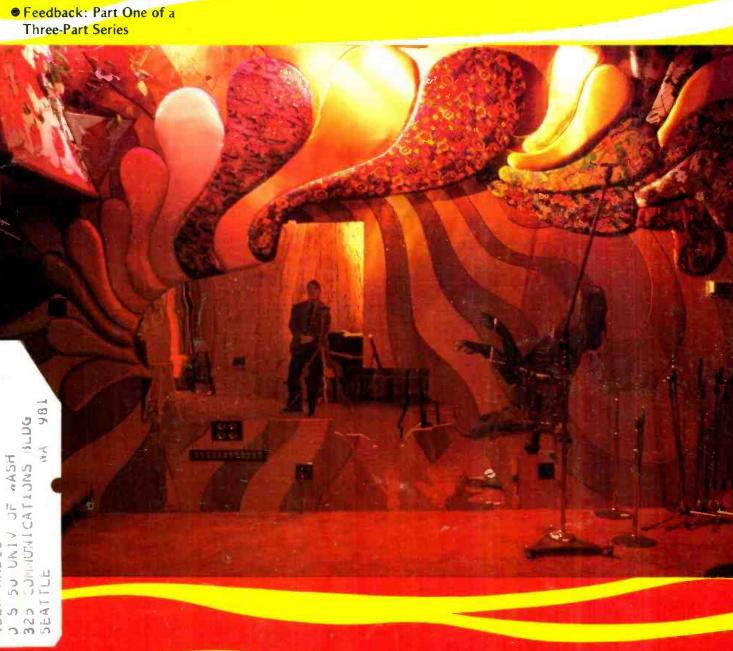
KAU1U

100



IN THIS ISSUE:

- Calibrate Microphones By Reciprocity
- A Simple and Superior **Microphone Preamplifier**







Better doesn't have to cost more.

Circle 10 on Reader Service Card

Accept our invitation to contact us and discuss your studio needs. 14045 Sherman Way, Van Nuys, California 91405 / (213) 873-444

20. The Audio Cyclopedia (2nd ed.). Dr. Howard M. Tremaine. New and updated, here is the complete audio reference library in a single volume. It provides the most comprehensive information on every aspect of the audio art. This new edition includes the latest audio developments including the most recent solid-state systems and integrated circuits. It covers all subjects in the fields of acoustics, recording, and reproduction with more than 3400 related topics. Each topic can be instantly located by a unique index and reference system. More than 1600 illustrations and schematics help make complicated topics masterpieces of clarity, 1760 pages, 61/2 x \$34.00 9% hardbound.

1. The Technique of the Sound Studio. Alec Nisbett. This is a handbook on radio and recording techniques, but the principles described are equally applicable to film and television sound. 264 pages; 60 diagrams; glossary, indexed; $5\frac{1}{2} \times 8\frac{1}{2}$; clothbound. \$14.50

13. Acoustic Design & Noise Control. Michael Rettinger. 1973. NEW, revised and enlarged edition covers physics of sound, room acoustics and design, noise and noise reduction, plus noise and its problems. Many charts and graphs. A practical and useful book. 562 pgs. \$22.50

16. Magnetic Recording. Charles E. Lowman. Reference guide to the technology of magnetic recorders used in audio recording. broadcast and closed-circuit TV, instrument recording, and computer data systems. Includes latest information on cassette and cartridge recorders; TV recorders; direct and FM signal electronics from low to wideband; servo-control and signal record/playback circuitry; capstan, reel. and head-drum servos for longitudinal, rotary, helical-scan, and disc recorders. Glossary, index. bibliographical information. 274 pp. \$17.50

28. Environmental Acoustics. Leslie L. Doelle. Applied acoustics for those in environmental noise control who lack specialized acoustical training. Basic information in comprehensible and practical form for solving straightforward problems. Explains fundamental concepts; pure theory minimized. Practical applications stressed. acoustical properties of materials and construction listed, actual installations with photos and drawings. Appendixes illustrate details of 53 wall types and 32 floor plans and other useful data. 246 pgs. \$21.75

39. Reference Data for Radio Engineers. *ITT Staff.* 5th Ed. The latest of one of the most popular reference books for radio and electronics engineers as well as libraries and schools. Complete, comprehensive reference material with tables. formulas. standards and circuit information. Contains 45 chapters. 1196 pages with hundreds of charts. nomographs. diagrams, curves. tables and illustrations. Covers new data on micro-miniature electronics. witching networks. quantum electronics. etc. **\$23.00**

24. Basic Electronic Instrument Handbook. Edited by Clyde F. Coombs, Jr. Hewlett-Packard Co. A basic reference background for all instruments. Offers saving in time and effort by having complete information in one volume on how to get the most benefit from available devices, how to buy the best instrument for specific needs. Reduces chances of costly errors. Ideal reference book, it is an excellent source for the beginner, technician, the non-electrical engineering man, or general non-engineering scientific and technical personnel. 800 pages. Hardbound. **\$29.50** 25. Operational Amplifiers-Design and Applications. Burr-Brown Research Corp. A comprehensive new work devoted entirely to every aspect of selection, use, and design of op amps—from basic theory to specific applications. Circuit design techniques including i.c. op amps. Applicatons cover linear and non-linear circuits, A/D conversion techniques, active filters, signal generation. modulation and demodulation. Complete test circuits and methods. 474 pages. \$18.00

33. Noise Reduction. Beranek. Designed for the engineer with no special training in acoustics, this practical text on noise control treats the nature of sound and its measurement, fundamentals of noise control, criteria, and case histories. Covers advanced topics in the field, 1960, 752 pp. \$26.00

32. Circuit Design for Audio, AM/FM, and TV. Texas Instruments. Texas Instruments Electronics Series. Discusses the latest advances in design and application which represent the results of several years research and development by TI communications applications engineers. Emphasizes time- and cost-saving procedures. 1967. 352 pp. \$18.00

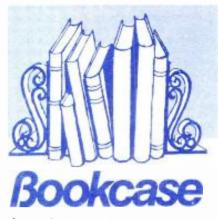
31. Solid-State Electronics. Hibbard. A Basic Course for Engineers and Technicians. An extremely practical reference book for anyone who wants to acquire a good but general understanding of semiconductor principles. Features questions and answers, problems to solve, 1968. 169 pp. \$12.50

35. An Alphabetical Guide to Motion Picture, Television, and Videotape Productions. Levitan. This all-inclusive, authoritative, and profusely illustrated encyclopedia is a practical source of information about techniques of all kinds used tor making and processing film and t.v. presentations. Gives full technical information on materials and equipment, processes and techniques. lighting. color balance, special effects. animation procedures, lenses and filters. high-speed photography. etc. 1970. 480 pp. **\$28.50**

40. Radio Transmitters. Gray and Graham. Provides, in a logical, easy-to-understand manner, a working knowledge of radio transmitters for quick solution of problems in operation and maintenance, 1961. 462 pp. \$17.50

37. Television Broadcasting: Systems Maintenance. Harold E. Ennes. Covers maintenance of the t.v. broadcasting system from switcher inputs to antenna. Theory and operation of systems, tests and measurements. including proof of performance for both visual and aural portions of the installation. Many illustrations. A thorough treatment of modern television maintenance practice. 624 pgs. \$16.95

41. Modern Sound Reproduction. Harry F. Olson. A basic text covers amplifiers, microphones. loudspeakers, earphones, tape systems, film sound, tv and sound reinforcement — the significant elements and systems of modern sound reproduction. Employs simple physical explanations which are easily understood without special engineering training. Highly recommended text and reference. 328 pages. \$18.50



As a service to our readers we are pleased to offer books from prominent technical publishers. All prices listed are the publishers' net. Shipping charges are included.

To order use the coupon below. Indicate quantity on the special instructions line if more than one copy of a title is wanted. Full payment must accompany your order. We cannot ship c.o.d. Checks or money orders should be made payable to Sagamore Publishing Company, Inc. Because of the time required to process orders, allow several weeks for the receipt of books.

Sagamore Publishing Company, Inc. 1120 Old Country Road, Plainview, N.Y. 11803

Please send me the books I have circled below. My full remittance in the amount of \$______ is enclosed, N,Y. State residents add 7% sales tax.

1	2	3 4	5	6	7 8	9	10	11
12	13	14	15	16	17	18	19	20
21	22	23	24	25	26	27	28	29
30	31	32	33	34	35	36	37	38
39	40	41	42	43	44	45	46	47
Nan	ne			-				-
Add	Iress		-				-	
City		invite:	17	-				-
Stat	e					Zip		
	cial ructi	ons						
-	-						-	
Gan	ada	and f	oreid	ID. A	00 51	.00 1	Jet D	OCK.

Ganada and foreign. Add \$1.00 per book

38. Television Broadcasting: Equipment, Systems, and Operating Fundamentals. Haroid E. Ennes. An extensive text covering fundamentals of the entire television broadcast system. Excellent for new technicians and operators and as a source of valuable reference data for practicing technicians. Discusses NTSC color systems, camera chains, sync generators, recording systems, mobile and remote telecasts, t.v. antenna systems. Tables, glossary, exercises and answers. 656 pgs. \$16.95

26. The Design of Digital Systems. John B. Peatman. Textbook for students desiring to develop a creative approach to design capability through a digital systems approach. Answers these questions: Under what circumstances it is desirable to implement a system digitally? What are some of the components available for implementing the system? How do we go about designing it? 448 pages. \$18.50

If you work with microphones, you need this book!

The most important microphone book ever published.

LOU BURROUGHS

LC #73-87056 ISBN #0-914130-00-5

licrophones: Design and Application

A practical, non-theoretical reference manual for those involved in the application of microphones for tv, motion pictures, recording and sound reinforcement.

At last, the practical aspects of microphone design and application have been prepared and explained in one concise, fact-filled volume by one of audio's outstanding experts. This book is so full of useful information, we think you'll use it every time you face a new or unusual microphone problem.

Perfect for Reference or Trouble-shooting

The twenty-six fact-packed chapters in this indispensable volume cover the field of microphones from physical limitations, electro-acoustic limitations, maintenance and evaluation to applications, accessories and associated equipment. Each section is crammed with experience-tested detailed information. Whatever your audio specialty — you need this book!

Along with down-to-earth advice on trouble-free microphone applications, author Lou Burroughs passes on dozens of invaluable secrets learned through his many years of experience.

- He solves the practical problems you meet in day-to-day situations. For example:
- * When would you choose a cardioid, omni-directional, or bi-directional mic?
- * How are omni-directional mics used for orchestral pickup?
- * How does dirt in the microphone rob you of response?
- * How do you space your microphones to bring out the best in each performer?

This text is highly recommended as a teaching tool and reference for all those in the audio industry. *Price: \$20.00*

THE AUTHOR

Holder of twenty-three patents on electro-acoustic products, Lou Burroughs has been responsible for extensive contribu-tions in the development of the microphone. During World War II, he developed the first noise cancelling (differential) microphone, known as the model T-45. Used by the Army Signal Corps, this achievement was cited by the Secretary of War. Burroughs was the creator of acoustalloy, a non-metallic sheet from which dynamic diaphragms are molded. This material made it possible to produce the first wide-range uniform-response dynamic microphone. Burroughs participated in the design and development of a number of the microphones which have made modern broadcasting possible - the first one-inch diameter wide-range dynamic for tv use; the first lavalier; the first cardiline microphone (which ultimately won a Motion Picture Academy award) and the first variable-D dynamic cardioid microphone. He also developed the first wind screens to use polyester foam. Burroughs was one of the two original founders of Electro-Voice, Inc. He is a charter member of the Society of Broadcast Engineers and a Fellow member of the Audio Engineering Society.

ORDER FORM

Sagamore Publishing Co., Inc. 1120 Old Country Road, Plainview, N.Y. 11803

Please send [] copies of MICROPHONES: DESIGN AND APPLICATION at \$20.00 each.

Name____

Address.

City____

Total amount \$____

Zip

N.Y.S. Residents add 7% Sales Tax.

State

Enclosed is check for \$_____ Foreign Orders add \$1 postage and handling

coming next month

• The source of the polished and impeccable programs Americans associate with British imports is examined in detail in THE STATE OF BRITISH BROADCASTING, by distinguished engineer Angus McKenzie.

• A BETTER WAY TO COPE WITH STEREO PHASING is offered by Harold E. Ennes, describing a simpler and more accurate indication of phase differences than the X-Y scope plot.

• If you've come across an expensive Catch-22 situation after obtaining a new transport with a d.c.-controlled servo system and discovering that your existing auxiliary equipment is not compatible, Brian C. Lowe has an ingenious solution, telling How to Con-VERT A 92C RESOLVER.

• Our regular columnists will be on hand with their special views of the audio panorama.



• An unusual view of Record Plant's plush studio in Sausalito, Calif., showing its "subdued freak" decor that was designed with the richer labels in mind.



- A SELECT/CANCEL CHANNEL CONTROL M. C. Volker
- 22 BEING PRACTICAL ABOUT FEEDBACK, part 1 Norman H. Crowhurst
 - A SIMPLE AND SUPERIOR MICROPHONE PREAMPLIFIER R. S. Mintz
- 3) CALIBRATE MICROPHONES BY RECIPROCITY Michael Rettinger
 - INDEX TO ADVERTISERS
 - LETTERS

16

28

2

4

6

- FREE LITERATURE
- 8 CALENDAR
- 10 THEORY AND PRACTICE Norman H. Crowhurst
- 13 SOUND WITH IMAGES Martin Dickstein
- 18 NEW PRODUCTS AND SERVICES
- 33 CLASSIFIED
- 36 PEOPLE, PLACES, HAPPENINGS

db is listed in Current Contents: Engineering and Technology

Larry Zide Robert Bach PUBLISHER EDITOR John Woram **Bob Laurie** ART DIRECTOR ASSOCIATE EDITOR **Eloise Beach** Hazel Krantz COPY EDITOR CIRCULATION MANAGER Lydia Anderson Ann Russell ASST. CIRCULATION MANAGER PRODUCTION GRAPHICS Crescent Art Service

db ine Sound Engineering Magazine is Published monthly by Sagamore Publishing Company. Inc. Entire contents copyright © 1975 by Sagamore Publishing Co., Inc., 1120 Old Country Road. Plainwew L.L., N.Y. 11803. Lephone (516) 433-6530, db is published for those individuals and firms in professiontal audiorecording, broadcast, audio-visual, sound reinforcement, consultants, video recording, film sound, etc. Application should be made on the subscription form in the rear of each issue. Subscriptions are \$7.00 per year (\$12.00 per year outside U.S. Possessions, Canada, and Mexico) in U.S. funds. Single copies are \$1.00 each Controlled Circulation postage paid at Harrisburg, Pa. 17105. Editorial, Publishing, and Sales Offices: 1120 Old Country Road, Plainview, New York 11803. Postmaster: Form 3579 should be sent to above address

BOZAK'S new model CMA-10-2DL Mixer/Preamplifier

Tailored for the DISCOTHEQUE; suited to the BROADCASTER

The stage at a discotheque, the console in a broadcast station—no matter where a disc jockey works, his success depends on three factors which are combined in Bozak's new mixer preamplifier—superb fidelity, outstanding control flexibility and unsurpassed reliability.

Combining four stereo inputs and 2 monaural inputs, the model CMA-10-2DL permits mixing audio signals from a variety of sources into a single stereophonic output. Changing among high-level tape. low-level phono and or microphone inputs is easily accomplished by changing plug-in circuit boards.

Reliability is at the highest level of studio equipment standards with allsilicon circuitry permitting continuous operation at full rated output.

The partial specifications tell part of the story; only your ears can tell the rest.

Specifications:

Gain-Microphone. 75dB Phono. 66dB High-Level. 27dB

Output. +18dBm

Frequency response (+1dB), 20 to 20.000 Hz

Distortion at 1kHz. less than 0.2%

Output hum and noise.below rated output.-125dBm

Input impedance-Microphone. 200 ohms High-level. 50k ohms

 $\begin{array}{l} {\sf Equalization-Bass.~\pm10dB} \\ {\sf Treble.~\pm10dB} \end{array}$

For complete specifications on the Model CMA-10-2DL or for a copy of our complete commercial sound equipment catalog. write:



Commercial Sound Department BOZAK, INC. Box 1166 Darien. Connecticut 06820

index of advertisers

Audio Designs .			•			12
Auditronics	•			С	ove	: 3
Bozak	•	•				2
dbx, Inc	•				,	14
Electro-Voice				•		7
ESE Digitals						32
Everything Audio				С	ovei	2
Garner Industries						4
Infonics, Inc.						6
Inovonics						20
J. B. Lansing						11
Jensen Tools						20
Lexicon						19
Orban/Parasound					21,	
Peavey Electronics						23
Philips Audio AKG						5
Polyline				•		20
Quantum Audio .						23
R. A. Neilson .						8
Ramko Research		•				9
Rauland-Borg .				•		18
Shure Brothers .						15
Sphere Electronics						14
Soundcraftsmen .				,		4
Sound Workshop			•			8
Standard Tape Labs						6
Stanton Magnetics						17
Teac Corp. of Ame	rica	a				3
Willi Studer						27
Woram Audio			•			35
Yamaha Musical.	•			С	ovei	4



) sales offices

THE SOUND ENGINEERING MAGAZINE

New York

1120 Old Country Rd. Plainview, N.Y. 11803 516-433-6530

Roy McDonald Associates, Inc.

Dallas Stemmons Tower West, Suite 714

Dallas, Texas 75207 214-637-2444

Denver

3540 South Poplar St. Denver, Colo. 80237 303-758-3325

Houston

3130 Southwest Freeway Houston, Tex. 77006 713-529-6711

Los Angeles

500 S. Virgil, Suite 360 Los Angeles, Cal. 90020 213-381-6106

Portland

2035 S. W. 58th Ave. Portland, Ore. 97221 503-292-8521

San Francisco

Suite 265, 5801 Christle Ave. Emeryville, Cal. 94608 415-653-2122

db September 1975

Circle 13 on Reader Service Card



The half-inch 8-track. \$3490.00, less console.



The best multitrack tape machine made can't help you at all unless you can afford it. We don't say our tape recorders are the best you can buy. We do say they're good enough to produce commercial product.

And that's what counts. Because quality is as much a matter of talent as tools. Because the true test of what comes off a tape is what goes on in the first place.

Now if you're rich and famous (or just rich) you probably don't need us. But if you have more talent than money, check out the TEAC/Tascam Series 70. You can find your nearest Tascam Series dealer by calling (800) 447-4700. In Illinois, call (800) 322-4400. We'll pay for the call.

> TEAC TASCAM SERIES

When you've got more talent than money.

The half-inch 4-track. \$1990.00, less console.



ω

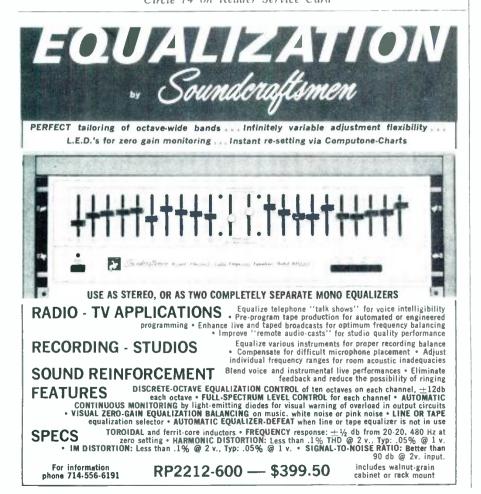
After you use the 1056, we'll know one thing about your dub quality: it just got better.

Professional studios that make lots of dubs for radio, welcome the speed and quality they get using the Garner 1056. It offers a whole new set of advantages for producers of reel-to-reel duplicates for radio, AV, or educational needs. Some of those are: • Single capstan drives the master and all five copies. • Solid-state electronics and special heads provide outstanding frequency response. • Two-speed drive allows either 30 or 60 i.p.s. duplicating. • Extra-fast rewind of master tape speeds production. • Unique forward tilt of transport mechanism aids threading. • Conveniently located controls feature push button operation.





GARNER INDUSTRIES 1200 NORTH 48TH STREET LINCOLN NEBRASKA 88504 Circle 14 on Reader Service Card





THE EDITOR:

Your article on San Francisco audio was a surprise, and I was anticipating something interesting when I turned to it. Too bad for me. I found it pretty trivial, and I think the author missed the point anyway. Those garage studios are probably the most important element in the San Francisco scene for anyone except people with a vested interest in expensive equipment and high overhead recording. I, for one, would have liked to have found out a good deal more about these studios-where they are, why the people are operating garage studios, what kinds of equipment and expertise they offer, etc. As a producer of records in the Bay Area, this information is valuable to me. That froth in the article about ". . . recording-in-the-round with a twist" is a drag.

I note that Leo's recording school is certainly not the only one in the Bay Area. Raccoon Studios in Tomales, Family Light in conjunction with the Church in San Rafael, and Blue Bear Studios all have offered recording courses in the past year, and. so far as I know, still do.

> ED DENSON Kicking Mule Record Co. Berkeley, California

THE EDITORS:

I have noted several errors in my article, "A Simple High-Quality High-Speed Tape Duplicator." which would be misleading to one making the modifications. First, the heavy vertical line of Figure 2 should not be discontinuous. This diagram shows how to apply the silver paint to the head, should it be necessary. The paint must form a continuous path for the grounding of the static charges that build up on the heads as a result of tape passage. Another item which may give trouble is Figure 5, where V7 and V6 are shown as 12AT7s. The tube indicated as V6 is not V6. If the real V6 were to stand up, it would be the tube directly above the one shown as V6. the one tube in the picture which has no designation. The one incorrectly called V6 is actually a 12AX7 which serves the purpose of meter amplifier and rectifier. If someone changes that one to a 12AT7. he will certainly do the recording circuitry no good and will be upsetting the metering circuit.

BOB S. WHITE Sumas, Washington 98295 We put some of our best heads together and came up with a mike that will satisfy almost everybody. The multi-module AKG electret condenser system.

It's an AKG condenser microphone. It's an electret. It can be phantom powered also, (Your choice.)

A twist of the wrist changes modules from a cardiold-to a cardioid/integral windscreen-to a omni-directional-to a lavelier.

To make sure the system performs at top specifications we make it with FET pre-amplifiers and gold-vapored Teflon^M diaphragms in the modules. Humidity can't hurt it. (Temperature/Humidity range from -5° F/99% to $+125^{\circ}$ F/95%.) And aging won't cause deterioration either.

It's a professional system in every way. Sensitivity variance between modules is remarkably low.

College stations and studios on a tight budget find the AKG compact electret system an easy step-up to AKG quality without straining the budget. All four modules are interchangeable on the power supply/handle. That saves equipment purchases.

The AKG electret system components are all available separately or in several economical combinations. They're perfect for young recording outfits that need growing room without obsoleting anything.

See your professional equipment dealer. Or write to us. Good things happen when you put our heads together.

AKG MICROPHONES • HEADPHONES distributed by: PHILIPS AUDIO VIDEO SYSTEMS CORP. AUDIO DIVISION 91 McKee Drive, Mahwah, New Jersey 07430 A North American Philips Company



Four heads are better than one.

Introducing the AKG multi-module miracle mike.





when you use **STL magnetic Test Tapes**

STL magnetic test tapes are widely used by major recording studios, equipment manufacturers, government and educational agencies throughout the world. The most comprehensive test tapes made they are offered in 1" and 2" sizes as well as flutter tapes and all other formats.

You know your system is in step with the rest of the industry, compatible and interchangeable, when you employ STL tapes, the most accurate reference available.

Write for a free brochure and the dealer in your area.

Distributed exclusively by Taber Manufacturing & Engineering Co.



STANDARD TAPE LABORATORY, Inc.

2081 Edison Avenue San Leandro, CA 94577 (415) 635-3805

Circle 17 on Reader Service Card

The duplicator !

We'll solve your tape duplicating problems . . . large or small ... whatever your operating budget! A complete, professional line of tape and cassette duplicating gear . . . reel-to-reel, reel-to-cassette, cassette-to-cassette.

System 200 offers professional quality cassette/tape duplicating. D-8 and RR-Series duplicators produce multiple reel-to-reel dubs for recording studios and commercial duplicating firms. The Model 102 Cassette Copier . . . for the office or library produces two superb C-30 cassette duplicates in one minute!

Call Infonics Sales (219 879-3381) today ... or write to: Infonics, P.O. Box 1111, 238 Hwy. 212, Michigan City, Indiana 46360

Export Representative: Singer Praducts Co., Inc. - One World Trade Center Suite 2365 - New York, NY 10048 - Telex: 423592 SPC UI



FREE LITERATURE

To obtain the free literature described. please circle the appropriate number on the Reader Service Card in the back of the magazine.

MUSICIANS' CLEARINGHOUSE

Musicians desiring to buy or sell equipment, to offer or learn about employment, may use the classified services of a new publication, FUSION. A free copy may be obtained. Distributor: Dimension Five

Circle No. 90 on R. S. Card.

PRODUCT ROUNDUP

Descriptive literature is available regarding the following products: #800A mobile f.m. signal generator; #940 audio intermodulation meter; #950 CATV measurement field strength calibrator. Mfr: McGraw-Edison Co.

Circle No. 91 on R. S. Card.

MULTI-TURN TRIMMER

A two-page color data sheet on model 64 3/8-inch multi-turn square cermet trimmer, with six product photos shows configurations, line drawings, specifications, resistance values, and ordering information. Mfr: Spectrol Electronics Corp.

Circle No. 92 on R. S. Card.

HIGH VOLTAGE POWER TRANSISTORS

A new line of 125 watt high voltage power transistors, with graphic reports, is listed in this 4-page data sheet. Mfr: International Rectifier

Circle No. 93 on R. S. Card.

PORTABLE AND BENCH TEST INSTRUMENTS

Portable and bench test instruments, including oscilloscopes, multimeter, function generator, curve tracer, f.e.t. multimeters, semiconductor testers, color bar generators, tube testers, CRT tester/rejuvenator, sweep and marker alignment generator, are listed in catalog 75CBA. Mfr: Hickok Electrical Instrument Company

Circle No. 94 on R. S. Card.

EMERGENCY CONTROL SYSTEMS

Audio evacuation and panic control systems are described in this brochure, which details means of conforming to fire safety requirements in high rise buildings. Mfr: Bogen Division, Lear Siegler, Inc.

Circle No. 95 on R. S. Card.

AUDIO COOKBOOK

Audio Cookbook is the name of a 20-page book describing standard audio tests using the 5L4N spectrum analyzer. Mfr: Tektronix

Circle No. 96 on R. S. Card.

G

Circle 18 on Reader Service Card

Sometimes you want lots of proximity effect.

"The Mike With Guts." E-V's New 671 Single-D Cardioid.

Proximity effect. It's that husky bass boost a singer gets working close to the mike. It's just one of the things our new 671 does better than other mikes. Make a comparison test. We think you'll find that the 671 provides greater gain before feedback than the mike you are using now—or any competitive mike. You'll also find that our sophisticated shock mounting assures superior rejection of handling noise. And it's got all the tough-as-nails ruggedness you expect from an Electro-Voice microphone.

Sometimes you don't.

"The Clean Mike." E-V's New 660/661 Continuously Variable-D Super Cardioid.

Successor to the famed 664 ("The Buchanan Hammer"), our new 660/661 minimizes proximity effect to deliver clear, crisp sound at any working distance. Frequency response, both on and off axis, is continuously smooth and uniform. Rear sound rejection capabilities are excellent. The 660/661 mike is the one mike for doing the most jobs best. The same professional performance as our famous RE series at less than professional price.

The 661 has a high/low impedance switch. The 660 and 671 have no-solder impedance change ethat takes less than a minute.



660 clamp-mounted...\$72.00, 661 stud-mounted...\$75.00; 671... \$68.70 (suggested resale net-slightly higher in western states). Electro-Voice, Inc. Dept. 652BD, 686 Cecil Street Buchanan, Michigan 49107 Circle 11 on Reader Service Card



CALENDAR

SEPTEMBER

- 15-17 NOISE-CON'75 National Conference on Noise Control Engineering. Gaithersburg, Md. Pre-seminar at the Shoreham Hotel, Washington, D.C. Sept. 11-13. Contact (914) 462-6719.
- 21-24 International MUSEXPO '75. Las Vegas, Nev. Contact: Roddy Shashoua, International MUSEXPO, 1350 Ave. of the Americas, New York, N.Y. 10019, (212) 489-9425.
- 28- SMPTE Technical Conference
- Oct. 3 and Equipment Exhibit. Century Plaza Hotel, Los Angeles. Contact: SMPTE Conference. 862 Scarsdale Ave., Scarsdale, N.Y. 10583.
- 29-30 N.Y. Chapter of ERA, Commercial Sound & Communications Show, Statler-Hilton Hotel. New York City. Contact: GIM Sales Corp., 375 N. Broadway, Jericho. N.Y. 11753 (516) 433-4080.

First Congress of the FASE on Acoustics. Groupement des Acousticiens Francaise (GALF). Paris. France. Secretariat: C.N.E.T., Issy-les-Moulin-eaux 92. Paris.

OCTOBER

Fall Conferences, National Association of Broadcasters, Contact: NAB, 1771 N St., N.W., Washington, D.C. 20036, (202) 293-3500.

- 12-14 Atlanta
- 15-17 Boston
- 21-26 International Audio Festival Fair. London. England. Contact: British Information Service, 845 Third Ave., New York, N.Y. 10022. (212) 752-8400.
- 31- Audio Engineering Society 52nd
- Nov. 3 Convention, Waldorf-Astoria Hotel, New York, N.Y. Contact: AES, Room 929, 60 E. 52nd St., New York, N.Y. 10017, (212) 661-8528.

NOVEMBER

- 4-7 Meeting of the Acoustical Society of America. San Francisco, Ca.
 NAB Fall Conferences. Contact: NAB, 1771 N St., N.W., Washington, D.C. 20036. (202) 293-3500.
- 9-11 New Orleans
- 12-14 Chicago
- 16-18 Denver
- 19-21 San Francisco

You can buy one channel of really good spring reverb for \$695

OR . . .

You can buy two channels of really good spring reverb for \$375.

The **Sound Workshop 242** STEREO REVERBERATION SYSTEM



Judge it by its' SOUND... not by its price.





1038 NDRTHERN BLVD., ROSLYN, N.Y. 11576 (516-621-0138)

Circle 21 on Reader Service Card

qp

September 1975



TEN DAY FREE EVALUATION AND 2 YEAR GUARANTEE INSURE YOUR UNCOMPROMISED SATISFACTION

TURNTABLE PREAMPS

Preamps costing almost 3 times more will not compare with these units. RIAA/NAB equalized ± 1 db, 0.5MV sensitivity at 1KHz (or + 4dbm out, balanced outputs, -75db s/n at 10mv in, 0.05% distortion, +21dbm max. out. Internal power supply. MP-8E Mono \$86 SP-8E Stereo \$137

MIC & LINE AMPLIFIERS

Dual function and superb performance. Inputs for mic and line, ± 0.5 db response 10Hz-20KHz, 67db gain on mic channel(s) ± 26 db gain on line inputs. Balanced inputs & outputs, ± 21 dbm out max, 0.1% distortion. Internal power supply. MLA-1E Mono \$98

MLA-2E Dual Mono/Stereo \$139

AUDIO DISTRIBUTION AMPLIFIERS

From 1 in/6 out to 20 in/80 out in one small package. Whatever your distribution requirements we have an answer. All units meet or exceed the following specifications: Balanced bridging/matching inputs, balanced 600ohm outputs, ± 0.5 db response 10Hz-20KHz, ± 3 db 5Hz-40KHz, 26db gain, -21dbm out, max. capability, 0.1% or less distortion, outputs isolated by 80db, hum and noise 90db down referenced to ± 21 dbm out. Internal power supplies.

DA-6/E	Table top. 1 in/6 out.	\$131
DA-6R/E	Rack mount. 1 In/6 out.	S149
DA-6BR/E	Rack mount. 1 in/6 out. Individevel controls for each output.	
DA-6RS/E	Rack mount. 1 in/6 out stereo In/12 out mono.	or 2 \$229
DA-16BR/E	Rack mount. 1 in/8 out stereo In/16 out mono. Individual o level controls, selectable meterin headphone monitoring.	utput
DA-2080/E	Rack mount main frame with prote	ected

	power supply, metering & headphone
	monitor. Will accept up to 10 slide in
	modules. Each module has 2 inputs
	& 8 outputs. Individual output level
	controls & selectable meter switch. Up
	to 20 in/80 out.
80/E	Main Frame \$150

DA-2080/E Modules 2 in/8 out \$135 ea.

DA-208

AUDIO CONSOLES & CONTROLLERS

Our new series 35 audio controller introduces a new concept in audio mixing. Allows separation of controls from the audio functions. Controls can be placed in any convenient location in the studio, while electronics may be mounted anywhere for easy maintenance & hookup. Remote DC control for completely unaffected audio.

This versatility gives you a custom designed console at a standard production model cost.

Features include; 8 channels, mono, dual channel mono, stereo, dual channel stereo, or combinations; paralleling 2 units for quad, fail safe power supply & plug in interchangeable cards.

Performance specifications are; 0.3% or less distortion, 124dbm equivalent noise on low level channels, approximately 25w power consumption, --70db crosstalk, balanced bridging/matching inputs & response within ± 2db 20Hz-20KHz. Series 35 audio controllers start at \$1200.

AUTOMATIC TAPE CARTRIDGE AND CASSETTE LOADERS

So easy to use & accurate that our largest winder competitor has been using one of these to load their own carts.

Eliminates guesswork. Set the dials to the length desired. The exact amount of tape is fed onto the cart or cassette hub and then shuts off automatically. Also has exclusive torque control for proper tape pack on different size hubs. Winds at 30 1PS. ACL-25/E \$185

60 series). Sa	come in higher speed mode ame operation as above but w pts 14" pancakes.	
ACL-60T/E	(tone stop only)	\$266
ACL-60B/E	(Blank tape loader)	\$331
	(for both prerecorded and ank tape)	\$375

STUDIO MONITOR AMPLIFIERS

Exceptional reproduction! Internal muting, $\pm 2db$ response from 20Hz-40KHz. 25w music power, 20w RMS into 8 ohms. Hum & noise 65db below rated outputs.

Distortion le	ss than 0.25% at less than	20w out,
1% or less a	t 20w. Works into 4-160hms	. Balanced
bridging inpu	its, variable bass contour, int	ernal over-
load & short	circuit protection.	
SMA-50/E	Table top (mono)	5125
SMA-500/E	Rack mount (mono)	5142

5MA-500/E	Rack	mount (mono)	5142
SMA-1000/E	Rack	mount (stereo-40w)	\$196

REMOTE POWER CONTROLLERS (DUAL)

Safe,	transient	free	mea	ins of	controlling	110V/AC.
Turnta	ables, on	the	air	lights,	etc.	
PR-2	(toggle	\$\vi	ch	on/off)	\$39
PR-28	(mome	ntary	cor	niact a	ctuation)	\$54

Give us a call or write today for further details. You'll be money and performance ahead.

CALL COLLECT --- (916) 392-2100 WRITE --- 3516-C LaGrande Boulevard Sacramento, California 95823

RAMKO RESEARCH



• Since I wrote the last column, another BYU workshop has come and gone, and a mighty interesting experience it was. I know now what Dean Austin means when he says that every year's group of participants is different. And I guess that the difference shows up more because of the way we handle the workshop, which is to stay loose, to provide what the students need, rather than having a cutand-dried presentation that we try to pour into them come hell or high water.

Which reminds me, there is some unfinished business that J want to talk about, relative to the relationship between learning and tests, or vice versa: what they are, as opposed to what they should be. The way it is, most people spend time preparing for tests with little thought about whether in doing so they are also learning something: just so long as they get that allimportant diploma!

I've said before what I believe the relationship should be. Testing should be part of the learning process—students should concentrate on learning the subject, of which taking the test would merely be a proof of success. But beyond that generalization, which may start you thinking. I'll go no further at the moment, because when I got back from BYU I found several letters waiting for me relating to what I said in the June issue, to which some experiences at BYU also have relevance.

DISTORTION-AN ILLUSION OF LOUDNESS

First, let me say, that while I prefer other forms of music. I am not averse to rock. But I am very definitely averse to the variety in which the whole effect is lost if distortion is absent. And this is not confined to modern rock. When I was a lot younger. I used to hear whatever was in vogue at the time blaring forth from the latest type of portable radio. It might have been good music, if the kid with the radio would have turned the volume down.

The set probably was capable of 500 milliwatts, tops, half a watt. But operated at the level to which its user turned it, the distortion level was about 50 percent and it sounded loud. That was, in fact, the only reason it sounded loud, because actually it wasn't loud.

An experience on the other side of the coin is also quite typical, and there

have been (and probably still are) several examples at BYU. A high quality sound reinforcement system (but not high fidelity, about which I still need to say more) has been demonstrated to a musical group (rock or otherwise) so they can see whether they need their own reinforcement.

Often, if the equipment is good and natural sounding, which is what it should be, they get the impression that there is no reinforcement system used, so they ask, "Is it on?" Only when it is switched off, and everything goes quiet, so the building seems to "soak them up," do they realize what a good system is, or should be.

Yet the natural sound they were hearing is probably many dB louder, as measured on a sound pressure level meter, than the little portable radio (now it would be what the kids call a *transistor*) we were talking about just now.

I will not argue, as one reader suggested that distortion is often, as in the case of rock music, in the ears of the listener. I know that some people find something they don't like loud simply because they don't like it. But I was not referring to that. And perhaps I was wrong to suggest that appreciation of purity, in sound or anything else, is something that comes with maturity. Maybe there is no essential connection between the two.

The same reader wrote, "If the material presented to the listener is as the artist desired, then no undesirable distortion is present." Maybe he has a point. After all, some rock groups employ fuzz boxes to introduce deliberate distortion, as we mere mortal electronic engineers would view the matter. But if that is the effect they desire to create, it is no longer distortion, in the sense of not being what they want.

However, that does not influence what I was saying, as I see it. Because it introduces such a splutter of spurious frequencies, by which I mean frequencies not natural to the guitar strings, or whatever musical original source is used, it does sound louder than it would if all the frequencies produced were what are usually regarded as harmonious.

Musically, any kind of generator that produces a single note, such as a guitar string, produces harmonically related frequencies. I am there using *harmonically* in its mathematical sense. not its musical one, in case someone wants to argue. However, if the musician, or perhaps the group playing together, produce multiple tones, whether such a combination also sounds harmonious is a musical question based on prior conditioning.

OVERLOADING PRODUCES SPURIOUS TONES

No argument there. I agree. What once would have been considered discordant may now be considered harmony. That was not what I was talking about in June, either. When you deliberately overload an amplifier, or make a device that is intended to overload, playing two discordant notes (or more) will usually result in spurious tones not generated by the instrument you are trying to record, or reinforce.

Have you ever played on a guitar amp that had some distortion not supposed to be there and found that, when you played two notes at a close musical interval, it sounded as if something was buzzing? First maybe you suspected a flaw in the guitar itself. Having checked that out, perhaps by turning the amplifier off, and finding that then the buzzing stopped. you next suspected the loudspeaker grill cloth of buzzing.

So you looked for the buzz. maybe changed the loudspeaker, or listened on headphones. Whatever you did, if the real source was distortion in the amplifiers. you still heard it. The only way to stop it was to remove the distortion or play some other combination of notes that didn't excite that particular distortion.

Maybe, apart from the unwanted buzz, the sound is not that loud. But the presence of the buzz inevitably makes the sound seem louder because it grates on your nerves—unless you happen to be some screwed-up rock musician who thinks that buzz is music!

Now, I'm not knocking rock for its own sake when I say that. Or even off-beat ideas of what constitutes harmony.

But many is the time, particularly when I've been operating a reinforcement system for young people, that I've been told the sound isn't loud enough. Maybe. from force of good habit, I was operating just below the distortion point. So I turn it up, maybe 2 dB, so the amplifier is clipping good and well, and they say, "That's better."

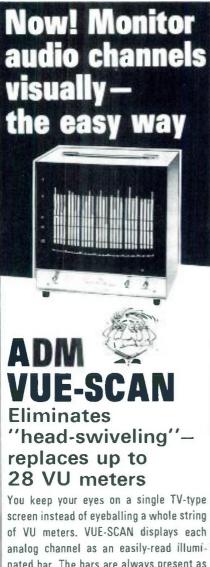
Now I happen to know that if I



Five monitors. One sound. Five JBL studio monitors. You could record with any one, play back on any other, and take your pick among the rest for mixing or mastering. The only differences are acoustic output, size and cost. No matter what size your studio is, you can cross refer-ence with any other studio using JBL's. But reading isn't knowing for sure. Come listen to one. Or two. Or five. JBL Studio Monitors from \$303 to \$1596.



James B. Lansing Sound, Inc. / Professional Division / 3249 Casitas Avenue / Los Angeles 90039.



nated bar. The bars are always present as a background reference. As the voltage level of a channel increases, the bar representing it increases in height and color intensity. Blue represents a normal operating condition—red immediately signals an overmodulated condition. VUE-SCAN can be used as a self-contained accessory with any Audio Designs or competitive console. Write us for details.



theory & practice (cont.)

had been 2 dB below clipping and I merely turned it up that 2 dB, these same people would have been unable to tell that I had turned it up at all. For anyone, a 2 dB change in loudness is barely perceptible. For someone whose measure of loudness is really, whether they know it or not. measured in percentage distortion, 2 dB change in level just isn't perceptible at all!

ENVIRONMENT AND HIGHS

Now, the other thing that raised questions was whether the difference between home high-fidelity quality, and public address wide range quality, is merely what we expect in a certain environment. It is not so much a matter of expecting. If you think about it, you will realize that our hearing faculty works more on an economic basis; it tends to reject redundant information.

In a small-room environment, the highs contribute useful information about where the sound comes from, so a high-fidelity quality loudspeaker (or two or four of them) adds to our appreciation of the original sound. makes it more natural. But in a larger auditorium, less highs are necessary for two reasons. If they were there, the wavelength and multiple reflection effects would rob them of any usefulness as information.

Secondly, the air, as a transmission medium, attenuates the higher frequency end of the spectrum, above about 1000 Hz, at 3 dB 'octave. So in a natural large auditorium—that is one without artificial reinforcement the high frequencies that you expect to hear, as my reader put it, in a smaller room, just aren't there.

This is something that should be taken into account in room equalization. If you install wide-range speakers, say in a cluster, the response will roll off due to the size of the room. even if the absorption characteristics of the walls and other room boundaries are perfect. So to equalize the room to have a level response will make it sound as if you have put in a 3 dB/octave high-frequency boost make it sound harsh.

A room should be equalized to sound natural. Keep in mind that highfidelity loudspeakers do not have the acoustic output capability that reinforcement types do. For high-fidelity use, there are two main approaches. The first is to get wide. uniform response at any cost, possibly at quite low efficiency (as in the case of the bookshelf type). The second method achieves high efficiency (which involves large size for the low-frequency end) based on the argument that a high-efficiency unit is inherently more flat in its response.

THE POWER FACTOR

Without getting into that argument. either alternative is acceptable for high-fidelity use in small rooms because the needed power is cheaper to get than the speakers it has to drive. For larger-scale distribution in auditoriums, we have different picture: now efficiency becomes important. To fill that large space with low-efficiency speakers would require kilowatts of audio power, where one tenth as much would be adequate with higherefficiency speakers, perhaps less than that.

Such large amounts of power not only cost more for the necessary amplifiers: even with solid-state amplifiers. that's an awful lot of heat to get rid of, and it has to be installed somewhere.

In your home environment, biamplifiers make sense, for various reasons that have been discussed time and again. Do they also make sense in a sound reinforcement environment, with all the extra wiring, as well as extra power requirement? Since nobody is going to hear 20 kHz, and possibly not even 10 kHz, in that environment, why provide it just to have it absorbed in air losses?

So it is not only what we expect to hear, it is also what we actually can hear, in different environments, as well as the whole economic question of providing something that is unnecessarily costly. For reinforcement use, the emphasis is on power, *acoustic* power, not just *electrical* power. For home high-fidelity use, the emphasis is on full-range quality, not power as such.

This brings up one more point about which many people seem inadequately informed. They talk about how many watts (meaning electrical or audio watts) it will take to fill a given room, without ever thinking about loudspeaker efficiency. What fills a room with sound of some specified level is not electrical, but acoustical watts.

With a speaker system, using units of 0.5 percent efficiency (not uncommon for home high fidelity use) 1000 watts of audio power will produce only 5 watts acoustic. Put the efficiency up to 10 or 15 percent, and a 50-watt amplifier will produce as much sound. That is quite a difference! Have you ever heard a loudspeaker with efficiency approaching 50 percent? Let's talk more about that another time.

 \simeq

db sound with images

• An occurrence recently at a large company in New York prompted a senior vice president to distribute an $8\frac{1}{2} \times 11$ -inch white sheet on which he had the art department print, in very large black letters, the words "Never, never assume!" When you stop a moment to think about it, very rarely does much time go by when an assumption is not made. Sometimes it works out, but the many times that it does not could lead to a lot of trouble and possibly a disaster or two. This holds true in audio-visual work, too.

Take the case of the people who went to make a presentation at a client's office. They had rehearsed diligently, gotten all the material together to take with them, including the software, such as slides, sync tape cassette, film, and boards, and then decided on what equipment to take along. They checked and found that the client had a film projector and a screen which would be available in the conference room the evening before the meeting to be set up by the presenters for rehearsal if they wished.

Late in the afternoon, they gathered with all the material, and took a sync cassette machine and slide projector because these worked well in rehearsal and there was no sense worrying about the slide projector at the client's when it was easy to take one. Fortunately, one of the men had a spare hand and off they went. It was close to their office so if anything was missing, they could race back for it.

When they got where they were going, sure enough, a film projector was available on a small rolling table, and there was also a tripod screen. No problem. Everything was as promised and all it would take was to set up.

After a couple of mishaps, the 6 ft.-wide screen was figured out and set up, the table was rolled into place, the projector was faced in the right direction, and the people got to work getting the software ready for setting up one more time.

It didn't take long before one or two things were found to be missing. First came the easel, at least one of which every conference room normally has. Well, that could be easily taken care of by asking for one from the client contact with whom the arrangements were made for the meeting. Then, where was the lectern which one of the men would need as he was working from notes during the

talk? The lectern had to have a light on it for use during the time the room was dimmed for slides. Also, they had to ask for a pointer to use during one of the chart talks. Speaking of dimming the lights, someone had to be assigned to do that before the slides began, and turn them on again during the board presentation, and then down again for film. One of the people could run the projector, since the film came during someone else's portion, but it would take a second person to work the lights, located near the door. quite a bit too far for the one working the projector.

WHERE WAS THE TAKEUP REEL?

So far so good—at least there was nothing major missing. The film was loaded on the projector—where was the take-up reel? Didn't anyone bring a take-up reel? Well, the client should have one of those, since he has a projector. Of course, if one was available, it might be locked up in a cabinet somewhere and the man with the key would not be in until tomorrow when the meeting was already in progress, or if one could be found, it probably would be too small for the amount of film.

Setting up the slide projector would be easy. Since the rolling table was too small for both projectors, and the shelf under the top was too close to the top to permit changing the drum, and the slide-sync cassette unit had to be set up, another table was necessary. The table that was found was too low, there was no a.c. extension cord to accommodate the three pieces of equipment, the 5 in. lens in the slide projector made the picture too large for the screen, and they had forgotten to take any extension cords for the remote control on the slide machine. Besides, how could one hook up the remote control at the same time the cable was hooked up from the slidesync machine?

The same person running the film would have to change plugs at the back of the slide unit. Then it was necessary to readjust the projector in case it moved during the plug change. Oh, by the way, where was the tape recorder to play the reel-to-reel tape one of the people was using during the presentation? And wouldn't it have been better if the sound for the film could come from the screen area instead of from the back of the room? And the same for the tape and cas-

Audio Measuring Instruments

Low Distortion Oscillator Series 3

A continuously variable frequency laboratory oscillator with a range of 10Hz-100kHz, having virtually zero distortion over the audio frequency band with a fast settling time.

SPECIFICATIONS:

Frequency range: 10Hz-100kHz (4 bands) Output voltage: 10 volts r.m.s. max.

- Output source resistance: 150 ohms unbalanced (plus 150/600 balanced/ floating)
- Output attenuation: 0-100dB (eight, 10dB steps plus 0-20dB variable)
- Output attenuation accuracy: 1% Sine wave distortion; Less than 0.002%
- 10Hz-10kHz (typically below noise of measuring instrument)
- Square wave rise and fall time: 40/60 n.secs.
- Monitor output meter: Scaled 0-3, 0-10 and dBV.
- Mains input: 110V/130V, 220V/240V Size: 17'' (43cm) x 7'' (18cm) high x
- 8%" (22cm) deep Price: 150 ohms unbalanced output: \$950.00.
 - 150/600 unbalanced/balanced floating output: \$1070.00.

Distortion Measuring Set Series 3

A sensitive instrument with high input impedance for the measurement of total harmonic distortion. Designed for speedy and accurate use. Capable of measuring distortion products down to 0.001%. Direct reading from calibrated meter scale.

SPECIFICATIONS:

Frequency range: 5Hz-50kHz (4 bands) Distortion range (f.s.d.): 0.01%-100% (9 ranges)

Input voltage measurement range: 50mv-60V (3 ranges)

Input resistance: 47Kohms on all ranges High pass filter: 12dB/octave below 500Hz

Power requirement: 2 x PP9, included Size: 17" (43cm) x 7" (18cm) high x 8%" (22cm) deep Price: \$770.00

AUDIONICS 10035 NE Sandy Blvd. Portland, Oregon 97220 Authorized U.S. Sales Agency for Radford Laboratory Instruments

â



sound with images (cont.)

sette recorders, right?

Not any of this is made up! In fact, much of this takes place from time to time at different locations, in various circumstances-sometimes happening even to the same people. This has all taken place but maybe not all at the same time. There are probably even more bizarre cases in your own experiences, to be sure. Many times, the people involved with the preparation of the presentation material are unaware of the complexities of the actual showing. Anyhow, they figure that there's always someone at the client's who can supply the pieces that are missing, or it's just a short walk back to the office so the forgotten items can be gotten easily, or. . . Many times everything finally works out, but how much less trouble there would be if a professional were asked for assistance, or if one of the people in the presentation just took the time to work up a list of essential items, considering everything as essential if it could be of any help at all in the presentation.

MAKE LISTS BEFOREHAND

Most important, never, never assume. For example, it's pretty easy to

DX

minates pe noise

dbx eliminates tape noise when connected to any reel-to-reel recorder. You can make original recordings with no audible hiss or background noise, which even the most expensive studio recorders cannot achieve using conventional noise reduction systems.

Circle 26 on Reader Service Card

dbx 157 is a two-channel simultaneous tape noise elimination system which permits

monitoring off tape while recording. It is fully compatible with all dbx professional studio systems and costs 8600.00 from professional audio dealers. Switchable (record or play) 150 series noise elimination systems start as low as \$187.50 per channel.

For complete product information and a list of dealers serving your area. circle reader service number or contact: dbx. Incorporated. 296 Newton Street. Waltham. Massachusetts 02154. (617) 899-8090.

make the list of accessories needed, according to the pieces of equipment. For the film projector, there must be a take-up reel sufficiently large for the film to be played: the open-reel tape recorder needs a take-up reel and an a.c. cord, if it is not permanently wired in; for the slide projector there should be an extra remote control unit, just in case, along with an a.c. cord if it is a separate item; and for all the equipment, an a.c. extension cord or two which will also permit plugging everything in conveniently by running only one wire.

For the professional, this is not enough. For the film projector, there also must be a spare lamp, a spare exciter, possibly a spare spring for rewind and take-up if the projector is one of the manual types, some tape to hold the film together in case of a break, and even such a simple thing as a three-way a.c. plug and a threeto-two adapter for the three-pin a.c. plug. For the reel-to-reel recorder there might be a splicing block and tape in case the tape broke, for the slide projector, an extra 25 ft. extension cord for the remote control device, and a spare lamp. For both projectors, he would require a zoom lens or at least a 7-in. lens for the slide unit to match the standard 2-in. in the film projector.

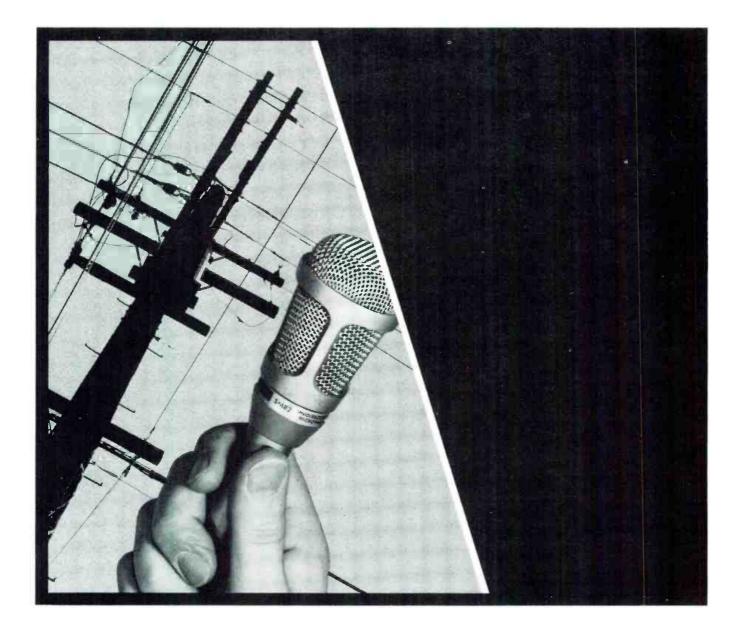
Then there's the possibility of having a separate speaker for the tape/ film sound which can be used for the projector and tape units with a switch box with proper inputs and output, or a "Y" connector with correct plugs although this method might end in a mismatch if not properly done, with a resulting loss of sound quality. For the slide projector there could also be a "Y" to permit operation of the slides by either the person presenting or by the slide sync machine.

The professional might also bring the kind of tape that will hold down cables to keep them from being kicked around and the people that step over them from falling. There are also projection tables which the a/v man might decide to take along to set up for best projection with no interference with the seating arrangement.

A good audio-visual man usually carries, or has accumulated, a bag of tricks. In addition to all the little spare parts and accessories, there could also be blank slides to put into the projector's aperture to be able to change drums without turning out the projection lamp, and various cable adapters to permit feeding sound into existing sound systems, and...

And one thing the professional a/v man carries with him at all times is the constant reminder—never. never assume.

Circle 27 on Reader Service Card



Hand-held hotline.



Finally there's a microphone that follows the news wherever it goes—to the end of a mile-long parade or to the center of a championship golf course. The new Shure SM82 super-rugged microphone—with its own built-in, linelevel amplifier—is the long-distance hotline designed for broadcasting. Used with a telephone line or voice coupler, the SM82 becomes a complete remote amplifier/microphone with line-level output and a self-contained peak limiter to prevent overloading. All in the new Shure SM82, the long hotline for the short deadline.

Shure Brothers Inc. 222 Hartrey Ave., Evanston, IL 60204 In Canada: A. C. Simmonds & Sons Limited



September 1975

qр

5

Manufacturers of high fidelity components, microphones, sound systems and related circuitry.

Circle 28 on Reader Service Card

A Select/Cancel Channel Control

Here's a versatile, simple, and inexpensive design that will accommodate any desired number of channels.

I NUMEROUS applications, you may want to select a particular input or channel while simultaneously canceling a previous selection. Additionally, you may desire to cancel all selections. (That is analogous to the push-button station selectors of car radios.)

An obvious broadcast console application is that of previewing a program source while canceling a previously cued selection. Here we present a very versatile and yet inexpensive means of providing such a control function on a single-wire common bus. The common bus feature permits the accommodation of any desired number of channels.

The inherent simplicity of the circuitry is most appealing. The circuit consists of two TTL open-collector NAND gates (e.g. 7401) and one JK Flip-Flop (e.g. 7473) plus an optional driver or buffer circuit. The circuit shown in FIGURE 1 operates as follows: If the pushbutton switch is pressed once, the output, Q, of FF2 is at logical 1. This output level can then activate, via a buffer or driver if necessary, an electronic audio switch, lamp, or relay, thus effecting the desired switching operation. At the moment when the switch is activated, the output at gate G1 is momentarily at a logical 0 level. This sets the FF2 outputs of all other channels on the bus to 0. The output of the circuit corresponding to the pressed switch will not go to 0 since inputs J, T, and K are all momentarily at logical 1. If the bushbutton switch is activated a second time, then FF2 will toggle, causing that output to clear. All other circuits on the bus remain at 0. That will cancel all channels.

M. C. Volker is president of Volker-Craig Ltd., Waterloo, Ontario.

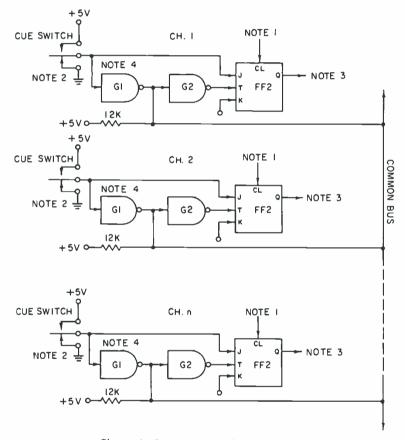


Figure 1. Channel select/control logic.

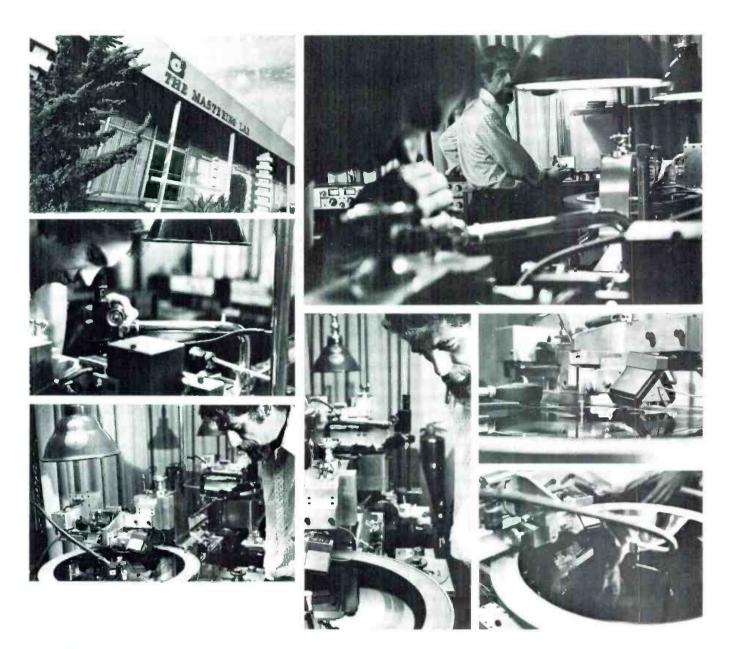
NOTES:

- 1. Optional additional clear Input.
- 2. Momentary Contact push-putton switch, normally at O.
- 3. TTL Output to electronic switch or relay driver.
- 4. NAND gates must be open-collector, e.g. 7401 type.

The availability of a clear input to FF2 permits remote canceling of that particular circuit. An application exists when the circuits are used for cueing program sources. In this case, if for example, a record or tape has been cued, then the control signal which starts the tape deck or turntable can simultaneously "kill" the cue.

This circuit has proven itself adaptable to a broad class of channel control requirements. Its simplicity and the single common bus feature render it practicable for many control situations.

ı



Top Disc Cutting Studios, like The Mastering Lab, rely on Stanton's 681-Calibration Standard in their Operations.

Not everyone who *plays* records needs the Stanton Calibration Standard cartridge, but everyone who *makes* records does!

At The Mastering Lab, one of the world's leading independent disc mastering facilities, the Stanton 681 Triple-E is the measuring standard which determines whether a "cut" survives or perishes into oblivion.

A recording lathe operator needs the most accurate playback possible, and his constant comparing of lacquer discs to their original source enables him to objectively select the most faithful cartridge. No amount of laboratory testing can reveal true musical accuracy. This accuracy is why the Stanton 681 Series is the choice of leading studios. When Mike Reese, principal disc cutter at The Mastering Lab, plays back test cuts, he is checking the calibration of the cutting channel, the cutter head, cutting stylus, and the lacquer disc. The most stringent test of all, the evaluation of direct to disc recordings, requires an absolutely reliable playback cartridge . . . the 681 Triple-E.

All Stanton Calibration Standard cartridges are guaranteed to meet specification within exacting limits. Their warranty, an individual calibration test result, comes packed with each unit. For the technological needs of the recording and broadcast industries, and for the fullest enjoyment of home entertainment, you can rely on the professional quality of Stanton products.



db September 1975

For further information write Stanton Magnetics, Inc., Terminal Drive, Plainview, N.Y. 11803

Circle 29 on Reader Service Card



the new standard in professional audio design

amplifiers



INCOMPARABLE solid-state amplifiers for continuous-duty performance, available in a broad selection to meet professional audio amplification requirements.

mixer-amplifiers



SPECTRUM SERIES 40, 80, 120 WATTS RMS

SUPERIOR mixer-amplifiers with less than 1.5% THD and program equalization provisions. Flexible, advanced design; professional in every sense of the word—for the most demanding applications.

input equipment



OPTIMUM mixing performance in maximum flexibility input equipment. Ideal for broadcast and recording use, theaters, auditoriums, and churches. Distinguished for ultra-low distortion and wide-range performance.

for equalization



PRECISE professional acoustical analysis and system correction in minimum time is now at your command with this all-new ½ Octave Active Filter Test and Correction Instrument-System.

WRITE FOR TECHNICAL BULLETINS

RAULAND-BORG CORPORATION 3535 W. Addison St., Dept. N, Chicago, III. 60618



AUDIO CONSOLE

• Console component SR101 is a

rack-mountable, eight-channel mixer/

preamplifier which can be custom-in-

stalled in a desk or console. The unit

accepts up to eight balanced low-im-

pedance microphones, each with indi-

vidual controls for volume, reverber-

ation, high- and 102-frequency equa-

Switchable inputs on two channels ac-

cept auxiliary high-level signals or high impedance microphones. The

SR101 provides simultaneous pro-

gram and monitor outputs, each con-

trolled by its own master volume con-

trol. Additional switches allow the

operator to route input signals in any

combination to either output. Dual

link jacks maintain connections to

Circle 50 on Reader Service Card

• Developed for public address and

entertainment applications, this 4 kHz

horn loudspeaker can provide wide

horizontal sound dispersion from a

single source, through the action of

its unique high but narrow mouth.

The acoustic distribution is over 150 degrees in the horizontal plane and

30 degrees in the vertical, with a

claimed maximum acoustic loss of 2

dB at 12 kHz from the response on

the axis. Made of heavy-gauge steel

treated with acoustic damping com-

pound, the horn is 61/2 in. long. It

has a heavy-duty mounting bracket.

Circle 51 on Reader Service Card

Mfr: Vitavox Ltd.

attenuation.

lization, and input

other components. Mfr: Shure Bros. Inc. Price: \$960.00

HORN LOUDSPEAKER





• Trimly fitting units of PD-11 series can be accommodated side by side on a 19-inch rack. Both the reproducer and recorder/reproducer feature direct-drive motor, air-damped solenoid, $\frac{1}{2}$ -inch thick aluminum deck, plug-in circuit cards, adjustable tape guides, micro-adjustable head assembly, and lubrication-free operation. Mfr: International Tapetronics Circle 52 on Reader Service Card

QUADRIPHONIC MIXER



Totally modular P/M-40 series quadriphonic mixer emphasizes flexibility, servicing any number of inputs, with modules expandable or removable at will. The unit contains a tri-state input l.e.d., dual power supply as a safety margin, phantom powers condenser microphones, and three submaster capability. B, C, and D mixes can be switched pre or post the A slider, are individually controlled on each channel with continuously variable settings that can be equal to, less than, or greater than the A slider. A, B, C, and D mixes can be used for combinations of functions, such as main, monitor, echo/reverb, submastering, recording, cue or quadriphonic p.a. A continuously variable pad on each channel provides attenuation for calibration to any microphone and most line inputs. The l.e.d. glows green and varies in intensity with the proper input signal, changing from green to red at clipping level. Threeway equalization gives a +20 dB of bass and treble and 3 dB or 6 dB shelving boost in the mid-frequency band.

Mfr: Malatchi Electronic Systems, Inc. Circle 53 on Reader Service Card



18

COMPACT STEREO CASSETTE MACHINE



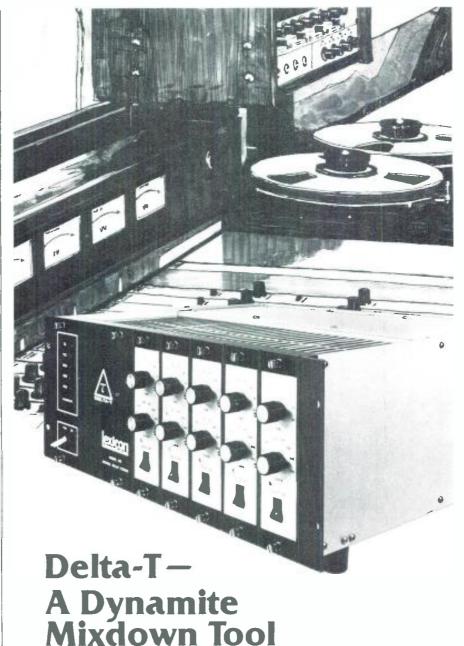
 Measuring only 7 x 2 x 7 in., CR134 stereo cassette machine can be used as a recorder with a built-in speaker or as a deck. driving two speakers directly with 1-watt output per channel. It can also be used in conjunction with an external amplifier or receiver. The head design incorporates four tracks in-line and has a photo-sensitive electronic control for the tape-drive mechanism, as well as automatic tape reversal. CR134, which operates on batteries, has a built-in condenser microphone and operates in both mono and stereo in both record and playback. The manufacturer claims wow and flutter less than 0.12 percent and frequency range of from 25 to 15,000 Hz, within 2 dB, and signal-tonoise ratio better than 56 dB. Mfr: Uher of America Price: \$378.00 Circle 54 on Reader Service Card

PORTABLE PATCHING CENTER



 Live recording, mixing, signal proccessing, and dubbing in up to four channels can all be handled by the QT-1 patching center. The unit has 72 RCAtype phono jacks on the back panel which permit the user simultaneously to connect up to four tape recorders and other accessories and leave them connected to the rear panel. Interconnection functions are performed by front panel switching or changing patch cords, making possible numerous applications-coding, mixing, intermixing, reversing and rearranging channels, limiting and equalizing, playback, copying, monitoring, etc. The device is passive, with no a.c. power or active circuitry, except at tape recorder outputs. Included are twelve patch cords. The case may be removed for rack mounting.

Mfr: Russound/FMP, Inc. Price: \$249.95. Circle 55 on Reader Service Card



That's what we provide in our new Series 102 Digital Delay Systems. We've been making high quality, reliable delay systems for five years and have learned how to do it better than anybody else.

Simply put, the Delta-T's 90 dB dynamic range and low distortion deliver a superb quality signal, leaving you free to creatively explore the powerful artistic potential of time delay. Discover for yourself, as leading studios such as Leon Russell's Shelter Studio have, how a Delta-T can thicken vocals and instruments, add slap or in-tempo percussive repeats, and provide ambience and spatial depth to the dry mono sources encountered at mixdown.

In the Delta-T 102 Series we have used our patented digital techniques to provide reliability, convenient features, and excellent performance at highly competitive prices. Let us help you define the configuration you need to get started. Call or write for more information.



Circle 36 on Reader Service Card

WANT TO BE INVITED?

If you have anything to do with discotheques—designers, sound and light installers, DJs, record promoters—call Judy Giddings at Meteor Light and Sound Company for an invitation to a showing on Sept. 12th of their inhouse discotheque/lighting theater. Call 516-364-1900



Circle 40 on Reader Service Card

products & services (cont.)

RECORDERS/REPRODUCERS



 Open reel series 1400 recorders/ reproducers are built on a die cast aluminum main frame with a d.c. servo drive system that assures, according to the manufacturer, a timing accuracy of less than three seconds deviation per 30 minutes of operation, regardless of line voltage variations. Units are available in full, half and quarter track head configurations and operate at 33/4, 71/2, and 15 in/sec. Solid state logic controls all tape transport functions. The amplifier provides balanced and unbalanced inputs and outputs. Units are equipped with dual tap transformers for 110-130 VAC and 220-260 VAC, 50 or 60 Hz. Rack mounted. Remote control stations are optional. Mfr: Telex

Price: \$1,645.00 up. Circle 56 on Reader Service Card



THE LIMITER FOR ALL NEEDS For recording, mastering, broadcast, film, sound reinforcement Average level and peak limiting functions operate simultaneously but independently Front-panel controls for all functions Variable peak attack/release time and average response action Unique open-loop control and distortion-reducing circuitry Model 201 Average and Peak Responding Limiter: \$480.



PORTABLE TWO-CHANNEL REVERB



• Based on the Torsion Transmission Line principle, model BX-10 portable two-channel reverberation unit offers independent decay-time adjustment, high and low frequency equalization and reverberation/dry signal mixing. Decay-time is adjusted silently through motional feedback, enabling adjustment during program material. Reverberation/dry signal mixing enables reverberation to be added to a dry signal without the need for a reverb-return mixing section in the mixing desk. The two channels are electronically and acoustically separate. The unit, equipped with two-point suspension, may be rack mounted or used as a portable system.

Mfr: Philips Audio Vidio Systems Corp. (AKG) Price: \$1,400.00 Circle 57 on Reader Service Card

PORTABLE EIGHT-CHANNEL CONSOLE



 Production mixing console model ESC-8, called the Eight Pack, handles eight stereo channels and claims less than 0.1 percent distortion and -123 dBm equivalent input noise as well as balanced low impedance inputs and variable input gain. Control features include separate monitor mix, pan, high and low equalization, and overload indicators on each channel. Provided are a 9 watt headphone amp, an internal reverb unit. and two peak reading level meters. Input/output features include preamp outputs (patching or post channel out, optional), reverb mix output, left and right effects input, bridging/auxiliary out, and monaural output. The unit weighs 33 pounds.

Mfr: Head Sound, Inc. Circle 58 on Reader Service Card

AUTOMATIC RELEASE CARTRIDGE MACHINES



• An automatic release deck is featured in series 3000 Spotmaster cartridge machines. Units in the series, including mono and stereo, record/ playback/delay, with all cartridge sizes and in a choice of desk or rack mounting, have gold-plated connectors, plug-in electronics, direct-drive motors, noise suppression, and airdamped solenoids. F.e.t. switching, transformer output, and remote control connectors are standard. Stereo units are equipped with head brackets having independent azimuth adjustment for tight control of stereo phrasing. Options include all tape speeds, secondary and tertiary cue tones, fast forward and various operating voltages. A tape splice/fault detector accessory is available.

Mfr: Broadcast Electronics Circle 59 on Reader Service Card

TAPE/TIME COMPUTER



• A simple and ingenious doubleslide cardboard calculator, called The Tape Measure, computes the amount of recording or playing time remaining on a given amount of recording tape. It will work for regular 5 in., 7 in. or 10½ in. open reel tapes. Mfr: Rothchild Printing Co. Inc. Price: \$1.49 Circle 60 on Reader Service Card

TAPE SPLICER



• An integrated self-sharpening shear mechanism replacing a razor characterizes this manufacturer's line of splicers. Model H-50-S, specifically for $\frac{1}{2}$ -in. tape, contains a precision dovetail aluminum groove which secures the tape during the splicing operation while Teflon-coated flat springs help keep it in position. The line also includes splicers for 1/4-in. and cassette tape.

Mfr: Nagy Research Products Price: \$24.95 Circle 61 on Reader Service Card

GAUSS REPLACEMENT HEADS



• Exact replacement heads for Gauss (Cetec) type high speed tape duplicating systems are made of hot-pressed glass-bonded ferrite construction. They are designed to replace the original metal 4-track cassette and 8-track cartridge heads in the systems, and are fully compatible with existing Gauss electronics. The manufacturer claims that the new heads will outwear the original metal heads by ten to twenty times.

Mfr: Saki Magnetics, Inc. Circle 62 on Reader Service Card

Introducing the \$299 Stereo Synthesizer.

We cut the fat from the package.

We redesigned the circuit with state of the art components.

We built in larger quantity.

The result: we were able to cut 60% off the old price without sacrificing performance.

Now the smaller FM stereo station can afford to process old records, announce mikes, and mono spots, while eliminating phase cancellation problems in mono reception. And any recording studio can save tracks by recording strings, drums, or horns in mono and spreading them in the mix.

The synthesizer will stereoize a mono reverb device (like our own model 106CX). And it's ideal for reissuing old mono masters in synthesized stereo.

The output is fully stereo/mono compatible -just add the outputs to obtain the original mono input. Setup is a snap, and the unit can be adjusted in seconds for different kinds of program material.

At \$299, the new Orban/Parasound model 245E is an essential addition to your collection of audio processing gear. An explanation of its operating principles and an extensive collection of applications are detailed in our brochure, available on request.

Also inquire about our Spring Reverb, De-esser and Parametric Equalizer, all with outstanding price performance characteristics.

orban/para*s*ound

680 Beach Street San Francisco, Calif. 94109 (415) 776-2808

Now every professional who needs stereo synthesis can afford the best.

Being Practical About Feedback, part 1

This is the first of a three-part discussion of feedback, its characteristics, and how to work with it.

ROM TIME TO TIME. we've covered pieces about feedback in my column, Theory and Practice. Also, a few months ago, I mentioned the more thorough treatment that I once covered, first in a little book on just that specific subject, and later in a textbook that is now out of print. Several readers wanted to know where they could get this information, and it seems to be unavailable in print. So your editor invited me to write this series, which will also update the information to apply to the new devices that have since come into use.

In this first part, we will discuss what feedback can do, within the somewhat limiting assumption that we are talking about something that can be either positive or negative—no in-betweens! In practice, that is never true because phase shift always gets into the picture somewhere. So that what starts out being negative feedback can turn itself around and become positive at some frequency or other. That is where trouble starts. We will avoid confusion by deferring that kind of discussion to the second installment.

EFFECT ON GAIN

Whatever else feedback may do, it will always be related to one primary effect: on gain. So we start with that. First, assume we have an amplifier (FIGURE 1) that has a gain we designate A. That will have the dimension of either a voltage gain or a current gain, or else a transconductance or transresistance. So if you want to give it a dB figure, it will be 20 $\log_{10} A$. Thus, if output voltage is 100 times input voltage, we would say that A = 100, or the dB gain is 40.

Now, for the moment we will assume that feedback is also designated by voltage ratio and is given by the symbol β . To work with voltage ratios in each case requires the configuration shown in FIGURE 1. Now let's put in some figures to make it easier to follow because letters are not all that easy.

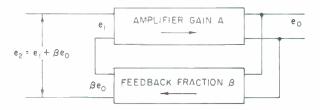


Figure 1. The classic feedback block schematic: voltage in, voltage out.

We assume, let's say that internal input voltage, e_1 , is 10 millivolts. Because the gain, A, is 100, the output voltage, e_0 , is 100 × 10 mV, or 1 volt. Now, let's assume that β , the feedback fraction, is 0.09, which is about 1/11. Then the fed-back voltage is β times 1 volt, or 90 mV. Because it is negative feedback, the external input, e_1 , must be the sum of these 10 + 90 = 100 mV.

So with feedback, an external input of 100 mV will produce an output voltage of 1 volt. The gain has been reduced from 100 to 10, or from 40 dB to 20 dB. As a formula, we write that:

$$A_{\rm F} = \frac{A}{1 + A\beta} \tag{1}$$

where A_F is the gain with feedback.

IHE NOCAL PROJECTOR A reasonable alternative to the "Black Box Blues"

After extensive research into the soundman's "response vs. portability" dilemma we've arrived at what we feel is the ideal solution...The Peavey Vocal Projector. The Vocal Projector system is a compact, highly portable system that maintains the wide frequency response of the bulky theater type "Black Box" enclosures.

These new enclosures are bi-amp compatible and employ matched low, mid, and high frequency components in a three-way system that is crossed over by a precision inductor/capacitor type crossover network.

Lows are handled by a super heavy-duty 15" driver employing a 3½ pound magnet coupled to a 2" aluminum voice coil assembly. Low frequency is enhanced even more by the Vocal Projector's ported, horn loaded enclosure.

The mid range of these systems is provided by a high quality folded horn. This horn is coupled to a motor/driver assembly with a 1½" voice coil wound on an extremely heat resistant phenolic coil form.

The extreme high frequencies are reproduced by the popular Piezo super tweeters, and add response to 20 KHz.

High quality, selected components in a well designed system make the Peavey Vocal Projector unmatched in value...a truly outstanding alternative to the "Black Box"



*Two 15" drivers, two mid horns, two Piezo tweeters in a 4 ohm enclosure. \$425. list price.

*One 15" driver, one mid horn, two Piezo tweeters in a 8 ohm enclosure. \$275. list price.

N

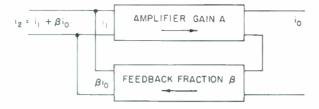


Figure 2. An alternative feedback block schematic: current in, current out.

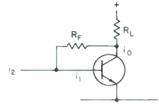


Figure 3. A typical transistor stage, that can be analyzed in terms of Figure 2.

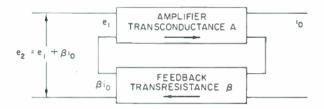


Figure 4. A third alternative feedback schematic: voltage in, current out.

That is the most important formula in all of feedback theory.

Before going further, we need to define terms. β is called the feedback fraction. The product $A\beta$ is called the loop gain. In the example we used, the loop gain was 9. It could be given in dB as 20 log₁₀ 9, which is about 19 dB. Finally, the important factor, $1 + A\beta$ is called the feedback factor. If it is expressed in dB, it will be called the dB feedback.

In this case, 9 + 1 = 10, and 20 $\log_{10} 10 = 20$ dB. If you substitute, 100 divided by 10 is 10, giving A_F as also 20 dB. Working in dB, you subtract the dB feedback from the dB gain without feedback to get the dB gain with feedback. But it is best to work with the formula directly, using the numbers and fractions rather than their dB equivalents because that 1 + has to get in there, which working in dB makes a little awkward.

TRANSISTORIZED AMPLIFIERS

Now, in FIGURE 1 we were working with voltage gain and a voltage fraction fed back. That is not the only form that feedback can take, although the formula is the same

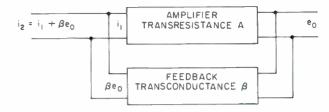


Figure 5. The fourth possibility: current in, voltage out.

whichever form we use. In the old tube amplifiers, input was usually in the form of a voltage. But with transistors, except the f.e.t. variety, input is essentially a current, although in a composite transistorized amplifier we may be back to voltage again.

Essentially, a transistor is a current amplifier; it gives an amplified current output for a smaller current input. The symbol β is also used for the current gain of a transistor. But as we are using that for feedback fraction in this series, we will avoid talking about the same symbol for transistor current gain, or else we will spell it out, and call it beta when we mean current gain.

If we redraw the block schematic of FIGURE 1 for current gain instead of voltage gain, it will look like FIGURE 2. Applied to a practical transistor stage, it might look like FIGURE 3. Here R_L is the collector load resistor, and R_F is the feedback resistor. The current through R_F will be $R_L/(R_L + R_F)$ of the transistor output current. That is the value β will have in formula (1).

Now, let us suppose we have a transistor with a working beta of 300, that R_L is 10 K $_{\Omega}$, while R_F is 90 K $_{\Omega}$. β calculates to 1/10, or 0.1. A is 300, so A β is 30, and the feedback factor 1 + A β is 31. Gain with feedback is 300/31, which is about 9.7, slightly less than 10.

To pursue that example a little further (although we will deal with this more fully later), a transistor with a nominal beta of 300 might be bracketed to have a range of betas from 150 to 450. Without feedback, that means current can vary over a 3:1 range, which is nearly 10 dB. But now let us look at what feedback can do to it, in theory.

If beta is 150, $A\beta$ is 15, and the feedback factor 1 + $A\beta$ is 16. Gain with feedback is 150/16, which is 9.375. And if beta is 450, $A\beta$ is 45, and the feedback factor is 1 + $A\beta$ = 46. Gain with feedback is now 450/46, which is about 9.78. So the gain without feedback, which changed by almost 10 dB, is reduced to a change of less than 0.4 dB. We will have more to say about this later.

We have shown two possible configurations for feedback. There are two more. The two we used were for straight voltage gain, or straight current gain. Now suppose the gain is given as a transconductance: current output for voltage input. Then feedback must be in the form of a transresistance: voltage fed back for current output. The configuration is shown at FIGURE 4.

The other possibility is one we can use in many transistor circuits: voltage output for current input, which is a transresistance. So feedback must be given as a transconductance: current fed back for voltage output. And the configuration is shown at FIGURE 5. The complementary quantities for transconductance and transresistance are micromhos and megohms. A transconductance of 2,000 micromhos means that an input of 1 volt produces an output of 2 milliamps. A transresistance of 0.5 megohm means that an input of 2 microamps produces an output of 1 volt.

For the moment, we will leave using specific examples of each kind until we've taken a look at the things we can expect feedback to do for us.

EFFECT ON DISTORTION

Effect on distortion is what feedback was first known for. So let us figure out what we can expect it to do. Suppose that without feedback the amplifier, for an input of e_i , produces an output that is $e_0 = Ae_i + A\delta e_i$ the first term being amplified fundamental, and the second term being a percent, δ of distortion, of designated harmonic content. Thus, if the distortion was 5 percent, δ would be 0.05.

If we proceed to apply the fraction β to this, we will have a distortion component in the fed-back signal that does not offset the input signal as the fundamental does. That distortion would be amplified again, and we would never solve the problem. We must assume that the external input, e_i , is what has no distortion, so that the input now consists of $\hat{e}_i = e_i - \theta e_i$, where θ is the component of distortion fed back, reversed in phase, because it has no offset in the input, e_i .

So the output will now be:

$$\mathbf{e}_{0} = \mathbf{A}(1+\delta)\mathbf{e}_{i} = \mathbf{A}(1+\delta)(1-\theta)\mathbf{e}_{i} \qquad (2)$$

And the fed-back voltage will be:

$$A\beta(1 + \delta)(1 - \theta)e_{i}$$

The input voltage, \mathbf{e}_i is the internal input voltage, $\mathbf{\hat{e}}_i = \mathbf{e}_i (1 - \theta)$ plus the fed-back voltage, or:

$$\mathbf{e}_i = [1 - \theta + A\beta(1 + \delta)(1 - \theta)]\mathbf{e}_i \qquad (3)$$

Multiplying this out, and keeping the fundamental and distortion terms separate, we get:

$$\mathbf{e}_i = [(1 + A\beta)\mathbf{e}_i + (A\beta\delta - [1 + A\beta]\theta - A\beta\delta\theta]\mathbf{e}_i$$
(3a)

The first term is the fundamental input, required in the absence of distortion. The second term contains all the distortion products, none of which is present in the external input, so that term must be equated to zero. Equating the first order terms to zero is easy, but the second order term, $A\beta\delta\theta$, represents a distortion of distortion component.

If the distortion is second harmonic, second of second is fourth, and if it is 5 percent without feedback and, say 0.5 percent with, 5 percent of 0.5 percent is 0.025 percent. We could go around again, and introduce a term for that, but it never ends. Similarly, if the distortion is 3rd harmonic, 3rd of 3rd is 9th. And if more than one order of distortion is present, you will have combined products, such as 6th, and so forth, but all second order and much smaller than the first order—in theory, at any rate. More of that later.

So we equate the first order term to zero:

$$A\beta\delta = (1 + A\beta)\theta \qquad (4)$$

From which

$$\theta = \frac{A\beta}{1 + A\beta} \delta \tag{4a}$$

	200 Park Avenue South New York, NY 10003 (Entrance on 17th Street)
Quantum Audio Inc. Professional Sound Equipment	
Tour—Experienced Staff	Telephone 212, 260 2300
Authorized dealers for: Altec Professional Products AKG Atlas BGW Beyer Cerwin Vega Cetec Community Light & Sound Crown Emilar Eventide Gauss Heil JBL Professional Series Kenwood	Manhattan Mavis Moms 2005 AD Otari Phillips Quantum Speaker Enclosures Revox RSE Sennheiser Sescom Shure SR Sony Switchcraft Tapco Terry Hanley Audio Systems
Expert consultation with demons	stration equipment on display

25

Substituting this into (2) and simplifying:

$$e_0 = Ae_i + \frac{A\delta}{1 + A\beta}e_i - (2nd \text{ order term})$$
 (2a)

Note that the distortion term δ has been divided by 1 + A β . That is the important result.

GAIN STABILIZATION

You could arrive at these conclusions by application of logic in different ways. For example, distortion occurs because of change in gain over different parts of a signal waveform. So the reduction in distortion could be regarded as being due to gain stabilization over the various parts of the waveform. From this we would expect both to be reduced by the same factor, $1 + A\beta$.

But here is another way to go. Gain is usually quoted in dB units, which is logarithmic. Using the symbol Δ for incremental change in gain, without feedback, this would be $\Delta \cdot \log A$. With feedback, it will become $\Delta \cdot \log A_{\rm F}$. So reduction in gain variation, due to feedback, will be:

$$\frac{\Delta \cdot \log A_{\rm F}}{\Delta \cdot \log A} = \frac{\Delta \cdot \log A_{\rm F}}{\Delta A_{\rm F}} \cdot \frac{\Delta A_{\rm F}}{\Delta A} \cdot \frac{\Delta A}{\Delta \cdot \log A} \quad (5)$$

Now substituting equation (1) for Ar, and treating A as the variable, as implied by equation (5), this reduces to:

$$\frac{\Delta \cdot \log A_{\rm F}}{\Delta \cdot \log A} = \frac{1}{1 + A\beta}$$
(5a)

Thus, adding the feedback reduces the change in gain, from whatever cause, by the feedback factor, $1 + A\beta$. This is something about which we will have much to say in later parts.

EFFECT ON IMPEDANCES

Feedback also changes input and output impedances of active devices or circuits. Look at FIGURE 1 or FIGURE 4 -that is, when you look at input impedance, it is the input configuration that is important. The current input is determined by the amplifier's internal input resistance or impedance. Suppose this is 1000 o and the input voltage is 10 mV.

Without feedback, the input current would be $10\mu A$. But now put in 20 dB of feedback, requiring a $1 + A\beta$ of 10, and only 1 mV of the 10 mV total input will be across the 1000 Ω . The other 9 mV will be supplied by the feedback. It does not matter what the feedback network's impedance is, from the viewpoint of change to input impedance, because the input current is determined by that 1000 Ω . And feedback has changed it so that, from the viewpoint of the 1000 Ω , the current has only 1 mV driving it, instead of 10 mV.

Thus, the effective impedance with feedback is 10 mV taking 1 μ A. This is 10,000 Ω , meaning that feedback has increased input impedance by the feedback factor, $1 + A\beta$.

That is for voltage feedback, or series injection of the feedback. Input impedance is multiplied by the feedback factor. What about shunt injection, which is more often used with current type amplifiers, such as ordinary transistors? That is represented in FIGURE 2 or FIGURE 5. Now the input voltage does not change, but the current does.

If the input voltage is 10 mV, and the input impedance without feedback is 1000 g, without feedback the input current will be 10 µA. But now, if 20 dB of feedback is applied, the A β will be absorbing an additional 90 μ A, to bring the total input current up to 100 μ A. So with shunt injection, or current feedback, the input admittance or conductance is multiplied by the feedback factor. If,

Sensual 14

Whether on record or in live performance, today's most commercially successful music is more visceral, immediate, and sensual than ever before. This impact has been achieved through advances in the musician's art, and through a quantum jump in the control available in

audio processing. The Orban/Parasound Parametric Equalizer, Model 621, has received outstanding acceptance since its introduction because it combines economy (\$340/ channel) with extraordinary control. Each of its four noninteracting bands permits continuous, stepless adjustment of bandwidth, equalization, and center frequency. Each band can be tuned over a 20:1 frequency range with no change in curve shape (unlike some competitors), and peak gain remains constant as the bandwidth is varied. The unique "constant-Q" equalization characteristic is more musical than the usual reciprocal curves, and lets the equalizer create infinite-depth dips to remove hum, whistles and ring modes — making it ideal for cinema and sound reinforcement as well as recording studio and

broadcast applications. Other outstandingly useful features include a front-panel gain control and a peak-stretching overload lamp which indicates clipping anywhere in the equalizer circuitry.

While our spec sheet (available from the address below) gives the details in cold black-and-white, it cannot describe the sensual interaction between man and machine which occurs when the frustrating limitations of conventional equalizers are finally overcome, and the user is given the power to create sound that feels really right. Our ability to deliver this power at an affordable price is the true reason for the O/P Parametric's success. But don't take our word for it - discover the Sensual Equalizer for yourself, soon.

For further information, contact

afa/ound

680 Beach St. San Francisco Ca. 94109 Or contact your local an/Parasound distributor like many of us, you find it easier to think in resistances or impedances, then this means that the input resistance or impedance is divided by the feedback factor for this form of input combination.

A similar set of deductions can be made about output impedance. One way to think about that is to consider what happens, in theory at any rate, when you change from the working load to either open or short circuit, one of which will effectively remove the feedback.

For voltage feedback, better called voltage-derived feedback, if the output is short-circuited, there is no voltage to feed back. Suppose the output with the normally connected load is 1 volt, with feedback, and that 20 dB feedback is used, meaning $1 + A\beta$ is 10, again. That means that removal of feedback would put the output voltage up to 10 volts.

So whatever current is flowing, with the normal load connected and feedback operative, will go up to a current limited only by the amplifier's' internal resistance, from 10 times the voltage when the output is short-circuited. That means the effective internal resistance, with feedback, is divided by the feedback factor, 10 in this example.

Similarly, if the feedback is current derived, open-circuiting the output will remove the feedback. So the output current will rise to 10 times, absorbed only by the internal conductance or admittance of the output stage. That means that feedback divides the internal conductance or admittance by the feedback factor, or multiplies the internal resistance or impedance.

NEGATIVE OR POSITIVE FEEDBACK

So far, we have, without saying so, been talking about negative feedback, so-called, because it offsets the original input, reducing gain. In applying feedback, we must ensure that it is negative, because if we get the phase wrong, we could be in trouble. To apply the formula we developed for negative feedback to positive feedback, we just put a minus sign in front of β . Thus, if β is a numeric fraction, such as to produce positive feedback, we rewrite equation (1) as:

$$A_{\rm F} = \frac{A}{1 - A\beta} \tag{1a}$$

Now, you will see that, if the loop gain product, $A\beta$, becomes 1, 1 - 1 = 0, and the reciprocal of zero is infinity, which means the amplifier achieves infinite gain, becoming an oscillator. For positive feedback, you can use loop gains of only less than 1, unless you want an oscillator.

If the loop gain is a loss of 6 dB, $A\beta$ is 0.5, then $1 - A\beta$ is also 0.5, and gain increases 6 dB as a result of feedback. If loop gain is a loss of 2 dB, which means $A\beta$ is approximately 0.8. then $1 - A\beta$ is 0.2, and gain is multiplied by 5, or an increase of 14 dB, as a result of feedback. Here it gets very critical.

WHAT ABOUT PHASE?

As stated at the outset, so far we have assumed that feedback is always simply negative or, just at the end, positive. But all practical circuits contain something beside resistances and conductances, such as reactances. And at some frequency or other, every reactance introduces phase shifts. Then the feedback signal is neither in phase nor out of phase with the original input signal, exactly. It must be added vectorially.

Whether we like it or not, this is a fact of life. We may design a circuit to have negative feedback all right, but somewhere, at some frequency, that negative feedback leaves off being simple negative, so that the factor $1 + A\beta$ ceases to be a simple addition.

STUDER a standard of excellence.

Complete information on STUDER Professional Audio Equipment and the name of the sales/service representative in your area is available from, Willi STUDER America, Inc., 3916 Broadway, Buffalo, New York 14227. Phone 716-681-5450. In Canada, Willi Studer Canada, Ltd. Phone 416-423-2831.

A Simple and Superior Microphone Preamplifier

This effective design uses an integrated circuit with few external passive components for far better performance.

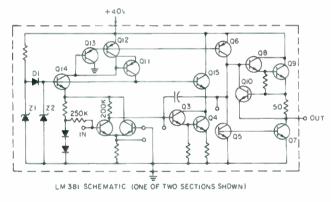


Figure 1. LM381 schematic (one of two sections shown)

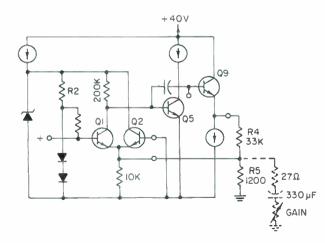


Figure 2. A.C. equivalent circuit.

ERE'S A MICROPHONE PREAMPLIFIER design that takes advantage of the newest technology to produce a simple, reliable, and effective way to improve existing audio equipment. And, it produces a new audio system with a performance that is superior to designs employing conventional transistors, tubes, hybrid components, and/or general-purpose op amps.

The preamplifier uses a National LM381AN integrated circuit and a minimal number of external passive components. FIGURE 1 shows the schematic diagram of the LM381. The operation can be described as follows:¹

A low-level input is coupled to the base of Q1 via a 0.1 microfarad external capacitor. Q2 is turned off to achieve the noise advantage of a single high-gain input stage rather than a differential input pair. Q3 and Q4 operate as a Darlington emitter-follower driving a common-emitter amplifier Q5 with current-source load Q6 and output current sink Q7. The voltage amplification of the first stage is 160 and that of the second stage is 2,000. Thus the open loop gain of the entire amplifier approaches 320,000. Capacitor C1 compensates the amplifier to unity gain at 15 MHz, with provision for additional compensation to limit high-frequency noise in audio applications. The output stage is a Darlington emitter-follower, Q8 and Q9, with transistor Q7 acting as a current sink. Transistor Q10 limits the short-circuit output current to 12 milliamperes.

Zener diode Z2, driven from a constant-current source Q11, provides a biasing reference. Q11 and Q12 provide the high current-source impedance necessary to maintain a power supply rejection ratio of 120 dB. (The LM381 has an internal voltage regulator which prevents interaction between several units operating from the same supply.)

AMPLIFIER A.C. CIRCUIT

FIGURE 2 shows the equivalent a.c. circuit of the amplifier. The output quiescent point is established by negative feedback through an external voltage divider, R4/R5. The a.c. and d.c. gain of the amplifier is equal to R4/R5. However, by shunting R5 with a capacitor and resistor in series, the a.c. gain of the amplifier will be equal to the ratio

28

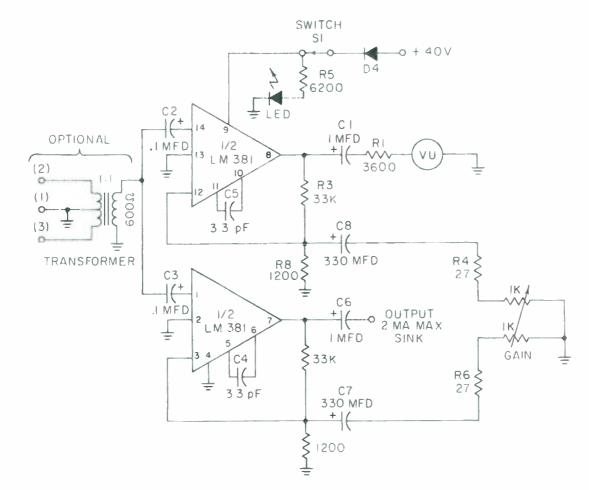


Figure 3. Microphone preamp schematic.

of R4 to the shunting resistance. The a.c. gain will approach the open loop gain of the amplifier (i.e., 320,000) as the shunting resistance approaches zero. However, the low-frequency 3 dB corner (cutoff frequency) is equal to:

open loop gain

2π imes shunting resistance imes shunting capacitance

Therefore, low-frequency gain is limited by the size of the shunting capacitance; for a gain of 1.000 and a low-frequency 3 dB point of approximately 20 Hz, a 330microfarad capacitor and a 27-ohm resistor are used. A 1,000-ohm audio potentiometer is placed in series with the shunting resistor and capacitor to enable the gain to be varied continuously from a maximum of 1,000 to a minimum of 33. (If the shunting network were removed, the minimum gain would be about 27.5, as established by the ratio of R4/R5.)

Since the internal noise generated by this amplifier in the single-ended mode of operation is extremely low, special considerations must be made to maintain ultra-low noise in the operating circuit.² FIGURE 3 shows the schematic diagram of the complete microphone preamplifier.

INPUT TRANSFORMER

In order to cancel out common-mode noise in long microphone lines, an input transformer is used with a grounded center-tap primary. Input and output impedance of the transformer is 600 ohms (nominal). The low secondary impedance provides an exceptionally smooth frequency response and a minimum of transformer-induced

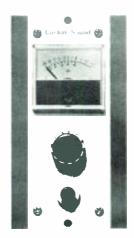


Figure 4. The finished package.

distortion. Since the secondary-coil resistance of the transformer is about 80 ohms, very low noise-voltage is generated in it. The total equivalent input noise of the preamplifier circuit is -130 dBV. An improvement of 20 dBV over this value could be achieved by an ideal input transformer with a gain of 10.

However, all input transformers of this type introduce a certain amount of signal distortion, and the low noise of the LM381AN allows us to avoid using them. Additionally, we have the advantage that even the loudest rock band would have difficulty overloading the input of the preamp circuit, since the input from the microphone may approach 300 millivolts without producing input distortion or output distortion with the gain set at its minimum

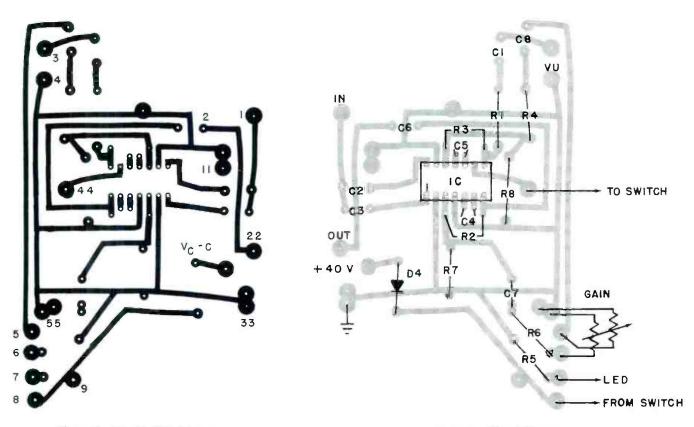


Figure 5. Printed wiring mask.

Figure 6. Circuit layout.

value. The input transformer may be left out of the circuit for unbalanced microphone lines.

FIGURE 4 shows the finished product packaged in suitable form for use in a recording studio. A Vector EFP module was used with a three-pin XLR-type connector on the rear for input and a phone jack for output. A dixson front-panel vu meter fits nicely, along with a gain control and an illuminated on/off switch. Since the vu meter is driven by a separate amplifier with 60 dB of isolation, no distortion is introduced by its use. The gain of both amplifiers is adjusted simultaneously by the use of a dual audio taper potentiometer. (The LM381 is a dual device.)

To avoid the amplification of high-frequency noise, a 3.3 picofarad capacitor is used as additional compensation. Input and output signals are coupled via small tantalum capacitors, and provision is made for the use of an l.e.d. in the on/off switch.

The LM381AN is sensitive to soldering heat, so it should be mounted in a 14-pin socket. A protection diode is provided to prevent a blown i.e. if the 40V power supply is accidentally hooked up with its polarity reversed. All components are mounted on a 3 x $4\frac{1}{2}$ -in. glass-epoxy circuit board which fits right into the enclosure. FIGURE 5 is a printed wiring board mask which may be used to make up your own boards, and FIGURE 6 is a component layout guide, both intended for use with radial-leaded tantalum capacitors.

ITS PERFORMANCE

In field tests by rock musicians, this preamplifier has performed admirably. It is very easy to build in any form since the components and layout are not critical. In use, its operation is almost foolproof, since the integrated circuit is short-circuit proof and will shut down if overheated. If built with military-grade components and packaged as shown, or in a well-designed console or rack, it should operate indefinitely and provide the best that modern technology can provide for a modern studio.

One word of caution: This circuit is primarily a voltage amplifier that provides up to 24 vu of output voltage, but very little current. The maximum output current you can expect the amplifier to deliver is only 2 mA. If you feed a high-impedance load, such as an amplifier, equalizer, console, etc., you won't have any problem. But if you feed a load of 6000 ohms or less, when the input signal increases to a point where the load would tend to draw more than 2 mA, the output voltage of the amplifier will not increase even if the input signal increases. In this case, harmonic distortion would result. Thus, when driving a 600-ohm line, for instance, the output voltage of the preamplifier would be limited. Since this preamplifier design is intended to be plugged into a high-impedance, high-level input, the problem would not normally occur.

If you intend to use several preamplifiers as a mixer, however, make sure that the input resistance of your summing network is at least 6,500 ohms for each preamplifier. Another solution would be using a voltage follower, or current-gain stage, after the preamplifier, such as the LH0002. The use of this voltage-follower will permit the present design to be expanded for use in driving 600-ohm lines or low-impedance devices. However, the circuit as described is normally adequate to improve the noise performance of any audio control console. It is a state-of-theart mic preamp in its simplest form, suitable for use in film, video, and in any application requiring such a device.

REFERENCES

- 1. "Linear Applications," National Semiconductor Corp.
- Mintz, R. S., "Noise Considerations in Audio Amplifiers," db, December, 1974.

30

Calibrate Microphones By Reciprocity

In place of a secondary microphone, rapid alternation of loudspeaker transducers calibrates sensitivity quickly and accurately.

N THE CALIBRATION of microphones, sound-level meters, "noise dosimeters," magnetic tape recorders, etc., one generally uses a secondary standard microphone. However, the reciprocity technique of microphone calibration has found great favor among acousticians because of its simplicity and the fact that it constitutes an absolute measurement system.

The preferred type of microphone for noise and reverberation time measurements is a non-directional transducer. The problem which sometimes arises with microphones for noise level evaluation is one of calibration. In the absence of a secondary standard which may be used for comparison, the unit must be sent to one of the few laboratories in this country which is able to calibrate it. (The United States Bureau of Standards is one such laboratory.)

A number of electroacoustic equipment manufacturers have marketed so-called "Microphone Reciprocity Calibrators." The fallacy in using these is that they are generally designed for the calibration of the transducers sold by the same manufacturer and do not lend themselves readily to the evaluation of other sound-pickup units. But there is a simplified method of checking the sensitivity and frequency response of microphones by the reciprocity method.

ALTERNATION IN A FREE FIELD

FIGURE 1 shows a schematic diagram of the free-field measurement system. S_1 and S_2 are two small identical loudspeakers located in quiet and dead surroundings. The backs of the units are enclosed to prevent radiations from the rear (as by diffraction) which would interfere undesirably with the test.

A double-pole, double-throw switch allows the rapid alternation of the transducers as receiver and transmitter, as is the case when the loudspeakers are single-coil, singlecone direct radiators.

Another double-pole, double-throw switch, ganged with

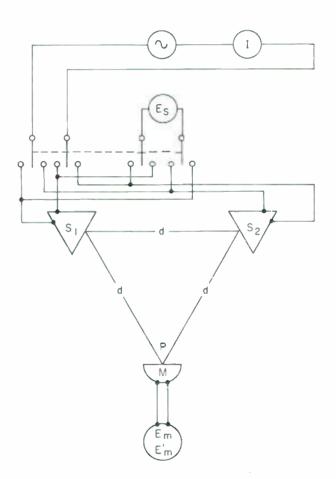


Figure 1. Schematic diagram of the free-field measurement system.

the first, allows the assessment of the open-circuit voltage of one of the two transducers acting as a microphone while the other operates as the sound-emitter.

An ammeter in line with the oscillator establishes the current at all times.

When both the current (1) and the open-circuit voltage (E_s) remain the same in the alternation of the transducer functions, it may be said that a free-field exists in which the reflections from the room boundaries are negli-

Michael Rettinger, a valuable contributor for many years to practical acoustical knowledge, is a consultant, working from Encino, California.



gible, and that S_1 and S_2 are alike in performance.

Note the slanting angles between the longitudinal axes of the three transducers, which are located at the apexes of an equilateral triangle.

THE MICROPHONE TO BE CALIBRATED

The microphone to be calibrated is assumed to be a pressure-actuated unit. For a velocity of pressure-gradient transducer, unless all units are several feet apart. corrections have to be applied for the fact that near a point source, the sound pressure and particle velocity of a sound wave are not in phase.

When S_1 and S_2 are switched in their functions as loudspeaker and microphone, the open-circuit voltage of the test microphone normally remains the same, so that $E_m = E'_m = E$.

The sensitivity of microphone (M), in volts per bar (dyne per square centimeter), is given by

$$\begin{split} K &= 0.013 \quad \sqrt{\frac{E_{m}E'_{m}d}{IfE_{s}}} \\ &= 0.013E \ \sqrt{\frac{d}{IfE_{s}}} \qquad \qquad \text{if } E_{m} = E'_{m} = E \end{split}$$

- where $E_m =$ open-circuit voltage of M when S_1 operates as loudspeaker
 - $E'_{m} =$ open-circuit voltage of M when S_2 operates as loudspeaker
 - d = distance between transducers, cm
 - f = frequency
 - $I = current for S_1 and S_2 (assumed equal), amperes$
 - $E_s =$ open-circuit voltage of S_1 and S_2 operating as microphones (assumed equal)

SOUND PRESSURE

To establish the sound-pressure at (P), we may write

$$P = \frac{E}{K} = \frac{E}{0.013E} \sqrt{\frac{IfE_s}{d}}$$

$$\frac{P}{0.0002} = \frac{1}{0.0002 \times 0.013} \sqrt{\frac{IfE_s}{d}}$$

$$SPL = 20 \log \frac{P}{0.0002} = 20 \log 77 \sqrt{\frac{IfE_s}{d}} + 74$$

$$= 10 \log \frac{IfE_s}{d} + 111.7$$

$$= 10 \log IE_s + 132 \text{ if } f = 1000 \text{ Hz and}$$

$$d = 9.25 \text{ cm}$$

As an example, when I = 0.01 amperes and E = 0.001volts, and a sound-level meter is put at P, it should read SPL = $10 \log 0.01 \times 0.001 + 132$

= 82 dB

The frequency response of many loudspeakers is beset by sharp peaks and dips; frequencies may not become accurately identified by reading the frequency dial of most common audio oscillators. Incorrect identification of frequency results in incorrect assessment of microphone sensitivity. Moreover, standing waves between the various transducers may result from their housing, cones, and diaphragms. Therefore, it is recommended, in this stepby-step procedures of calibration, to employ narrow frequency bands of pink noise for the signal rather than single frequencies for the achievement of valid and accurate measurement results.

BIBLIOGRAPHY

Olsen, H. F. Acoustical Engineering, p. 427. D. Van Nostrand Co., Inc.

d b classified

Closing date is the fifteenth of the second month preceding the date of issue. Send copy to: Classified Ad Dept. db THE SOUND ENGINEERING MAGAZINE 1120 Old County Road, Plainview, New York 11803

Rates are 50¢ a word for commercial advertisements. Employment offered or wanted ads are accepted at 25¢ per word. Frequency discounts: 3 times, 10%; 6 times, 20%; 12 times, 33%.

FOR SALE

DYMA builds roll-around consoles for any reel-to-reel tape recorder. Dyma Engineering, Route 1, Box 51, Taos, New Mexico 87571.

SPLICE TAPE FASTER, BETTER, BY SHEARING. Experts recommend Nagy splicers. Quality long-lasting instrument. Reasonably priced. Details, NRPD, Box 289, McLean, Va. 22101.

CUSTOM CROSSOVER NETWORKS to your specifications; a few or production quantities. Power capacities to thousands of watts; inductors and capacitors available separately; specify your needs for rapid quotation. Also, PIEZO ELEC-TRIC TWEETERS—send for data sheet and price schedules. TSR ENGINEERING, 5146 W. Imperial, Los Angeles, Ca. 90045. (213) 776-6057.

AMPEX, SCULLY, TASCAM, all major professional audio lines. Top dollar trade-ins. 15 minutes George Washington Bridge. PROFESSIONAL AUDIO VIDEO CORPORATION, 342 Main St., Paterson, N.J. 07505. (201) 523-3333.

TASCAM REVERBS---\$500; Tascam mixing consoles---\$2.350; Tascam ½-inch recorders; \$1.990; Tascam 8-track recorders---\$3.490. All shipped prepaid/ insured, including free alignment/equalization/bias/calibration. Music & Sound, Ltd., 11¹2 Old York Rd., Willow Grove, Pa. 19090. (215) 659-9251.

→ Note Special Prices ~

AKG, ADVICE, Audio Research, Beyer, Crown, dbx, Infinity, Integrity, Koss, Maxell, Nakamichi, Phase Linear, PML, Revox, SAE, Scotch, Sennheiser, Service, Shure, Sony, Stanton, Stax, TDK, Tascam, Technics, Thorens, etc. Hi-Fi Haven, Inc. 28 Easton Ave., New Brunswick, N.J. 08901. (201) 249-5130.

MIDSUMMER SOUND HAS RADFORD. Pre-amplifier and integrated amp.

NEW YORK'S LEADING PRO AUDIO/VIDEO DISTRIBUTOR for audio, video, broadcast, public address, and hi-fi systems; representing over 200 audio/video manufacturers, featuring such names as Ampex, Scully, Tascam, Sony, J. B. Lansing, Neumann, Altec, McIntosh, AKG, Dynair, T.V. Microtime, UREI, 3M, and other major brands; the largest "in stock" inventory of equipment, accessories, and parts; competitive discount prices; factory authorized sales, service, parts, systems design, installation. Write for free catalog! Martin Audio/Video Corporation, 320 W. 46th St., New York, N.Y. 10036. (212) 541-5900.

\$2 MILLION USED RECORDING EQUIP-MENT. Send \$1.00 for list, refundable, to The Equipment Locator, P.O. Box 99569, San Francisco, Ca. 94109. 94109.

PROFESSIONAL CASSETTE DUPLICA-TION, any length, stereo or mono.. any master acceptable., Dolby. Cantor Productions, 26 W. Nottingham Rd., Dayton, Ohio 45405. (513) 6571.

MIDSUMMER SOUND COMPANY. Specialists in Revox and Beyer.

MULTI-TRACK

- * SERIES "B" MIXING CONSOLE
- * VARI-BAND 5 SECTION PARAMETERIC EQUALIZER
- * DUAL EQUALIZED REVERB
- * LONG & SHORT THROW SLIDE FADERS
- * HIGH BALLISTIC VU METER

P.O. BOX 3187 HOLLYWOOD, CA 90028 (213) 467-7890 ACTIVE BUTTERWORTH FILTERS for electronic crossovers: 4 po'es, highpass or lowpass. 24 dB per octave rolloff, 1½ x 2 inches, \$14.95; 8 pcles, lowpass, 48 dB per octave rolloff, 2 x 3 inches, \$19.95. Specify cutoff frequency 500 Hz to 20,000 Hz. FREE catalog. (601) 323-5869. MFJ Enterprises, Box 494, Mississippi State, Ms. 39762.

MCI . . . DOLBY. Two great names! Two great products! For authorized factory representation in the progressive Midwest. contact: Jerry Milam, Milam Audio Co., 700 W. Main St., So. Pekin, III. 61564. (309) 348-3112, 9605.

EAR PROTECTION DEVICES. Bilsom Swedish Ear Down is an easy to use, safe, hygienic. disposable synthetic fiber now being used by many major rock bands and sound engineers. Save your valuable hearing. Send \$9.95 for 20 pocket boxes, plus a free guide on hearing conservation. Listen Up, Inc., 685 S. Pearl St., Denver, Colorado 80209.

AUDIO BEEPER, new low cost model for all audio-visual systems. Call or write C-TRONICS, P.O. Box 84, East Brunswick, N.J. 08816, (201) 254-9487.

ONE WAY NOISE REDUCTION for cutting rooms/tape copies; retains highs. rids hiss/surface noise & clicks/pops by a full 10-14 dB and costs \$170 up per channel! Music & Sound, Ltd., 11¹/z Old York Rd., Willow Grove, Pa. 19090. (215) 659-9251.

CAMBRIDGE, B & W, and AUDIONICS speakers at MIDSUMMER.

THE LIBRARY . . . Sound effects recorded in STEREO using Dolby throughout. Over 350 effects on ten discs, \$100.00. Write, The Library, P.O. Box 18145, Denver, Colorado 80218.

MCI . . . Now the best selling multitrack recorder!

MCI . . . only from Audiotechniques, Inc. in the great Northeast!

Tape recorders from one track to 24-track.

Recording consoles up to 40 input

MCI sales-service factory trained technicians Studio design and construction service

AUDIOTECHNIQUES, INC. 142 Hamilton Avenue Stamford, Conn. 06902 (203) 359-2312

33



SYNTHESIZERS—find out about all systems from us. Tell us your needs; we'll fill them. And for EMS systems, we're it! Write or phone: EMSA, 460 West St., Amherst, Mass. 01002. (413) 256-8591.

4-TRACK STUDIO FOR SALE. Ampex, Tascam, dbx, UREI, Pultec, Electrodyne, Gately. Phase Linear, Advent, etc.; mint; studio is in a home; very little use. Write Box 144, Roslyn, N.Y. 11576.

MICROPHONES — Advice, sales and rentals—MIDSUMMER SOUND.

AVAILABLE SERVICES. Milam Audio Co. specializes in every phase of professional studio wiring, from complete systems to individual pre-wired parts and components. Available from stock: patch bays, custom mic panels, multi-paired cabling and harnesses, etc. Milam Audio Co., 700 W. Main St., So. Pekin, III. 61564. (309) 348-3112, 9605.

DYNACO RACK MOUNTS for all Dynaco preamps, tuners, integrated amps. \$24.95 postpaid in U.S., \$22.50 in lots of three. Audio by Zimet, 1038 Northern Bivd., Roslyn, N.Y. 11576. (516) 621-0138.

AMPEX CAPSTAN MOTORS. Factory replacement for model 300 or 3200 duplicators, 1800/3600 rpm, \$140.00 xchange. Tellet Communications, 8831 Sunset Bivd. W., Hollywood, Ca. 90069. (213) 652-8100.

FOR SALE: EIGHT FOOT TEAKWOOD HARPSICHORD; double keyboard; five pedals; excellent condition. Must be seen. \$2.300.00. ODO Sound Studios, Inc., 254 W. 54th St., New York, N.Y. 10019. (212) 757-3180.

MCI TAPE RECORDER FOR SALE. 4track machine with 2-track heads. Machine is eight months old and has both remote control and built-in auto locater. Asking \$48.000; was \$65,000. For more information. please call (212) 226-4278.



B.B.C. REFERENCE MONITORS PRE-EQUALIZED J.B.L./Altec monitors; Dahlquist phased arrays; I.M.F. transmission lines; Infinity electrostatics: Crown/McIntosh 16 Ω/bridged bi-amps; Scully/Revox A-700 recorders: Micmix/Parasound/Multi-Track reverbs; Eventide flangers/omnipressors; Lexicon digital delays; dbx/ Burwen N.R. companders; Little Dipper hum/buzz notch filters; Cooper Time Cube echo send; moving coil Denon/Ortofon; B&O/Rabco straight line arms/cartridges; Studer/AKG/ Sennheiser condensers; Beyer ribbons, U.R.E.I, comp/limiters/crossovers; Gately pro-kits; Q.R.K. turntables; White equalizers; 1000s more. Music & Sound, Ltd., 111/2 Old York Rd., Willow Grove, Pa. 19090. (215) 659-9251. → All Shipped Prepaid + Insured ←

AMPEX FR-1100 1/2 -inch tape transport

with 3- and 4-track heads. \$950.00. Will swap. James Einolf, 12149 N. Piney Lake Rd., Parker, Colo. 80134. (303) 841-2105.

LOWEST PRICES IN THE COUNTRY on microphones, power amps, speakers, musical instruments, P.A. equipment, including Peavey, Heil, J.B.L.; huge stock, top brands only. Gracin's 110, 606 Rte. 110, Huntington, N.Y. 11746. (516) 549-5155.

BODE FREQUENCY SHIFTERS since 1963 . . . Announcing the new model 735 with exclusive (patented) features and amazing new sounds. Continuous single-sweep detuning +5 kHz to -5 kHz (and the inverse), including smaller ranges. In-step detuning with voltagecontrolled keyboard; voltage controlled frequency shift modulation; unlimited spiraling echo effects; plus all known frequency shifter features. Special price: \$900.00 f.o.b. N. Tonawanda, For details and information on other models, write to: Bode Sound Co., Harald Bode, 1344 Abington Pl., N. Tonawanda, N.Y. 14120. (716) 692-1670.

SOUND SYSTEMS — Planning, equipment, installation by **MIDSUMMER SOUND**.

SCULLY. ELECTRO-VOICE, NEUMANN. Shure, Spectrasonics, Quad Eight, Masterroom, ARP, Pultec, Mictrotrak, Russco, dbx. Interface, EMT, and others. The Audio Marketplace, Div. United Audio Recording, 5310 Jackwood, San Antonio, Texes 78238. (512) 684-4000.



AUDIO PRO CLEANER

The sensational new cleaner that eats up studio grime, makes your equipment look like new. cleans heads, tape guides, capstans . . . and everything else in the studio. Send \$11.95 for one gallon plus 8 oz. sprayer post paid to:

> AUDIOTECHNIQUES, INC. 142 Hamilton Ave. Stamford, Conn. 06902 (203) 359-2312

LOOKING FOR HIGH QUALITY "PRE-VIOUSLY OWNED" (and new) INDUS-TRIAL & PROFESSIONAL AUDIO EQUIP-MENT? Ohwerke is involved in the consultation, marketing, brokerage and referral of industrial and professional audio components. All merchandise warranted for ONE YEAR. Call or write Ohrwerke--40 High Street, Guilford, Conn. 06437 (203) 453-5285.

MODERN RECORDING TECHNIQUES by Robert E. Runstein. The only book covering all aspects of multi-track pop music recording from microphones through disc cutting. For engineers, producers, and musicians, \$9.95 prepaid. The Great Northern Recording Studio, Ltd. Box 206, Maynard, Mass. 01754.

MIDSUMMER SOUND CO. (212) 966-4150, 147 Mercer St., New York, N.Y. 10012.

TUNED ROCK P.A.s. Customized high intensity touring/permanent installation sound systems, including narrow band (5Hz!) feedback suppression, detailed regenerative response Acousta-Voicing/environmental equalization (\pm 1 dB at your ears), room design / measurement / treatment ≤ 15% articulation loss of consonants; 1000s of customized professional products, including fiberglass horns, consoles, comp/rms/peak limiters, 18 dB continuously variable electronic crossovers, digital/acoustic delays, omnipressors, flangers, reverb, echo, doubling/tripling effects, P.A. noise reduction; piezo transducers; frequency shifters from J.B.L./Altec pro, Tascam, U.R.E.I., Eventide, Gately, Studer, Beyer, Crown, Community, Mom's Audio, McIntosh, Bozak, Allen & Heath, Gauss, Cetec, Electrodyne, Multi-Track, Parasound, White, etc. All shipped prepaid/insured. Music & Sound, Ltd., 111/2 Old York Rd., Willow Grove, Pa. 19090. (215) 659-9251.

Inventors/Engineers

MONITOR EQUALIZERS for your Altecs & J.B.L.s are a steal at \$75/channel

FREE ROOM EQUALIZATION with purchase of 1/3 octave filters. This is not a misprint. Music & Sound, Ltd., 111/2 Old York Rd., Willow Grove, Po. 19090. (215) 659-9251.

→ S.M.E. Domping Mods-\$30.00 ←

MEASURE REVERB TIME IN REAL TIME —instantly! New, easy-to-use RT-60 delivers precise, instant real time digital readout. Eliminates chart recorder analysis. Only \$395.00. Write: Communicotions Co., Inc., 3490 Noell St., Son Diego, Co. 92110.

NEW & USED 1/2-in. recording tape on metal reels. New Ampex 11/2 mil 2500' \$7.75 each; 1 mil, \$8.25 each; used Scotch, 206, \$7.00 each; new and reconditioned metal reels, (all sizes). Soundd Investment, P.O.B. 88338, Atlonto, Go. 30338.

WANT TO GO BI-AMP?

DeCoursey Electronic Dividing Networks are available from \$89.10 (monaural blamp) to \$205 (stereo tri-amp). Price includes plug-in Butterworth filters; 6, 12, or 18 dB per octave at any desired cutoff frequency. Write for brochure. DeCoursey Engineering Loborotory, 11828 Jefferson Blvd., Culver City, Ca. 90230.

WHATEVER YOUR EQUIPMENT NEEDS —new or used—check us first. We specialize in broadcast equipment. Send \$1.00 for our complete listings. Broodcast Equipment & Supply Co., Box 3141, Bristol, Tenn. 37620.

DYMA builds custom studio consoles, desks, enctosures, studio furniture. Dyma Engineering, Route 1, Box 51, Taos, New Mexico 87571.

STUDIO SOUND—Europe's leading professional magazine. Back issues available from June, '73. \$1 each, postpaid. **3P Recording, P.O. 99569, San Fran**cisco, Ca. 94109.

AMPEX 300, 352, 400, 450 USERS—for greater S/N ratio, replace first playback stage 12SJ7 with our plug-in transistor preamp. For specifications write VIF Internotional, Box 1555, Mountoin View, Co., 94042. (408) 739-9740.

NEW MODELS: Ampex AG440C 2-track, servo capstan motor; Scully 280B; used AG440Bs; used Scully 280, 8-track, new heads, in console, excellent condition; Sennheiser mics. Immediate delivery from stock. Malaco Recording, Jackson, Miss. (601) 982-4522.

ORTOFON

DYNAMIC MOTIONAL FEEDBACK mono disc cutting systems. Complete with drive, feedback, and feedback-playback monitor amplifiers and cutterhead. All systems guaranteed. Spare cutterheads available for exchange/repair. Albert B. Grundy, 64 University Ploce, New York, N.Y. 10003. (212) 929-8364.

CASSETTE LABELS, blank and custom printed. Free samples. Torzoc, 638 Muskogee Ave., Norfolk, Vo. 23509.

RENT AN 8-TRACK SCULLY by the day/ week. Contact F. Rubin, (212) 631-5919.

WANTED

WANTED: Top \$ paid for U-47s (tube type), M-49s, KM-56s, U-67s, and Tetefunken 251s. Sigmo Audio Systems, P.O. Box 2469, Hollywood, Co 90028. (213) 462-8515.

WANTED: USED 12+ in, 8 out recording console in good condition. P.O. 2468, Phoenix, Az 85003.

WANTED: Educational public broadcast station needs surplus studio equipment, particularly magnetic tape and mixing equipment. Contact Boulder Community Broodcast Association, 885 Aropohoe, Boulder, Colorado 80302.

EMPLOYMENT

FILM SOUND MIXER/ENGINEER Complete Boston commercial film sound studio seeks experienced career person to take over while staff mixer goes on holiday. Operations include recording, transfer, screening, and 4-track 16/35 mixing with rollback. Please submit resume to: The Cody Co., 24 Done St., Somerville, Mo 02143.

ENGINEER, to design audio equipment. Ace Audio Co., 25 Aberdeen Dr., Huntington, N.Y. 11743. (516) 549-1233.

PROFESSIONAL RECORDING PERSON-NEL SPECIALISTS. A service for employers and job seekers. Call today! Smith's Personnel Service, 1457 Broodway, N.Y.C. 10036. Alayne Spertell. (212) WI 7-3806. USED GEAR SALE: Ampex MM-1100 16track, \$14,000.00; Ampex MM-1000 8track, \$7,500.00; Gately 16 x 8 console, \$9,500.00; Scully 284 8-track, \$7,500.00; Ampex/MCI 8-track, \$6,000.00; EMT 140S echo chamber, \$4,000.00; AKG BX-20G echo chamber, \$2,600.00; Ampex/ MCI 2-track, \$2,000.00; Tascam 8 x 4 console (demo), \$2,000.00; Tascam 8 x 4 console (demo), \$2,000.00; Tascam S70 4-track in cabinet, \$2,750.00; new Scully 280Bs; Klipsch monitors, Heresy & Cornwall in stock. Ask for Dave, Tom, or Emil. Studio Supply Co., P.O. Box 280 Noshville, Tenn. 37202. (615) 327-3075.

MOVING?

Keep **db** coming without interruption!

Send in your new address promptly.

Enclose your db mailing label, too.

Write to:

Eloise Beach, *Circ. Mgr.* db Magazine 1120 Old Country Rd. Plainview, N.Y. 11803



WORAM AUDIO ASSOCIATES

Consultants in Studio Systems Engineering, Design and Installation

---offering---

A COMPLETE CONSULATION SERVICE FOR STUDIO PLANNING AND CONSTRUCTION

FREE-LANCE RECORDING SERVICE IN THE NEW YORK AREA

212 673-9110 64 University Place New York, N.Y. 10003

dlb people/places/happenings





DEOVLET

THOMPSON



HANSEN

• Uher of America, Inc. of Inglewood. California. will distribute the complete Lenco line of turntables in the U.S.A. The exclusive rights to the line were established in an agreement between Uher Werke Munchen and Lenco of Switzerland. Also news from Uher is the appointment of Larry Deovlet as national sales manager.

• Direction and supervision of operations at Technical Services of the **RCA Service Company** has been undertaken by Martin H. Rubin, recently appointed director of field operations. He will be headquartered at Cherry Hill, N.J. Mr. Rubin, who has been with RCA since 1948, was formerly manager of the Long Island Consumer Services.

• Coil Specialists, manufacturing coil and inductor custom items and offering custom sub-assembly work, has opened a new facility in Valentine. Nebraska. Jerry Riewe is the founder of the new firm.

• Two new appointments have been announced by the Cannon Electric Division of International Telephone and Telegraph Corporation of Santa Ana. California. Robert J. Trivison has been named vice president-director of operations. Mr. Trivison was formerly with Trivex, Inc. L. Wayne Oliver has been promoted to director of business development replacing Mr. Trivison.

• Benjamin B, Bauer, vice president and general manager of the CBS Technology Center, Stamford, Connecticut, was awarded a "Debby" by the Society of Audio Consultants for development of the SQTM quadriphonic system. The award, co-sponsored by High Fidelity Trade News, was given in recognition of outstanding service to the audio industry. • Robert B. Thompson has been appointed regional sales manager of Columbia Electronic Cables, a division of Avnet, Inc. of New Bedford. Mass. Mr. Thompson has been with Columbia for five years. Prior to that, he was with the Grinnell Corporation.

• Telex Communications, Inc. of Minneapolis has announced the appointment of Norman H. Hansen to the position of product manager in the professional audio products group. Mr. Hansen will be involved in expanding amateur radio and citizens' band product sales as well as with the sale of professional tape equipment and headsets. Before joining Telex, Mr. Hansen was affiliated with Webster Electric Communications.

• Professional acoustical consulting services associated with architectural acoustics, open office landcape design, electronic sound masking systems design, and tuning is offered by newly formed Acoustical Design, Inc. Their office is at 24 Pine St., Morristown, N.J. Principals in the new firm are James E. Sulewsky, Donald R. Cummingham, and William E. Shinnick.

• Dr. Peter C. Goldmark, president of Goldmark Communications Corp. has received the 1975 Trustee Award from the National Academy of Television Arts and Sciences. The award was presented on the basis of Dr. Goldmark's technological contributions in communications. Dr. Goldmark developed the long-playing phonograph. the first practical color television broadcasting system. and electronic video recording.

• Auditronics Systems Division, of Memphis. Tennessee has announced the appointment of **Bill Brock** as regional sales manager, based in Nashville. Mr. Brock's career includes sales and instructional work at Scully and at the Ampex Corporation.

• William Krucks has been elected by the Rauland-Borg Corporation as Chairman of the board and chief executive officer. Mr. Krucks is a specialist in the field of corporate finance. taxation, and management. He has been associated with the transportation industry, most recently with the Chicago and Northwestern Transportation Company as corporate treasurer.

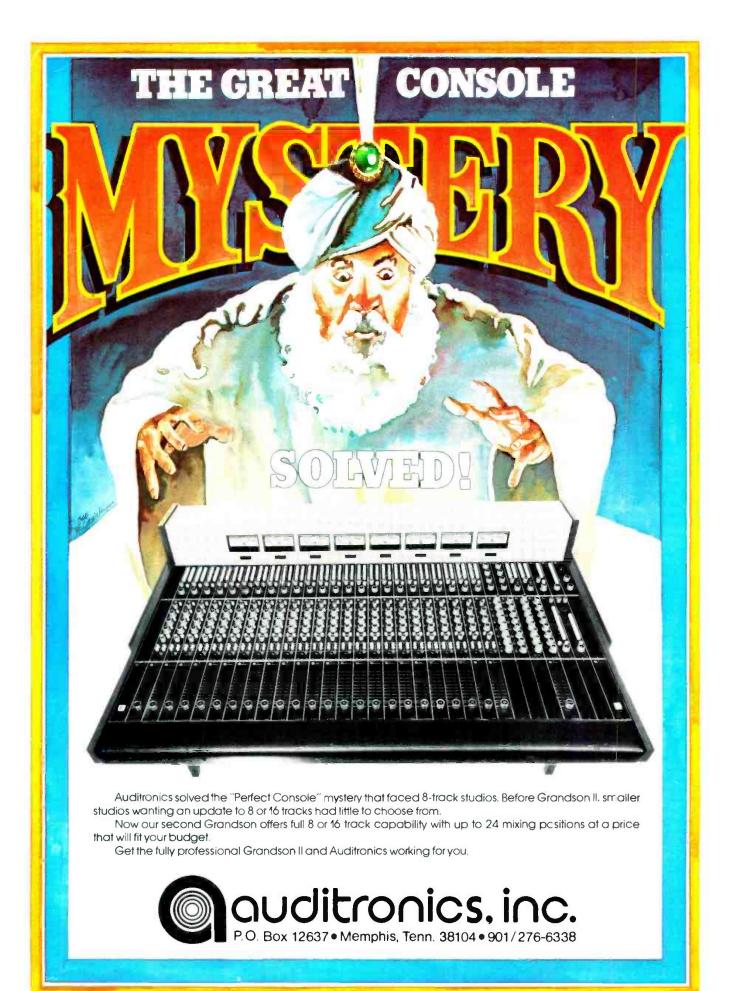
• Four additions have been included in the staff of Sunwest Recording Studios, Inc. of Hollywood. California. Bill Lazarus has been appointed vice president and director of engineering, Cheri Wagner as traffic manager and assistant to the vice president. Phil Seretti and Tonn Harvey will serve as maintenance engineers.

• A milestone 50th anniversary was celebrated by **Shure Brothers, Inc.** of Evanston. Ill. The firm was started as the Shure Radio Company on April 25, 1925 by **S. N. Shure**, who still heads the company. Beginning as a distributor of radio parts. Shure now manufactures microphones, high fidelity phono cartridges, and sound reinforcement components. The products are sold in over 100 countries.

• Sony Corporation of America has established a broadcast service unit within its v.t.r. division The new department will cover marketing and engineering, as well as service and parts. Operation of the department will be limited to television and radio stations as well as production facilities servicing the broadcast and recording industries. David K. MacDonald will head the new department.

• A new Los Angeles agency, the **R. A. Neilson Co.**, will offer marketing services in research. advertising, sales promotion, industrial design and technical writing. Also, a personnel search and placement service will specialize in personnel for the audio/ visual field. The agency is headed by **Ron Neilson**, formerly marketing manager at **Quad/Eight Electronics**. The address of the new agency is 3378 Oak Glen Drive, Los Angeles.





Circle 19 on Reader Service Card

The audience can't see you, but they hear you.

They depend on you to get that music to them. Undistorted. Balanced. Correct.

That's an art.

Presenting the Yamaha PM-1000 16 x 4 mixing console.

A technological wonder.

Chock full of professional controls and features: The exclusive 4 x 4 matr x with level controls gives greater control of the sound than by driving speaker amps directly from the bus output.

It's standard equipment.

Like transformer isolated inputs and outputs. Dual echo send busses. And an input level attenuator that

takes +4 dB line level to -60 dB mike level in 11 steps. Plus five-frequency equalization.

The PM-1000 keeps you flexible so that an unexpected glitch won't hang you up.

It helps you maintain because you know your mixer won't let you down in the middle of somewhere.

We make the PM-1000 for the studio, for the road, and for the stage. We make it right.

Because we think that some of that applause belongs to you.



THINK OF IT AS YOUR MUSICAL INSTRUMENT.