you're into video? Simple. The Falcon030 might beat Amiga at its own game, or Commodore's CDTV might take off (but don't bank on it) and grab market share by sneaking in as a home computer disguised as an interactive-CD machine, which could recharge the entire Commodore line. But that's all up in the air right now.

DECISIONS, DECISIONS

So where does all this leave us?

•You're probably going to need more than one computer platform (sorry) because no one computer does it all, and you often need to be compatible with the rest of the world (Murphy's Law states that your most important client will always use a different machine). • The handwriting is on the wall: one platform will almost surely have to be a Windows-compatible machine.

• Invest in emulation and bridge software so you can get files from one platform to another.

• If you need a computer now, get something that will make enough

money to pay for itself inside of two years, when the scene will have (hopefully) settled down a bit. Then, commit some big bucks to a third-wave computer system that will last for another decade...and get ready to go through the whole merry-go-round all over again.

16 channels of 3-band parametric equalization in 3 rack spaces



You've never seen this much EQ in this little space before 16 More importantly, you've never heard a more musical equalizer 16 Speck optimized the response for electronic musicians, so 16 sounds fall into place quickly and easily 16 No more repatching 16 It's the answer to your limited-eq console 16 It's available in 16 mono channels, 16 stereo channels, or an 8/8 combination Get to your dealer today for a live audition... and while you're listening, ask him to tell you the price. That'll be music to your ears, too.

The model EQ16

Some of us need more equalization than others

For more information: 619-723-4281 925 Main Street Fallbrook, CA 92028



CIRCLE 59 ON FREE INFO CARD

Digital Tape Wars: 8mm vs. S-VHS

Format wars have characterized the consumer video business since its inception. First it was Beta versus VHS. Then it was VHS versus 8mm. And now that consumer video technology has made low-cost digital multitrack recorders a reality, these format wars have found their way into the studio scene as well.

There's a new videotape format war brewing and this time low-cost digital multitracks are the battleground BY RICK SCHWARTZ

With units already on the market from Akai, Alesis, and Yamaha, and Tascam's much talked about digital 8track just around the corner, users will be faced with some tough choices. But that's not all. Fostex recently announced its support for the ADAT digital multitrack format through a licensing agreement with Alesis. The various machines have hardware characteristics all their own. But beyond these differences is the question of which tape format is best for your needs. The tapes differ in terms of cost, storage capacity, size, and more

From a technical standpoint, recordists need to consider much more than just the tape format a digital multitrack uses. From the point of view of standardization, however, there is quite a bit at stake. In a world where compatibility is a major issue, where studio operators need to exchange tapes easily with other musicians, producers, and commercial facilities, the tape format that builds the larger marketshare may turn out to be the long-term winner.

For the time being, however, a look at the tape formats themselves offers a comparative view of how several leading manufacturers see the future of low-cost digital multitracking. Let the chips fall where they may.

BOLD CLAIMS

One of the most popular debates involves tape width. Are the newer narrow tape formats like 8mm less reliable than the descendants of VHS?

According to Fostex's vice president of marketing David Oren, "S-VHS has more tape passing over the head drum each second than 8mm. Almost four times the tape surface per second at standard speed and over eleven times as much at ADAT speed each



Ilustration by Normal Rockhard

Like the dinosaur, conventional mixers have bodies that are bigger than their brains. But now there's a leaner, smarter, faster species of mixer for both stage and studio: the DCM-100 Digitally Controlled Mixer and MIXTABTM Controller.

THE DINOSAUR Mixer Is About To Become Extinct.

RUSTEN

11534

World Radio History

Say goodbye to the dinosaur's big footprint. The DCM-100 packs a full-featured, 16 channel (8 channel stereo) mixer in a single rack space. Cascade three of them for up to 48 inputs.

Say goodbye to high cost. With MIDI parameter automation you don't pay for zillions of knobs and switches, most of which you set and forget anyway.

And say goodbye to mixing with a mouse. The MIXTAB human interface provides 8 channels of hardware knobs, switches, and faders for level, muting, EQ, panning, and more. It works and feels like a traditional mixer. Use it to program your sequencer quickly and painlessly, change selected parameters in *real time* to add spontaneity to your mixes, or to control the DCM-100.

There's more: 100 "snapshots" you can recall with MIDI program changes. LED null indicators for easy level matching before you record new moves. Exceptional sonic integrity. And a mercifully short learning curve.

The DCM-100/MIXTAB combination is a whole new kind of animal – after all, dinosaurs belong in museums. To find out more about the latest evolution in mixers, see your favorite Fostex dealer.

FOSTEX





second." To this, Alesis vice president Allen Wald adds: "It can be proven that a single piece of dust can interrupt the linear path of the tape past a rotary recording head and cause irreparable errors. ADAT is liberally over-designed to account for this with strict regard for large bit size and wide track width."

Paul Young, product planner with Tascam, says that 8mm is simply a more modern medium. "As technology has progressed, the physical storage medium has gotten smaller and denser. That goes for videotape, RAM, digital storage, anything. The fact remains that our track width with 8mm tape is actually one and a half times the track width of RDAT. And, in terms of reliability, RDAT is a proven format and the accepted professional standard. In terms of error correction, we have over-engineered our product to prevent any problems. Besides, the advanced nature of the 8mm shell makes an extra provision for dust prevention, covering the tape on both sides during storage. With VHS, the tape is exposed in the shell."

Clearly there is a difference of opinion here, but who is right?

MAGNETIC ROOTS

"Of course there are differences in tape formats," says Mike Johns, market development manager at 3M. "Hi8 is based on a format that came out ten years later than the original VHS

tape." The S-VHS format, launched in 1987, represented an evolutionary step in the development of the VHS system. The use of better tape resulted in improved picture resolution." According to the Ampex Guide to Media & Formats, when the 8mm format was introduced in 1985 it made use of an embedded tracking servo rather than the longitudinal control track, thus promising perfect tracking and eliminating the need for a tracking control.

Although the benefit of using narrower stock is a smaller cassette size, it does not come without a price. According to Steve Smith, marketing man-

ager of audio tape at Ampex, "When you reduce the size of the media, the system needs to make up for it by using more sophisticated electronics and higher output tape." This is why 8mm tape systems use metal tape formulations to store more data per cubic or square inch of tape (see chart 1). Smith adds that Ampex manufactures a military airborne recorder that uses S-VHS T-120 cassettes and offers 10.4 Gigabyte storage capacity.

VHS is, in fact, almost twice as thick as Hi8 in a tape with the same play time (10.9 vs. 18.1 microns). However, according to Bruce Fellows, senior technical service engineer at 3M, the transport design, not just the tape thickness, ultimately determines how well the tape format will perform for digital audio recording.

Still, the basic specifications of the tape formulations are quite revealing. Chart 2 compares the coercivity (the amount of magnetic force necessary to return a tape that has had signal recorded on it to its original nonmagnetized state) of all four consumer videotape formats. As you can see. Hi8 has almost twice the coercivity of S-VHS tape. According to Smith, the higher the coercivity, the harder a tape is to record or erase. It also means there's a higher retentivity (the measurement of the residual magnetization after a tape is returned to its non-magnetized state), which is another one of the attractions of the



VOU buy any brand of 16-channel mixer, consider why the CR-1604's become the rack-mount mixer of choice for top pros (request our literature packet and we'll include an astonishing list of touring and session players, studio producers, engineers, song writers, sound track composers and late night talk show bands who rely on Mackie).

-1 design features. One reason we specialize in mixing boards is so we can be maniacally meticulous about details otners overlook, Some details help make the CR-1604 rugged and roadable. Others contribute to the mixer's excellent sound quality. And some, like our unique qain architechture and mix amp topology aren't really details at all, but rather represent a fundamentally better way to mix music.

Fengineering means phenomenal sound. Compare our specs and features with anyone. But remember that the bottom line is what your ears tell you. Put a CR-1604 through a hands-on work-out and you'll understand why several major-label CD releases have been recorded and mixed solely on Mackie mixers.

7

OUTPUT-O-RAMA: Man steres outputs with ano an over the state the laboration or unimations input. In fact, an outpute have sufficient lead to where any spat. All outestermut are hay prote the

TRIM matches any size of from matching levels to -KarSer even pro-to +4454 pro-action

7 SENDS and above and stores of the former of a second at the seco STEREO AUX RETURNS Headroom and welse and emograph alt and with di-1.

> MUSICALLY-USEFUL EQ. Har Hundar Los

our 2014, 25 Hat our 2014, 25 Hat 21 Hat El point e 21 Hit de Lin not emp south sweet but in morner calleder framer conduced cutlicard -11.ra. r

COLD-ROLLED STEEL CASE BUS 3&4

ASSIGN

STEREO IN-PLACE

SOLO TRATERINA atternal percentation for all and crannels 3 com.

UNITY PLUS IN CARE AND A conter set or a manufact repopul houroun, provae 20 als more auto allow willy to reduce constant frie in adjustment as input in all change

BETTER-THAN-DIGITAL. and when the design of the second sec works constitutes as ours do. Congress which and you'll opport all the families of the second state of the second state of the SK-ES/-45 species. And arry arrantizer so vesity from millers that merely make a

SEALED ROTARY

STUDIO-GRADE MIL

Carland and a

PREAMPS. Day De Chicos

PREAMING, any dig score and the many score as a score a second and the memory transition and the score and only our skip preaming defer - CBalline E.N. 10000 Tell yet can base a lightly may must without move for a part for any state to be preaming and part for any score and address of the score and the present and address of part for a score and the present and address of the score and the present and address of the score and the present and address of the score address of the score and address of the score address

P ODSTRACT HOARD DOARD PREMITING

POTENTIOMETERS termination from shell, encode legiste à mi CO-attal PED antihout the use of dans y sourt envir prenche inschame, al part a (a moor securitie), protrient le many, consertement misse designed. Plan, car CONCILIES. ENERGY-ABSORBING KNOB DESIGN COMPANY OF THE STATE from pase and main circuit board, areally aling possible durings.

LARGE INTERNAL POWER

SUPPLY cintuity, his subject the systems and A sophisticates enhancemente such as one CRO incorporation expansion & internal fundamentation with automation research the expry tota only emittance recent and fault parts & her large "but enter of of their hurr-inducing magnetic felide, for



GNARLY CIRCUIT BOARDS.

or the or the prierule. h r's at all or at a AUDIO MARK, PPOULINI HOSE platest therefore boards and mark multiply me . U to used in the series 1 in in truth rates a -1

MIX AMPS WITH TWICE THE HEADROOM.

Stanals from all report or net a received on reading of the standard ייד איז אייר א you ניד אינער און גייי ז trainess. Our sails structure solves the problem for author better sound. Cram the GK 100-F with 10 hot speaks until it still have more here froom them other me ers norming 3 inputs. No worder over is the invert of choice among top persons signified the Chester Teampion (Com-encil, Santon Constant (Wilting resiston) S FIRE FOR CHUMESTICN

ULTRA-DENSE CIRCUIT TOPOLOGY. Le re publies d'en a l'a . Lorre : Mitt passie : in al

and input and output built and a CK:NSCH is compatible with a much when ralles of desized.

12230 WOODINVILLE DR WOODINVILLE, WA 98072 FAX 206/487-4337 COMPUSERVE: GO MACKIE MACKIE DESIGNS INC 1-800-258-6883

> CIRCLE 45 OH FREE INFO CARD World Radio History

TECHNIQUES DIGITAL

metal tape formats (see chart 3).

Fellows explains that higher coercivity tapes are better for applications requiring greater information densities. He says, "S-VHS does not use a metal particle tape. It uses a cobalt-modified iron pigment. It is a high-coercivity tape but Hi8 is a much higher one."

Alesis makes its system's transport move three times faster than a normal VHS transport (see chart 4) to compensate for these differences. What's more, Wald adds that "it is a faulty assumption to consider that coercivity and retentivity are significant factors in choice of media for digital multitrack recording. They are not. In fact,

had we designed the ADAT to make maximum use of the coercivity and retentivity of S-VHS, and had we pushed the format to deliver maxi-



"GOING TO CMA WAS "I was a disc jockey, but I knew that I needed strong professional THE BEST THING training to get my media career going." "I chose the Center for the Media Arts and learned THAT EVER Recording A to Z. I've used this training on over 75 albums, including 11 gold **HAPPENED TO ME**." and platinum efforts. And I worked on Color Me Badd's double-platinum - Mervyn Jordan, Songwriter, producer, engineer, remixer, 'I Wanna Sex You Up?" edit specialist and Owner

of Funkmaster Productions At CMA, you spend most of your time applying what you learn with actual hands-on experience on the equipment you'll use as a professional.

If you're trying to get your career going in Recording, Television Production, Broadcast Announcing, Graphic Technology or CAREER GOING NOW. Photography, nothing beats the hands-on experience of Center for the Media Arts. CALL CMA TODAY!



World Radio History



mum bit density, we could have had longer recording time or even more channels of audio. But reliability and interchangeability would have suffered."

To which 8mm supporters like Tascam's Young responds: "Coercivity and retentivity are significant factors. Digital audio recording requires high density and, as Fellows says, higher coercivity tapes are better for applications requiring greater information densities. Retentivity is also important for the longevity of the tape."

"The narrow width of 8mm systems shouldn't worry you," he adds, "considering the fact that these drives are used extensively for computer data backup. And, with computer data, if one bit of information is lost you could have a serious problem. In practice, unrecoverable errors are rare. All systems included in this survey use a Reed Solomon type of error correction, which minimizes the effect that dropouts have on tape performance. The fact that Exabyte has built one of the most successful data storage companies based on 8mm-cassette technology should indicate how reliable these products really are."

To which Wald counters,"With tape data storage devices it is not required to re-record a specific portion of data without affecting data around it. In multitrack recording, this is exactly what must happen when punching in and out. Data backup tapes simply have to record a large block of data without retaining any data on the tape. When recording multitrack audio, however, a single channel of audio must be able to record or re-record in a very specific location on the tape without the danger of altering or erasing any other channel of audio. In addition, data backup tapes have the luxury of rewinding and re-reading a section of data as many times as required in order to recover the information. Because multitrack recording is a real-

continued on page 80

You Can Take It With You.

Full of Features and Convenience.

The Shure VP88 MS Stereo Condenser Microphone marks a new age of convenience in the art of live and recorded stereo.

While reducing the setup complications normally associated with producing quality stereo, the VP88 offers a variety of stereo signals – from a single source – that never degrade if combined into mono.

The portable and rugged VP88 utilizes two condenser cartridges a forward-facing cardioid (Mid) and a perpendicular bidirectional (Side) - with a switchable built-in matrix to convert Mid and Side signals into discrete left and right outputs. The VP88 lets you choose from three selectable degrees of stereo spread, or provides a classic MS signal for complete control of width and ambience.

Popper and a series of the ser

Excels In Every Environment.

Wherever your successes take you – from the studio to the stage to the great outdoors – you will never want to be without your VP88. The VP88 faithfully reproduces the spaciousness and subtleties of orchestras and ensembles. It handles the most demanding percussion and the softest strings. It adds ambience to live house tapes and expands recording capability.

The VP88 makes it easier than ever to produce stereo DAT recordings so realistic you'll swear you are there.

With both phantom and battery power capability, there's no place your VP88 can't go.

And the VP88 is a remarkable value. No other stereo microphone gives you so much versatility and quality at such an affordable price. Audition Shure's VP88 MS Stereo Condenser Microphone.

Audition Shure's VP88 MS Stereo Condenser Microphone. Hear how much life can be brought to your productions through two ears rather than one.

For the VP88 dealer nearest you call 1-800-25-SHURE.



The Sound Of The Professionals® ... Wor dwide.

CIRCLI: 56 ON FREE INFO CARD World Radio History Erotica's coproducer brings us into the inner sanctum of Madonna's latest labor of lust

THE MADON

remember when Madonna and I first started working together on *Erotica*. We were listening in my home studio to one of the first songs and I turned to her and said, "It's great, but it's no 'Vogue'." She told me that not every song could be "Vogue" not every cut could emerge as the top-selling record of all time. She was right, but I pressed my case anyway: "I guess I'm always trying to out-top myself," I told her, "the next thing should be bigger than the last." Madonna just turned and looked me straight in the eye. It had been a long time since I'd been star-struck by her, but she was glowing differently now. "Shep," she began, "no matter how fierce something is, you can't ever do the same thing twice." She sat down to record the final vocals on "Erotica" and looked out onto the terrace and into the New York City night.

"Ever," she repeated.

By Shep P

World Radio History



WORK THIS THING JULY -AUGUST 1991

DOES ANYBODY

I wanted to start writing again. The last project I had worked on with Madonna was *The Immaculate Collection* but that was just a month and a half of working with that QSound stuff. I knew I could do something great after "Vogue" and "Rescue Me," so I just started putting tracks together with my assistant, Tony Shimkin. I wanted to have a few songs for Madonna to listen to when I went out to Chicago, where she was filming *A League Of Their Own*. I had no idea that she was planning to do an album at that time, but then again, neither did she.

KNOW HOW FO

Editor's Note: The Polaroids in this article were taken Juring the Erotica sessions and carry Madonna's personal handwritten annotations.

I arrived in Chicago on July 8th and gave Madonna a cassette. I told her to give it a listen and tell me what she thought. She said she'd listen to it in the car, in the trailer, wherever she could. A few days later, I heard back from her. Madonna liked all the songs — three out of three. I decided to work on a few more.

Usually, when I sit down to write, it isn't as if I have a specific person in mind for any one song. By the time I get to a certain place in the music, it begins to mold itself an identity and I think, "Hey, this per-

son would like that." At the time, Cathy Dennis, Taylor Dayne, or Madonna were the primary inspirations for a variety of songs.

OCTOBER - NOVEMBER 1991

Madonna returned to New York and we began to work on demos in my apartment. It's cool working at home. It's convenient, cozy and there's no studio time ticking by. Plus, if you wake up in the middle of the night and you have an idea, you just go upstairs, turn on the equipment, and go. Our schedule was kind of sporadic in the beginning. I'd work with her for a week and then she'd go off to work with Steve Meisel on her book (*Sex*) for two weeks. Occasionally, Madonna would meet with Andre Betts, her coproducer on "Justify My Love." While she was away, I would spend time coming up with other tracks or work on Cathy Dennis and Taylor Dayne material. At this point, I wasn't working on any remixes — just writing.

"Deeper and Deeper," "Erotica," "Rain," "Bad Girl," and "Thief of Hearts" made up the first batch of songs we worked on together. I did the music and she wrote the words. Sometimes I'd give her some ideas lyrically and she'd go: "Oh, that's good," or: "That sucks." I remember when I gave her some ideas lyrically for "Vogue" and she said, very curtly, "That's what I do." Essentially, her songs are her stories. They're the things she wants to say.

I did everything upstairs in my home studio: keyboards, bass lines, and vocals. Depending on the mood I was in, I chose from an Oberheim OB8, Korg M3, or a Roland D-50. On the sampling side, the Akai S1000 was our prime workhorse. We used it to

ix Maestro

IF REMIXING WAS THE music industry's latest monster, then Shep Pettibone was its Dr. Frankenstein. After tailoring his craft as a part-time DJ and record-store owner, Pettibone plunged headfirst into the remix craze in the 1980s by re-creating hardhitting versions of songs for a variety of dance queens including Loleatta Holloway, Gloria Gaynor, and Alisha. Suddenly, Pettibone found himself in the eye of the pop hurricane, where New Wave acts such as Pet Shop Boys, Thompson Twins, New Order, and Erasure were spinning out an enormous web of dance songs ripe for the remix. Not surprisingly, clubs immediately took to this new form of dance music, but radio still handled the genre with kid gloves. Essentially, they needed an excuse to play New Wave. Pettibone gave it to them.

By inventing powerful dance versions of "Chains of Love" (Erasure), "Something About You" (Level 42), and "West End Girls" (Pet Shop Boys), Pettibone transformed the way radio listeners heard music. It seemed as if anything, no matter how staid or played, could be given an enormous backbeat and transformed into a dance hit. As a result, remixes began to influence the way songs were marketed, and, more important, how they were created.

By the time the 1980s came to a close, Janet Jackson, Paula Abdul, MC Hammer, Lionel Richie, Prince, and Cyndi Lauper had entrusted their work to Pettibone. In the meantime, an icon-in-waiting by the name of Madonna approached him to remix her song "Get Into The Groove," thus initiating a relationship that would span several hit records, including "True Blue," "Where's The Party," and "Causing A Commotion."

"By the time I worked on 'Like A Prayer' and 'Express Yourself,'" says Pettibone, "it looked as if Madonna liked the remixed versions better than the ones that were on the album. That was great, but producing was still at the top of my wishlist."

Pettibone got his wish when he was asked to write and produce a B-side single for Madonna's *I'm Breathless* album. That B-side eventually became "Vogue," which, in turn, became the largest-selling single of 1990. Pettibone had arrived on the production scene, in a big way.

With a bonafide smash under his belt, Pettibone went back to work for today's hottest entertainers, while juggling a client list of post-modern disco divas that included Cathy Dennis, Taylor Dayne, Mariah Carey, and of course, Madonna. Of all his compatriots, however, it was Madonna who most encouraged Pettibone to continue writing, spurring on what would eventually evolve into Madonna's current tour de force, *Erotica.* With a ground swell of multimedia publicity, the album entered Billboard's Top 200 at #2 in its first week.

-Jon Varman
The ENSONIQ ASR-10 brings the features you've always wanted to a sampling workstation.

Announcing the latest *ENSONIQ* boom-the new ASR-10 Advanced Sampling Recorder. With features like stereo sampling, dynamic effects processing, time compression/ expansion, an integrated sequencer and 31-note polyphony, ENSONIQ brings innovation to the world of samplingagain. To start with, the

Io start with, the ASR-10 supports up to 16 megabytes of sample memory– eight times the memory of our previous sampler, The extensive sound library for the ASR-10 contains over 1000 sounds- with new collections being released every month.

and uses SIMMs for

industry-standard easy expansion.

For on-board effects, the 24-bit digital effects processor has 50 algorithms, many derived from the DP/4 Parallel Effects Processor. Input sampling is easily routed through the effects processor. You can re-sample in stereo through the effects, and

even process external signals while you play

live. And for drum loops and other time-based samples, use the ASR-10's time compression/ expansion facilities to change the length f a recording with

of a recording with-

CIRCLE 33 ON FREE INFO CARD

EXPLOSIVE FEATURES!

out changing the pitch.

The ASR-10's quad-density (HD) disk drive increases storage capacity and speeds load times. Add the optional SP-3 SCSI interface, and you can connect up to seven hard drives and CD-ROMs for easy access to an extensive library of sounds.

Together with a flexible, integrated 16-track sequencer, Poly-Keytm Pressure keyboard and Performance Presets that provide instant recall of sound combinations, you have one powerful workstation.

Harness the power of an ASR-10 for your music. Play one today at your Authorized ENSONIQ Dealer. For the one nearest you, call 1-800-553-5151.



THE TECHNOLOGY THAT PERFORMS

					-														
Ple	ase send	me mor	e inform	nation	on the	□ ENS	ONIQ A	SR-10	Also,	please s	end me	information	on	I ENSONIQ	Synthesizer	Workstations	DP/4 Pa	rallel Effec	ts Processor

Name	Address			
City	State Zip	Phone (1	and the second second
Mail to: ENSONIQ, Department Q-32, 155	Great Valley Parkway, P.O. Box 3023, Maivern, PA 19355	-	-	1 ::: : # -: : :
			THE OWNER OF THE OWNER	DP/4 Parallel Effects Processor
-1.32 Voice Music Production Synthesizer	SO-2 32 Voice Music Production Synthesizer	KS-32 Weighted Action MH	Of Studio	SO B PLUS 32 Voice Synthesizer Modular



1600 Broadway, NY, NY 10019 (212) 315-1116/Fax: (212) 315-0825 sample snake charms for "Words" and Kool & the Gang horns for "Erotica."

When it came time to record demos, we laid down a track of SMPTE on the last track of my 8-track Tascam 388 Studio 8 reel-to-reel, which has dbx. Usually we'd put the track down on tracks 1 and 2 in a stereo mix, and then bring Madonna's vocals in on 3 through 7 — a lead, a double lead, the harmonies, and the background parts. Ninety-eight percent of the time, the vocals recorded in my apartment were the keeper vocals, the ones you hear on the album.

It took about two or three days to write a song from beginning to end. Still, sometimes even after they were done we'd want to change the flow of the song and ask the song a few questions: Where should the chorus hit? Should it be a double chorus? Sometimes Madonna would call me in the middle of the night and say "Shep, I think the chorus should go like this," or "I hate this verse, fix the bass line." "Deeper and Deeper" was one of those songs she always had a problem with. The middle of the song wasn't working. We tried different bridges and changes, but nothing worked. In the end, Madonna wanted the middle of the song to have a flamenco guitar strumming big-time. I didn't like the idea of taking a Philly house song and putting "La Isla Bonita" in the middle of it. But that's what she wanted, so that's what she got.

DECEMBER 1991

"I hate them." That's what she said to me when we listened to the first bunch of songs we'd recorded. I thought it sounded great because some of the songs had a New York house sound and some of them had an L.A. vibe. "If I had wanted the album to sound like that, I'd have worked with Patrick Leonard in L.A.," she told me. I got the point pretty fast.

Madonna wanted *Erotica* to have a raw edge to it, as if it were recorded in an alley at 123rd Street in Harlem. She didn't want some light, glossy production to permeate her sound. I got back into my usual style of mixing, which is pretty bass-oriented, analog, hit-you-over-thehead kind of stuff. When you're recording songs for Madonna, the attitude is: Either make a song work, or it's not going to be on the album. That's that.

Typically, Madonna would get over to my place by one in

the afternoon and we'd work until eight or nine at night. Improvising vocals took one or two passes and by the time the third pass came around, she'd get on the mic and say "Let's go." Madonna has an incredible mind; she locks the melody into her head and memorizes the words

read the words off the paper when she's singing. The only problems were during sequencing, when we had to do something on the Mac that would take some time. Two minutes into it, Madonna would ask us: "What are you guys doing that's taking so long!" — and this was just after the first few minutes. We'd tell her to go downstairs and make some popcorn or phone calls so that we could put the song together and she'd do that for about five minutes before screaming: "Come on, guys, I'm getting bored!" I had to keep things moving as fast as possible because it's one of my jobs to keep Madonna from losing interest in what she's doing.

immediately. She doesn't even have to

mR.+ MRS. PETTY BVTT



FEEL BETTER?

For \$795, we have trouble convincing some of you that this board has outstanding sound quality. There's an easy solution, though. 'We can raise the price. Or, you can go to your nearest authorized dealer and listen for yourself.



The CardD is an AT-compatible bus board that gives you:

- Real time direct-to-disk stereo recording and playback
- True professional- quality 16-bit audio
- Stereo analog inputs and outputs
- 32kHz, 44.1 kHz, and 48 kHz sampling rates



The EdDitor is a fully non-destructive stereo waveform editing program with powerful features:

- Extremely fast cut and paste operations
- Overlays
- Customized fades and crossfades
- Multiple undo's
- Very easy to learn
- On-line help is always available



Add the I/O CardD to give you the S/PDIF (IEC) digital interface, allowing direct digital transfer to and from your DAT machine.



14505 21st Avenue North, Suite 202 Plymouth, Minnesota 55447 (612) 473-7626 Fax (612) 473-7915

CIRCLE 27 ON FREE INFO CARD World Radio History

WHICH RED KNOB? As far as the music went, it was getting a

little melancholy by that point. It definitely wasn't up-and-happy music. Maybe I inspired songs like "In This Life" and "Bad Girl" because they were written in a minor key. But Madonna's stories were getting a lot more serious and intense and she was definitely driving the creative direction of the songs into deeply personal territory.

JANUARY - FEBRUARY 1992

I spent Christmas on vacation in Jamaica and when I got back on Janu-



After it was done we thought: "How about if we get a male Jamaican rapper in here to do some stuff on the record?" We found this guy, Jamaiki, who runs a Jamaican record store uptown. He was this big guy with a real deep-ass voice. When we were trying to explain the song to him, he just looked at us and said, "Do you have any rum, man?" By the time Jamaiki was laying down the tracks in my studio, he was dancing around, swigging rum and spilling it everywhere. We ended up not using the track because it sounded too rough for the song, but it was a very fun day - completely different.

By this point, people had begun to realize that Madonna was recording in my penthouse. All day her fans



Now, the ultimate demo tape is a CD. An optical format that's digital in quality and virtually indestructible. In this competitive industry it pays to be absolutely

sure the music you promote will not only sound great but will guarantee an audience. For more information, call Frank Kramer at 1-800-TDK-TAPE (Ext: 200).



would wait outside, even though it was freezing, just to catch a glimpse of her or take a picture. One particular day, when I walked her down to her car, the lobby was filled with building residents getting the mail, hanging out at the front desk, sitting on benches. It was weird because usually the place is empty. After I walked her outside and ran across the street to get the day's newspaper, I came back to find nobody there. People were coming downstairs to the lobby just to get a look at her, even if it was out of the corners of their eyes.

MARCH 1992

Now I knew we were doing an album. We had fifteen songs demoed and she liked them all. The last song we did was for the movie, A League Of Their Own. Madonna just started singing a melody over and over again into the Shure SM57 microphone while the Mac with Vision was playing strings, organ, piano, and a basic rim-shot loop. It sounded really timeless, very nostalgic. I spent all night filling in the verses and the song became "This Used To Be My Playground."

The day after "Playground" was finished, Madonna went to Oregon to work on her next film, Body Of Evidence, with Willem Defoe. This gave me some time to wrap up work on some songs with Cathy Dennis and Taylor Dayne at Soundworks Studios in New York. The workload had grown quite intense since the beginning of the year and it showed no signs of letting up. Thanks to my manager, lane Brinton, we were able to coordinate all the ongoing projects without a hitch.

MAY 1992

1 met Madonna at Oceanway Studios in Los Angeles to complete the



A CLOCKWORK MADONNA 31/2 MONTH

Miking Madonna

WHEN MIKING a singing superstar, do you pull out all the stops? Or do you just stick to the basics, record in your apartment studio, and employ the help of a beat-up old mic?

If you chose option two, then you're striking the same studio pose as Shep Pettibone and Madonna. Simply put, Madonna would sit in the middle of Pettibone's cozy apartment studio and sing into an old Shure SM57, while two Yamaha NS-10 nearfield monitors blasted out the goods over Manhattan's Third Avenue.

"Madonna didn't use headphones," says Pettibone. "We had these monitors pumping out the music, which definitely makes for more leakage. But who knew we'd be using the demo vocals in the final mix." When it came down to isolating vocals for remixed vocal dubs, Pettibone used Drawmer gates to eliminate drumbeats and sounds that would otherwise show up on a track because of leakage.

orchestra parts for "This Used To Be My Playground." We had to record a string arrangement — something I was excited about but had never done before. Madonna chose Jeremy Lubbock to do the arrangements because he had done such a good job with her I'm Breathless material and came highly recommended. Everything went fine until the point when the orchestra played their parts; we didn't like what we heard. Madonna and I had to change the whole arrangement, right there in the studio, with a full orchestra sitting there getting paid for taking up space — around \$15,000 for three hours, \$3,000 for every half hour over that. And of course, Lubbock was talking to two people who didn't know a C from a B natural. The pressure was on.

I can only sing the notes I hear at the moment, so that's what I did. Madonna and I stood there over my little Mac, singing the notes, and Lubbock would go, "Oh, that's a G; Oh,



OFINSANITY

When smooth melodies were called for, a Yamaha SPX 900 or Roland DEP-5 came into play, while spoken parts were recorded dry. For one track, "Bye Bye Baby," Madonna and Pettibone searched for an old 1940s feel, which would make the vocals sound as if they were coming out of an antique radio. They found their sound with a Pultec HLF filter.

Although the *Erotica* production team would eventually go hi-tech behind Soundworks' 48-channel SSL board, the mind-set was still mostly one of low-tech standards. One example came during the overdub sessions for "Bad Girl," when Madonna came up with some new lyrics for the song. "We had to record new vocals for the second verse," explains Pettibone, "and they had an SM57 handy. The only problem was that it was a brand new microphone and we had recorded the original vocals with the old SM57 back at my apartment. We knew the song wouldn't sound cohesive, and it didn't." Pettibone ran back to his studio and returned to Soundworks with the tried-and-true mic. "Sometimes," he adds, "older is better." —Jon Varman

that's a B" and that's how it got done. We completed the session in 2 hours and 58 minutes — two minutes away from another three grand.

The last day of recording fell on Memorial Day. Madonna wanted to do the lead vocals again, insisting that it would sound better. It did. I finished off some edits before going over to a party Madonna was throwing in her Hollywood mansion.

JUNE - JULY 1992

The schedule for recording at Soundworks in New York went something like this:

- •June 8 "Erotica"
- •June 9 "Words"; "Why's It So Hard"
- •June 10 "Why's It So Hard"; "Thief Of Hearts"
- •June 11 "Thief Of Hearts"; "Goodbye To Innocence"



CIRCLE 64 ON FREE INFO CARD

•June 15 — 8-track dumps w/no time-

•June 16 — "Deeper and Deeper"

And so on and so on...

We transferred everything we had on the Tascam 8-track onto 24track. I decided to produce the tracks 15 ips with Dolby SR because it has this warm bottom in the bass and I wanted to capture that for *Erotica*. Plus, I was listening to some of my old remixes, which were recorded at 15 ips, and was amazed at how much

more you could feel the music. Compact discs seem to move you one step away from the music, while



Ask about the KABA CREDIT PLAN. Discover how easy and profitable it can be to duplicate your own cassettes.

See us at AES in San Francisco, Oct 1-4, Booth 1320



records put you right in the mix. So I figured that if I overemphasized that LP feeling, it would rub off on the CD, which is the primary format manufactured for American audiences today. Strangely enough, our country can't get any LPs of *Erotica*, while the rest of the world can.

On July 7, we did the mixing for "Erotic," the ode to S & M that Madonna wanted to include in her book, Sex. She felt it should sound the same as "Erotica" (the song on the album), with just a bass line, her voice, and some sensuous Middle Eastern sounds. But by then I had seen the book and had come up with an interesting idea. "You have all these great stories in the book," I told her, "why don't you use them in the song?" I knew that Madonna was developing a 1930s dominatrix-look for Erotica, but I didn't realize how far she was willing to go until I saw Sex. It contained stories authored by her mysteriously dark alter ego, Dita.

Madonna took the book and walked out of the room and didn't come back until about half an hour later. Suddenly she was on the mic, speaking in this very dry voice. "My name is Dita," she said, "and I'll be your mistress tonight." I knew that the original "Erotica" would never be the same again, and it wasn't. The chorus and bridge were changed entirely and the whole psyche of the song became sexier, more to the point. It seemed as if Dita brought out the beast in her, actually serving as a vehicle for the dangerous territory she was traveling. Actually, it was the name Madonna used when she'd stay in hotels around the world. Not anymore.

When July 10 came, I felt my thirty-something years hit me full-force. It *continued on page 108*



Jack Renner and the ATM35 on

Bass,

Brass,

Jack Renner needs little introduction. As co-founder, Chairman, and Chief Engineer of Telarc International he has been nominated 11 times and won 5 Grammies just for classical engineering alone. Jack Renner and the Telarc team pioneered digital orchestral recording and the minimalist microphone technique that is now the standard for premiumquality classical CDs.

But when Jack recently decided to create a series of jazz CDs to complement the superb Telarc classical library, he knew - from his years of location recording - that he would need a new approach to solve the difficult acoustic and performance problems posed by live recording in jazz night clubs.

After an intensive search, he has found exactly the string bass sound he needs...with the Audio-Technica ATM35. This small cardioid condenser microphone can be clipped directly to the instrument. Its unique mount is readily positioned for best balance and consistent level all night long.

The cardioid capsule can also be easily replaced with an omni element when needed. Its low noise is impressive and the smooth extended response is both realistic and musical. Everyone loves the results: the musicians, the technicians, the record buyers, and Jack. But he also found the ATM35 has another role to play.

Model ATM35 Cardioid Condenser Microphone

When Jack Renner clips the ATM35 to a trumpet or trombone he finds it equally musical, even with very intense playing. He gets very low distortion and noise and consistent high quality no matter how much the musicians move around. Jack's coal of recording brass exactly as it is heard in live performance has been met simply and precisely.

and all that jazz!

He notes that while the microphone is actually guite close to the instrument it sounds farther away, with acoustic "air" that provides an amazingly natural perspective. And the ATM35 offers almost no restriction to the freedom of movement and expression that is so important to live jazz. Location photos: Frank Mice

World Radio History

You can hear the superb performance of the ATM35 for yourself. Just listen to this sampling of recent Telarc CDs that have gained critical acclaim: To Diz With Love (Live at the Blue Note) Dizzy Gillespie, trumpet, Peter Washington. bass

Lionel Hampton (Live at the Blue Note) Milt Hinton, bass Bobby Short (Late Night at the Cafe Carlyle) Beverly Peer, bass Manhatian Mambo sax, horns, vocals The Count Basie Orchestra (Live at El Morocco) brass solos...and more

releases in this live jazz series are due shortly.

Put the mighty little ATM35 to work today. Whether used for quality sound reinforcement or the most critical recording, this versatile microphone is outstanding. Just ask Jack Renner and Telarc! Write, phone, or fax for more details or see your Audio-Technica sound specialist today.



Available in the U.S. and Canada from Audio-Technica U.S., Inc. 1221 Commerce Drive, Stow, OH 44224 (216) 686-2600 • Fax (216) 686-0719

CIRCLE 14 ON FREE INFO CARD



The following chart is your easy-to-use shopping guide for new recording and sound gear that was introduced at last fall's AES in San Francisco. Scan the next few pages for gear you can use and fill out this issue's Free Literature Card to receive more information directly from the manufacturers.

BRAND	MODEL	TYPE	DESCRIPTION	PRICE	TEL	LIT #
BRAND	DX 4000	Digital system controller	Program- cross sys- eq: time align delays	S2,490		
Adamson Acoustic Design Corp.	UBI	Loudspeaker	Compact, 2 way tropezoidal loudspeaker	\$550	416-420-6279	121
Additisen Acoustic Design Corp.	AFM212	Active floor monitor	2 AW12 waafers; 50x40 deg. coverage Acoustic Woveguide	\$1825 (no HF)	410 420 0277	
	DD1000i	Disk-based digital audio editor	Stereo Tecorder/editor; removable MO disks	\$14,995		
Akoi	501	Digital sampler	16-bit, 8-vaice, multi-timbral sampler	\$995	817-336-5114	122
	Blue Line Series	Condenser mics	Moduler mic components	N/A		-
AKG Acoustics, Inc.	Tri Power Series	Dynomic mics	Rugged, high-output stage mics with clean, clear top end	N/A	510-351-3500	123
	BRC	Master remote cantrol	Controls up to 16 ADATs; record track selection	N/A		
Alesis	RMB	Remote meter bridge	Provides 32 channels of remote LED metering	N/A	310-558-4530	109
HIESIS	Al-1	Digital interface	Provides digital interface for interfacing ADAT	N/A	310 330 4330	107
Allow & Marsh	GL3	Mixing console	For live sound reinforcement; 6 selectable oux, busses	From \$3495	801-268-8400	124
Allen & Heath				\$495	405-324-5311	125
Alter Lonsing Corporation	Acousta QWIK	Speaker design saftware	For architectural spaces; equipped with room model templates	\$15,000	403-324-3311	125
Amber Electro Design	7000	Measuring system	Analog und digital audio generator/analyzer		514-333-8748	126
	5500	Prog measuring system	Front panel controllable; modular construction	\$6,900		
A 10 FW (T + C	Einstein Super E	Mixing console	A large frome/patchbay version of the AMEK Einstein	\$65,000	919 509 0799	197
AMEK/TAC	TAC SR6500	Monitor SR5600 companion	All modules are designed for foldback	\$49,500	818-508-9788	127
	9098	Mixing console	Designed by Rupert Neve	\$500,000	415 3/7 0457	100
Ampex Recording Media	489DM	Digital audio mastering tope	S-VHS tape for audio digital moster recording	N/A	415-367-2457	128
Anatek	RADAR	Multitrack digital recorder	24-track digital recorder in 4 space rackmount case		800-665-4175	110
Aphex Systems	104	Aural Exciter	Aurol Exciter with Big Bottom	\$349	818-767-2929	129
	Legacy Series	Mixing console	Modular discrete console	\$125-\$900,000		120
API Audia Products	"Six Pock" 500-86	Powered rack	Six-slot version of the API Lunchbox	\$849	703-455-8188	130
	525b	Signal processor	Compressor/limiter gote-de-esser	\$1,195		-
	AE-8B	Stage monitor	High-output floor monitor	\$2,640	707 770 0007	121
Apogee Sound, Inc.	CRQ-12	Porametric EQ	Porametric EQ with 2 outputs and 6 filters per channel	\$2,680	707-778-8887	131
	MPTS-1	Cinemo System	MPTS-1 is a five-channel cinema and postproduction system	\$18,000		
A.R.A.S. Marketing, Inc.	X-52	Digital audio processor card	1 x AES/EBU/SPDIF input; 1 x AES/EBU/SPDIF output	\$3,999	313-572-0500	132
	X-54	Digital audio processor card	1 x AES/KBU/SPDIF input; 2 x AES/EBU/SPDIF output	\$5,499	71/ 40/ 0700	100
ART	Phontom Series	Professional consoles	16-, 24-, and 32-input config: 4-bond EQ; 8 Aux	N/A	716-436-2720	133
Ashly Audio, Inc.	CFT-1800	Convection cooled amp	MOS-FET smplifier with 300 wott/chonnel	TBA	716-544-5191	134
ATM Fly-Wore	AMFS-DML1122-T	Hordware-flying orray	AMFS series loudspeaker flying system for EV DML1122	\$150	310-639-8282	135
	AMFS-DML1152-T	Hordware-flying orray	AMFS series loudspeaker flying system for EV DML1152	\$154		
ATR Service Co.	ATR-LB	Conversion kit	Upgrodes on Ampex ATR-104 to do 1" Type C Audio	\$6,995	415-574-1165	136
	CM-50	Auto locator	For Studer, Ampex, Toscom, Fostex and Otari recorders	\$1,090		
AudioControl Industrial	PCA-200	Signal processor	Subharmanic synthesizer with complete controls	\$299	206-775-8461	137
Audiomotion, Inc.	990M/913	Uptown outomation kit	Opto-isolation interface kit to handle 8 foders	\$200	508-443-8053	138
And the second second	990M/911	Control ponel	Control panel for 990 Uptown Automation System	\$200		
Audiomotion, Inc.	600	Uptown Automation	Screen-bosed moving foders	\$7,000	508-443-8053	138
Audio Precision	Fastrig	DSP Softwore	For audia quality measurements	N/A	503-627-0832	139
	Engineered Series	Microphones	Mini and sub-mini mics far pro sound	N/A		144
Audio-Technica	PRO 43R	Mini boundary microphone	Smoll and inconspicuous with a controlled unidirectional	N/A	216-686-2600	140
	AT808G	Talk-back microphone	Over 12dB of rejection from the rear	N/A		
	D Series	Microphones	Transformerless, dynamic mics for instrument miking	\$249-289		
Audix Corporation	Speakers	Powered speakers	25 watts/channel for mixdown and playback	\$429 pr	510-828-7529	141
	HRM-30	Speakers	Double 6 1/2" woofer and proprietary dome tweeter	\$899 pr		
Aurica Saund Labs	Analog technology	Patented circuitry	Allowing extension of bandwidth beyond 300 kHz	N/A	408-241-2600	142
	ELF-1	Dual channel integrator	Low frequency integrator	\$2,388		
Bog End Loudspeaker	S18E-C	Single 18" ELF toudspeaker	Super quick 18" transducer in a sealed, compact	\$598	708-382-4550	143
	TA12jr-O	Time Aligned® loudspeaker	Oak veneer version of two way, Time Aligned laudspeaker	\$690		
BBE Sound	862 Maximizer	Sound Enhancer	Dual channel unit with law contour and process controls	\$599	714-897-6766	144
	386	Acoustic instrument pre-amp	High impedance input; parametric EQ; notch filter	\$299		
BEC Technologies	AudioPlex AD16	Digital transmission	16 channel A/D transmitter	\$3,000	206-632-2431	113
BEC Technologies	AudioPlex DA16	Digital transmission	16 channel D/A receiver/repeater	\$3,000	206-632-2431	113
BEC Technologies	DatoPlex FBIt	Fiber optic transceiver module	Hondles up to 4 AD16s and 4 DA16s	\$1,800	206-632-2431	113

16 Channels, 4 Busses, 6 Sends, 8 Returns, 20 True Inserts. \$899



19"

Flexibility is the horsepower behind any professional recording mixer. You need all the

channels, sends, returns and inserts you can get your hands on in the heat of a mix. Check out these front and rear panel shots of the Alesis 1622 and you'll see professional features that make other mixers pale by comparison.

To suit your tracking needs you can set up for 2 bus/2 master or 4 bus operation. All 16 channels have 6 sends: 2 pre-fader for your choice of dual mono or stereo monitoring and 4 postfader for effects. Use all 6 sends for effects-heavy mixdowns or set up three separate monitor

mixes plus 4 effects for a live recording. And the 8 returns handle any combination of stereo or mono sources.

Every channel and all 4 busses feature professional TRS inserts for individual processors.

And they're true pre-fader inserts so you can change levels without disturbing critical effects

settings. Master outputs are switchable +4 or -10 to drive pro or semi-pro mixdown decks, and channels 1 through 8 feature XLR mic inputs and direct outs. There's even a headphone output for convenience.

All this flexibility wouldn't mean anything if the 1622 didn't sound great. Specs like "distortion plus noise: too low to measure"*, and the cleanest, quietest signal path we could possibly devise, truly put the 1622 in a class by itself. In fact, it's the absolute best mixer

under \$2000 for recording with our ADAT Digital Audio Recorder. Your ears will prove this.

Flexibility, superb audio, unbeatable price. Ultimate value. See your Alesis dealer for a demo and start mixing today.

*Keyboard Magazine July '90



Alesis Corporation 3630 Holdrege Avenue Los Angeles CA 90016

CIRCLE 03 ON FREE INFO CARD

BRAND	MODEL	ТУРЕ	DESCRIPTION	PRICE	TEL	LIT #
	MC834	Lorge diophrogm condenser mic	Cardiald vocal and instrument mic w/3 position roll-off	\$999		
beyerdynamic	DT150	Monitoring headphone	5=35 kHz response; ambient noise isolation; rugged	\$199	516-293-3200	145
	MCE87VS	Shotgun microphone	Condenser shotgun mic for video camero interface	\$299		
DOW C	SPA-2	Signal processing amplifier	Two-way amplifier; crossovers; delays, parametric EQ	\$2,199	310-973-8090	146
BGW Systems	U86	Computer	Rack mountable 386/486 PC computer	\$3,995	310-773-0070	140
P	APE Series	For 8&K 4003 and 4006 mics	Function as spatial and spectral EQs	N/A	519-745-1158	147
Bruel & Kjaer	4011	Cardioid microphone	Designed to deliver accurate pattern characteristics	N/A	317-743-1130	147
BSS	FCS-920	Slave unit	Up to 32 ch of a parametric EQ	N/A	510-351-3500	148
Carver Corporation	PDR-10	Compoct disk recorder	CDR with AES/EBU in/out seriol/parollel remote	\$8,500	206-775-1202	149
Carver Corporation	PMCX-1	Active crossover network	Stereo 2 way/mono 3 way-4 way onboard limiters	\$899	200-773-1202	147
Circuits Maximus	CMCI ITE-20	Wearoble monitors	In-the-ear wearable stage monitors	\$479	703-276-0125	150
Circuits Maximus	1000 Series	Wired stoge monitor system	Stage box and control beltpack for monitors	\$1,500	703-270-0123	130
	MX-816	Mixing outomotion	16 chonnels of rack mount autamation	\$780		
CM Automation	FX-100	Automation control console	100 millimeter automated faders	\$449	818-709-4020	151
	MIDI Mix	Software (Moc)	Automation control softwore for Macintosh	\$379		
	N-Series FB	Nearfield loudspeaker	3-way tropezoidol live/playback	\$1,340	1-	
Community Pro Sound Sys	R5 Jr	Reinforcement loudspeaker	LF/1-12"; MF/2-6 1/2"; LF/2-6 1/2" HF/1-1"	\$745	215-876-3400	152
	R\$660	Reinforcement loudspeaker	Compact 3-way trapezoidal live/ployback	\$1,863		
	РСНІ	Patch cord manager	Hand turned in hard wood	\$63	203-375-5577	153
Connectronics Corp	J822 MJ	Patchbay	44 point, normal, 1/2 normal or straight	\$207:14	203-3/3-33//	133
6	CSPA-1	Mic preomp	A bottery operated, two-channel mic preamp	from \$990	714-248-1361	154
Cooper Sound Systems	Aux module DS-1	For CS 106 + 1 audio mixer	Converts CS 106 + 1 into a 6 x 4 + aux mixer	\$895	714-240-1301	1.54
	Century GT	Front-of-house mixing console	16 to 52 inputs; 4-band sweepable EQ; two Matrix Mix out	\$5- \$30,000		
Crest Audio, Inc.	Century SP	Sound reinforcement consule	16 to 52 inputs; 8 Aux Mix buses; full function returns	\$5 - \$30,000	201-909-8700	111
	Century LM	Monitor mixing consoles	16 to 52 inputs; mono and stereo designs	\$5- \$30,000		
	MA-5000VZ	Power amplifier	5,000 watts of power; takes up only 3 rack spaces	\$3,295		
Crown International	Power-Tech 1	Power amplifier	Delivers over 300 watts per channel into 4 ohms	\$1,049	219-294-8000	155
	Power-Tech 2	Power amplifier	Delivers over 400 watts per channel into 4 ohms	\$1,299		
Could start and	C\$\$389	Anolog-to-digital converter	107 dB, stereo 18 bit NY A/D converter	\$42 in 100s		154
Crystol Semiconductor	C\$4303	Digital-to-onalog converter	107 dB, stereo 18 bit 👋 D/A converter	\$29 in 1000s	512-445-7222	156



Cause.

BRAND	MODEL	TYPE	DESCRIPTION	PRICE	TEL	LIT #
	CAD Equitek E-200	Multi-potterned microphone	3 patterns - Omm. Figure 8, Cordioid	K/A		
CTI Audio Inc.	MegaMix Series	Automation system	Transparent VCA circuit, easy user interface	N/A	800-762-9266	157
	172 SuperGate	Gate, expander, ducker	Full-featured professional gate	5869		
dbx Professional	760x	Mic preamplifier	2 channels; phantom power and polarity	\$349	510-351-3500	116
	120XP	Subharmonic synthesizer	Separate subwooler aut; built-in crossover; 2 control bands	\$299	510 551 5500	
	Forum "PA"	8-bus console	24 to 40 inputs	\$11,995/24in		
DDA	Forum "Motrix"	8 bus cansole with 8x8 matrix	24 to 40 inputs		s 516-249-3660	158
	Forum Composer	24-track recording console	24 to 40 inputs; 24 track monitoring	\$14,500/24in		130
Demeter Amplification	VT275HF	Tube power amplifier	All-tube, 75 watt-per-channel power amplifier	\$1,795	818-986-7103	159
	DTR-80PR	Portoble DAT	Digital I/O, mic input	\$900	010 700 7103	
Denon America, Inc.	DN-961FA	CD player	Reads Orange Book CDs	\$1,350	201-575-7810	160
Desper Products, Inc.	Spotiolizer	Real-time processor	24-channel 360 degree real-time virtual imaging system	\$8,450	818-986-9673	161
besper ribberis, inc	CDR-63	Compact disc	63-minute recordable compact disc; holds 540 MB of data	\$38	010-700-7073	
DIC Digital	CDR-74	Compact disc	74-minute recordable compact disc	540	201-244-9344	162
bie bighti	AR-4000	DAT drive cleaner	Non-abrasive, dry DAT drive cleaner	\$15	201-244-7344	102
	Expansion Chassis	Pro Tagis accessory	12-stat NuBus card chassis for Digidesign cards	\$2,295		
Digidesign	ProMoster 20	2-channel editing system	20-bit version of Sound Tools II	\$5,995	415-688-0600	163
bigiocalgi	DINR	Noise Reduction softwore	For PraMoster 20, Sound Tools II, and Pra Tools	\$995	473-088-0000	103
Digital Audio Labs	Ver. 3 of EdDitor	Softwore	New Windows 3.7 version of hard-disk editing software	S250	612-473-7626	164
Digital Expressions	SoftSplice	Digital audio editor	4 trocks; built-in DSP; SMPTE support	\$2,995	206-389-9895	165
orgital expressions	Digital Master EX	Digital audio workstation	4-channel recorder /editor	\$4,995	200-307-7073	105
Digital F/X	WaveFrame 401	Digital audio workstation	8-channel recorder, editor, and mixer	\$14,995	415-961-2800	166
	TSR-24	Digital reverb	Programmoble, hordware/software expandable reverb	5800		
DigiTech	DHP-55	Harmony pracessor	Dual DSP, 5-port oversampled harmony processor	5995	801-268-8400	167
Digitech	Vocolist H	Human vaice processor	Five-part human voice processar	5799	001*200-0400	167
Dołby Laboratories	740	Spectral processor	2-chonnet dynamic EQ		415 550 0270	1/0
Doremi Labs, Inc.	DAWNIE			\$1,760	415-558-0270	168
Doremi Labs, Inc. Dynocord	DRP15	Digital Audio Workstation	All-digital version of original DAWN 24-bit signal processing: 17 different basic effects	N/A	818-966-2454	169
Dynocord	MMS Series	Multi effects processor		\$1,495	516-249-3660	170
Dynatek Automation	MMS Series Raider	Multiple media system	34 rackmount data storage depot	\$3000 /\$8000		
Dynutek Automation		High capocity arroy	Free standing hord disk arroy, up to 10GB capacity	\$20-\$40,000	416-636-3000	171
	CDO	Dato storage	24 storage subsystem incorporates CD-ROM	\$3,500		

1. What would independent multiple effects be like if you couldn't apply them separately to the different inputs? Like Rogers without Hammerstein.

2. True stereo inputs let you do two independent reverbs at once at once. Got it got it?

3. Also because of true stereo inputs, you'll have both gate and ambient reverb to play around with simultaneously.

4. Can you say ambience? Then you can also simulate the use of an ambient microphone while recording.

5. You'll not only be able to utilize up to five independent delay lines, but also set intervals up to two seconds apiece without compromise.

6. How do four independent pitch shifters, each with a four octave range, sound? Great, that's how. 7. Join this rotary club and you'll simulate a classic rotating speaker with separate rise/fall times for the lows and highs.

8. Severely detuned signals. Mild flanging. And everything in between. Ladies and gentlemen, the SE-50's stereo flanger.

9. For a mild phase to one that's truly radical, the stereo phaser is sure to come in handy.

10. Plug a mic into one input and a keyboard into another and you're set for vocoder.

11. Shhh! The SE-50's noise suppressor is about to kick in.

12. Having separate EQ and noise suppressors per input is one thing. Being able to add reverb, delay and chorus to either or both is quite a cool other. 13. The SE-50's chorus recalls the renowned Roland space chorus circuitry.

14. Naturally, you'll be able to call up the classic gated reverb effect with the SE-50.

15. The compressor evens out dynamic levels and produces more sustain on your guitar.

16. For everything from heavy metal to blues to country, the SE-50 is ready to kick into overdrive or even distortion.

17. You've also got a line driver with the SE-50. Drive carefully.

18. The limiter prevents clipping or overloading on a mixer, an amplifier or even a tape recorder.

19. To simulate panning, move this magazine from one ear to the other while flipping the pages. 20. Add more high end, more low end—you know, more clarity—with the SE-50's enhancer.

21. Need enough reverb to fill a small room? How about the Royal Philharmonic Amphitheater?

Now consider you'll have 28 powerful preset algorithms for combining up to eight effects simultaneously, full MIDI control, not to mention 128 memories. All in a sleek half-rack design.

The SE-50 Stereo Effects Processor, part of the BOSS Pro Series.

It could effect you in a serious way.



BOSS, Roland Corporation US, 7200 Dominion Circle, Los Angeles, CA 90040-3647. 213 685-5141.

-					
BRAND	MODEL	TTPE	DESCRIPTION	PRICE TEL	LIT #
East West/Saundwarehouse	E ₩-040	CD/CD-ROM	Drums 2, the sequel to the sample library, Drums 1	\$129/\$399 213-8	48-3436 172
	MH Series	Engineered sound systems	Virtual array mid/high systems	\$1950 - \$2900	
ŁAW	JF Series	Ultra-compact full range	High output yet extremely compact	\$495 - \$1650 508-2	34-6158 173
	KF691i	Virtual array system	90 degree horizontal coverage	\$2,695	
E-Mu Systems, Inc.	SP-1200	Sampling percussion system	12-bit digital sampling drum machine	\$2,795 408-4	38-1921 174
	Emulator IIIxp	Digital sample editing software	For the Ellixp digital sampling system	\$3,995	
Euphonize	The Cube	Modular analog routing matrix	Adds aux sends, mix buses, and electronic patchboy	From \$15,000 838-7	66-1666 175
Eventide	DSP-4000	Effects processor	Digital I/O harmonizer with reverb; AES/EBU; SMPTE	\$5,000 201-6	41-1200 102
Fostex America	DCM-100 MIXTAB	Digitally controlled mixer	MIDI-controllable mixing system	N/A 210.0	21-1712 106
	MC-102	Mixer	Rockmount mixer with cassette	N/A 510-9	21-1112 100
Frazier	F1620/F1630	Loudspeaker systems	60" x 40" cooxial system/90" x 40" cooxial system	\$2960 ea. 501-7	27-5543 176
Gefen Systems	Pioneer CAC-V3000	300 CDs outochanger	Library starage for CS sound effects	\$6,000 818-8	84-6294 177
Gold Line	DSP-30	Audio spectrum analyzer	Portable 1/3 octave RTA w/ 85 dB window	\$1,500	
our me	APT-2	Phose/polarity analyzer	Analyzer to read phase/polarity	203-9 \$249	38-2588 178
Gotham Audio	UM 925	Microphone	Microtech Gefell studio condenser microphone	\$2,295	
Gornom Abdio	DC 1	Digital signal processor	Cedar Aucio De-Clicker	\$19,995	55-3=10 179
	MP-1 and MP-1s	Vocuum tube mic preamp	XLR input; DB-9 input for the Groove Tubes Model One	\$650/\$1200	
Groove Tubes	ST-1	"Sbock Therapy" microphone	5" dia. olum. basket w/2 rubber-lined damps		51-4500 180
	PS-1 and PS-2	Vocuum tube power supplies	For use with Model One vacuum tube mic	\$150/\$250	180
	Trans Nova 9300	Power amplifier	Three stage amp as apposed to typical 5 stages		4.1410
Hafler Pro	Trans Nova 9500	Power amplifier	Three stage amp as apposed to typical 5 stages		181
HEDCO	A5R-8X8	Audia serial router			6-1619
	1000		Stereo &x& router module for the X plus Series		1-9673 182
HME/Portland	700-16B	Integrator	Intercom		0-0137 183
Huch Sustant		HME	Intercom handset; new optional venture	\$105	
Hush Systems	Hush ISP	Intelligent sound processor	Noise reduction, compression, dynamic enhancement		3-3055 184
Innovative Electronic Designs	4400	Automatic microphone mixer	4 gated and 1 ouxillary inputs; 1 EIA rackspace	\$1,392 502-26	7-7436 185
	4800	Automatic microphone mixer	8 goted and 2 ouxillory inputs	\$3,045	
	4893	Compact subwoofer system	Addition to the Array Series System	N/A	
JBL	4894	Two way loudspeaker system	Addition to the Array Series System	N/A 818-89	3-8411 186
	M712	Compressor/limiter	Control of Threshold, Ratio, Attack, Release	N/A	
Jensen Tronsformers	58 JT	Rock-mount record electronics	For Magna-Fech	S2k per ch 213 87	6-0059 187
	PR03700	Automation apgrade pockage	Upgrade for the Tascam M-3700 recording console.	\$995	
JLCooper Electromics	datoSYNC	MIDI time code synchronizer	Designed to work with Alesis ADAT.	5349 310-21	3-3064 188
	SoftMix	Soundcraft Spirit cutomation	Record MIDI events, automate up to 16 software subgroups	\$500	
Josephson Engineering	C-700A	Studio condenser microphone	Dual-capsule, variable-pottern microphone	\$2,400 408-23	8-6062 189
JRF Magnetic	Otori DP1610	Cassette QC assembly	Fully adjustable head assembly for track placement	\$395	
and mugnerit	Ampex	Replacement heods	Flux mognetic heads far most Ampex machines	201-57 N/A	9-5773 190
JVC	XD-PIPRO	Portable DAT	Portable DAT; no SCMS; digital mic	N/A 201-79	4-3900 107
	RTDS-4CD	CD source for cossette dup	4-track doubletime duplication of audio cossettes	\$825	
KABA Research and Develop	55-RHA-LC	Record heads	Longer life (5x); low cross-talk heads (15 dB better)	\$250 415-88	3-5041 191
	RTDS-4CA	Master contraller	4-channel distribution amp flat to 80 kHz		2 204E 141
	DN728	Digital audio delay line	Delay line with 64 memories; 18-bit AD/DA conversion	\$1,195	
Klork-Teknik	DN800	Configurable active crossover	Stereo 3-way or 4-channel 2-way	53,295 516-24	9-3660 192
	K-1132	Compression driver		\$1,995	
Klipsch and Associates	9040/6040/4020	and the second se	New technology features space-age materials	S417 501-77.	7-6751 193
		Large format horns	Hybrid Trodrix technology	\$583- \$747	
Korg USA	Soundlink V 2.0	Digital audio workstation	8-track random occess recording and editing system	\$37,000 516-33	3-9100 194
	Marshall JMP1	Programmable tube preomp	100 great Marshall sounds in a box	\$999	
	7000	Nearfield manitor	2-way, very occurate neorfield monitor	\$989	
KRK Monitoring Sys	9000	Nearfield monitor	2-way, phase-coherent, accurate nearfield monitor	\$1,750 714-84	I-1600 195
	13000	Midfield monitor	3-way, phase coherent monitor achieving SPL of 112 dB	\$2,699	
Kurzweil	K2000R	Rack module	Incorporates sample playback	\$2,895 617-890	-2929 196
	DR195	Wireless receiver	Ratio diversity UHF/VHF wireless mic receiver	\$3,800	
Lectrosanics, Inc.	T195	Wireless transmitter	Hond-held UHF/VHF wireless mic transmitter	\$1,600 505-892	1-4501 197
	DC1	Audio signol processor	Digitally controlled processor	\$2,395	
	ADA-881MB	Aadio distribution amplifier	Standalone audio D/A with 8 low-impedance outputs	\$315	
Leitch Incorporated	ADA-883MB	Stereo distribution cmp	2 channels with 4 low-inspedance outputs per channel	\$355 800-231	-9673 198
	ATG-880MB	Audio tane generator mix box	Standalone with stable switch-selectable uudio tones	\$320	
Lexicon		Analog-to-digital converter	2-channel; 20-bit/4-channel; 18-bit	\$3,495 617-736	-0300 199
Magdio Americo, Inc.		DAT tope	Customized DAT length	\$4.50 - \$8.50 714-595	
		Six ch mixer/four tr recorder	Four balanced mic connections, four insert points dbx, HX Pro	5899	200
Marontz Professional Products		Compact disc recording system	Rack mountable unit, full Philips Orange Book compatibility		4900 001
		Portable DAT recorder	4x2.5x8.5"; under 3lbs, 1-bit A/D conversion at 48kHz	\$7,500 708-820	-4800 201
			Conversion of the start and th	\$2,500	

48 DECEMBER EQ



THERE ARE ANY NUMBER OF PARTS OUT THERE BUT ONLY ONE NUMBER WHERE YOU CAN FIND THEM.

You're in the middle of a late night session. Your gear goes down and you need a hard-to-find part to get up and running again. What to do?

Just fax or call AudioTechniques and your problems are over. Our expanded parts department has the largest inventory around -- including all the major lines.

And each part is backed by AT's legendary service and support. Because we have the experience and know-how to answer any questions you have about parts.

Fast turnaround? Fax us directly from your midnight session at our 800 Fax Order Line and we'll

-800-PRO-AUD

respond by 12:00 noon of the next working day! Or call us any weekday between 9:00 AM and 6:00 PM for fast service.

Whatever the number of parts you need, now there's only one number to fax and one number to call. AudioTechniques.

 Telephone Order Line: 1-800-PRO-AUDIO (776-2834)
 A

 Fax Order Line: 1-800-775-2834
 In NY State, call 212-586-5989

 CIRCLE 15 ON FREE INFO CARD

A Manny's Company 1600 Broadway, New York City. N.Y. 10019

AudioTechniques

BRAND	MODEL	TYPE	DESCRIPTION	PRICE	TEL	LIT #
Mark of the Unicorn	UNISYN	Saftware	Universal editor/librarian for the Macintosh	\$395	617-576-2760	202
	3080	Microphone coble	AIS/EBU digital microphane cable	S0.38 per foat		-
Marshall Electronics	3082	Mogami speaker cable	CL-2 14 gauge OFC cooxial cable	S0.42 per foot	310-390-6608	203
	3076	Mogami SCSI II cable	Long distance computer cable and assemblies	\$1.10 per faot		
Martech/Martinsound	EMT-140	Upgrade	Electronics, transducer, and cabing upgrade for EMT plates	\$2,495		
	EMT	Vibration damper	Mounting bases to absorb floor vibration	\$250 pr	818-281-3555	204
Medio Technologies	DCC Mastering	Mastering/text editing	Basic timecode and PQ codes used for DCC	N/A	516-563-1721	205
Metrosoft	MetraTracks	Software	Multitrack software for NeXT	\$595	619-488-9411	206
Meyer Sound	HD 2	Studio monitor	High definition mid-field monitor	\$6,000	510-486-1136	207
	950 UHF	Wireless mic system	Ten-channel UHF rackmount with 120 dB dynamic range	\$2,496		
Nady Systems, Inc.	RW-3 UHF	Wireless mic system	Four-channel UHF rockmount with 120 dB dynamic range	\$996	510-652-2411	208
	301 UHF	Wireless mic system	Four-channel UHF with 120 dB dynamic range	\$800		
lagra/Phi Technologies	NAGRA-D	Digital field recorder	Portable, 4-channel digital tape machine	N/A		
togray i in recimologies	Nagra T-	Audio timecode	2-channet, center track timecode	N/A	40 5- 521-90C0	209
leotek	Esprif	Mixing console	8 Group busses; mix minuses	\$23,900 >	312-929-6699	210
	KU 100	Binoural microphone	Third generation dummy head: "Fritz III"	\$5,995	6 X 4 4	
leumann (USA)	KFM 100	Stereo microphane	Spherical-surface sterea microphane	\$5,595	203 434-9190	211
	U 67	Tube microphone	Reissue of classic tube micraphone	\$4,995	203 434-7170	
	A2	Audio test equipment	Dual channel audio generator/onalyzer	\$5,900		
leutrik USA, Inc.	Combo	Audio connector	The first XLR and 1/4" jack in ane connector	N/A	908-901-9488	212
	Bontom potch cords	Patch cords	One-, two- and three-foot pre-assembled patch cards	N/A		
VISION, Inc.	NV5000	Master sync generator	Locks to NTSC, PAL, AES/EB, and SDIF-2	\$7,500	916-265-1000	213
	Galaxy V 1.2.2	Macintash MIDI software	Integrated universal librarian with editors for MIDI devices	\$399		
prode Systems, Inc.	StudioVision, V 1.4	Macintash MIDI software	Integrated MIDI sequencer with digital recording/editing	\$995	415-856-3333	214
	Studio 4	Mac MIDI interface/processor	8-in/10-out, 128-channel MIDI interface	\$495		
ptical Medio International	TOPIX	CD recording system	Desktop CD recorder and softwore	\$10,000	408-376-3571	215
	DTR-90T	R DAT recorder	4-heads; offers tape monitoring as well as insert editing	\$8,995		
to ri Corporation	CB-149	Editor	Provides simultaneous edit contral of two DTR-90Ts		415-341-5900	216
	4.0 upgrade	ProDisk software upgrade	GUIDE operating system and visual display	N/A		
aktec Automation	JC7000	J-Cord/U Cont inserter	Assembles J-Cord or U-Cord at 100 pieces per minute		206-745-5580	217



with the NEW Furman PUNCH-10 Bass Enhancement System!

Add super bass PUNCH[™] to your mix or sound reinforcement system with Furman's PUNCH-10 Bass Enhancement System. No mere equalizer, the PUNCH-10 is a true subharmonic processor that actually *creates* bass content by synthesizing harmonics one octave below the original bass. It takes you from bass you can hear to bass you can *FEEL*! The PUNCH-10

includes a limiter for speaker protection. With its adjustable high

UNC

pass filter, it can be used either with or without a subwoofer Best of all, the PUNCH-10 is *very* affordably priced. See your dealer for a demonstration, or contact:

RMAN anal (inversina

Furman Sound, Inc 30 Rich St. Greenbrae, CA 94904 USA Phone: (415) 927-1225

CIRCLE 37 ON FREE INFO CARD

Take Control

When factory presets alone are just not nough. When you know it's time to exercise more ontrol over the sound of your music. Move up to **MIDIVERB®III.** The powerful simultaneous multiiffects processor that's the latest incarnation of the ward winning* MIDIVERB from Alesis.

ALESIS

We know better than anyone else how to nake great sounding digital effects. That's why MDIVERB III has 200 programs in 16 bit stereo with 5kHz bandwidth. And features the kind of prorammability you need to create powerful mixes. It this and 4 effects at the same time make MDIVERB III the first choice in digital processing. For the ultimate in creativity, MIDIVERB III's extensive MIDI control lets you alter effects in real time while you're mixing or performing. Control reverb decay time with aftertouch. Alter the modulation speed of the chorus with a foot controller. Record the changes into your sequencer for an automated mix. This is the kind of control you need to make a personal statement with your music.

It's time to stop compromising. Get on the inside of your effects processing. Take control of a MIDIVERB III at your Alesis dealer now.

Midiverb II won the prestigious 1988 TEC Award for technical achievement.



llesis Corporation 3630 Holdrege Avenue Los Angeles CA 90016

				PRICE	TEL	LIT #
BRAND	MODEL	TYPE	DESCRIPTION			
Ponasonic/Ramsa	WX=RP410/RP700	ENG/EFP wireless microphone	Receiver and transmitter feature 30-channel selection	\$4,800	201-348-7846	218
T one some y wants a	Ramsa WZ=DM30	20-bit digital multiprocessor	Comp/limiter, graphic eq, 4-way crossover,4-bond poro eq	\$3,600		
	PRM 3085i	Nearfield studio monitor	3-way reference monitor; switch-selectable response	\$399.95 ea		
Peavey Electronics	DPM SXII	16-bit stereo sampler	256 kbytes on board, RAM expandable to 32 MB;	\$400	601-483-5365	219
	APB 8000	Audio patchbay	64-input patch points; front panel jacks internally normalled	N/A		
Burn and Ciles Inc.	PGF7000	Endless belt output controller	100mm stroke; belt with LED indicators under the belt	\$70	310-452-4995	220
Penny and Giles, Inc.			Measures and verifies the audio content and bitstream	\$5,535	800-231-9673	221
Pro-Bel	6510	Digital audio analyzer		\$499		
Professional Audio Systems	PermTech P100	Loudspeaker (contractor)	Full range outdoor loudspeaker system		619-591-0360	222
	TOC tm \$1-122M	Stage Loudspeaker	TOC system 12" coaxial stage monitor	\$867		
QSC Audia Products, Inc.	USA Line	Professional power amplifiers	Includes three models, all with full complementary circuits	\$578 to \$998	714-754-6175	223
	Q'SControl	Audio Control System	Uses Virtual Network Operating System (V-NOS)	N/A		
Quested Monitoring Sys	Q208	Nearfield monitor speaker	Compact 2-way nearfield monitor	N/A	510-351-3500	226
Q Up Arts	Heavy Hitters Hits	CD audio sound library	Over 350 heavy metal drum samples	\$149	408-688-9524	224
Radion Audio Engineering	MMBTA	Reference monitor speaker	Time-align, nearfield monioring system	\$990 pr	714-693-9277	225
	CM 86	8x4 compact mixer console	With 4 output buses and expandibility via Flex Bus	N/A		
			3 stereo ins for instrument mic with Phontom/Piezo Pickup	N/A	206-355-6000	101
Rone Corporation	MAP 33	Prog. acoustic inst. processor		N/A		
	FVL 22	Flex VCA limiter	2-ch remote level control interface w/limiters each channel		_	_
Renkus-Heinz, Inc.	QUBE-2	Two-way laudspeaker system	Coaxial; 15" woofer w/mounted 1" HF horn/driver	\$788	714-250-0166	227
	QUBE-STK	Two-way loudspeaker system	Constant beamwidth horn and 1" HF driver	\$675		
	Stage Monager	Cue software for MIDI control	MIDI sequencer for live performance	N/A	604-732-1234	228
Richmond Sound Design Ltd.	MMC-16	MIDI-to-contact converter card	Uses SET command of MSC for 27 contact outs;	\$2,034		
	SRC-2	Dual sample rate converter	Mixes 2 stereo digital signals at different rates	\$2,595		
Data Data Andi (Midaa Carua	DM-80-L	Locking Resolver	Lacks DM-80's sample clock to SMPTE frame edge	\$995	213-685-5141	104
Roland Pro Audia/Video Group				\$2,895		
	SP-700	Sample playback module	Access to CD-ROM Sample Archive via SCS1		818-504-6294	229
Roldex Industries	MIK-2	Muiti functon silk screener	Silkscreens on a flat surface from video down to DAT	\$22,500	810-304-0274	117
RPG Diffusar Systems, Inc.	Omnitfusor	Fibreglass reinforced gypsum	2' x 2; non-combustible wall ar ceiling mounted	N/A	301-249-0044	230
	VAMPS	Modulor performance shell	Avoilable in partable, rollable, or fully rigged formats	N/A		
	Intelliverb	24-bit virtual room processor	Virtual Room reverberation effects	\$1,299		
RSP Technologies	Studio Gote	Quad noise gate	4 gates with individual wide range controls	\$379	313-853-3055	231
	Studio Q	Four-band parametric equalizer	Offers state variable symmetrical peak/dip filtering	\$379		
Russion Dragon	RD-R	Timing accuracy meter	Measures timing accuracy of 2 sounds; polarity check buttons	\$499	512-525-0719	232
Kossion Drogon	FBX-1200	Feedback Exterminator	DSP-based filtering device with 12 independent filters	N/A		100
Sabine Musical Monf				\$600	904-371-3829	233
	FBX-900	Feedback Exterminator	Spectral variation <± 1.5dB, 20Hz to 17,000Hz	\$795 each	_	-
	VPR 2 A/B	Audio post 1"	Replaces Ampex metal heads with ferrite			024
Saki Magnetics	VPR 6 R/P	Audio post 1"	Heads with 6x life expectancy	\$795 each	818-880-4054	234
	VPR 80 R/P	Audia past 1"	Heads with 6x life expectancy	\$795 each		
	2242	Mixer	22 inputs, 4-bus, 4-band EQ; 6 aux sends; left/right bus	\$1,130	516-932-3810	235
Samson Technologies	VRS/VTS	Selectable VHF wireless	74 available frequencies, 11 for simultaneous use	N/A		
	Vitalizer	Signal processor	Psycho-ocoustic equalizer	\$1,500		
Sascom Marketing Group	Optifile LC	Console outomation	Retrofit console outomation system	\$7,800	416-420-3946	236
Justom murkening Group			Raindirk's new console	N/A		
	Symphony II	Recording console		\$555		
	Schoeps CMC 6	Microphone omplifier	12-48V Universal phontom-powered omplifier body		010 040 0707	007
Schoeps/Pasthorn Recordings	Schoeps VST 62	Stereo mic amplifier	12-48V Universal phontom-powered amplifier bady	\$1,165	212-242-3737	237
	Schoeps AMS 22	M-S stereo shockmount	Mid-side stereo elastic suspension	\$189		
	CPA-658	Foder	65mm conductive plostic	N/A		
Sellmork Electronics	CP-601	Rotary pot	Conductive plostic	N/A	508-443-8053	238
	CP-2500	Ratary pot	Conductive plastic	N/A		
	EM 1046	UHF receiver	8-channel modular receiver	\$59,698		
Sennheiser Electronics	SK 50	UHF tronsmitter	50 mW transmitter	\$3,165	203-434-9190	239
Jennierser Electronics				\$3,990		
	SK 250	UHF transmitter	250 mW transmitter			
Siemens Audio, Inc.	Neve Capricorn	Digital mixing console	Digital signal path, reset, dynamic automation, stereo pairing	N/A	203-744-6230	240
	SoundField MkV	New single-point mic system	Improved electronics, systems diagnostics, and ergonomics	\$5,850		
Singular Solutions	A/D64x PRO	A/D converter	2-channel 16-bit A/D w/mic preamp; audiophile quality	\$1,395	818-792-9567	241
Solid State Logic, Inc.	Scenaria	Digital a/v production system	Automated routing system and random access video	N/A	212-315-1111	242
	Custom colors	Acoustical foam	Now available in any color from postels to neons	N/A	800-662-0032	243
Sonex Acoustical/ilibruck	ProSPEC	Vinyl barriers for isolation	Provides same density os lead sheeting	N/A	000.001-0031	.45
	SonicNet	Network card	FDDI SCSI node with 100 MB network top	\$4,500		
Sonic Solutions			Time compression/expansion without pitch change	\$995	415-485-4800	244
	TimeTwist	Softwore option				
	DPS-F7	Dynomic filter plus	Dynamic EQ; compression/limiter; exciter; tope simulation	\$1,600		110
Sony Corp. of America	PCM-2700A	DAT recorder	DAT recorder with 4 heads, foder stort, and AES/EBU	\$2,900	201-930-1000	112
	CDK-3600	CD changer	Dual-transport 360 disc CD changer	\$8-9000	201-930-1000	112
Soundcraft USA	Delta SR	Four bus console	Available in 8-, 16-, 24- and 32-input versions	N/A	818-893-8411	245

Sani Ash New York, NY (212) 719-2640

Manny's Music New York, NY (212) 819-0578

Gand Music Chicago, IL (708) +16-1265

Goodman Music N. Hollywood, CA (213) 845-1145

Direction Sound Los Angeles, CA (310) 276-2063

Westlake Audio Los Angeles, CA (213) 851-9800

West L.A. Music Los Angeles, CA (213) 477-1945

Ace Music Center N. Miami, FL (305) 891-6201

YRS MIDI Systems Fern Park, FL (407) 331-6333

Richard Audio Montreal, PQ (514) 733-5131

Adcom Electronics Toronto, ON (416) 251-3355

Saved By Technology Toronto, ON (+16) 928-6434

Steve's Music Toronto, ON (416) 593-8888

High Tech Annex Vancouver, BC (60+) 689-6639

Sound Technology Letchworth, UK (0462) 480-000

Soundware Aarhus, Denmark (086) 11 89 12

Creative Software Oslo. Norway (02) 64 14 30

Ceritec Utrecht, Holland (030) 316 353

And Music Wicsbaden, Germany (0611) 928 050

M. Casale Bauer Bologna, Italy (051) 76 66 48

SCV Audio Paris, France (0148) 63 22 11

Giant Electronics Bienne, Switzerland (032) 22 52 78

The Right Connections



Macintosh



Hybrid Arts, Inc.





World Radio History

SPECTRAL

DynaTek Automation Systems Inc., 15 Tangiers Road, Toronto, Ontario, Canada M3J 2B1 Tel.: (416) 636-3000 Fax: (416) 636-3011 CIRCLE 30 ON FREE INFO CARD

an | []]

127

				PRICE	TEL	LIT #
BRAND	MODEL	ТУРЕ	DESCRIPTION			Elt w
Soundcroft USA	Venue III	Console	Available in 16: 2 32- and 40-input frame sizes	N/A	818-893-8411	245
		In-line recording console	Available in 20, 28, 36 or 44 sizes	N/A	818-893-8411	
Sound Ideas	Universal CD Library	Sound effects library	A classic collecton of motion picture sounds	\$595	416-886-5000	246
	Jode	Mixing console	DSP microprocessor technology; dual inputs on every module	S45k - S85k		
Soundtracs	Solo Monitor	Mixing console	Designed for stage; 10 dedicated monitor sends	\$6,000	516-932-3810	247
	Trocmix II	Automation system	Foder/mute system for up to 64 ch locked to SMPTE	\$10,000		
The second second	StudioTrocks V 1.5	Softwore	Windows-based hard disk recorder/editor system	\$1,295		
Spectral Synthesis, Inc.	ADAX 8818	Digital audio conversion	8 in/8 out A/D, D/A convertor	\$4,000	206-487-2931	249
	SynClock 6400	Timing hardware	Studio timing reference convertor for digital audio	Price TBA		
Stellovox Digital Audia	Stelladat	RDAT portable tape recorder	Demonstrated modular plug-in timecode board (\$2000)	N/A	703-378-1515	250
	PA-1800	Amplifier	A two rack space, 17 lb, 1800 watt power amplifier	\$1,699		
Stewart Electronics	PRO-DAC	Digital to analog converter	Single rock space digital to analog converter	\$799	916-635-3011	251
	the second s	Hord disk-based multitrack	Allows real-time crossfades in all editing and recording modes	\$28,950		
	Dyoxis II		High spooling speeds of up to 400 times play speed	\$7,400	615-254-5651	252
Studer Revox	D780 R-DAT	Digital R-DAT tope recorder		\$9,995		
	Dyoxis Lite	Hard disk recording/editing	Digital editor/controller w/tape type interface			
	P7	Mixing console	MIDI inline recording console — 16x8x16	\$4,195		050
Studiomoster, Inc.	Diamond Pro Series	Mixing consoles	16x4x2x1 live reinforcement console	\$1,495	800-878-7883	253
	Diamond Pra Series	Mixing console	24x4x2x1 live reinforcement console	\$1,995	_	
Summit Audio	MDSP-200	Peak and overage meter	Modular two-channel high resolution DSP-based meter	\$495	408-395-2448	254
Johnin Addio	MMP-125	Mic preamp	Vacuum tube modulor mic preamp	\$595		
	601	Digital voice processor	Mic or line level analog signals, converts to digital (18 bits)	\$1,995		
Symetrix, Inc.	421	AGC-Leveler	AGC section maintains constant output level	\$549	800-288-8855	117
	402	Dual output delay	1 input/2 output; >100dB dynamic range	\$695		
and the state of the second second	System 6	Studio monitor	8" reference dual concentric	\$795 pr		255
Tannoy.	PBM-5	Playback monitor	5" playback audia monitor	\$295 pr	519-745-1158	
	DA-88	and in contrast, and the second se	8-track digital multitrack using 8mm format Hi8 tape	\$4,499		
		8-track digital multitrack		\$5,499	213-726-0303	105
Toscam	RA-4000	Digital recorder/editor	2-channel, 4-track random access standalone dedicated unit		213-720 0303	103
	DA-60	Synchronizable DAT	4-head DAT recorder w/sync for the film, video,audio	\$5,499	005 030 1000	054
TC Electronic	M5000	Digital signal processor	User upgradeable processor w/AES/EBU, SMPTE, and MIDI	\$4,240	805-373-1828	256
TDK Electronics	DA-R16	DAT cassette	Tailor-made for short running time studio applications	\$9	516-625-0100	257
	CD-W12	Recordable compact disc	63 minutes ofdirect-to-disc stereo recording	\$30		_
	TCM-1121	SR speaker	2-way 10" system with Apaxial Array Technology	\$4300 each		
TAD	TWM-1622	Wedge monitor	2-way 16" wedge monitor w/ Apaxial Array Technology	\$4100 each	310-952-2415	258
	TWM-1122	Wedge monitor	2-way 10" wedge monitor w/ Apaxial Array Technology	\$4500 each		
Techron - Tef Products	AcoustoEQ	Software	For use with TEF 20 analyzer and a parametric EQ	\$300	219-294-8000	259
- 24	ACC Series	Audio tope duplicator	Duplicator with expandability and audio level controls	\$1800-\$2600		
Telex Communications	Copyette 1.2.1	Audio tape duplicator	Economy line duplicator	\$419	612-884-4051	260
	Copyette 1.2.3	Audio tope duplicator	Economy line duplicator	\$1,099		
Timeline	and the second second	Timecode module	Wide-bond timecode reader	N/A	619-727-3300	274
TimeLine	Lynx-2	and the second	150 W into 8 ohms; 300 W into 4 ohms	\$1,950		
Times One Amplifiers	RFM 500	Mono block amplifier			310-247-4848	261
	PS 400	Stereo omplifier	125 W into 8 ohms; 250 W into 4 ohms	\$1,595	415 500 0520	24.0
TOA Electronics	WT-780	Wireless microphone system	The VHF unit operates between 168 MHz and 216 MHz	\$498	415-588-2538	262
Trident Audio USA	Trident 90	Mixing console	24 multitrack busses; 10 aux busses, in-line monitoring	\$99,500	310-533-8900	263
Tube-Tech	Tube-Tech LCA-2A	Tube stereo comp/limiter	Separate comp/limiter for each channel; after Fairchild 670	\$2,993	212-586-5989	264
Turtle Beach Systems	Woves	Editing for Windows 3.1	Cut/poste editing; 3-D freq analysis; 4 band para eq	\$149	717-843-6916	265
Valley Audio Bradente Inc.	Autogote	Noise gate	Two-channel frequency selective noise gate	\$679	800-800-4345	266
Valley Audio Products, Inc.	Advontage 310	Noise and level meter	Av. RMS & Peak Detector Resp., 70 dB scale in two ranges	\$399		
Vego	Q600	Wireless intercom	A UHF Q-Plus wireless intercam system	\$14,207	818-442-0782	267
Westloke Audio, Inc.	4 AWG Series	Low-resistance speaker cable	Contains 800 strands of capper per wire	N/A	805-499-3686	268
	TEN-4	Tester	Speaker cable tester; Newtrik speakans and 1/4" connectors	\$125		
Wireworks Corporation	815	Multipin lotches	Broadway lotching system; new style multipin lotches	N/A	908-686-7400	269
interioras corporation	G5/160i	Connector	New multipin connector stondord for cobling systems	N/A		
		- A - A	Stage monitor cansale, 22 mixes, 44 or 52 inputs	N/A		
	PM4000M 44/52	Mixing console			714-522-9011	270
Yomoha	MC3210M	Mixing console	Stage monitor console, 10 mixes, 2 oux, 32 inputs	\$7,500	714-322-9011	270
	MC2410M	Mixing console	Some as above, except 24-inputs	\$5,700		
	Audiopro 1212 /16	1200 watt powered mixer	12 and 16 channels respectively; weight less than 50lbs	\$1212-\$244		
Yorkville Sound	EX-350	2-way BBC loaded	1x12" with horn; 102 dB sensitivity	N/A	716-297-2920	272
	Aeroquip	Flying hardwore options	6,000 lb roting/trk; +10 to -25 degree cabinet pitch adjust	N/A		
	9001	Multi-effects processor	40 presets/40 user patches; 20 effect types	\$399		
Zoom Corporation	9120	Sound environment processor	Sterea reverb, 99 recollable presets; 10 effect types	\$599	415-873-5885	108
	9200	Reverberation processor	31 effects with up to 297 presets	\$2,995		

Until every facility and touring system gets one, you'll want to keep your SuperGate nearby. That's



because dbx's real engineering advances mean your 172 will pull sparkling, clean output from material other gates can't touch. You won't want to let go of your SuperGate because it sounds better, sets up faster and does more for you than those other gates can.

You'll be more selective with Voltage Controlled Filters.

The SuperGate uses sophisticated VCFs to produce steep slopes that are TWICE as selective—dramatically reducing false triggering. They're

also faster to set up because the PARAMETRIC design lets you zoom in on exactly the instrument or part of the attack you want to catch.

You'll stop time with Transient Capture Mode."

You can see and hear the difference TCM^{*} makes compared to other gates. With real-world filter and threshold settings, it attacks so quickly you'll actually catch those critical transient wavefronts that give percussive signals their life and definition.

You'll control time

In One-Shot

mode, the 172

gives muscle tone

to that drummer

with One-Shot"

mode.

FILTER SLOPE COMPARISON



with the not-sosteady hand or dbx with TCM

Input Signal

foot, so you get solid, even output from erratic input. There's more-

we'll fax you the whole story. But with all of the people who want to get their hands on a SuperGate, it

When you do get yours, you may find yourself going to extremes to hang onto it.



SuperGate, it may be awhile before you can test one for yourself.

Brand A

172 SUPERGATE



A DIVISION OF AKG® ACOUSTICS, INC. • 1525 ALVARADO STREET • SAN LEANDRO, CA 94577 • TEL: 510-351-3500 • FAX: 510-351-0500)



CIRCLE 23 ON FREE INFO CARD



Fill your rack without emptying your pockets.

We build more equalizers than anyone else in America. and we sell them all over the world.

And now we're introducing the new SeriesII, with 5 EQ's and 5 signal processors. More than just a new look and feel, DOD SeriesII EQ's are engineered and built "smart", using the latest Computer Aided Design techniques, and the highest manufacturing standards.

We've even lowered the price to give you more value, and made them the best performing units in their class.

RTA Series II Real Time Analyzer \$359.95

> Quick, easy and accurate setup of your system to almost any environment. Includes calibrated microphone and pink noise generator.

These new models offer 12dB of boost & cut on ISO standard center frequencies, and FET switches for the lowcut and bypass functions to eliminate annoying "thumps".

With a signal-to-noise ratio greater than -90dB, and harmonic distortion less than .006% @ 1kHz. they're ideal

> CIRCLE 31 ON FREE INFO CARD orld Radio History

for PA, club, or lounge installations, or for your studio recording projects.

Come and listen to all the new SeriesII products at your favorite dealer today... or call or write us for more

information.

Fill your rack with DOD and you can keep the change in your pocket.



5639 South Riley Lane • Salt Lake City, Utah 84107 (801) 268-8400 • Fax (801) 262-4966 Int'l Fax (603) 672 4246

A Harman International Company





STINKING UP THE GREAT INDOORS

Sweats." Recently, the group finished a 10-week reunion

tour in celebration of its 25th anniversary and their latest

album, Break Like The Wind.

The tour took Tap across

America, up into Vancouver, Canada, and finally to a two-

night homecoming stint at

the prestigious Royal Albert

The group's house sound

Hall in London.

WHEN THE NAME Spinal Tap is mentioned, people think of one thing. All right, two things - but we're not talking about an often-painful back examination here. We are talking about the legendary band popularized in Rob Reiner's mock-documentary This Is Spinal Tap, and that features comedianscum-musicians Michael McKean as guitarist David St. Hubbins, Christopher Guest as lead guitarist Nigel Tufnel, Harry Shearer as bassist Derek Smalls, and more drummers than a percussionist's convention.

Throughout the years the group has given the world such memorable tunes as "Big Bottom," and "Christmas With The Devil," as well as the pungent "Stinking Up The Great Outdoors" and the profound "The Sun Never

THE BAND THAT MAKES THE FANS SCREAM AND THE CRITICS CRINGE CELEBRATE 25 YEARS OF TAP

BY TONY SAVONA



tour went pretty smoothly. "Of course," Paterson adds, "there was a constant fear that the drummer would explode at any second. Luckily, this never happened, although he did break his ankle and had to play half the tour using only one foot."

Paterson has worked with many of heavy metal's finest acts, including Judas Priest and Black Sabbath. His rack usually includes a large stable of gear (see the October 1991 *EQ*) but because Tap played in smaller venues, Paterson brought only a small part of his rig. Included in this abridged rack was a Yamaha PM3000 board, a Roland tape echo and 3000 digital delay, an AMS digital delay, and some gates and compressors.

His most important tools, Paterson explains, were the arsenal of Shure mics that he packed. "Tap's philosophy regarding their sound can be summed up simply: loud. Very loud. I needed microphones that could give a lot of gain before feedback, and the Shure Beta Series did the job."

Paterson used Beta 58 microphones to capture all of Tap's vocals and used AKG D12's on the guitar amps. Shure SM91A, SM98A, SM81, SM94, and Beta 57 mics were used on the drums. For the opening song, where the band flies in on wires and lands on the stage (oftentimes missing the stage entirely), Paterson used Shure LS114 wireless guitar systems.

A real challenge arrived in the form of Tufnel's custom-made Marshall amp the JCM9000. "The JCM9000 is 12-feet high and has 24inch speakers," recalls Paterson. "Nigel's old amp was able to go up to '11'; this one reaches *infinity*. Not only did we have trouble getting enough electricity to power the monster, but we needed a mic that could capture the full effect without distorting. Fortunately,

continued on page 61



RANK C



Introducing the breathtaking performance of Crown Power-Tech.* Designed spanifically for the professional musician. Power-Tech delivers the power and reliability you've come to expect from Crown at an affordable price.

When its all on the line, there's no room for comprimise. Here's the scenario: The club's starting to cet crowded and you need more gain. You've got the mater maxed-out, but there are still a couple or clicks left in the amp. No problem! It's a good thing Crown is on the name plate. With



Power Tech, dependability is a given. You can crack it with confidence.

Why? Because Power-Tech is fully protected from everything that can go wrong, including piot error. Patented ODEP* circuitry protects the amplifier from over-stress problems and

Fault Protection design monitors

the output devices. Additionally lindependent power subplies provide assurance that your music will keep pumping even under the most adverse conditions.

In other words with Power-Tech, your performance will never be compromised.

For additional information see your local Crown dealer or contact Crown directly at 1-800-535-6289



BEG	1 # 1	C A T	1 0 N 5
Model	4 ohms	8 ohms	bridged mone
Power-Tech 1	300 watts	220 watts	575 watts
Power-Tech 2	440 watts	320 watts	965 watts

iummineed Excellence CIRCLE OF ON FREE SHID CARD



BOYS NEXT DOOR



SOUND COMPANY: Jason Sound, North Vancouver, British Columbia **HEAD GUY: Jeff Lilly** ENGINEER: Jody Perpick CURRENT PROJECT: Bryan Adams Waking Up The Neighbors Tour CONSOLE: Soundcraft Europa 48 x 8 RACK: Jeff Lilly explains, "We're using White 4700 programmable 1/3 octave equalizers and BSS FCS 926 programmable parametrics.

MEET MY RACK

JASON SOUND HELPS BRYAN Adams & Company wake up the neighbors

How to Get From Here...







Rear-panel connections include Low-Z In, Loop Send/Return, Stereo Output (all audio connections are unbalanced 1/4-inch phone jacks), MIDI In and Out, and Remote In (for footswitch; compatible with Macintosh printer/modem cables). The front panel includes a hi-Z input that overrides the rear panel input jack and headphone jack. Power comes from a "wall wart" AC adapter.

CHECKING OUT THE SOUNDS

One of the tests I like to do on a piece of gear is see how far I can get into editing without consulting the manual. The Zoom 9030 did well in that respect. As it turned out, I was able to select and edit patches with just a cursory skim of the manual.

Selecting patches is easy, in as much as there are two dedicated Patch Up/Down buttons. Since it takes a while to scroll all the way from 01 to 99, pressing on the Down button while pressing Up (or vice-versa) speeds up the scrolling.

The best user interface features are the four knobs that let you adjust four parameters per display page instead of the usual one (all right!). Although the knobs work best with tiny fingers, they beat scrolling through parameters and using a single slider.

The factory patches are very impressive, but tend to be melodramatic in spots—you go "wow" when you first hear them, but some aren't so usable in a musical context. Not to worry. I had no problem taming the patches into something a bit more "meat-and-potatoes" instead of ultrabright, swirling, animated nifty-keen special effects. One pleasant surprise: reverb, usually the weak point in multieffects, is surprisingly good, as is the pitch-shifting.

The editable parameters reflect a synthesizer designer's influence. The "Metallic" effect includes ring modulation, the tremolo incorporates multiple LFO waveforms, and several parameters respond to input level—not just the usual filter cutoff frequency, but also things like flange sweep, ring modulator attack, and pitch detuning. The last is great for nonperiodic chorus effects. There are also some truly novel effects, such as bomber (pick a note and hear a grainy, rumbling explosion) and zitar (a virtual sitar emulator—certainly not the real thing, but interesting, nonetheless).

Programs can be recalled by MIDI program change commands via four separate program change tables that match incoming commands to 9030 programs. An interesting "learn" mode lets you program the tables by sending program change commands from an external source (e.g., keyboard, footswitch).

Regarding continuous controllers, each page of parameters has an associated page that lets you pick one of the parameters (with some exceptions, such as delay time) for MIDI control, along with a controller number. The initial parameter setting determines one end of the controller sweep; a range parameter scales the incoming controller, positively or negatively, to add to or subtract from the initial value. The higher the range parameter, the greater the sweep.

Real time modulation controllers are 1-5, 8-31, 33-37, 40-69, 80-90, 92-95; 7 controls master volume, 70-78 turn the various effect groups on and off, and 91 bypasses all effects. This extra capability is welcome, although the 9030 does not respond to pitch bend or pressure data, nor do any parameters sync to MIDI clocks—adinittedly a rare feature in signal processors, but one that would suit the 9030 well. Finally, there's a MIDI monitor that indicates the channel, controller number, and value of incoming changes—most handy.

Space prevents a detailed analysis, but Zoom offers the 8050 companion footswitch for remote control. It features 9 footswitches, and two inputs for expression pedals to change parameters in real time.

OPINIONS

The Zoom 9030 packs a lot into its halfrack size. It sounds squeaky clean, and can make "gimmicky" as well as straightforward sounds. Of course, sound quality is very subjective — but running through the presets gives a good indication of what the unit can do. Programming is pretty easy, although some of the parameter abbreviations may seem cryptic until you become familiar with the unit, and the MIDI implementation is well-done. Overall, the 9030 will build on Zoom's reputation of providing good things in small packages.

-Craig Anderton



Join the Audia Institute's successful working graduates or learn how to build your own studio. Career guidance. Diploma. Job placement.

SEND FOR FREE INFORMATION: Audio Institute of America 2258-A Union Street, Suite AN San Francisco, CA 94123 For A Sound Education TM

CIRCLE 12 ON FREE INFO CARD

MAXELL'S MUSIC MASTERING MACHINES New Professional MS Master Studio Cassettes



DEMO TO THE MAX!!

•Maxell High Bias XLII Formulation.

•Available in 20,30,60, & 90 Minute Lengths.

•Call or Write for Our A/V Catalog.

Distributed Nationally by:



THE LEADING DISTRIBUTOR OF MULTI-MEDIA PRESENTATION PRODUCTS 575 Eighth Ave. New York, NY 10018 TEL. 800-274-7666 FAX 212-564-9488

CIRCLE 57 ON FREE INFO CARD

IN REVIEW

Boss EH-50 Stereo Enhancer



MANUFACTURER: Boss, product of Roland Corporation US, 7200 Dominion Circle, Los Angeles, CA 90040-3647. Tel: 213-685-7141.

APPLICATION: Half-rack space signal processor that accepts instrument and line levels.

SUMMARY: Dual-channel enhancer and much more, including an EQ and instrument preamp.

STRENGTHS: 5 EQ presets, three dynamically controlled, plus an instrument preamp.

WEAKNESSES: Like all processors, this unit can be abused, so be cool.

PRICE: \$260

EQ FREE LIT. #: 119

THE BOSS EH-50 is a dual-channel audio signal processor. While its name, "Stereo Enhancer," implies that it can modify the stereo image, "Enhancement" actually comes in five flavors of preset equalization: Presence, Voice, Attack, Expansion 1, and Expansion 2. The last three presets are also dynamically controlled by an expander.

GLOBAL

The Enhance Level control adjusts the amount of processing as indicated by an LED bar graph. A pair of LEDs serve as overload indicators on the Left and Right channels. Processing can be switched in or out by the Enhance switch. Quarter-inch jacks access the inputs and outputs. The Left input also serves as a Mono mult to both channels.

BONUS FEATURES

The EH-50 is capable of handling a wide range of input signals, from that of musical instruments (guitar, bass, etc.) to professional line levels. A switch on the rear panel selects between -10 and +4. The input circuitry includes an FET (Field Effect Transistor) that matches the input impedance with that of musical instruments. For this reason alone, the EH-50 serves as an effective preamplifier for such devices, preserving

the full body of sounds they produce. A good alternative to a direct box, the EH-50 bypasses all of the additional circuitry of active DI's and mic pre's, getting the signal up to line level. From here you can go straight to multitrack, a fader input or a multieffect processor.

EQ AND EXPANSION

Equalization is a very subjective area. What I prefer, you may not, and vice versa. I found Expansion 1 to be great for tired-sounding rhythm tracks. The EQ boosts the bottom and the top, the expander gives it punch. If you are bouncing tracks on a personal multitrack, this preset may be the ticket. If you have an overcompressed track, Expansion 2 is an across-the-board expander. Attack restores transients to percussive sounds.

The other two presets, Presence and Voice, are high-frequency boosters that help "lead" instruments and vocals, respectively, to cut through dense arrangements. I experimented with many different sources: drum samples, analog voice tracks, a Wurlitzer electric piano, bass guitar, and entire mixes.



100 DECEMBER EQ



We have an array of compressor/limiters: Focusrite ISA-131's, Drawmer 1960's and DL 231's, and three BSS models — DPR 404's, DPR 402's, and DPR 901's. We're using Drawmer DS 201's along with the gates on the Europa console for gating and expansion. For reverb, we have Lexicon 480L's and LXP15's. To round out the system, we have Focusrite ISA 115DHP microphone preamp/ equalizers; a Carver PSD 36b CD player; Tascam DA-30 DAT recorder; and BSS TCS 804 time correction."

LOUDSPEAKER SYSTEM: "We use our own customdesigned J60 system that consists of five different loudspeakers. The J61 is a horn-loaded bass unit that employs two JBL 2241 drivers in a proprietary "pushpull" configuration. Full-

power, low-frequency output is down to 35 Hz in a proper array. Each of the 24 J61's on the tour is powered by one channel of a Carver PT2400 amplifier. There are 22 J62's with us. These are low-frequency systems that cover 70 Hz to 170 Hz and feature two JBL 2240 drivers and can handle 1200 watts of continuous RMS power. Thirty-two J63's handle the mids and highs. These loudspeakers cover 170 Hz to 20 kHz through four JBL 2202, three JBL 2445+2380, and four JBL 2402-05 drivers. In addition to the 161s, we have 12 J1 sub systems which feature 2240 drivers. The J23 is a full-range, compact utility speaker that uses a JBL 2445J two-inch compression driver mated with a JBL 2380A flat front biradial horn and JBL 2202 low-frequency cone transducer. These enclosures are of trapezoidal design and we have 12 of them on the road with us."

AMPLIFICATION: Carver PM 1200's and PT 2400's. All Jason loudspeaker systems are powered by customized and calibrated Carver amplifiers.

TIPS: Lilly says, "It's important to make sure that the clients get what they need and want. And the system must be totally reliable. We designed this system together with Bryan. We've worked with Bryan for many years now, and we know what he likes and what he expects from us. At this level of touring, nothing can be left to chance, and when there's 15,000 people or more waiting for the show to start, we must have the highest level of confidence in our equipment." A.

SPINAL TAP

continued from page 58

the Beta 57 did the trick."

Another hurdle presented itself at the Royal Albert Hall. In addition to the shows being taped for a television broadcast, the band wanted the distinct sounds produced by the hall's pipe organ. "The pipe organ is declared a national treasure," states Paterson, "and when some other guy went to record it, he bent some of the pipes. So now they won't let anybody near it. I had to mic around it — literally."

Despite the challenges and superloud sounds, Paterson is interested in working with Tap again. "They are a talented group of guys," he says. "Besides, Nigel is married to Jamie Lee Curtis, and it's always nice to see her walking around backstage."

....TO HERE



QMR is a new, competitively-priced and highly professional mixing console from DDA. Its flexibility and outstandingly clean signal path ensures that, no matter what your signal source or your recording format, you can be sure of an accurate recording.

With QMR, nothing is added to, or taken away from, the original performance. Its signal transparency will leave you thinking that there is, literally, nothing in between the perfect performance and the perfect recording.

WITHOUT LOSING, OR ADDING, A THING IN BETWEEN CIRCLE 24 ON FREE INFO CARD



MR



THE PERFECT LIVE GIG COUNTDOWN



TO SOME experienced musicians (hundreds of gigs), sometimes a gig is so bad, it would have turned out better if you hadn't shown up! Every gig presents a new collection of variables that can test even the most seasoned performer, and the key to soaring over these obstacles is control over these variables. To help get things under control during blastoff, follow these tips during countdown

T-MINUS 7: THE RIG

• All equipment, regardless of cost or quality, must be periodically maintained and checked over for possible problems. Find a tech (or learn enough about your gear) to do regular maintenance.

• Separate and secure all cables in the back of any racks before going on the road. Cable ties with screw holes for mounting on wood or metal, which are available at electronic supply stores, are ideal for securing cables.

• Use clip lights to illuminate the back of a rack while setting up (I've often wondered why someone doesn't invent a light pack that illuminates the back of the rack, where it's dark, instead of the front).

• Use a three-ring binder to keep info on your equipment — reset procedures for microprocessorcontrolled gear, repair records, schematics (worth their weight in gold on the road), etc. Owner's manuals and manufacturer/service facility/music store phone numbers are also useful. Trying to get a piece of gear repaired in an unfamiliar city is difficult; replacement equipment can be shipped out next-day air to many locations through UPS, Federal Express, DHL, or the US Postal Service.

T-MINUS 6: POWER HOOKUPS

• Carry an adequate supply of power strips, long and short extension cords, and ground lifts (use these to isolate ground problems, not to keep grounds permanently lifted while playing). A power outlet beyond the reach of your gear can be disastrous.

• Before hooking up, test the AC outlets. Radio Shack makes a suitable tester for \$5.99 (part number #22-101).

• Line filters and surge suppressors can prevent those embarrassing crashes of computer-based equipment (synths, drum machines, etc.). Check that any line conditioners have enough wattage to handle the load of your gear.

• A ground tree (see sidebar) can literally be a lifesaver by reducing the potential difference (voltage) between grounds on different pieces of gear. Nothing puts a damper on a guitarist's performance faster than being electrocuted.

• Though most topquality venues have adequate power hookups, some clubs may try to cut costs by using anyone with a pulse to do wiring, rather than hiring an electrical contractor. This can lead to bizarre wiring problems. The most common ones due to bad AC lines are:

— Reversed polarity

IF YOU WANT YOUR BAND-IN-A-VAN-TECHNICIAN CAREER TO SOAR, BE PREPARED BEFORE THE LAUNCH

By Mr. Christy





PROGRAM HIGHLIGHTS WILL INCLUDE...

- Music mixing for concerts and special events
- The application of concert-quality sound systems to theme parks and houses of worship
- System control integration: digital signal processors and computer linkage
- New stage monitoring technologies
- Industry trends and career opportunities

Program Chairman: Will Parry, SIGNAL PERFECTION, LTD. Facilities Coordinator: David Scheirman, CONCERT SOUND CONSULTANTS

Join the live sound industry's top professionals in a hands-on learning and networking experience. The program features free-ranging discussions and small-group intensive sessions.

WHERE: Chapman University Campus, Orange, CA WHEN: January 12-14, 1993

(preceding the NAMM Winter Market, Ar aheim, CA) **REGISTRATION FEE: \$650, includes meals and workshop materials**

FOR MORE INFORMATION CONTACT:

 SYNERGETIC AUDIO CONCEPTS

 TEL: (812) 995-8212
 FAX : (812) 995-2110

 12370 W. County Rd. #100N, Norman, IN 47264

PHOTOS BY JOE SPIEGE

CO-SPONSORED BY PRO SOUND NEWS AND SYN+AUD+CON

World Radio History

Mixing Art with Science

The 4th in a successful series... The concert sound industry's premier educational event! Featuring an instructional staff that includes veteran representatives from these top touring sound firms... • <u>AUDIO ANALYSTS</u> • <u>ELECTROTEC</u> • <u>MARYLAND SOUND</u> • <u>SHOWCO</u> <u>PLUS: Guest World-Class</u> Live Sound Engineers









(neutral and hot lines are switched)

— One of the nonground lines connects to ground

— There's a refrigerator, dimmer pack, or neon sign hooked up to the same circuit to which your gear connects (this causes a nasty 60 Hz buzz) — Improper breaker rating or inadequate wiring (if your band hooks up and a fuse or circuit breaker blows, look out!)

T-MINUS 5: CARRY SPARES

The following crucial items are prone to breakdown or loss, so make sure you have spares: • Fuses (keep a list of the fuse requirements for all of your equipment; tape a spare inside the case, for emergencies)

• IEC type (noncaptive) line cords

- Tubes
- Strings
- Picks
- Sticks
- Drum heads
- Microphones

• Cables, plugs, and jacks (never use jacks with plastic cases, go to the extra expense for metal)

• Batteries (to monitor battery life, keep a log and place a sticker showing the installation date on the side of each battery-powered device)

• Light bulbs for those remaining pieces of gear, such as optoelectronic footpedals, that don't use LEDs

- Power amps
- Speakers

• Disks, cartridges, and the other media used for synth patch storage

T-MINUS 4: SET LIST

Each band member (including the sound person and any techs) should have an accurate set list so that everyone starts together. Incidentally, the Nady Song Starter is a great tool; it gives a visual countdown for a song so that all band members can begin in sync.

T-MINUS 3: MAPS

This may sound obvious, but know where you're going *before* you get in the vehicle. Since group members are often coming from different directions, make sure they all have accurate *continued on page 78*



A ground tree is a central grounding location for all AC equipment on a particular circuit. Most AC receptacles are not grounded to earth ground, but to the circuit panel that, in turn (supposedly), goes to earth ground. Often, if there is a resistance between grounds on two lines, different voltages and ground loops can result, causing danger to the musician. Companies such as Furman Sound, Inc. (tel: 415-927-1225) make commercially available power supplies that properly ground equipment. Contact any of these companies for a products list, or consult with a good electrician on how to build your own ground tree.



IDP is not a food preservative.

ntegrated Dynamics Processing... the concept behind the new Symetrix 425 Dual Compressor/Limiter/Expander. Three totally *priority-interactive* processors in each of the Dual/Stereo channels provide the essential level control tools you need for any mixing, recording or sound reinforcement task. And it's dead quiet— super smooth.

Demo the new 425 and the power of IDP from the the company that has specialized in dynamics processing for over 15 years. Because, audio problems don't come in neat little boxes— Symetrix solutions do.

- Transparent soft-knee compressor
- · 'Brick wall' infinite ratio peak limiter
- Noise reducing downward expander
- Separate LED monitoring of all functions
- Functional ease of operation

Set a new standard.

Symetrix 4211 24th Avenue West Seattle. WA USA Tel 206-282-2555 1-800-288-8855 Fax 206-283-5504 CIRCLE 61 ON FREE INFO CARD

POWER INCARNATE

a' 15

SYSTEM 8000

Artist Systems proudly introduces the newest member in its ultra-high performance family of loudspeakers: the System 8000. Designed for portable use in larger clubs and discos, the 8000 is a bi-amped loudspeaker featuring a 3-way design of two 15" woofers, one 10" horn-loaded midrange driver, and a 1" liquid cooled compression driver for the high end. For power to pump up the house, get the new System 8000. Then bring the Power to your next gig.

Frequency Response: 38 Hz—17kHz ± 3 dB Power Handling: Low—800 W cont.; 3200 W peak Mid/High—300 W cont.; 1200 W peak Sound Pressure Level: 130 dB continuous; 136 dB peak



1150 Industrial Avenue Petaluma, CA 94952 707.778.8893 fax 707.778.6923 CIRCLE 10 ON FREE INFO CARD



Let's cut through it.

This sonically superior 17 lb. package pumps out an honest 1800 watts of real, full bandwidth, low distortion, FTC-rated power! With no compromises and no excuses.

The PA-1800 is a two rack space amplifier that uses next generation Switch Mode Power Supply technology. That means it's significantly more efficient than amplifiers with non-active linear supplies.

It delivers more than 400 watts per channel into 8 ohms, 650 watts per

channel into 4 ohms, and 1800 watts into 4 ohms bridged.

You get top quality sound, smooth and uncolored, the equal of any amp. With the best highs you'll ever live through. And unbelivably deep, detailed low end.

The PA-1800. It's lightweight. Powerful. It's virtually indestructible. It does the job.

And that's the truth.

All Stewart products are made in USA.



ELECTRONICS

11460 Sunrise Gold Circle Suite B Rancho Cordova, CA 95742 Phone(916) 635-3011 FAX (916) 635-1787

© 1992 Stewart Electronics

CIRCLE 71 ON FREE INFO CARD



LIVE MIXING FOR \$3000 OR LESS

SOMEBODY OUT there is listening. Say what you want about the music-electronics industry, but don't accuse it of not paying attention to the end user. At least not in recent years, and definitely not when it comes to mixing consoles.

During the past five years the industry realized that its consumers — across the spending spectrum were becoming more sophisticated, more demanding. Nowhere was this more true than in the lowerspending brackets. Consequently, companies had to get serious about adding useful features, and, yes, quality too, to lower-priced equipment. Bands, live mixers, solo artists, and producers all needed greater technology and signal-path flexibility...at a very competitive price.

And now they've got it. Even low-range boards offer the power of sweepable EQ, and switchable pre/post aux sends — and even submasters. In addition, many of these consoles provide for both studio and live sound applications — a fact which adds to their value. A look at some of the latest low-cost mixers will give you an idea of just how far we've come.

THE ART OF THE DEAL

Applied Research and

Technology (ART) has been making very competitive and versatile signal processors for years. They recently used this expertise to build a series of compact, inexpensive mixers incorporating features for both recording and sound reinforcement. The Phantom series of mixers includes the 1608, with 8 balanced mic/line inputs and 8 auxiliary line inputs; the 2408, with 16 balanced mic/line inputs and 8 auxiliary line inputs; and the 3208, with 24 balanced mic/line inputs and 8 auxiliary line inputs. Inputs feature 4-band equalization, 8 auxiliary pre/post sends switchable between four level controls. phantom power for condenser mics, and input muting switches that mute both the mix bus and auxiliary sends.

The outputs can be

configured into 4 subgroups. The master outputs also sum to a mono output so that controls are at hand for those situations where you have a central cluster or delay loudspeakers and must provide a convenient feed from your masters.

Distortion figures are very low, typically 0.007 percent, even in the most important signal path: from the mic input to the Main output. Noise buildup is still below -93 dB for that same signal path. Remarkable figures, especially for a mixer that includes an internal power supply. ART has given the all-important feature of roadworthiness special consideration in the design, including an allsteel chassis that can stand up to being thrown in and out of the band's van.

THE LATEST WORD IN ECONOMICAL MIXING CONSOLES By Wade McGregor with Hector G. LaTorre





BIG MACK(IE)

Mackie Designs made a tremendous impact in the low-cost mixer market with its first mixer, the 1604. Now they are about to go into production on the new 8-Bus Series. Available in early '93, this new mixer offers a 16-input mixer with 8 sub-groups for less than \$3000. The new 8-Bus series recording mixers are easily adapted to sound



The Ramsa WR-S4400

series of mixers

Silent Type

Take a good, close look at the strong silent types from Yorkville. When the going gets tough, Audiopro amps deliver tremendous punch without fuss, without strain and without driving you into bankruptcy.

In the middleweight category the AP1200 has several innovative features such as a switchable sub-sonic filter and switchable internal limiters that manage the gain for maximum dynamic range. The limiters automatically prevent audible clipping while allowing the transient spikes through unaffected. If this looks interesting on paper, wait till you hear what it sounds like! You can even have the AP1200 configured to deliver its maximum power into either 2 ohms or 4 ohms. The AP1200 delivers 1300 watts* of solid, clean power for the full 12 rounds, night after night, with complete reliability and uncompromised fidelity.

With 500, 1300 or 2400 watt models, there's an Audiopro Amp to suit your needs and budget. See and hear them at your dealer soon and, oh yes, bring your checkbook. You're going to like what you hear.

*Continuous Average Power Bridged into 4 onms

YORKVILLE SOUND LTD. 80 Midwest Road Scarborough, Ontario M1P 4R2



YORKVILLE SOUND INC 4600 Witmer Industrial Estate, Unit #1 Niacara Falls, New York 14305

CIRCLE 75 ON FREE INFO CARD

AUDIOPRO PULSE 🛲 Mirmaix STAGE Maina Stage Maina Stage State BETA

reinforcement use and feature a sweepable midrange EQ - the high-mid is fully parametric, adding a variable Q control. There are activity (-20 dB) and peak indicators on the inputs, just above the 100mm fader and mute switch. Channel assignments to pairs of submasters or the output masters allow flexible routing for more sophisticated FOH work. The balanced XLR main outputs are capable of up to +28 dB before clipping, which is very important when working with dynamic live performers. The 8-Bus mixers will also be available in 24- and 32input versions.

TWO FOR THE ROAD

Yamaha has just upgraded its 4-bus sound reinforcement mixers, now called the MC04II series. Yamaha is well known for building robust mixers for touring, and the MC04II series uses many of the ideas from the more expensive Yamaha mixers. One of these features, unusual in this price range, is the matrix output. A 6 x 2 matrix mixer is fed from each of the group and master faders, allowing two independent mixes to be created for central clusters, backstage, or recording feeds. Another feature that can be very handy, especially for the warm-up act, is the ability to bus two consoles together without using up inputs or auxiliary returns. This can also allow all the PFL/solo audio to be carried through to the main mixer. The input channels include 4-band EQ with sweepable mids, a high-pass filter, 4 auxiliary sends, and inserts on input, groups and masters. The series starts at about \$3000 and includes frames from 12 to 32 inputs.

The Ramsa WR-\$4412/16/24 series of mixers features 100mm faders for mixing control of the 12,

TAS N THE MIX

1 Part Sonic Transparency • 1 Part User Transparency • 1 Part Comprehensive Control • 1 Part Ingenuity

MegaMix InBoard Series Automation Systems

Every now and then the wonders of art and technology combine to create audio bliss. The new MegaMix InBoard Series system is one such wonder. With MegaMix on your existing console, your studio will come to life with ex-panded features and comprehensive control. The InBoard Series will allow true creativity to transpire in your mixes and at the same time give your studio a new competitive edge. With enhanced traditional features and new innovations like recallable, automatable dynamics on each channel, the MegaMix InBoard Series will land you in mix nirvana.

Conneaut Audio Devices



MegaMix InBoard Series Software is easy to use in real time and offers more off-line editing than all the others combined

MegaMix InBoard Series VCA Faders offer sonic transparency and a completely automated dynamics section, with 10micro second attack time on each channel. Ali MegaMix systems utilize only the MTA1537 'Class A' VCA.

The new MegaMix MR! system is a full featured, in-console automation system, easily installed at a low cost.



Conneaut Audio Devices CIRCLE 25 ON FREE INFO CARD

For more information on Amage MegaMix call 1-800-762-9266

0200G

oco?

Conneaut Audio Devices, a division of CTI Audio, Inc. Harbor and Jackson Streets, Conneaut, OH 44030 Phone: 216-593-1111 . Fax: 216-593-5395

EVEN LOW-RANGE BOARDS OFFER THE POWER OF SWEEPABLE EQ AND SWITCHABLE PRE/POST AUX SENDS. IN ADDITION, MANY OF THESE CONSOLES PROVIDE FOR BOTH STUDIO AND LIVE SOUND APPLICATIONS.

16 or 24 balanced inputs, 4

subgroup, and 2 master bal-

anced XLR outputs. The

wide gain-range (64 dB) of

the inputs will allow a con-

tinuous control that accepts

signals from low-level mics

to line-level sources without

resorting to input pads. A

dedicated PFL meter allows

the user to check signal levels throughout the mixer without losing sight of the output levels. The input equalization is 3-band, with sweepable midrange. Special care has been taken in

the design of these mixers to reduce the switch breakthrough that has been

known to cause problems -

BOOD

especially for very high-gain sound reinforcement systems. It is

best to keep your muted channels out of the main mix, isn't it?

BUT WAIT! THERE'S MORE

There are a number of other mixers in this price range, such as the Ross Systems RCS802, 12, 16 and RCS2402 series; the TASCAM M1016 and M1024; the DOD Elec-

tronics 822, 1222, and 1642; and the Soundcraft Spirit Series, including the new Spirit 4, which adds 4 subgroups and 5 auxiliary sends to the list of Spirit features. Yorkville Sound has the AP series of mixers — from the AP208 to the AP216 — that also includes a digital effects processor.

Fostex has a series of recording consoles in this price range and has recently announced its new DCM-100 rack-mount audio mixer and MIXTAB. The MIXTAB is the controller with dedicated switches, knobs, and faders for sending MIDI control messages to the DCM-100. Using a long-distance MIDI link, this system could allow all of your audio signals to remain onstage and simply bring the controls (faders, EQ knobs, etc.) out into the audience for mixing. The future of live sound? The basic system with 8 stereo channels and snapshot automation is expected to cost less than \$1400.

The Yamaha MC320411 mixing console is part of its MC0411 series of mixers.

SERIES COMES TOGETHER !

7000

- HIGH PERFORMANCE ANALOG & DIGITAL AUDIO GENERATOR / ANALYSER
- COMPLETE SELF-CONTAINED
 PROGRAMMABLE SYSTEM
- COMBINES STATE-OF-THE-ART ANALOG
 & DIGITAL PROCESSING WITH POWERFUL
 COMPUTING RESOURCES

HIGH PERFORMANCE AUDIO TEST EQUIPMENT

* Optional





- PORTABLE *BATTERY OPERATED
- VERY HIGH PERFORMANCE, LOW COST

Amber Electro Design

\$3 300

- COMPREHENSIVE MEASUREMENT
- & WEIGHTING FILTER OPTIONS • LIGHT WEIGHT

• PROGRAMMABLE (IEEE 488)

\$6 900

- MODULAR CONSTRUCTION, EASY OPTION CONFIGURATION & EASY MAINTENANCE
- FRONT PANEL CONTROLLABLE




Pick up your room.

-10 12 5 3 6 6 fg

ting up with untidy products. Now's e been getting away with it too long th RSP ncredible new Intelliverb¹¹. he hance to tidy up and reach a new livi verb features a 24 bit professional DSP vienau Room Proce u create your own rooms for reverbs. iver 50 reverb parameters, you determine the room size, p ing source, and position the listener room!! Build more rooms without increasing your mortgag emory locations to store your presets, telliverb Virtual Room algorithms are second to none, and easy to control with complete mixing capabilities in a controllable package. Our unique "seperation" parameter en ou to control the stereo width! You've r had it so neat and tidy before. You'll be astonished by the clinity of the intelliverb. Not only do you get the most natural room sounds available, but 2.5 seconds of memory provide impeccable time domain effects, including chorus, delay, pitch shifting and ducking. The Intelliverb provides an incredible dynamic range of 105dB, and features digital HUSH¹¹¹ noise reduction licensed by HUSH Systems. Our 27MHz processor puts over 80 million operations per second at your fingertips. Sigma-Delta A/D conversion provides 64 times over sampling, and a

POON

44.1KHz sampling rate feeds the high speed, zero wait state SRAM data memory for maximum performance. Get yourself the best in sound quality and control, and lose those messy reverbs. Pick up your room at your RSP Technologies dealer.



2870 Technology Drive Rochester Hills, MI 48309 313 853 3055

HUSH Licensed by HUSH

CIRCLE 49 ON FREE INFO CARD

YAMAHA

Sound will obey.

The dog will never be the intelligent, obedient and loyal servant you would like him to be. Your music, however, will do whatever you want. How come?

Quick Edit. It's a mode of programming in the SY85 AWM synthesizer. It allows you to, very simply, create any sound you can imagine. How simply?

You can layer, split, and cross fade up to four voices at once. Eight control sliders let you modify any parameter you like. It's easy and it's quick, hence the name. You need more control? While playing live those same sliders can modify the effects, filters, attack, release, and balance, all in real-time. And the SY85 has SIMMs as well as battery-backed, expandable RAM, MIDI capability and, of course, our distinctly superior sound quality and playability. It is, after all, a Yamaha.

Would you like more information? Call us. Obediently, we'll send you our brochure. 1-800-932-0001, extension 100. **The SY85**.

WORKSHOP NOISE



o, really, I don't get annoyed at everything. It just sometimes seems that I do. At times, being the owner of a mid-sized project studio, venting my emotions in a rather loud fashion on occasion seems like an inherent right of mine. Several months ago I found myself *quite* annoyed and I realized I had reached the point where a major technical overhaul was due.

First, a little background. Oxford Studio evolved over the course of about three years. When I found a good piece of equipment, used or on sale, that I needed and I had the funds available, I bought it. Through this method of equipping the studio, I developed several philosophies. One of which is: if keyboard X was good

> How I made my unbalanced studio virtually noise free

enough for hit records three years ago, I can put it to good use now. Don't get me wrong, I would not be opposed to owning all the hippest toys out there. But for me, building my studio has been a process of deciding on what I need, then going out and finding it at or under budget.

Throughout this evolution, I've always tried to keep my studio as quiet as possible. I always believed I had done a reasonable job of it. No out-of-control humming, buzzing, or other assorted nonsense. Then one day, after adding some new gear, I began hearing the dreaded buzze, or, more correctly, the dreaded buzzes. After spending too much time finding a relatively minor problem, I decided it was time to go through the entire studio from the ground up.

TWISTS AND TURNS

The first thing I did was to empty all my racks and cabinets. It was amazing how, over a period of three years, all my cables — audio, power, MIDI, and video — had become severely knotted. After the untangling process, I realized that this project was going to be a little more involved than I had originally thought. Off I went to the local library and bookstores to arm myself with as much relevant material as I could find. Fortunately for me, a bookstore in my area carries almost the entire Howard Sams catalog, a series I highly recommend. Next I locked myself up in the studio and read.

There is a tremendous amount of material out there covering balanced equipment. Unfortunately, my studio is unbalanced, as are many other project studios. To these "other" studios, workable solutions are very difficult to come by. So, between the texts, talking with other studio owners, and consulting with friends who are well respected in the audio field, I arrived at what I now feel to be a quiet unbalanced room.

Most engineering professionals would probably argue with that last statement. For many in the pro audio world a "quiet



unbalanced" studio is an oxymoron. But, once again, philosophy comes into play here. Or maybe it's a financial position. Okay, how about a philosophically-oriented, financially-motivated juxtaposed concept? Anyway, for most real world applications, I know a quiet unbalanced studio can exist so I began rebuilding at the bottom.

FIRMLY ON THE GROUND

An ideal ground would have the capacity to source and sink infinite amounts of energy within any resistance. The ground should reach to all parts of a facility and to the equipment which requires the ground. This is, of course, an *ideal* ground. My grounding method involves what is called "brute grounding." I designed my facility with what some may consider overkill in the grounding department.

Installing the 3/0 copper cable was a little bit of a challenge. It is a fairly thick (1/2-inch) copper strand cable. I actually ran two lengths of this, about 25 feet each, off of the same eight-foot ground rod. Inside the studio, the ground wire is mounted on the walls behind all the rack-mounted equipment. I then attached about 50 eightgauge copper wires to the 3/0 ground with micro pipe clamps. My thinking was that I wanted everything to be *very* accessible, with all cable runs as short as possible. Again, I knew using a wire this thick for my ground was overkill, but there were two reasons why I did this:

— Unbalanced equipment is very prone to noise. More than that, it is going to *make* noise. The best you can do is to keep it to what you feel is an acceptable level.

— Troubleshooting. Whenever any noise does pop up now, I can eliminate my ground as a possible suspect. To be that sure of something in a smaller studio where you are in charge of everything is a great comfort.

Once I had an adequate ground installed, I then turned to the power.

ELECTRICAL SYSTEM

Here I must stress the importance of consulting with someone who knows what he or she is doing — preferably a licensed electrician. The cost for a con-

on a

Song

Production

Contest

1st

Annual

sultation will be more than made up for in the safe operation of your studio. Fried samplers generally produce less than desirable results. Fried engineers are totally unacceptable.

The power scenario in my studio is as follows: The entire room is run off a single 30 amp fuse; the electricity terminates in several duplex outlets around the room (all two-conductor outlets with no ground); the total draw of my equipment is about 2400 watts, excluding a laser printer and a microwave oven (for the coffee).

Once it was determined that I had an adequate amount of electricity coming into the room, the next step was deciding on how to distribute it. It was unfortunate that all the power to the studio came off one circuit, having an isolated line to work with comes in handy when working with audio noise. On the other hand, having the entire room fed from one circuit eliminates the possibility of additional loops being formed because all of the electricity in the studio is now on the same phase power line.

WIN MACKIE'S NEW 24x8 CONSOLE OR AN OTTO-1604 MIDI AUTOMATION ADD-ON BOARD!

FREE COMMEMORATIVE T-SHIRT

JUST FOR ENTERING! It's no secret that our mixers are already being used to record and mix down major label CD productions. But we'd also like to hear what YOU'RE creating with our mixers! So we're holding the 1st Annual "Mixed On A Mackie" Song Production Contest. Grand Prize is our brand new 24-8-2

console. Four runners up will receive OTTO-1604 add-on MIDI Automation boards for the CR-1604. All five winners will be featured on a Mackie compilation CD. This contest is open to anyone who owns and uses a Mackie CR-1604 or MicroSeries 1202. Because sound quality is a major judging criteria, we're asking for DAT tapes only. But because we're musicians ourselves and know how much those little bundles of rust cost, we'll return all DAT tapes after the contest is over....along with a free multi-color "Mixed on a Mackie"T-shirt. Send for your Official Entry Packet today!

ANNOUNCING A SONG CONTEST JUST FOR MACKIE CR-1604 & MS1202 MIXER OWNERS!

If you've done a killer mix on a CR-1604 or MS1202, we'd like to hear it. Mackie Designs and our Panel of Celebrity Judges will be evaluating submissions primarily on the basis of sonic quality and mixing skill, so you can send us anything from an original sequencer masterpiece to a live big band recording.

Entries must be: I) recorded & mixed on a Mackie; 2) 5 minutes or less in length; 3) submitted on a DAT tape. Plus, we'll RETURN your DAT tape and send you a free commemorative T-shirt just for participating! NOTE: You **must** use our Official Entry Packet. Circle the reader card number below or call our toll-free number for faster response. Entry deadline is **January 31** ** **1993**.



SOME Sound Forecasts...

...Today's low...20Hz to 35Hz Whether you're looking to custom tailor that mix or optimize the acoustics of a really tough room, nothing gives you the control of Gemini PVX Series equalizers!

...Today's high...16kHz to 20kHz

Both PVX Series Equalizers feature:

- Single space rack mount
- ±12dB of boost/cut
- Center detented potentiometers
- LED level indicators
- Low cut & bypass switching
- Gain control to compensate for the equalization
- Rugged steel chassis construction

Corporate Offices: 1100 Milik Street, Carteret, NJ 07008 ■ Phone 908-969-9000 ■ FAX 908-969-9090 Florida Branch: Gemini South ■ 2848J Stirling Rd, Hollywood, FL 33020 ■ 305-920-1400 ■ FAX-305-920-4105



CIRCLE 38 ON FREE INFO CARD World Radio History





LEARN THE ART OF RECORDING

THE RECORDING WORKSHOP IS THE WORLD'S LEADING SCHOOL FOR "HANDS-ON" TRAINING IN MUSIC RECORDING TECHNOLOGY. IN OUR INTENSIVE, 5-WEEK PROGRAM, YOU'LL LEARN THE CREATIVE OPERATION OF PROFESSIONAL RECORDING EQUIPMENT. WE'LL TEACH YOU THE JOB SKILLS NEEDED TO SUCCESSFULLY START YOUR CAREER AS A RECORDING ENGINEER, PRODUCER OR STUDIO MUSICIAN.

- ► 6 STUDIOS FILLED WITH THE LATEST EQUIP-MENT: DIGITAL RECORDING TO HARD DISK, DAT MASTERING, AUTOMATED MIXING, MIDI AND MUCH MORE
- SMALL CLASSES, PERSONAL ATTENTION



- ► INTENSIVE, 5-WEEK PROGRAM GIVES YOU MORE QUALITY, IN-STUDIO EXPERIENCE THAN MOST LONG-TERM SCHOOLS
- JOB/INTERNSHIP PLACEMENT ASSISTANCE
 ► FINANCIAL AID AVAILABLE
 ► LOW-COST, ON CAMPUS HOUSING
- REQUIRED LOW-COST, ON CAMPUS HOUSING FOR FREE BROCHURE, PLEASE CALL OR WRITE TODAY

THE RECORDING WORKSHOP 1-800-848-9900 1-614-663-2510 445-J MASSIEVILLE ROAD, CHILLICOTHE, OHIO 45601 OHIO STATE BOARD OF PROPRIETARY SCHOOL REGISTRATION #80-07-0696T WORKSHOP NOISE

GETTING WIRED

I decided that it would be more costeffective in my situation to attempt to isolate the electricity myself. I used two Tripp-Lite line conditioners with four isolated duplex banks. Each bank covers a certain section of the room. This way, if there is any noise problem, I can isolate it quickly. Since the power coming into the room had no ground, I tied the line conditioners into the 3/0 ground wire and used that as my electrical ground. Of course, my electrician assured me that this was safe. I grouped "like" kinds of equipment together when deciding what to plug into which isolated bank as follows:

----Video-related gear in a bank of their own. (This included the Mac.)

- -Signal processing equipment.
- -Sound modules and keyboards.

By doing this, several problems have been eliminated. My computer no longer produces that nasty little highpitched video whine through my consoles. My reverbs have less noise to process now. Strange intermittent noises in the system have all but been eliminated. What it comes down to is this: one more set of variables has been defined. I know what the system will sound like when I power it up.

GROUND LOOPS

A ground loop is, simply put, a conductive loop that is formed when the earth or grounded conductive elements become part of the circuit. Everyone I spoke to on this topic had their own views on how to handle it. My answer was covered earlier in the section on the grounding system. My reasoning is this: in an unbalanced environment, using some, most or all semi pro equipment, you are going to have loops. Add several patch bays providing complete interconnection and it's a fact of life. My solution was to provide a path to the ground so solid and oversized that rather than forming an audible loop, any stray noise would choose the path of least resistance - the 3/0 ground.

TAKING THE RIGHT ROUTES

One of the problems I ran into before the studio was overhauled was wire. All the wire. Audio met MIDI which then crossed over to the power and video lines. It was not a pretty sight. It wasn't planned that way, it just evolved. Getting any piece of equipment out of its rack quickly and simply could not hap-

continued on page 78

CIRCLE 68 ON FREE INFO CARD

When it comes to Music Equipment Mail Order Companies, there's only one number one, and ...

We're #1 HERE'S WHY!

#1 In Shipping...

 We ship more music gear than anyone — over 12,000 shipments a month to over 125,000 satisfied customers.

#1 In Service...

- Our 30-day money back guarantee is legendary. This no-risk advantage lets you buy with complete confidence.
- Toll-free ordering, knowledgeable technical support, and customer service that's dedicated to your complete satisfaction.
- On-line order entry for immediate up-tothe-minute inventory and pricing info.
- Huge multi-million dollar inventory assuring you of fast delivery — most orders are shipped within 24 hours!
- Two-day Federal Express delivery for just an additional \$5.95. Order it by noon and we ship it the same day.

#1 In Product Presentation...

- Quality photography with lots of color photos, so you get to see every knob and button before you buy.
- Copy information that's written in musicians terms, to help you make intelligent buying decisions.
- Full page product reviews on the latest products, keeping you up to date on all the hot, new gear.
- Magazine quality product articles give you more indepth information, and great ideas for getting the best from your setup.

#1 In Circulation...

• We mail three or more new editions a year totaling over 2,000,000 catalogs to musicians like yourself.

FREE 1 Year Subscription ...

Now's your chance to join hundreds of thousands of satisfied musicians in receiving the #1 catalog, absolutely free! Call 503 772-5173 or mail in this coupon to Musician's Friend, P.O. Box 4520, Dept. 107, Medford, OR 97501.

Please send me the next 3 editions of Musician's Friend totally FREE of charge!



Here's a sampling of just a few of the quality brands featured in every Musician's Friend catalog ...

)ıgiTec	h <u>ADA</u>	KORG
Marshall	Rola	nd D	3055	Jender
TASC	M Striff Seyr	mour ican tups	FOS	tex Ex
exicon SAN	SON W	Jackso	Charate	L.C. electroni
WASHBURN	Epiphone	Gibson	FURMAN	l banez

n	NAME	
satisfied musicians in receiv - 5173 or mail in this coupon ord, OR 97501.	Address	
Friend totally FREE of charge!	Сіту	ZIP

GIG COUNTDOWN

continued from page 64

maps with the name of the club, a contact person, phone number, and address. A late band member can be a disaster; be at the gig at least one hour ahead of time to prepare. AAA membership can be invaluable for the map and road services they provide. Don't drive around and hope to find some place — stop and get directions the second there's any confusion.

T-MINUS 2: STAGE SETUP

• Don't experiment with positioning! Measure off the ideal distances (for both large stage and small stage setups) between different band mem-

bers' equipment for ideal audio and visual contact.

• A m plifier power should be equal. All too often there is one powerczar that just has to have enough volume coming out of his or her rig to fill the whole arena.

One approach is to have enough power to fill the stage and let the PA do the rest of the work. The other approach is for each performer to have a PA system that can fill the entire audience. The first approach is more realistic and far less costly.

• The sound engineer should have a standard, familiar cabling and configuration setup. Hand out several copies of this configuration to the stage crew so they can work in tandem with the engineer.

T-MINUS 1: THE SOUND

• A good engineer knows what the arena will sound like before ever turning the power on. This assessment is based on the type and quality of the gear, and on positioning, the arena's architecture, and speaker placement. Try to use a consistent setup so you can develop an intimate knowledge of your gear.

• Don't use one piece of gear to compensate for a problem elsewhere in the audio chain. For example, don't boost certain frequencies with an equalizer when the real problem lies in an improper crossover frequency or poor mic selection. Another example is under- or overdriving the mixer's input stage by improper input attenuation, then trying to compensate with the channel fader.

• Avoid feedback at all costs. Feedback desensitizes the ear to musical subtleties. It can take a whole evening (or lifetime) to recover from a deafening howl. Decibels are a serious business that can have such serious consequences as tinnitus, or temporary or permanent deafness. For expert advice on this subject, contact the House Ear Institute at 213-483-9930, or your nearest hearing specialist.

• All musicians should be able to hear themselves on stage since music requires a constant interaction between the fingers and ears. Proper monitor and amp placement is the key. Slightly elevating floor monitors can eliminate floor resonance, help

direct the sound to the ear, and minimize feedback.

• Turn down rather than turn up. If someone can't be heard, first turn down the elements that are too loud. This also applies for equalization — cut,

rather than boost, to solve response problems.

• All band members must be able to see each other and the sound engineer. Eye contact plays a most important role in starting, stopping, soloing, speed, and inflections. Music may end up being aural, but there should always be a visual conductor.

• Always have a sound check. Regardless of sleet, hail, rain, snow, or drunken stupor, the group (and engineer) should know the venue before you start playing. Always have a test flight, or don't test the parachute on the first jump!

BLASTOFF!

It can take a whole evening

(or lifetime)

to recover from a

deafening howl.

As someone who's had his share of good and bad gigs, I hope these tips help you along. As to making sure you get paid...well, that's a whole other subject!

Mr. Christy is a guitarist, film scorer, and instructor. He does a lot of club gigs in Southern California with some of the musicians from Oingo Boingo, and has had every possible live gig disaster and triumph happen at least once. When he's off the road, he scores films.

UNBALANCED

continued from page 76

pen. Troubleshooting was nearly impossible. One of the most effective ways to prevent this from happening to you is to plan ahead. What I did was as follows:

I have a "U" shaped area where I work. I installed two-inch Electrical Metal Tubing (EMT) through the surface where my consoles, keyboards, and racks sit. Although EMT provides almost no shielding from magnetic fields, it does provide fairly good shielding from electrical fields. It also ensures that the wires will cross each others' paths at 90 degree angles. The tubing for the MIDI, for example, passes through the audio and power raceways. Since all of my wires carrying a different type of signal now pass each other at perfect right angles, signal interference is kept to a bare minimum. All of the raceways themselves are mounted to the walls and separated from each other by about 18 inches. I would have preferred more space between them but, given the room I had to work with, I felt that this was an acceptable compromise.

REASSEMBLING

At this point, I felt that I had a good framework and that I could get all my equipment off the floor and back into the racks. Working backwards from the power amp, I installed a single piece of equipment at a time, listening for any noise that should not be there. With nothing plugged into my boards except the power amp and the speakers, I had a good idea of what the minimum amount of noise to expect was to be. Doing this provided a good reference point for me.

Physically wiring everything back in was made much easier due to the fact that I had all the vertical EMT and the raceways in place. No trying to figure out what wires should go where. During the reassembly I kept notes on which gear gave me any problems as far as noise was concerned. Some of my synths and samplers are just inherently a little noisier than others. It helps to know what everything in your studio is or is not capable of.

Now that it doesn't look like a construction site any more, Oxford Studio is running much quieter and more efficiently. When the next crisis arises (it always does), tracking it down will be faster, which means fixing it will be faster. And contrary to some popular myths, you *can* have a quiet unbalanced studio.

World Radio History



THE PROFESSIONAL'S PERSONAL PROCESSOR

THE LEXICON PCM-70: BIG STUDIO PERFORMANCE ON A PERSONAL STUDIO BUDGET

For over 20 years, audio professionals have relied on the superb quality and control of Lexicon Digital Effects Processors. So much so that today, 80% of the most successful productions are processed with a Lexicon.

Designed especially for the discriminating musician and recording engineer, the legendary PCM-70 offers brilliant sound quality (*ii has the* same algorithms as the acclaimed 224XL) — and Lexicon's exclusive Dynamic MIDI "control. Since 1986, the artist-friendly '70 has set he standard for processing effects and MIDI control which remains unrivalled by any other brand.

On the job, the '70 gives you the vital sound effects you need. Superb, lush reverbs. Shimmering 6-voice chorusing. Mind-altering multi-band

delays. With a PCM-70 in your rack, any ordinary instrumental or vocal track can be instantly rescued from mediocrity – effortlessly, because the '70's front panel

is designed for creative people who want unique, useful results. Does this mean there's a Nerd-gratifying interface with a plethora of inspiration-killing buttons? Of course not. The '70 is renowned for its reverb; chambers,

The '70 is renowned for its reverb; chambers, halls, plates, gates and inverse room - great sounds for primary lead vocals or percussion tracks. Undoubtedly, you've heard the PCM-70 on chart-topping songs or seen it lurking in a favorite player's stage rack. It's no surprise that in the world's best recording studios (*large or small*), the PCM-70 has earned a reputation as the professional musician's most useful effects processor.

When you audition a PCM-70 at your Lexi-

con Pro Audio dealer, check-out our surprisingly affordable LXP Series too. Whichever you choose, *any* Lexicon will deliver the finest sound available.



60000

CIRCLE 52 ON FREE INFO CARD

LEXICON INC., 100 BEAVER STHEET, WALTHAM, MA., 02154-8425 TEL: (617) 736-0300 FAX: (617) 891-0340

EQ DECEMBER 81

World Radio History

Tooling a "Tone Tweaker"

Where's the all-important midrange control?

Glad you asked. The Tone Tweaker is a single-stage parametric EQ that sweeps over an approximate 200:1 range (switchable between 80 Hz-16 kHz and 40 Hz-8 kHz), with up to \pm 15 dB boost or cut at the selected frequency. This is a true parametric, not a pseudo-parametric, so it also has a resonance control with bandwidth from 0.2 to 4 octaves. Patch this handy device in between an instrument output or tape track and your mixer, and you'll regain control of the midrange.

HOW IT WORKS

The Tone Tweaker places a state-variable bandpass filter (made up of IC2 and IC3A) in op amp IC1B's feedback loop. R15 determines whether IC1B will boost or cut the filtered signal, to produce a peak or notch respectively.

Input stage IC1A provides a high input impedance (100k) and inverts the signal's polarity. Since the filter itself causes a second inversion, the output signal's polarity matches that of the input. IC3B is unused, but wired as a unity-gain buffer to prevent it from oscillating or doing other nasty things.

BUILDING IT

This circuit can run at fairly high gain,

A single-stage true parametric EQ can add range to your midrange BY CRAIG ANDERTON so use short leads where possible (especially the leads to the pots and S1), avoid running inputs next to outputs, and run shielded cable to the input and output jacks. For best results, lay out a circuit board or build the circuit on a small piece of perf board. (If you're not familiar with electronic construction techniques, check out my book *Electronic Projects* for Musicians, published by AMSCO.)

With S1 open, the Tone Tweaker covers around 80 Hz to 16 kHz. Closing S1, which parallels C4 and C5 with C2 and C3 respectively, drops the range an octave. These capacitors should be matched, or at least lowtolerance, precision parts.

To simplify the schematic, IC power connections are not shown. However, each op amp requires +15V at pin 8 and -15V at pin 4. Any \pm 15V power supply capable of delivering \pm 50 mA will do. Battery operation and lower voltages can also be used, down to about \pm 6V; however, this decreases the headroom and can result in more noise (and possibly clipping distortion). No matter what kind of power supply you use, add a 100 nF (0.1µF) capacitor from each supply lead to ground at the point where the power supply leads feed the circuit board.

The circuit is small enough that four of them can fit behind a rack panel (in fact, I use three sets of fourstage parametrics in my own studio). A stand-alone unit, however, also does the job.



USING THE TONE TWEAKER

Use the Tone Tweaker as you would any midrange equalizer. To avoid "plastic-sounding" EQ, don't boost too much — try to fix the problem at the source itself. Lots of resonance and boosting can lead to distortion; the maximum input signal level for any frequency under these conditions is about 2V peak-to-peak. With minimum resonance and no boost, you can feed in at least 20V peak-to-peak.

Remember, too, that cutting is as powerful a tool as boosting, if not more so. If you need to boost the treble and bass, try cutting the midrange instead. This will reduce the risk of distortion and often give a "sweeter" sound.

You can also put two or more Tone Tweakers in series (i.e., the output of one feeds the input of the next) for a multi-band parametric EO. For example, one stage could cut the lower midrange, while the other boosts the upper midrange.

FINDING PARTS

Many parts are available at Radio Shack (see parts list); you can substitute the slightly noisier TL082 for the 5532 if the latter is unavailable.

The hardest part to find is R16. the dual ganged pot. Ideally, you would want it to have a reverse audio taper, but such pots - especially dual pots - are rare. Instead, you can either use a dual linear pot and deal with the fact that the lower frequencies cover most of the pot's travel, or use an audio taper pot and wire it in reverse. This gives the right taper, but turning the control clockwise lowers the filter frequency, which is the opposite of what you'd expect.

Finally, Paia Electronics (3200 Teakwood Lane, Edmond, OK 73013; tel. 405-340-6300) offers a kit of parts for a four-stage parametric EQ. Although not exactly the same design as this one (the Paia version uses quad op amps to save space, eliminates the input buffer, and does not include a range switch), you'll get pretty much the same results without having to chase down parts. Paia also makes a ±15V power supply suitable for powering the Tone Tweaker, as well as

Parts List

Radio Shack part numbers are indicated with #.

CAPACITORS

(50 o	r more working volts, mylar or polystyrene
prefe	rred except as noted)
ต่	75 pF ceramic (#272-121, 50 pF, is also acceptable)
C2- C5	Matched 10 nF (0.01µF) metal film or mylar capacitors (#272-1053)
C6, C7	220nF (0.22µF) ceromic or mylor (#272-1070)
68	10pF tantalum or electrolytic (#272-1913)
RESIST (5 pe resist	rcent tolerance, metal film preferred for fixe
	470Ω (#271-019)
k3, k4 R5	1k (#271-023) 22¤ (#271-038)

many of the other projects covered in this column.

So what are you waiting for? Plug a Tone Tweaker into your budget mixer and give some snap to a synth. make a vocal more intelligible, or eliminate the annoying "honk" from a low-budget piano sample - and that's just the beginning.

	NUNFT	100k (=271043)		
	R15	100k linear pot (#271-092)		
	R16A+B	Dual ganged 100k pot (#271-1732; see text		
	R17	500k linear pot		
ne		(#271-211, 1 Meg, is acceptable)		
	MECHA	NICAL PARTS		
	S1A+B	Double-pole, double-throw		
		(DPDT) switch (#275-663)		
	31	Mono, closed circuit 1/4" phone jack		
		(#274-255)		
	12	Mono, open circuit 1/4" phone jack		
3)		(#274-252)		
fixed	OTHER PARTS			
IIKOU	101-103	NE5532 or TL082 (#276-1715)		
		internally-compensated dual op amp		
	441	16 1 (1071) 20052 (1)		

R6-R14 100k (#271-045)

IC sockets (#276-1995), perfboard, rack case, Misc wire, knobs, power supply, etc.

Craig Anderton, EQ's west coast editor, is the author of eleven books including MIDI for Musicians and has worked on 10 recordings. His latest, Forward Motion (distributed by MCA). was done with new age keyboardist Spencer Brewer.



Texas los rum pts

Aerox Co

Mc Yosh Lans

ha mo

Duron La cilitories Motoro a

General Ellistic

Phone: (619) 451-1799

FAX: (619) 451-2799

Figure

Tascam's new 8mm digital audio products get the industry's attention

THE TASCAM division of Teac is the leading producer of recording equipment in the world, so it is not surprising to find it bringing a cost-effective digital multitrack tape recorder to market. Of course, the main competition is the Alesis ADAT, but Tascam already has more to contend with: at the recent AES show Alexis announced an agreement with Fostex, which should result in a Fostex-produced digital recorder that conforms to the ADAT format. This doesn't seem to have intimidated Tascam, however; it has its own clear concept of how this type of recorder should be built, and has pursued this vision in the form of the new DA-88 (list price: \$4499)

The Tascam unit is due to hit the streets sometime in early 1993, at which time this magazine and others will undoubtedly put it through its paces. Before that happens, it still pays to look at how Tascam has taken the Alesis challenge and at how it expects the project recording marketplace to do its digital recording in the years ahead.

First off, the DA-88 does its recording on 8mm Hi8 videotape. Hi8 is the highest grade of 8mm video tape, and was chosen, according to Tascam, because of the extremely high density of the metal-particle medium, which allows long recording times with low dropout and error rates. In fact, the DA-88 is capable of recording up to 100 minutes on a 120-minute tape. Another consideration for Tascam engineers was tracking; the Hi8 format embeds its control track within each helical scan of the head, rather than as a separate longitudinal track According to Tascam, this is how they achieve perfect tracking and interchangeability between machines (see "S-VIIS vs. 8mm" story in this issue).

A front panel switch is provided for switching between 44.1 and 48 kHz sampling rates, with variable pitch of $\pm 6\%$ available at either rate. Since most of the time the DA-88 will be used for creation of multitrack recordings that are then mixed down through analog mixers, the 48 kHz rate is likely to get the most use (because of better frequency response). At some point, alldigital mixers may become available, in which case it could be cleaner to use 44.1 so that a sample rate conversion can be avoided when mastering to CD.

TOUGH TRANSPORTS

Tascam makes a point of talking about the ruggedness of its 8mm transport. The company does have quite a bit of experience in the production of this transport format, since it makes an 8mm unit that's the heart of the "black box" flight recorders in many commercial aircraft, and other specialized units for NASA (including the space shuttles). You can't get much tougher than that! While the particular transport in this unit is not the one found in the black boxes, at least the knowledge gained in production of rugged compact transports has been applied to the DA-88. The entire transport is built by Tascam at its factories. Tapes may be preformatted, or can be formatted on the fly — a real convenience when you're dealing with 100 minutes of recording time.

The front panel will seem instantly familiar to anyone who has operated a Tascam recorder of recent vintage. Two memories hold location points that can be used for Rehearse, Repeat, and Auto Punch In and Punch Out. The large display shows absolute time in hours, minutes, seconds, and frame and also shows memory location, pitch change amount, SMPTE timecode, and SMPTE offset (the SMPTE



The heart of Tascam's digital audio system — the DA-88 8-track digital recorder.

World Radio History

you've waited this long to decide upon the perfect home studio...



THE WAIT IS OVER!

MR-66



Frequency Response: 20Hz-18kHz, ± 3dB THD: 0.5% (1kHz nominal level) S/N Ratio: 95dB

RMC-88



Input: TAPE 1-8, LINE 1-8, MIC 1-4, GROUP A/B, MASTER L/R, PHONES Frequency Response: 20~20 kHz ± 3dB THD: Less than 0.1% Noise Level: -120dB (EIN)

The new Vestax RMC-88 Recording Mixer is a breakthrough in mixing technology, now available at an affordable price. Manufactured in the U.K., the RMC-88 is an 8 input/8 program output mixer dedicated to home studio recording. Each channel's signal can be selected for direct Record Out or routed through its Sub-Master outputs and allocated back to any chosen channel for Record Out. This provides the greatest possible flexibility for recording with multitrack recorders and eliminates the struggle of changing patch-cables. Ideally suited for any 4, 6 or 8 track recorder, the RMC-88 can also be used as a conventional mixer for live applications using P.A. Featuring 4 XLR Mike inputs, 3 band E.Q., 3 auxiliary sends, Pre-fader listen, Peak indicator LED's and is 19" rackmountable.



Vestax Musical Electromics Corporation, 2860 Cordelia Road, Suite 120 Fairfield, CA 94585 Tel: (707) 427-1920 Fax: (707) 427-2023

CIRCLE 69 ON FREE INFO CARD World Radio History

SPECIALISTS IN MIDI SYSTEMS AND HOME RECORDING SWEEDWALET SOUND SWEEDWALET SOUND NO HASSLES For 12 years, Sweetwater Sound has been dedi-

For 12 years, Sweetwater Sound has been dedicated to providing the latest technological breakthroughs at affordable prices. Isn't it about time you found out why musicians and engineers around the world have come to depend on Sweetwater Sound for all their equipment needs?





WORKSHOP SYSTEMS

features apply to the optional synchronizer). A shuttle wheel simplifies location of a particular spot, and operates bidirectionally. A switch below it takes it out of the circuit if desired. Variable speed is set by selecting pitch change on the display and using up/down keys to achieve the desired setting.

Tracks can be placed into record in several different ways. The main Record transport control can be used to place tracks that have been individually armed into record in a single-button operation. The Auto Record function will similarly drop preselected tracks into and out of record using the two memory points. Or you can simply roll in Record mode, and use the individual track record selects to place one or more tracks in and out of record — very convenient when replacing several sections at once, like redoing all the background vocal parts in every chorus.

GETTING CONNECTED

Up to sixteen DA-88 units can be locked together simply by connecting a 15-pin D-sub connector between each unit, without using up any of the available tracks. Digital I/O utilizes a 25-pin D-sub connector that facilitates copying tracks between units in the digital domain. Word Sync I/O is provided on BNC jacks. This will allow Tascam's forthcoming hard-disk-based RA-4000 random access recorder and its timecode-capable DA-60 DAT recorder to lock up to the DA-88 without additional sync gear. It's also a way to lock to and with Digidesign Pro Tools. Two optional digital interfaces have been announced, one in AES format, the other in S/DIF-2. There was some mention of an 8-channel digital interface at the AES show; stay tuned for further developments. Prices for the digital interfaces are "to be announced."

cations for any audio recorder these days is synchronization to video sources. Tascam has covered this base nicely with the optional SY-88 synchronization board (list price: \$799). It plugs neatly into the rear of the DA-88, and allows the recorder to function as either master or slave. A sub code channel is used for the code, so no tape tracks are used up. There's more to the SY-88 than just SMPTE sync; you'll also find an RS422 port, Video Sync, and MIDI Machine Control capabilities built in. Only one synchronizer board is required, regardless of the number of DA-88's locked together.

Two remote controls are available as options. The basic RC-808 (\$175) provides transport and record selects for one machine. The RC-848 (\$1499) will directly handle up to six DA-88's; Tascam doesn't think that most end users will have need for more tracks, at least not in this format, so it has optimized the remote for this number. Record selects for all 48 tracks are available at the top of the unit without having to select the unit first. The RC-848 serves as a full-function autolocator for up to six machines, and can set pre- or postroll times when locating. It also is capable of offsetting machines against one another, and handles Repeat Play and Auto Punch functions. The final available option is a 24-track remote meter bridge, the MU-8824 (\$899).

Of course, the bottom line on any recorder is the sound; the jury will have to remain out on this one until production units ship, which Tascam expects will be in the first quarter, with some of the options following along a bit later. But as this preview indicates, there's no question that the engineers at Tascam have worked long hours to come up with an affordable digital 8track recorder that's competitive with anything on the planet.

One of the most important appli-



Tascam's DA-60 DAT recorder (above) can hook up to their DA-88 without additional sync gear.

Listen to the sound of silence..... the D&R Orion.

Everything you ever wanted.....and more!

In-line format • high definition electronics • 4 band sweepable equalizer with monitor and channel switching • 8 mix subgroups • 8 busses with 4 **"Floating Subgroups"** • 8 aux send busses • aux sends on monitor or channel • stereo in place solo or pfl on monitors and channels • three inputs per module (2 usable at same time with EQ and aux sends) • dual stereo effect return modules with EQ, aux sends, mutes, solos, and bussing • LED bargraph meters (optional high resolution LED bargraphs or VU meters) • three frequency oscillator • +4 dBu or -10 dBv interface built-in • extensive communications • phase meter • frame 30 (72 mix inputs), frame 38 (88 mix inputs), frame 51 (96 mix inputs) • custom size frames available • balanced inputs and outputs • balanced TT patchbay built-in or optional • optional pedestal base • automation ready • free installation consultation • regional product support • and **"Factory Direct"** pricing!



the Orion

(starting at \$12,500)



D&R. Every sound under control!

D&R West (818) 291-5855 • D&R Nashville (615) 661-4892 • D&R Southwest (409) 756-3737 • D&R USA (409) 588-3411 D&R Electronica b.v. • Rijinkade 15B • 1382GS Weesp • The Netherlands • Ph. (--) 31 2940 18014

> CIRCLE 21 ON FREE INFO CARD World Radio History

FELDMAN'S BASICS

OH NO — Noise!



Everything you always wanted to know about noise reduction (except for what we'll be publishing next issue)

NOISE IS EVERYWHERE. On city streets. In stadiums, classrooms and bars. Even in great recordings. A good definition of "noise," at least as far as audio sound reproduction is concerned, is any reproduced sounds that are not part of the original sounds picked up by microphones during the recording session. Generally, even though harmonic and other forms of distortion such as IM fit that definition, we usually exclude such coherent signal components from our definition of noise and confine the definition to random noise. Clearly, an objective of any recording is to have as little noise as possible.

What seems like a very long time ago (in the dark ages of tape recording), when tape hiss was the enemy of both recording engineers and consumers who made their own tapes, a brilliant engineer by the name of Ray Dolby came up with the first double-ended noise reduction system. By double-ended, we mean that the system (dubbed Dolby A) involved a dynamic form of EQ encoding at the recording end of the chain, and a reciprocal or opposite kind of decoding during playback of the finished tape. Dynamic EQ means that certain frequencies were boosted during recording by an amount dependent upon their amplitude. Since tape "hiss" consists of upper middle and high frequencies, it was those frequencies that were boosted during recording, effectively getting those frequencies up and further away from the generated noise floor. When mid- and high-frequency amplitudes were low (closer to the noise floor), the Dolby system boosted them more than when they were loud. Thanks to the psychoacoustic masking effect, when you hear loud sounds they tend to mask low-level noise and hiss. When mid- and high-frequency sounds are soft, however, the tape hiss comes through loud and clear, so those signals have to be boosted much more.

The playback process in a doubleended noise reduction system such as Dolby A involves cutting or attenuating those mid- and high-frequencies by exactly the same amount that they were boosted during recording. The attenuation not only restores "flat" response to the reproduced music, but cuts down the residual tape hiss and noise generated during the recording process by the same amount. Prior to the invention of Dolby noise reduction, single-ended noise reduction systems such as highcut (low-pass) filters were also able to reduce tape hiss, but they did so at the expense of fidelity. In a single-ended noise reduction system, if you knock down treble frequencies by 10 dB or so to reduce noise, you're also going to knock down the high-frequency musical content in the programming. To be sure, some single-ended systems, such as one developed by high-fidelity pioneer Herman H. Scott early on, worked dynamically, too. His system sensed the loudness of high frequencies. If they were very loud, no attenuation took place, but if they were softer, they were cut in much the way that a treble control cuts down highs.

DOLBY FOR THE MASSES

In the early 1960s, after the cassette tape format was introduced by Philips as a "strictly business dictation" format, Ray Dolby applied his ideas to reducing noise in the then "low-fi" cassette medium. He simplified his professional Dolby A noise reduction (NR) system to make it easier to use in consumer products and began licensing the system to manufacturers of consumer cassette decks. To differentiate the consumer version from the pro system, he called the new system Dolby B, by which designation it goes even today. Many audio authorities believe (and I completely agree) that were it not for the introduction of Dolby B to cassette decks, that format would never have achieved the popularity that it did - especially among high-fidelity stereo enthusiasts. Early cassette tapes exhibited signal-tonoise ratios no better than around 45 dB, with some better early grades approaching the 50 dB mark. Anyone who has listened to a tape having no more than 45 dB of signal-to-noise ratio will readily agree that the tape hiss is intolerable and detracts from enjoyment of the program material. Dolby B improved the signalto-noise ratio of cassettes by approximately 10 dB, turning tapes that inherently had signal-to-noise ratios of 45 dB into tapes capable of 55 dB signal-tonoise. In terms of noise power, that's a reduction of noise by a factor of ten-toone. In terms of audibility, the residual tape hiss was cut in half!

Many years after the invention of Dolby B, Dolby Laboratories decided that it was time for further improvement in the consumer noise-reduction system. That's when they came up with Dolby C, a system that improved signal-to-noise ratios by a total of nearly 20 dB. Figure 1 shows the residual noise level (referred to as maximum recording level) of a highquality cassette, plotted against frequency. The upper curve was taken with no Dolby noise reduction applied. The next

Do it Drawmer

KIN

DRAWMER

DS404

DS404 QUAD GATE

DRUM GATES

. . .

No C

KEY

1 >

Distributed in the USA exclusively by QMI 15 Strathmore Road Natick, MA 01760 Tel: 508 650 9444 Fax: 508 650 9476



Drawmer Distribution Charlotte St. Business Centre. Charlotte St., Wakefield W. Yorks., WF1 10H England Tel: 0924 378669 Fax:: 0924 250 460 CIRCLE 67 ON FREE INFO CARD

Isn't it time that you discovered what Drawmer really means?

World Radio History

DRAWMER DUAL GATE

DS201 DUAL GATE



lower curve shows how Dolby B lowers the mid- and high-frequency noise content, while the lowest curve shows the additional improvement in signal-tonoise at those frequencies when Dolby C is used. Notice that Dolby C improves the signal-to-noise level down to somewhat lower frequencies than Dolby B. One of the advantages of Dolby B (despite the fact that it is not as effective as Dolby C) is the fact that you can play a Dolby Bencoded cassette tape on an inexpensive cassette deck that's not equipped with any Dolby NR and it will still sound reasonably well-balanced, response-wise. Try that with a Dolby C-encoded cassette, and you sense that something's not quite right. Highs seem entirely too bright. That's probably why most commercially available music cassettes use Dolby B, even though it's not as effective as Dolby C.

Anyone who has tried to transcribe a well-recorded digital CD onto audio cassette tape knows only too well that the task is not easy - and rarely very successful. To be sure, makers of audio cassette tape have tried their best to come up with magnetizable particles that are smaller, and that can therefore be more densely packed for higher recording levels and correspondingly greater dynamic range capability. Despite these efforts, the dynamic range available on even the best grades of normal and even high-bias tapes falls short of what's required to record program material from most CDs.

In an effort to solve the problem, Dolby Laboratories once again turned its attention to analog recording technology. What it has come up with has been dubbed Dolby S-type. With CDs now dominating the audio software scene, the need for greater dynamic range in a tape recording system becomes apparent. Equally apparent is the need for a recording system that suppresses tape noise even further than either Dolby B or Dolby C noisereduction systems have been able to do. Dolby S-type has been developed not only to provide superior sound rivaling that of digital recordings, but also to ensure that consumers will be able to easily and consistently enjoy the superior sound of the new program sources when those sources are transferred to cassette tape. To achieve this goal, a new signal-processing system that dramatically increases available dynamic range while maintaining excellent signal stability has been combined with a set of performance standards encompassing the entire group of specifications for tape equipment.

The heart of the Dolby S-type is its new signal processing circuit, derived almost directly from the professional Dolby Spectral Recording, or SR system, that was introduced to the professional recording and film industry. Dolby SR increases dynamic range by up to 24 dB at high frequencies and 10 dB at low frequencies.

I'll tell you more about Dolby SR and Dolby S next time, and will also talk about other noise reduction systems such as dbx. Until then, let's keep the noise down, OK?

If you're gonna wear a tie...

guitar fretboard on a necktie. Looks as great at a gig as it does at the office. The perfect gift for yourself or any musician. 100% black silk, made in USA. Send check or money order for **\$19.95** plus \$3.00 shipping and handling to:

Fretware P.O. Box 520 Dept: E Kingston, NJ 08528 Visa or MasterCard call 609-426-6932 24 hrs/7 days Dealer inquiries welcome World Radio History



DYNAMIC DUOS: JIMMY JAM & TERRY LEWIS, C&C MUSIC FACTORY, L.A. REID & BABYFACE PROJECT STUDIO DESIGN TIPS ANNUAL STAGE MONITOR BLOW OUT SOUND LIBRARY UPDATE VOICE OF THE PEOPLE: SAMPLERS EXCLUSIVE EQ REVIEWS

HID OUT MORE BY CALLINGE

Kathleen Mackay, Director of Advertising Sales; Daniel Hernandez, Matt Charles, Andrea Berrie, Advertising Sales; Tel: 212-213-3444; Fax: 212-213-3484

Membership Has Its Privileges

More than just a convention organizer, the AES is a fullservice association



OUESTIONS FREQUENTLY ASKED at educational functions, professional meetings, conventions and other audio industry gatherings are: "How do I obtain and hold the respect of my fellow audio practitioners?"; "How do I stay informed on the latest developments in the audio and recording industry?"; and "Where can I go to 'network' with like-minded individuals interested in personal and project studios or other audio topics?" The answer to all of these questions, and a major source of personal and professional enrichment to those who already belong, is membership in the Audio Engineering Society (AES).

There is no other mark of respect that is so widely acknowledged amongst audio professionals as membership in the AES. It informs all who are interested that you have met the Society's criteria for acceptance as an audio "engineer" at the level appropriate to your current skills Whether you are a student, associate, regular member, fellow, or life member, your affiliation confirms your dedication to the art and science of sound recording.

TESTIMONIALS

If all of this sounds just a bit exaggerated, consider the following: "I had entered the Army to avoid dealing with

my future. I was bored with school, and the military seemed the one way to put off decision making and to continue 'party time.' But once in, I discovered that my MOS (Military Occupational Specialization) was in the kitchen feeding thousands of troops each day. I knew that I wanted to do something in audio rather than make thousands of donuts, so I approached the Armed Forces Radio (AFR) chief at our base. He was sympathetic to my plight but insisted on some proof of my supposed audio proficiency. I had joined the AES in college and I whipped out my AES card. He pulled his wallet and matched my card with his and a smile! Within a week I was forever relieved of the kitchen mess and well on my way to an audio career. Today, I own and manage a major recording studio and I could say I owe it all to the AES."

One hears from job seekers who report over and over again that the presence of an AES membership on their applications validated their professional resumes. Yet there is no secret handshake amongst AES members to justify this respect. What is acknowledged is a matrix of nearly 500 local section meetings in the United States and elsewhere each year. The local meetings offer a forum for the discussion of specific topics of audio interest involving audio technology, audio operations, or audio trends. They provide local access to the latest developments in recording, broadcasting, sound reinforcement, consumer electronics, maintenance, acoustics, signal processing, and a thousand and one other topics. What makes this program so attractive is that the individuals who invented and/or pioneered a specific development will frequently be the ones to address the local section meetings. The studio facility that has had the first breakthrough installation of an exciting new technology will host AES sections for a discussion and tour.

GET-TOGETHERS

AES conventions represent the yearly meeting of what is virtually the entire world audio community. Hundreds of audio- and recording-equipment makers exhibit their wares. Over one hun-

dred technical papers describing new developments are presented each year, and a dozen or more workshops are offered on topics that range from career options for women in audio to the latest techniques in digital audio editing on workstations. Attendance frequently exceeds 15,000 individuals. In addition, it is the venue chosen for the simultaneous gathering of the Society of Professional Audio Recording Services (SPARS); for seminars of the National Academy of Recording Arts and Sciences (NARAS); and as the annual meeting place of both audio educators and music industry educators. Numerous meetings are held by the various AES committees entrusted to develop and establish standards for the audio industry in such areas as digital audio interconnection, sampling rates, music sound levels, etc.

The AES publishes a journal that presents the best of the technical papers submitted at the conventions or directly for publication. In addition, convention activities are both previewed prior to the gathering and reported on in depth afterwards. So AES members who cannot attend the annual meeting in the U.S. or the similar yearly European conclave receive a full report within the journal. Section-meeting summaries expose the reader to all the new and interesting concepts that were introduced in the U.S. and around the world. Industry and AES news highlights, information on standards activities, and updates on a host of other topics assist the member in keeping "track" of the audio business.

There is much that the AES offers, but perhaps the most important value for many members is the pleasure of personal contact with the hundreds and thousands of other audio practitioners who attend the local meetings and yearly conventions. A bonus to be sure, but an important one at that!

For more information about the AES and its activities, contact the Audio Engineering Society, 60 East 42nd St., Room 2520, New York, NY 10165; tel: 212-661-8528; fax: 212-682-0477. Tell them EQ sent you.

Making the Most of Your Mics

Yes, you need to maintain your mics to get the max from your mouthpiece BY DR. RICHIE MOORE, PH.D.

LOOK AT THAT FUNNY-looking thing over there gathering dust in a box. I used to remember what that was, 'er, did. Oh yeah, that's an MD, um, a KM, oh wait...an SM-50 something. What the hell — it's a *microphone*. Out of sight, out of mind. That is the way it is with a lot of old tried-and-true equipment.

Currently, the most common sight in studios are direct boxes. Everything is sampled and synthesized into digital storage. Drums, keys, guitars, and S many other tones are all put on tape or hard drives through direct boxes. This is all well and good, but how do those samples into get these wonderful devices? You got it

m

microphones. Oh yeah, and until DFVC (direct from vocal cord) systems are available, artists will still have to use microphones to put their vocals on their storage medium of choice.

OF MICS AND MEN

Microphones are the recording engineer's and artist's palette of sound color. With the proper mic in the proper place, you can get all the tone you need without much tone processing. Every mic ever made has a distinctive color of its own. The trick is to match the desired tonal effect with the proper transducer (mic). You have to cater to the whims and personalities of mics to get consistent results. The secret of being able to get these sounds at will lies in the care given the microphones.

Mics are basically an electromechanical transducer. A transducer converts one form of energy into another. Acoustic sound pressure arrives at the capsule (diaphragm) and changes the air pressure into voltage. Mics come in all varieties, but can essentially be categorized as dynamic, condenser, ribbon, small diaphragm, or large diaphragm types. Pretty simple in theory, but many other things affect the operation of a mic.

Mics of all types are subject to changes in barometric pressure and humidity. I have found that when barometric pressure is high, a microphone is not as sensitive to soundpressure variants. When humidity is high, the tone becomes darker (fewer highs). These are just a few of the reasons why no matter how

much care you take in writing down or storing the console settings, the mic always sounds different the next time you use it.

CARE PACKAGE

As I said before, mics require some care to maintain their quality and be reliable. The first thing to remember is that a mic is a fragile item. The capsule is made of very delicate material and the electronics are also delicate. I know that people swear that you can hammer a nail with an old Electro-Voice EV-666, and I can personally attest to that, but the longevity of the mic will be severely diminished if you do so. You have to be gentle in the midst of the audio hurricane.

Microphones should be stored in their original cases whenever possible. This protects them from dust, dirt, moisture, and impact. It also allows an easy inventory assessment at the end of a session. In lieu of the original cases, a sealable flight case with foam cutouts shaped to each separate mic will do.

Moisture is a mic killer. Humidity and moisture can erode the capsule and make a mic unusable. Storing mics in a sealed case is therefore very important, especially with condenser (electret or phantom powered) mics. These capsules are very sensitive to moisture. Store condenser mics in a zipper storage bag with a packet of silica gel inside to absorb moisture. You know what the silica gel packet looks like; you get one with practically every piece of electronic gear you buy. If you must leave a mic up overnight, put a baggie over it with the silica gel pack inside, and seal it with a rubber hand

The electronics inside a mic don't need any special care beyond the exercise of normal common sense. Don't throw a mic in a wash basin filled with water while the mic is attached to the console and phantom power source - that sort of thing. The connectors can use some care. Modern FET (field-effect transistor) mics use phantom power usually supplied by the console at +48 volts DC through a 3-pin XLR-type connector. Valve (tube) mics usually have a dedicated power supply with five or more contacts through which the audio signal and the tube voltage are carried. The connectors get dirty and corrode with time. A little Cramolin® on a Q-tip applied to the connectors every six months can help prevent signal loss from corrosion. You might also do the same for the mic cables that you use with the

Illustration by Milton Reyman



IF ONLY GETTING IN SYNC WAS THIS EASY FOR EVERYBODY.

You see it every day. The quarterback lofts the ball over the wide receiver's head.

The cab pulls up to the terminal as the airplane pushes back from the gate. The guy on tv moves his mouth before the words come out. Classic cases of a world slightly out of sync.

In the audio environment, it's the same problem, times ten. The machines you use, most often don't speak, read, or respond to the same language. You can't just line them up, push the "go" button and have them all operate in unison. How, then, do you get the various formats of audio and video tapes to work in sync?



Well, you probably know about those magic "black boxes" called synchronizers that the guys with the megabudgets use. What you might not know is that this same magic is now available, and easily affordable, for any home studio.

The breakthrough comes thanks to TASCAM's ATS-500, priced below \$800.* It's a twomachine chase-lock system with a SMPTE time code generation capability that can resolve to video sync using your existing VCR and 8 track. It's got all the professional features an offset function, wide-band reader,

jam sync—yet it's easy to understand and operate. It's also autocalibrating and requires no mastery of hidden screens or functions. And, it not only works with TASCAM's serial control transports, but by adding TASCAM's IF-500 interface, it'll work with other parallel transports as well.

To see the ATS-500 first hand, just sync up with your nearest TASCAM dealer. He'll show you how easy synchronization can be.



*Suggested Retail Price C 1992 TEAC America, Inc., 7733 Telegraph Road, Montebello, CA 90640 213/726 0303

CIRCLE 55 ON FREE INFO CARD World Radio History



CIRCLE 66 ON FREE INFO CARD



best trained of anyone entering the engineering field. I give IAR my highest endorsement. Dan Healy, Chief Engineer The Grateful Dead 800-544-2501 NY, NJ, CT 212-777-8550 Lic. by NYS Ed Dept/HS Diploma or GED Required App for Veterans Training / Financial Aid if Eligible microphone.

While you are cleaning the connectors, you are in a perfect position to check the connectors for loose screws. The constant plugging and unplugging of connectors will loosen the best of them. There is nothing so frustrating as trying to disconnect a mic from a cable and yanking out the internal wiring — and mics don't have very long wire service loops with which to attempt easy repair. Inspect the mic carefully and use a jeweler's or flat-head screwdriver to tighten the connector.

MAINTENANCE

Always check the cable connectors as well as the internal ones. When an XLR connector starts getting wide around the rubber insertion holes, replace the connector; a loose one can cause all sorts of problems, and connectors are cheap. I personally prefer Neutrik NC3FXB and NC3MXB goldplated connectors when replacing damaged XLRs. Many microphone problems are directly attributable to poor connections.

Most microphones can and should be repaired by the original equipment manufacturers. They really contain few user-serviceable parts, or even many parts that can be replaced by a seasoned studio field tech. In the case of vintage tube microphones, Stephen Paul in Los Angeles and Klaus Heyne of German Masterworks near Portland, Oregon, are your best bets. Klaus has worked on all the microphones I use, and I swear by him. Stephen Paul is no slouch either, and provides quality craftsmanship, but Klaus used to live in the San Francisco Bay Area very near me. But be prepared — an excellent craftsman's work doesn't come cheap.

One final note of microphone maintenance: keep the mic stands and mic clips in a decent state of repair. When a \$10 clip goes bad, it can damage a \$500 microphone. Duct-tape, on a mic stand, is no substitute for tight fittings. Parts are available to replace worn ones, and, once again, an ounce of prevention... Regular cleaning and tightening of your clips and stands will serve you well.

All these maintenance items will help you get the best out of your mics for years to come, and protect a major investment.

continued on page 109

CIRCLE 41 ON FREE INFO CARD

titute of Audio Research

64 University Place, Greenwich Village, New York, NewYork 10003

How to turn your 8" to 15"



Aphex Aural Exciter® Type C² with BIG BOTTON

It's the all new Aphex Aural Exciter Type C^2 ! This 2-channel model with **BIG BOTTOM** gives you twice the excitement. **BIG BOTTOM** enhances *bass* the same way the Aural Exciter process enhances the high end. Make your 8" speaker sound like a 15" woofer ... load up your tape with powerful drums, huge synth bass, fat pads and thunderous bass ... without all the mud.

Best of all, **BIG BOTTOM** does its job without vastly increasing amplitude or generating subharmonics which can damage your speakers or overload tape. **OVERHANG** and **GIRTH** give you flexible control on how BIG you want your BOTTOM to be.

Of course, the Aural Exciter is still the premier way to enhance highs, too. The Type C^2 utilizes the latest refinements in Aural Exciter technology and is easier to use, and is quieter than ever. It does the job by recreating and restoring harmonics ... musically and dynamically. A patented process that many have tried to emulate, and none have succeeded.

Hear how the Aural Exciter Type C^2 with **BIG BOTTOM** brings new levels of *excitement* ... live, recorded or even sampled ... at selected dealers around the world.

APHEX SYSTEMS 11068 Randall Street Sun Valley, CA 91352-2621 (818) 767-2929 Aphex is proudly American... 100% owned, engineered and manufactured in the U.S.A.

Aphex. Aural Exciter and Big Bottom are trademarks of Aphex Systems Ltd. *Patents applied for. Aural Exciter is covered by U.S. Patent No. 4.150.253, and other patents is long and pending.

CIRCLE 13 ON FREE INFO CARD



CIRCLE 39 ON FREE INFO CARD

Ad Index

P46	BRAND	01	PHONE#
14 02	Acoustical Solutions	01	800-782-5742 415-351=3500
6,7, 51, 45, 13	Alco	08, 06, 03, 04	213-467-8000
70	Amber-Electro Design	05, 05, 05, 01	514-333-8748
05	Amek	13	818-508-9788
95	Aphe <u>m</u>	13	818-767-2929
65	Artist's Systems	10	707-778-8893
09	Ashly A <u>udio</u> , Inc.		716-544-5191
99	Audio Institute of Americo	12	415-931-4160
43	Audio Technica	14	216-686-2.00
49 96	Audio Techniques	15	212-586-5989 415-457-7-00
17	Banagas At Lorge Behringer	17	516-932-3180
83	Coig Lobortories	17	619-451-1799
32	Center for the Media Arts	19	212-807-6670
69	Conneaut Audio Devices	25	216-593-1111
101	Conservotory Recording	20	602-496-6508
59	Crown	09	219-294-8000
93	D & R	21	409-588-3411
30	DAT Audio Gollery	22	213-829-3429
55	dbx	23	415-351-3500
60,61	DDA	24	516-249-3660
115	DIC Digital	26	201-224-9344
39	Digital Audio Labs, Inc. Disc Mokers	28.20	612-559-6104 215-232-4140
12, 14 56	DISC MOKERS DOD Electronics	28, 29	801 268-8400
89	Drowmer	67	508-65D-9444
53	Dynotek Automotion Systems	30	416-6 2 3000
76	Electron Artistriles	32	513-746 4283
37	Ensonig	-33	215-647=3930
23	Fishmon Tronsducers	34	617=938=8850
29	Fostex	35	213-921-1112
90	Fretwore	36	609-426 6932
50	Furman Sound, Inc.	37 .	415=927 1225
75	Gemini Sound, Inc.	38	718-851-6000
96	Goldline Loft	39	203-938-2588
30 94	Ulbruch Institute of Audio Research	40	800-662 0032
116	JBL Professional	XX	212-677-7580 818-893-8411
81	JRF Magnetic Sciences	43	201-579-5773
42	KABA	44	800-231-8273
107	Lexicon, Inc.	52	617 736-0300
31, 74	Mackie Designs	45, 46	206 487-4333
77	Musicion's Friend	47	800-776-5173
10, 11	Peavey	48	601-483-5365
50	Polyline		708-390-7758
16	Q Up Arts	51	408-688-9525
108, 109	Rone Corporation	53	206-355 6000
46, 47	Roland	54	213-685-5141
71 33	RSP Technologies Shure Brothers	56	313 853 20 4 708-866-2527
33 99	Sonocroft	57	800-274-SONO
03	Sony Recording Media	XX	800-635-SONY
27	Speck Electronics	59.	619-723-4281
66	Stewart Electronics	71	916-635-30
86	Sweetwater Sound	60	219-432-8176
64	Symetrix	61	206-282-2555
87, 104, 105	Tascam/TEAC America, Inc.	55, 62	213 / 26-0303
40 ₀ 41	TDK	63 _a 63	516-625-010
38	Tech 21	65	212-315-1116
94	The DAT Store	66	310 828-6487
76	The Recording Workshop	68	614-663-2510
85	Vestox Corporation	69	707-427-1923
101	Westloke Audio	70	805-499-3686
101 20 21, 63	Whisper Room, Inc.	72	410-997-1440
20, 21, 83 68	Yamaho Pro Audia Division Yorkville Sound, Inc.	75	714-522-9011 716-297-2920
79	Zoom Corporation	15 162	415-873 5085

World Radio History

IF YOU MISSED MUSIC TECH EXPO LAST YEAR THEN YOU MISSED:

Exhibitors: AKG Acoustics Alex Music AMEK AshlyAudio Atari Audio Engineering Society Audix **Basement Recording** Behringer beyerdynamic **Bose Corporation** Bruel & Kjaer Carver Center for Media Arts Cerwin-Vega! Commodore/Amiga Community Pro Sound Crown dbx/A Division of AKG Demeter Digidesign **Digital House** Dr. T's Dynacord (Pinnacle Audio)

Electro-Voice Ensonia Corp. EQ Magazine ETA Lighting Systems Fostex Gotham Audio Corp. Hot House Institute of Audio Research Jazzica IBL Kimber Cable Klipsch Korg USA Kurzweil Lexicon Manny's Midivox Marketing NYPC User Group **Opcode** Systems Panasonic Passport Designs, Inc. Perraux Pinnacle Audio (Klark Teknik) Pro Sound News

QSC Radio Partner Custom Earset, Inc. Recording Industry Sourcebook Sam Ash Music Stores Sam Ash Soundlight Samson Technologies Soundtracs Stanton Steinberg/Jones Studiomaster Tannoy Tascam Tech 21, Inc. Thoughtprocessors Triad Marketing T-Shirt Distributors Twelve Tone Systems Videography Voyetra Technologies

Seminars:

Craig Anderton David Andrews Wendy Carlos Larry Fast Tom Gartland Ted Greenwald Kevin Killen Hector LaTorre Paul Lehrman Bob Ludwig loe Mardin Howard Massey Moby **Roger Nichols** Michael O'Reilly Gene Perla Phil Ramone David Scheirman Tim Tully Stephen Ward Larry Zinn

TechnoMusic:

Joey Beltram Lee Botix DJ Repeat Money Penny NICO Omar Santana Toxic Two



Zoom 9030 Effects Processor



MANUFACTURER: Zoom, 385 Oyster Point Blvd. #7, South San Francisco, CA 94080. Tel: 415-873-5885.

APPLICATION: Processes and adds special effects to guitar, bass, voice, keyboards, and other signal sources.

1/1

SUMMARY: Compact, cute, and easy-to-use, the 9030 gives good sound with a useful complement of editable parameters.

STRENGTHS: Good visual feedback, four knobs for simultaneous parameter control, small size, clean sound quality.

WEAKNESSES: Doesn't respond to pressure or pitch bend messages; small knobs and wide parameter ranges require finesse when setting parameters; order of effects is fixed.

PRICE: \$749 (matching 8050 footswitch, \$249) EQ FREE LIT. #: 118

200N Corsora

ZOOM HAS ZOOMED on the scene as a maker of nifty, cost-effective guitar effects boxes (although they're also suitable for bass. mics, electronic drums, etc.). The first thing you notice about this half-rack device is a very high "cute factor"— striking blue 40-character fluorescent display, gray mottled case, small LEDs to show which effects are selected — but dig beneath the hip packaging, and you'll find that there's more involved than good looks.

FACTS & SPECS

The 9030 includes 47 individual effects; up to seven can be combined in up to 99 user-programmable patches. Like most manufacturers, Zoom's definition of an "individual effect" is somewhat liberal for example, compression and limiting count as separate effects, as do chorus, flange, phasing, etc. Basically, you have nine groups of effects, arranged serially in the following order (the number of choices for each group is given in parentheses): compression (2 types), effects loop (for inserting that favorite vintage effect you can't live without), analog distortion (3 types), EQ (3 types), amp simulator (2, guitar and bass), misc. effects I (12 time-based effects including tremolo, octave divider, and wah), misc. effects 2 (12 that include variations on misc. effects I but also offer different effects), special effects (7 including advanced timebased effects and some real "out" sounds), and reverb/tapped delay (6 types).

The order of effects is fixed; you can't do some things, such as putting the EQ in front of the distortion. There are, however, four different signal routings (parallel/series, stereo/mono) that are selected automatically, depending upon which effects make up a patch. For example, if the special effects group is off and the amp group is set to stereo, misc. effects 1 and misc. effects 2 are patched in parallel, and followed up by stereo reverb. Although the routings are not particularly flexible, they do give you another set of options.

A matrix of nine legended LEDs show the status of the effects blocks. Red means active (turns to amber when the unit is bypassed) and green means inactive. A two-digit LED shows the patch number, and there are also two input-level blinky lights. It's easy to bypass an individual effect to see how it affects the sound, which is helpful as you start learning how to edit the factory patches (let's see, is that sharpness coming from the amp effect or the special effects...).

BB

100

98 DECEMBER EQ

THE NOISE FACTOR

The EH-50 is very quiet; however, a sound source with background noise, processed by the EH-50, will become an enhanced sound source with enhanced background noise. For best results, try to get the cleanest signal into the EH-50. In extreme cases, an external low-pass filter can be used to roll off high-frequency noise. Placing the filter either before or after the EH-50 can also make a difference. Another suggestion is to use the EH-50 when committing to analog tape, because enhancing recorded material may exaggerate tape hiss.

GAIN STRUCTURE AND DISCIPLINE

Since the EH-50 has the sensitivity to serve as an instrument preamp, it is very important not to overlook the Level switch located on the rear panel. Paying attention to this feature will ensure not only the optimum signalto-noise (S/N) ratio but also the correct range of the enhancement control. For example, setting the switch to -10 with a signal that is a nominal +4 will make the Enhance Level control overly sensitive. I suggest experimenting with this switch by leaving it in the +4 setting except when a musical instrument is connected. Unless you find this arrangement yields poor S/N, it might also serve as a sort of discipline to avoid excessive enhancement.

One "problem" is that the EH-50 really does work well with some material. Addictive personalities might feel, therefore, that if a little bit of enhancement works, more enhancement would be even better. In spite of the Enhance Level display and the overload indicators, it is possible to get nasty results. Be sure to use the Enhance switch regularly to compare treated signals with untreated ones.

CONCLUSION

If the EH-50 were my child, it would be named "Instrument Preamp and Dual-Channel Enhancer." This not only describes the EH-50 more accurately, it also emphasizes the hidden feature: an instrument preamp with an FET front end. I think Boss could sell the EH-50 based on this feature alone, especially in light of the recent proliferation of vacuum-tube preamps. (An FET is the semiconductor equivalent of the vacuum tube and, as such, is the perfect interface between musical instruments and electronics.) By adding the FET to the design, something Boss could easily have left out, the EH-50 is a very acceptable instrument preamplifier, especially for the price. As for EQ and enhancement, the EH-50 definitely increases your options.





CIRCLE 72 ON FREE INFO CARD



There are thousands of exciting, good paying jobs in the music inclustry for quality sound engineers and assistants. But no one can just walk into a major recording studio and ask for a job. The audio recording industry demands specialists. You must have the right training and we give you that training in just a matter of months. So don't waste time, call tocky, **1 800 562-6383**.



EQ DECEMBER 101

World Radio History

EQ is the brand new magazine that covers the real-life world of today's recording studic.

rom recording techniques to studio and career management, from programming and performing to hands on and howto, EQ is for engineers, producers, and musicians who are serious about recording. Each and every bi-monthly issue is packed with pages and pages of ...

Hard-hitting reviews of today's hottest new recording gear, MIDI equipment, software, and more.

Fascinating visits to artists' and producers' studios, as well as commercial and home facilities all over the U.S. and the world.

Audio-For-Video you'll learn the inside workings of music video, film, and television sound production.

Workshops and clinics, on everything from Arranging In The Studio to Digital Recording, will give you the competitive edge.

And that's not all. You get advice from leading engineers, producers, and artists as they talk about their latest projects. EQ will also help you get a real-world perspective on how recording professionals make their moves into the job they want, and how independent "wizards of the board" build their own studios into profit centers. Plus, monthly columns covering basic recording tips, computers, tax and legal advice, studio maintenance, and much more.

A ake advantage of the special CHARTER SUBSCRIBER rate that saves you up to 40% off the regular rate. Be part of the only magazine for serious creative recording.

BECOME A CHARTER SUBSCRIBER TODAY

Aixing Nirvanu Sy Butch Vig Peter Gabriel's Studio Hideaway Anderton's NAMM Not Picks

YES! Enroll me as a CHARTER SUBSCRIBER to EQ. \$19.97 for 1 year (6 bi-monthly issues), 29% off the basic rate! * \$33.97 for 2 years (12 bi-monthly issues), 39% off the basic rate! * \$49.97 for 3 years (18 bi-monthly issues), 40% off the basic rate! *

Prices guod in U.S. only. Canadian and international surface mail subscriptions add U.S. \$5 per year. International air mail subscriptions add U.S. \$20 per year. All non U.S. orders payable in U.S. dollars by VISA, MasterCard, American Express, or International Money Order only. Allow 4-6 weeks for delivery of first issue. "Basic rate: 1 year (6 bi-monthly issues) for \$27.95. 50 E E 9

INREVIEW

Lexicon LXP-15 Effects Processor



application.

MANUFACTURER: Lexicon, Inc., 100 Beaver Street, Waltham, MA. Tel: 617-736-0300.

APPLICATION: Cost effective digital effects with considerable options for remote control.

SUMMARY: An inexpensive delay and reverb unit for practically any

STRENGTHS: A very capable reverb processor with a wide palette of rooms, plates and delay effects. Front panel controls are fast and easy to use.

WEAKNESSES: Fewer parameters for adjustment than many competitive units and the chaining of effects is fixed. Slight click is produced when changing algorithms.

PRICE: LXP-15 - \$1050, MRC - \$399 EQ FREE LIT. #: 120

LEXICON HAS DEVELOPED some of the best reverb devices available. It has even developed a sophisticated architectural reverberation system, called LARES, based on its top-of-theline 480L reverberation hardware, that enhances the acoustics of theaters and concert halls. Taking an interest in the sound of the best concert halls in the world (including Symphony Hall in nearby Boston) has allowed Lexicon to develop algorithms that can be mixed directly with the acoustic reverberation produced in a good concert hall. Lesser reverb units sound distinctly anemic when put to this test.

In 1989, Lexicon began producing the LXP series of effects processors for those who could not afford its 480L. but still wanted that Lexicon sound. The most recent in the series is the LXP-15. This one-rack-unit digital effects processor uses a soft-key approach to couple the lower line of the two-line LCD display with the five buttons directly underneath. The function of each button is labeled to correspond to the function on that page of the LXP menu. Beside these buttons is the View button, which toggles the display between parameter selection and value display modes.

A large Adjust knob is then used to modify parameter values or select a preset program. Beside the Adjust knob is the Page knob, which selects the page of parameters to be adjusted. The menu of parameters is never buried within another menu; instead it is just the twist of a knob away. On the opposite side of the front panel are input and output level controls and a single four-segment LED array indicating input and regeneration levels, from -24 dB to clipping, in both channels.

The rear panel is simply a row of connectors: four phone jacks for the unbalanced stereo input and output; five phone jacks for foot switches or pedals; the usual trio of MIDI jacks; and an IEC-type power connector. The input is summed before processing, so either input can be used for a mono source, knocking the 50k-ohm input impedance in half. The adjustable 600-ohm output is nominally +4 dBu and is specified to deliver up to +14 dB into 10k-ohm inputs.

The multieffect processor concept of the LXP-15 differs from most other units of this type. Instead of stacking effects selected at will from a palette of algorithms, the user is presented with a fixed chain of effect parameters and simply decides on the setting for each one. There are five basic sets of algorithms; Delay/ Reverb. Pitch Shifting/Delay, Gating, Plate, and Chorus/Delay.

Unlike the half-rack LXP-1, all of the parameters of the LXP-15 are adjustable from the front panel. The parameters are assigned one at a time to the continuous-control Adjust knob to make changes. I found the process a little tedious when moving quickly from one end to the other, which typically involves 128 steps.

The manual uses clear and concise signal flow diagrams to show how the parameters interact and where each of the mixing/level adjustments is made. The trade-off for less flexibility in the chaining of the effects is made up for in the way parameters are assigned — the LXP-15 always presents the same types of parameters on the same page in every preset. This consistency makes a big difference in learning to quickly modify presets.

The parameters for modifying presets may be less numerous than in other processors, even in the low- and midprice range, but all of the most important ones are included. Creating a reverb that suits an instrument or voice is straightforward, and you won't be skipping over page after page of obscure parameters.

Assigning the parameters to external real-time control, such as Lexicon's Dynamic MIDI®, is also fairly easy and comprehensive. Controllers are assigned by name and not just parameter numbers. Parameters such as Gliding Delay are smooth and glitch free, whether controlled by input level, a pedal, or through MIDI.



Gliding Delay can also be controlled by the internal LFO for applications where a small amount of pitch-shifting is required to reduce the resonant quality of the digital reverb.

Using the Lexicon MRC (see Sidebar) with the LXP-15 gives the user four MRC faders to adjust four parameters per "page," moving between pages of parameters randomly with a numeric keypad. This means that even parameters such as decay time and panning are only one keypress away from each other.

The LXP-15 draws from the development of world-standard studio reverberation devices, having the range and density that you would expect from Lexicon. Each of the different reverberation settings has qualities that make it distinct. The Hall settings decay to the sides and then to the center as the signal dies away, typical of a shoebox-shaped concert hall. The presets for Rooms include more early reflections and are more colorful. The Small Rooms are particularly impressive, including a preset called No Room, which can take an absolutely dry mono recording and give it the spaciousness of a stereo mic pickup without destroying the presence or tone of the original. The Plate settings have a density and sparkle that can add a real fullness to stark tracks. Gated reverbs are flexible, cut off smoothly, and are easy to adjust to match the beat.

The equalization won't replace your parametric, but it is wellbehaved and powerful enough to pre-EQ reverb sends and control the decay/regeneration. Unlike your parametric, however, the EQ can be controlled remotely using MIDI or a footpedal — even in combination with other parameters. There is also a useful range of modulation effects such as flanging, chorusing, multiple delays, pitch-shifting, panning, etc.

Lexicon has given careful consideration to the signal flow within each preset group. There is access to the mix of delay and reverberant effects and even the option of using diffusion to thicken the sound of your delays. Parameter values have not been truncated to protect the naive and will allow the adventurous (or desperate) to create effects that go over the top and all the way to the other side!

The LXP-15 may not offer the long list of effects that other processors in this price range do, but it does deliver a variety of high-quality effects that can be manipulated in a quick and convenient fashion to suit all but the most jaded producers.

-Wade McGregor

64 CHANNELS. CABLE READY.

Sometimes it seems like you can't get there from here. You've got a thousand great ideas, and just about as many plugs in your hand. What you don't have is enough input channels.

Well, allow us to give you some input about a new way to solve your dilemma. It's a Tascam M3500 in-line mixing console. Choose either the 24 or 32-track mixer and by simply flipping a switch, you can double it to 48 or 64 mix positions.

And, with a suggested retail price of \$7,999 for 24 inputs or \$8,999 for 32, it won't take up a lot of your budget, either.

If you're planning to build a 24-track development studio, here's another advantage: The M3500 is the perfect match for the MSR-24, Tascam's oneinch 24-track recorder. Together, they make the most cost effective studio available.

It just may be that you don't need a huge console to enlarge your capabilities. The M3500 offers you a new, more effective approach to traditional mixing that is both compact and low cost. And when you need more inputs, all you'll have to do is switch channels. From 24 to 48. Or from 32 to 64.



© 1992 TEAC America, Inc, 7733 Telegraph Road, Montebello, CA 90640, 213/726-0303.

CIRCUL 62 ON FREE INFO CARD World Radio History

INREVIEW

LEXICON MRC

The Lexicon MRC (MIDI Remote Control) is a powerful device. I recommend spending some time coming to grips with the range of control the MRC can provide. Thankfully, Lexicon has prebuilt some configurations that can be immediately put to use or modified to suit a number of applications. Once you have become familiar with the MRC menu system, the tools are all there to allow very quick access to different menu layers and parameters, using the faders, pushbuttons, or a numeric keypad.

The MRC is the size of a hard-cover book, and includes our soft switches, four assignable sliders, a number pad, and eight function switches. The two-line-by-24-character LCD display makes an annoyingly audible whine. It is easy to read, though, allowing up to eight characters to label each fader or soft-switch function. The display is verbose enough to prevent having to memorize MIDI controller numbers and those assigned to functions are displayed by name, such as "Volume" instead of the typically terse "CC7." There is also a very handy mode for troubleshooting your MIDI system, where the display will name MIDI messages received. Stored MRC configurations are called Machines. There are prebuilt Machines to control the Lexicon LXP series of effects processors, the E-mu Proteus 1, Korg M1, and the Eventide H3000, as well as some generic MIDI messages. The generic MIDI setup can have the first-time user adjusting MIDI volume, starting and stopping sequencers, sending notes, and making program changes within minutes of connecting the MRC to a MIDI network. The generic control also acts as a template for building customized MIDI controls.

The MRC has been designed to fit easily into even very sophisticated MIDI systems. The MRC's 55mm sliders lack resolution beyond 128 steps each, which can be a problem when making a precise adjustment using MIDI, but does prevent slider movements from clogging the MIDI data stream. The MRC has so many powerful applications — including remapping incoming MIDI messages — that at first programming, it can seem overwhelming.

The MRC is capable of making the LXP-15 front panel follow you to the mixer but, when used with the LXP-1 or LXP-5, the MRC allows two-way communication. which lets you awake the sleeping giant within those simple-looking effects boxes. With a little programming effort from the user, the MRC can do as much for nearly any MIDI device. — Wade McGregor





VOICE OF THE PEOPLE gives you a forum to sound off about different pieces of gear. Here's what some readers recently wrote about DAT machines they've owned, hated, and loved.

YOU CAN'T GET THERE FROM HERE

When trying to do digital transfers among a Panasonic SV-3500, Sony TCD-D10 Pro, and Tascam DA-30, I ran into digital/SCMS hell! Here's why:

• The DA-30 always writes the "11" status code on any recording it makes (unless you do a mod)

 The Sony TCD-D10 Pro will roll, but not play, a tape with an "11" code (unless you do the factory mod)

• The Sony TCD-D10 Pro will monitor, but not record, an SPDIF input with an "11" code (unless you do the Digital Domain mods)

 The Panasonic's "secret switch" was in the prohibit mode

Other complaints: the D-10 "Pro" can't record at 44.1 through the analog inputs. It doesn't write absolute time (unless you get the factory mod), and the digital output is nonlinear at low levels (fixed by one of the Digital Domain mods). -Ken McGee

COMFORTABLY CLOSE

I chose the Panasonic SV-3900 for its remote control, which lets me locate short pieces of dialog quickly from the comfort of my Dyaxis keyboard.

Likes: AES and SPDIF ins and outs. Error display. Good meters. Peak display (though not as nice as the "margin indicator" on the Sony 2300). Fast transport. Good sound quality.

Dislikes: Won't record (via analog inputs) at 32 kHz. Can't set default for Remote/Local switch. Clumsy case design makes cleaning heads a 10minute job. Shuttle knob could be better at slow speeds. Main deck should also have a shuttle knob, like the 3700. On top of that, it has a skimpy manual.

I'd love an SV-3900 that would record/play and chase-lock to time code without any external boxes. I'm pretty happy with the SV-3900, though, and have recommended it to several people. —Jeff Kreines. DeMott/Kreines Films

THE POWER OF THREE

I use the Sony ES75, Sony PCM-2700. and Technics SV DA-10. The Sonys seem to have more robust transports and better-sounding converters. The ES75 is old and has seen the most tape, but it is still going strong. All functions are available from the front panel so you're not forced to mess with the remote.

I purchased the PCM-2700 because of the Read-After-Write heads - a must for live recording and mastering. The Technics is basically a Panasonic SV-3700 with consumer analog I/O; I modified it to record at 44.1 kHz from the analog inputs. Unfortunately it only made it for about a year before transport problems appeared, and I was less than thrilled with Panasonic's service.

I miss the lack of error indication on these machines. Perhaps they don't want us to know just how fragile DAT is! -Richard Bugg

LOVE OF MY LIFE

I love the Tascam DA-30's fidelity, although there does seem to be a little high-end noise in the playback. Also, when recording low level signals, the high end suffers considerably, maybe because fewer bits are available to represent the waveform.

-Michael Firis

SHARPER IMAGE

My first DAT (over two years ago) was a Sharp SX-D1000 - RCA ins and outs, slow, noisy transport, and the stereo field practically collapses. Recently I bought a Panasonic SV-3700. Much better converters, but the headphone jack is wired out-ofphase -- don't use it to monitor an open mic, or you will hear a hollow sound. Renumbering cuts and relaying absolute time will work only up to a certain point in each tape. Incidentally, despite their common IEC ports, the two units will not digitally tie to each other, in either direction. -Dan Popp

MY FIRST SONY

The Sony 2700's repro monitor allows me to monitor tape during the record process, which saves me countless hours of QC work. The time-search feature allows me to zoom to any exact ABS time, and pause. And, maybe best of all, I appreciate the copy prohibit on/off switch. The only feature I could do without is the manual ID write rehearsal. If you know your start times, you really don't need to listen twice and then wait for it to write the index. -Jeff Laughinghouse

POWER TRIO SHOOTOUT

We recently did a shootout between the Panasonic SV-3700, Sony PCM-2300, and Tascam DA-30. They all sounded pretty good. The Panasonic sounded closest to the test CD (in deference to Roger Nichols, it was Steely Dan's "Hey Nineteen"); the Fascam DA-30 sounded a little harsh in the high end, and the Sony gave the best imaging — but also seemed to emphasize certain frequencies.

Regarding digital transfers, the Tascam DA-30 and Panasonic SV-3700 had AES/EBU balanced and IEC-958 coax ins and outs, the Sony only had IEC-958 unbalanced. Going between the DA-30 and SV-3700 was no problem; the SV-3700 IEC out worked with the Sony IEC in, but not the other way around. -Rob Bernstein

TELL US ABOUT SAMPLERS

Next issue's Voice of the People is on samplers. Check out last issue for the rules. Send your views to: Voice of the People • EQ Magazine, 939 Port Washington Blvd., Port Washington, NY 11050. Or fax to 516-767-1745.

Remember: If we publish your response, you'll be the proud recipient of the hippest T-shirt in the biz.

EXCLUSIVE REPORTS DIRECT FROM EARWITNESSES

Tiny Wunderboxes Descend Into Hands Of Astonished Musicians

"I Knew I deserved a miracle, but three is beyond Belief!"

1000 4001

By Mickey O'Callahan

Special Correspondent

Politicians, militer nel, noted scie even speci plain' P

L to

Τ. arriva digital has the community tarists, keybe bassists, voca and other music abuzz

The first astonis the appearance of the. es," as the press is ca that never before have heard such powers of exp such compact packages, with represents a first. For electric own a high-que acoustic quitar processor, many and bass too! plenty of use for an bass too!

Equally astonish manner in which appeared - experts that such technology and size was years of. the pre

Hamilton, a motorcyel

Stage or studio optional-

FIRST SIGHTINGS OCCUR ON

Mere Coincidence - or Is The mg to Send a Message? Those who remember Elvis Presley fondly tend to dismiss accounts of bi recent sightings. But even the most skeptical among us would find it hard to dithe miraculous appearance of three different and mysterious boxes-on the annised his demise! Chris Albi and Neil Hamilton, long-time bodyguards for th edged in their unauthorized biography "Viva Las Elelectronic wonders. 'I can still hear him saying 'F Albi, who is "temporar" · . M

sent! Yet here we are, in the present, ud musicians he world over are 't with these most ming in' ch also provide For mult nce as to the ght of their

vocalre

arcat f

ee new mily. 'e is ome s well as sal and live mance; another uzed for elec coustic gu he .hird "dre

١r

050000

The New 9000 :

OF ELVIS' DEATH!

Our newest sourceles are materialising at vojit, local Zoom Dealer - heck them ont or cor-

Scientists Baffled by Expressive **Quality of Flying Wedge16-bit** Stereo Boxes: "Whoever built these little things obviously knows something the rest of us don't."

While musicians rejoice the unexpected arrival of these digital processors, members of the world's scientific community are scratching their heads. How can such such tiny parcelsproduce such expressive tone?

According to Dr. Niels Hartvig-Nielsen, of the Institute of Advanced Intelligence in Ludevisk. Finland: "First we were inpressed by the efficient industrial design. Then we plugged

in. What we heard was beyond the realm of Herstanding."

and others most perplexed is "sive "expressive" 1 to these

is that unable

The Killer-Tone

RE UFOs?

ich Schmardo Germany, points toward he sighted the first wonderbox, which "zoomed nto my hand. It seemed friendly enough, so I just d it right in. Never in all of my days have I or my ors in Wolfsnar[£] heard such beautiful sounds."

> Musician: "I Still Eat Tuna, But Now I Don't Sound Like One!" **Reports from corresponden** around the globe indi sudder change

Believe It



High-Tech Tools for Artistic Expression

Zoom Corporation of America 385 Oyster Point Blvd., #7, South San Francisco, CA 94080; tel (415) 873-5885. World Radio History CIRCLE 82 ON FREE INFO CARD

MADONNA

continued from page 42

was the day of reckoning — my birthday, and yet I was stuck in the studio with Madonna, Tony Shimkin, and an animal-balloon-twisting clown to celebrate it with. It was fun for about five minutes, until Madonna said, "Shep, you gotta get back to work."

AUGUST 15, 1992 - MO'S BIRTHDAY

One of the tracks, "Goodbye To Innocence," just wasn't working. There was something about the song that didn't grab Madonna, so we had to fix it. I worked overnight in my studio and came back to Soundworks with a brand new bass line that seemed to do the trick.

Madonna put on headphones and got ready to lay down the vocals for "Goodbye To Innocence." But instead of singing the original words, which were written last year, Madonna started toying with the lyrics, singing the words to the lounge-lizard act staple, "Fever." At first we thought: "This is cool," and it was. It sounded so good that we decided to take it one step further and actually cover the tune. Too bad no one knew the words. What we needed was a copy of "Fever" if we were going to record it that day. So, Madonna got on the phone with Seymour Stein at Sire Records, and, within an hour, we had the lyric sheets, the Peggy Lee version, and the original version of the song in our hands. I was really impressed by how quickly we got it all. That was the last track on *Erotica* and we finished mixing it just in time to celebrate another birthday — Madonna's.

That night, she had a birthday party on a boat circling Manhattan. Picture about 50 people dancing on a boat with disco blasting out of the portholes and you get the idea. In between dancing and celebrating, I spent the time reflecting on the album. I was confident that it was a great compilation of songs, but I was wondering how people would react to it. It was definitely a different album for her in that it was a dance/pop album, instead of a guitar-laden pop album designed just for top 40. That was a conscious decision on her part because it seemed that the more pop she went, the fewer of her albums people were buying. This time, she's giving the people what they want. And it could feasibly be the bestselling album of all time. We'll see.

RANE

SEPTEMBER - OCTOBER 1992

After three and a half months of working in the same studio and hearing the same songs day after day, it was relief to have the album finished. Everything went smoothly except the last two songs, "Why's It So Hard" and "Words," both of which we had to recall for changes. On September 12, I walked out of Soundworks with the completed master of *Erotica* in my hands.

A month later, I went to the Sex party. The Erotica blitz was about to hit in music, video, and book form and a variety of stars were coming out for the party. Madonna herself surveyed the scene during the midnight hour. I walked over to meet her in the DJ booth.

There was all this wild stuff going around us: people tattooing one another, couples simulating sex — it was crazy. And when I went to talk to Madonna, who was in the middle of it all, our conversations turned to music. For all the multimedia extravaganzas that were braying for her attention, it was still the music that mattered and it was the record that we fawned over. I realized that no matter how far I've come, I still feel the same way that I always did.

And then she put the handcuffs on me. NOT!

INST (Pzo)

INSTRUMENT

4of7 (Out)

Pan

-C-

INSTRUMENT

MicPzo

П

Ē

Now Acoustic Instruments Can At Last Share The Forefront With Electric Instruments Live On Stage, With Full Acoustic Timbre and Minimum Feedback.

PURF

MY FIDELITY

continued from page 114

Jung does on his DMP label. The difference was about the same as the difference between a DAT recording and a 2-track analog recording. DISQ won't need to put out a book with all of the DISQ-equipped studios, I'm just going to have the list tattooed on my forearm.

I pushed the button that

switched over to the

digital-domain processing.

As soon as I hit the button,

my whole life and the life of

someone I didn't even know

passed before my eyes.

QUANTUM LEAP

For you hightechies out there, Stanford University and a few other universities with linear accelerators have been taking the accelerators out of mothballs recently. Apparently, Tesla (radios, AC generators, etc.) was doing

THE

Jut L-Ins-R OP

(C)⁽¹⁾

INSTRUMENT 3

some work on particle beams in the late 1930s and early 1940s. His work was top secret and not many people knew about it. Some papers recently became unclassified and scientists

INSTRUMENT

HEADPHONE

went to work. Electrons are accelerated into an electromagnetic field in close proximity to a permanent magnetic field. A particle beam is regenerated when the magnetic fields are disturbed by the intrusion of the bombarding electrons. The new beam contains a trillion watts of power. So far, who cares, right? Well, here comes the good part: They discovered some time-domain anomalies during the tests. Some reactions seemed to take

place just before the electron hit its target. It was as though a hole appeared in a target before the trigger on the gun was pulled. The beginning of time travel had finally arrived.

You know what this means, right? At next year's AES show we may see prototypes of the first dig-

ital advances (as opposed to delays). Now anybody can be five milliseconds ahead of his or her time.

Ta ta for now, or didn't I just say that? EC

> VOC/AU) 1 IN

CIRCLE 53 ON FREE INFO CARD

FOOT

MAINTENANCE

continued from page 94

AUDIOS AMIGOS

This article will be my last one for EQ. It has been an honor and a thrill for the past three years to write about those things that make a recording studio a technically better place and less of a hassle to work in. I hope that in some small way I have been able to add to the knowledge base provided to you by the other EQ writers.

After nearly 28 years as an engineer, producer, and technician in this marvelous industry, it is time to move on to other pursuits. I am leaving the day-today studio scene to concentrate on my growing studio facility design and Macintosh desktop multimedia company. Maybe someday, I can pass along the information I learn from this endeavor to a whole new segment of EQ readers.

Thanks for having allowed me to be part of EQ since its inception. I am especially grateful for having been allowed to pass on information that was so freely given to me during my career. We are truly the sum of our parts. To play on words, I would like to say, Audios Amigos! EC

MIDI/PROGRAMMABLE ACOUSTIC INSTRUMENT PROCESSOR

Introducing The First Programmable Technologically Advanced Preamplifier/Processor Dedicated to ACOUSTIC INSTRUMENTS.

Rane's renowned signal processing and digital expertise have combined to create a quantum breakthrough in acoustic instrument amplification. The MAP 33 effectively preserves complex acoustic timbre through a unique system that processes two different pick-ups separately with an impressive arsenal of programmable functions: from EQ to separate monitor mix. With 64 configuration memories and full MIDI control, the MAP 33 can satisfy the most demanding venues with a sound quality of unprecedented purity, strength and flexibility! Call us for a detailed brochure and your nearest MAP 33 dealer.

STOR

RANE CORPORATION 10802-47th Ave. W. Mukilteo, WA 98275 • (206) 355-6000



World Radio History

EQ CLASSIFIED





THE **MONEY-MAKING READ FOR YOUR RECORDING &** SOUND **CUSTOMERS** (312) 427-6652



"The most accurate microphone preamps we've ever used." Jack Vad, Chief Recording Engineer, San Francisco Symphony

Optional 130 volt **B&K** Inputs and Plug-in, 18-bit A/D Converters

Millennia

The Brüel & Kjær **Specialists** 916-363-1096

SYMPHONY-CLASS HIGH VOLTAGE MICROPHONE PREAMPLIFIERS OF UNRIVALED TRANSPARENCY AND SPATIAL ACCURACY

Send your sounds direct to disc.

Finally. An affordable way to transfer all of your audio to compact disc.

Take advantage of the latest in audio storage & playback technology.

You and your clients benefit from this short-run CD recording service.

Get all the advantages of tight cues, digital storage, and long life on the most durable format available.



Call for FREE brochure: (800) 524-5706, Ext. 155 **RACK AND CHASSIS BOXES**



MUSIC FOR C SEQUENCING - From \$39 to \$99 NOTATION - \$99. TRAINING - From \$39. FREE CATALOG BEST PRICES MIDI & SOUND BOARDS - \$79. **SINCE 1981** OPTRONICS + (503) 488-5040 Box 3239 Ashland, OR 97520

THE IBM

TIPS 100 HOT

Drastically improve the sound of your recordings! 100 production techniques used by industry professionals. Will work in any environment. Hear the difference! Send \$5.00. Check or money order only. Allow 4-6 weeks for delivery. Payable to: S&S Marketing, Dept. EQ/PD Box 5454/Cayce, SC 29171





Windmark will take your 1/2", 1/4", DAT or cassette recording and transfer it to CD for only \$150!!



Full Audio Production Facilities Call (804) 464-4924 4924 Shell Rd., Virginia Beach, VA 23455

World Radio History

EQ CLASSIFIED

Midivox[®] Pro Voice Controller Factory Direct 1-800-433-MIDI Growl a Trumpet–Croon a Sax Hum a Bass–Vocal some Bongos **Become a Human Sequencer!** 2 Year Warranty \$1,595.00 30 Day Money Back Guarantee!



MUSIC PRODUCER lew book reveals exclusive details about today's top producers' classic sessions and trademark sounds. Features Daniel Lanois, Bill Laswell, Jeff Lynne, Hugh Padgham, Rick Rubin, Don Was and 18 others. From the publishers of *Mix* magazine. \$19.95 ppd. Mix Bookshelf, 6400 Hollis St. #12-T, Emeryville, CA 94608; (800) 233-9604.





HEAR THE FUTURE



Join the OFFICIAL user's group of Korg USA. Subscription to our quarterly publication. Korg Connection, is only \$14 in the U.S, \$20 in Canada (U.S. funds only). Make checks payable to Korg Connection. Write to us if you'd like additional info: Korg Connection, P.O. Box 43742, Tucson, AZ 85733-3742.

FREE MANNY'S 1ST EVER MUSIC CATALOG. MANNY'S BRINGS 48TH ST. TO YOUR DOORSTEP. **BECOME A CHARTER** SUBSCRIBER. SEND NAME AND ADDRESS TO **AUDIOTECHNIQUES C/O** MANNY'S MAIL ORDER 1600 BROADWAY, #3 NY, NY 10019

DISC MAKERS



Instru Ipportunities . Recording Services ce • Studios for Sale • Equip cessories • Sounds & Software • Instruction • Employment Opportun REAL TIME DUPLICATION



All master tape formats accepted • Every tape checked!

For a free brochure: 1-800-456-GLAD(4523) TEL 716-532-5098 • FAX 716-532-5061

ALL MATERIALS AND WORKMANSHIP GUARANTEED! MASTERCARD, VISA & AMEX ACCEPTED.

GLAD PRODUCTIONS 13758 Rte. 62 • Collins, N.Y. 14034

EQ CLASSIFIED



World Radio History

434 West Levers Ploce, Orange, CA 92667 (817) 847-5400 FAX (817) 847-6319

1-800-365-TAPE(8273)

My Fidelity



The final word on the AES show (for this issue, anyway)

BY NOW you've already read EQ's Blue Ribbon winners from AES. Personally speaking, there were quite a few things that turned my head more than once (besides the Italian blonde astrophysicist).

— JVC showed what somebody called the "Swiss Army DAT." It is a small portable that resembles my kid's GO-BOT. The A/D converter slips off the DAT and can be replaced by a stereo microphone with a built-in A/D converter. Now you have a one-piece unit for interviews or for whenever you don't want to deal with detached mics and mic cords. It was pretty cool. It will also record from the analog inputs at 44.1 kHz. Only some expensive portables will let you do that.

- A PC-based hard disk audio editing system called SADiE knocked my socks off. (Of course, they did need changing after four days of wandering around the AES show.) I will give you more details soon. (About the SADiE, not my socks.)

— Roland showed its new inexpensive sample rate converter. It also has two asynchronous inputs that can be mixed together digitally. You can also VSO the input. Way hip.

— Sony showed its Super Bit Mapped CD. I could hear a little difference between the normal 16-bit and the Super Bit Mapped versions of program material, but the only way you can really tell is to use them under fire. How can you tell which is the closest to correct unless you can compare it with the original? Or maybe it shouldn't matter any more. Maybe all of these new digital things should sound better than the original. It seems to be happening a lot these days.

— Meyer Sound showed its new HD-2 studio monitors. They use a horn-loaded titanium dome tweeter and a larger 12-inch woofer. They go 10 dB louder than the HD-1's and seem to extend lower in frequency response. By the time you read this, I will have used them to mix Donald Fagen's new album. Sometimes the only time you can tell what's going on is to jump in with both feet and hope you don't break your ankles.

 I saw Tascam's entry into the 8-track digital arena. As you've read elsewhere in this issue, it is based on 8mm videotape instead of on the VHS format. Multiple units can lock up to sample accuracy, and there is an optional remote for controlling multiple units, a similar arrangement to that of Alesis. One neat feature is the plug-in SMPTE sync card that will let a single machine chase SMPTE without having to purchase the remote. If you have a stack of Tascam 8-track digitals, you only need one SMPTE sync board to lock up the whole bunch, flock, herd, gaggle, brood, or whatever you're supposed to call a group of 8-track digitals.

— Oh, by the way, Fostex announced that its 8-track digital will be completely compatible with Alesis.

I have a sneaking hunch that it actually will be an ADAT with a Fostex face plate. That sure sweetens the pot for the Alesis format, doesn't it?

BEST IN SHOW

Overall, the best thing at the show was the AT&T digital mixing box, called "DISQ." AT&T demonstrated the product with a Sony PCM 3348 playing a Linda Ronstadt tape that George Massenberg supplied. The 48 track was hooked up analog through a Neve VR console with GML automation, and monitored on Meyer HD-1 speakers. The 48 track also had a set of digital cables going to the DSP-3 box. The digital audio was mixed internally and sent to a pair of 20-bit D/A converters.

The only thing the DSP-3 used the Neve console for was control surface input. The total-recall computer would read the position of an EQ knob, for instance, and tell the DSP-3. The DSP-3 would then emulate, in the digital domain, whatever EQ, dynamics, effect sends, returns, and level control were wanted.

The tape rolled and we listened to the mix through the Neve console. That George Massenberg really pisses me off! If his recordings start sounding any better I'm going to have his fingers broken! (Sorry, I got carried away for a second.)

Anyway, the tape sounded great. Nice horns, smooth vocals, clean orchestral percussion, what more could you ask for. I then pushed the monitor button that switched over to the digital-domain processing of the same material. I got ready to try really hard to hear the difference. As soon as I hit the button, my whole life and the life of someone I didn't even know passed before my eyes. I thought my teeth were going to fall out on the floor (and they're not even false). Everyone in the room gasped. It sounded great! We started soloing tracks to see if the difference was as pronounced with individual instruments. Yup, the difference was just as pronounced. There was a clarity that you only hear on direct-to-2-track digital recordings like the ones Tom

continued on page 109

John Arrias and DIC///DAT...

achieving the Ultimate in Sound!

DICIDAT

DICIDAT

The Ultimate in Sound

222 Bridge Plaza South. Fort Lee, NJ 07024 201-224-9344 • 800-328-1342 • Fax: 201-224-9363 Technology in its highest form. That's the John Arrias trademark witnessed on albums he engineered for Barbra Streisand, Bob Seger and countless other artists who demand his distinctive sound of perfection.

John developed his own audio preservation system which transforms older analog recordings into crisp, clear digital sounds. And to back up these and new recordings, he uses four digital audio tape computer drive units. Whether mixing down or backing up, John relies on DIC Digital's MQ series DAT and 4 mm data grade metal particle tape products.

The only cassette made in the U.S.A., the MQ shell has increased stability and structural integrity so it can stand up to rigorous studio use inflicted by the most astute engineers.

MQ's proprietary metal particle technology is enhanced with a formulation finish that reduces friction in the tape to head interface. Combined with its dust-busting design, the MQ cassette virtually eliminates environmental contamination that leads to recording errors and tape jamming.

For impeccable recordings, today's engineers use DIC Digital's DAT tape products for the ultimate digital sound.

CIRCLE 26 ON FREE INFO CARD

© DIC DIGITAL SUPPLY CORPORATION 1992

The 4200 Series. Designed For The Control Room, Not The Living Room.

Today's recording studio has evolved into a multi-function facility which simultaneously addresses the specialized needs of music recording, film and video post, and radio production. In this environment, where the most critical listening often occurs in the final mix, close proximity monitors are often more important than the mains. The problem: most console top monitors, unfortunately, were designed for the living room not the control room. Until now.

With the 4200 Series we're taking our stand from where you sit: right where you work at

the console. Designed, engineered and tested from this position, the 4200 Series is the first console mount monitor created specifically for the professional recording environment.

Both models give you pin-point imaging by delivering high and low frequency information to your ears at precisely the same instant. By virtue of their symmetrical design the 4200 Series monitors are mirror imaged.

And so nothing gets in the way of your music,

the 4200 Series introduces our uniquely sculpted Multi-Radial[™] baffles incorporating newly designed pure titanium tweeters and low frequency transducers. The combination of these technologies successfully corrects time arrival anomalies and eliminates baffle diffraction distortion.

4200 Series: console top monitors designed in the studio, for the studio, with sonic performance rivaling much more expensive monitors. 4200 Series: the shape, and sound, of things to come. Available at your local authorized JBL Professional dealer.





JBL Professional 8500 Balboa Boulevard, Northridge, CA 91329 ➡ A Harman International Company









Distortion vs. Frequency (Model 4208) 96 dB at 1 m, typical console listening levels (distortion raised 20 dB)

World Radio History