



TYPICAL audio system is not suitable for disco applications. It lacks the mixing, monitoring, and microphone preamps normally found in such facilities. The Disco Mixer presented here is a special-purpose audio preamplifier/ mixer with a number of attractive and unusual features. It has two independent phono preamplifiers employing the new IEC equalization characteristic, two IC-buffered auxiliary inputs, a low-noise microphone preamplifier, switching and mixing capability for multiple inputs, and the traditional preamp's bass, treble, balance, and volume controls. The Disco Mixer also contains a monitor circuit that allows the user to cue records or listen to one program source while another is driving the system's power amplifier.

You can build either a preamp/mixer or a preamplifier only. A kit for the preamp/mixer is \$110, while one for the preamp alone is just \$70.

The Disco Mixer is designed around six integrated circuits. Thanks to the advent of specialized IC's, signal processing functions previously performed by dozens of discrete components can now be accomplished by single chips. In many cases, both stereo channels can be handled by one IC.

High-level signals in this project are processed by members of a new family of high-performance op amps that are fabricated by "BIFET" technology. This is a process which allows diffusion of

## Build a Disco Preamp/Mixer

### **BY JOHN ROBERTS**

# Provides multi-source inputs and mixing/fading for your home disco.

both junction-field-effect and bipolarjunction transistors on the same chip. These op amps exhibit the excellent input characteristics of JFET's and the highly desirable output characteristics of BJT's—literally, the best of both worlds. BIFET op amps have higher slew rates and cause less TH and IM distortion than common bipolar IC's (see Fig. 1).

About the Circuit. A block diagram of the complete mixer/preamplifier is shown in Fig. 2. Functionally, the Disco Mixer can be considered to be made up of three types of circuits: input conditioning, high-level processing, and output conditioning. Let's examine each.

Input conditioning for the line-level auxiliary inputs, which would typically be driven by a tuner or tape deck, consists of simple unity-gain inverting buffers. One of the four buffers included in the preamp is shown in Fig. 3. An RC network at the buffer input acts as a highpass filter to prevent the passage of a dc level or infrasonic ac signals. One section of a quad BIFET operational amplifier, IC3, presents an inverted version of the input signal to the switching matrix composed of S2 and S4. This buffer displays a relatively high impedance (about 50,000 ohms) to the program source and a very low output impedance to the switching matrix. This avoids loading down the signal source and prevents interaction in the mixing process.







The signal conditioning stage for the microphone input must be able to amplify signals generated by the microphone by 60 dB (1000 V/V) or more. An NE5534N integrated circuit was selected for this task (Fig. 4) because it has a high open-loop voltage gain and a very low input noise voltage. The open-loop gain-the gain of the device in the absence of feedback-must sufficiently exceed the closed-loop gain selected by the proper choice of feedback components if the amplifier is to exhibit gain accuracy and low distortion. (An amplifier's closed-loop distortion is its open-loop distortion divided by the ratio of openloop to closed-loop gain.) The microphone preamp's input noise voltage is of critical importance because this noise signal will be amplified along with the millivolt-level microphone output signal.

With an open-loop gain of 6000 V/V at 10,000 Hz and less than one microvolt of input noise, the NE5534 (*IC7*) meets the foregoing requirements easily. The noninverting microphone preamp fea-



tures a pc-mounted trimmer potentiometer (*R43*) which allows the user to adjust the stage's gain to suit the sensitivity of a particular microphone. As is the case with the auxiliary input buffers, signals are capacitively coupled to the op amp. The microphone preamplifier has a high input impedance that will not load down the microphone, and presents a low output impedance to the mixing stage.

Similar gain and noise requirements must be satisfied by the phono preamplifiers. In addition, these stages must contain an equalization network which properly compensates for the preemphasis introduced in the recording process to increase dynamic range and suit the constant-velocity characteristic of the playback transducer (phono cartridge). At present, there is some debate as to the ideal de-emphasis (playback) curve. The existing RIAA does not clearly specify exactly how a phono preamplifier's gain should roll off at low frequencies. This has resulted in many phono preamps with very high gain at infrasonic frequencies, a situation which can cause serious problems when warped records are played or acoustic feedback combines with turntable rumble. The phono cartridge, preamplifier, power amplifier and speakers try to reproduce the warp or rumble as if it were a valid audio signal. However, many speakers are not designed and lack the ability to generate such strong, very low frequency output and can be damaged while attempting to do so.

Therefore, a feedback network comprising *R3* through *R7* and *C7*, *C9* and *C10* has been incorporated into the phono preamplifiers (Fig. 5) so that these stages exhibit a frequency response which agrees with the International Electrochemical Commission's proposed amendment to the RIAA characteristic. The deviation from the RIAA curve is slight and only at the lowest audible frequencies, and the improvement in infrasonic signal attenuation considerably outweighs the almost imperceptible (-3 dB at 20 Hz) low-frequency rolloff.

The LM387AN dual low-noise preamplifier has been chosen for *IC1* and *IC2* because of its excellent (high) open-loop gain and (low) noise characteristics. That there are two independent, identical amplifiers in one 8-pin DIP helps simplify the pc layout. The phono cartridge drives the noninverting input and is loaded by *R2*, a 47,000-ohm resistor and *C6*, a small disc ceramic, glass, polystyrene or silver mica capacitor. The user should consult the manufacturer for the recommended capacitance for his particular cartridge. In most cases, the value will lie between 10 and 300 pF.

Now that all input signals have been amplified to reasonable levels and impedances normalized, mixing can be performed in a straightforward manner. Two four-station interlocked switch arrays (S2 and S4) assign any one of four input signals to two mixer potentiometers, R46 and R47, as shown in Fig. 6. The wipers of these potentiometers are connected to one quarter of IC6, a quad BIFET operational amplifier, which is used as a unity gain inverting summer. This allows the user to mix any two of the four inputs or pan back and forth between the two as a disc jockey at a discotheque would. A third potentiometer. R51, mixes in a portion of the microphone preamp output so that the disc jockey can voice over his mix.

The inverting summer drives both the output conditioning circuitry of the preamplifier and a special monitor circuit which has been designed into the mixer section. This circuit allows the user to listen to other program sources without affecting the main preamplifier output, an especially convenient feature when



Fig. 2. Block diagram of combined preamplifier and mixer showing inputs, outputs, and controls.

**POPULAR ELECTRONICS** 



### PARTS LIST

- R1\*, R13, R40-10 ohms
- R2\*\*-47,000 ohms
- R3\*\*-43,000 ohms
- R4\*\*-360 ohms
- R5\*\*-33,000 ohms
- R6\*\*-390,000 ohms
- R8\*\*, R9\*\*, R10\*\*, R11\*\*, R12\*\*, R15, R17\*, R20\*, R27\*, R28\*, R29\*, R30\*,
- R31\*, R32\*, R33\*, R34\*, R35\*, R36\*, R37\*, R38, R41\*, R42-100,000 ohms
- R16\*, R26\*-100 ohms
- R19\*-24,000 ohms
- R21\*, R22\*, R23\*-5600 ohms
- R14, R24\*, R25\*-1800 ohms
- R43-10,000-ohm linear-taper printed circuit trimmer potentiometer
- R44-100,000-ohm linear-taper potentiometer R45 through R48-50,000-ohm dual audio-
- taper potentiometer R49, R50-50,000-ohm dual linear-taper po-
- tentiometer
- R51-50,000-ohm audio-taper potentiometer
- S1, S3-Dpdt pushbutton switch
- S2, S4, S5—four-station dpdt interlocked pushbutton swich
- T1-24-volt, 100-mA center-tapped transformer (Signal Transformer Co. #241-4-20)
- Misc.—Printed circuit boards and standoffs, IC sockets or Molex Soldercons (if desired), shielded cable, LED holder, hookup wire,

cueing up special record cuts or verifying the desired operation of a program source before routing it to the output.

A four-station interlocked switch array, S5, selects the monitoring mode. The user can monitor the A mixer input only, the B input only, the main preamplifier output (highly desirable if he is in a booth away or acoustically isolated from the sound system), or monitor in the "Auto-Cue" mode. Auto-Cue monitoring means that the user is listening to the exact opposite of his mix settings. For example, if MIX A potentiometer R46 is at its maximum setting and MIX B potentiometer R47 is at its minimum setting, he will hear the MIX B signal through the monitor if S5 is in the Auto-Cue position. Then, if the MIX A potentiometer is rotated fully counterclockwise and MIX B fully clockwise, MIX A will be heard. This mode is very useful in two-turntable systems because it allows the deejay to always listen to the one being cued.

The level of the monitor signal is controlled by MONITOR LEVEL potentiometer *R48.* A dual 2-watt audio amplifier, *IC5*, amplifies the monitor signal and delivers it to *J4*, a stereo phone jack. Either a pair of headphones or small monitor speakers can be plugged into *J4*. The audio *IC*, an LM377N, has internal current limiting and thermal protection so that, if overloaded, it will shut itself off until it

suitable enclosure, machine hardware, line cord and strain relief or grommet, solder, etc.

- Note-The following are available from Phoenix Systems, 375 Springhill Road, Monroe, CT 06468, (203) 261-4904: Complete kit of parts including enclosure for preamp/mixer, No. P-1130-DM, for \$110.00; complete kit of parts including enclosure for preamp only, No. P-1130-PA, for \$70.00; etched and drilled preamplifier pc board, No. P-1130-BPA, for \$7.50; etched and drilled mixer pc board, No. P-1130-BDM, for \$7.50; power transformer T1, No. P-1130-T, for \$6.00; dpdt pushbutton switch, No. P-1130-SW1 \$1.00; four-station dpdt interlocked switch, No. P-1130-SW2, for \$4.00; BIFET quad op amp, No. P-1130-C-TL074CN, for \$3.50; dual lownoise amplifier, No. P-1130-C-LM387AM, for \$3.50; dual 2-watt audio amplifier, No. P-1130-C-LM377N, for \$3.00; and lownoise, high-gain op amp, No. P-1130-C-NE5534, for \$3.00. All integrated circuits are fully tested. Please allow six weeks for delivery. Connecticut residents add 7% sales tax; COD charge \$0.85. Handling charge is \$1.00 for orders of less than \$10.00. No shipping charges within continental U.S.
- \*—two of each component required for full stereo preamp/mixer.
- \*\*—four of each component required for full stereo preamp/mixer

C1, C2, C3, C20\*, C25-1000-µF, 16-volt

C5\*, C21, C22, C26-0.1-µF, 50-volt disc

C8\*\*, C11\*\*-4.7-µF, 16-volt electrolytic

C9\*\*-2200-pF, 5% tolerance polystyrene

C10\*\*-8200-pF, 5% tolerance polystyrene

C12\*\*, C13, C17\*, C18-1-µF, 16-volt elec-

IC1, IC2-LM387AN dual low-noise pre-

1C3, IC4, IC6-TL074CN BIFET quad opera-

IC5-LM377N dual 2-watt audio amplifier

IC7-NE5534N low-noise preamplifier (Sig-

The following are 1/4-watt. 5% tolerance car-

amplifier (National Semiconductor)

tional amplifier (Texas Instruments)

J2\*\*, J5\*, J6\*, J7\*, J8\*-phono jack

LED1-20-mA light emitting diode

C4, C14-220-µF, 35-volt electrolytic

C6\*\*-10-10-300-pF (see text)

C7\*\*-22-µF, 16-volt electrolytic

C15\*-0.1-µF, 10% tolerance Mylar

C16\*-0.01-µF, 10% tolerance Mylar C19\*-100-pF disc ceramic

C23, C24—0.01-µF disc ceramic C27\*—10-pF disc ceramic

D1 through D6-1N4001 rectifier

FI-1-ampere fast-blow fuse

(National Semiconductor)

J1\*\*-insulated phono jack

J4-1/4-inch stereo phone jack

J3-1/4-inch phone jack

electrolytic

ceramic

trolytic

netics)



cools down, thus avoiding thermal self-destruction.

The output of the mixer is applied to

the output conditioning section of the preamplifier (Fig. 7). Switch S3 provides tape monitor facilities. Potentiometers R44 and R45 are balance and volume controls, respectively. A BIFET buffer amplifies the signal and presents it to the output stage in which the tone controls are found. Drive signals for the system's power amplifier(s) are available at jacks J7 and J8.

The excellent power supply rejection of the integrated circuits employed in this project eliminates the need for a regulated supply. As shown in Fig. 8, two full-wave rectifiers and filter capacitors furnish the  $\pm 15$  volts required by the BIFET operational amplifiers and the microphone preamp. The  $\pm 15$ -volt line is tapped to power the monitor audio driver. A voltage doubler composed of *C3, C4, D5* and *D6* develops the  $\pm 25$  volts required by the phono preamps.

Construction. Printed circuit assem-



Fig. 7. Output stages of the preamplifier for one channel. Conventional bass and treble tone controls are included. as well as a tape monitor loop.

Fig. 8. Schematic of power supply, which delivers three voltages required by the disco mixer stages (+15, -15 and +25 V).

N4001(2)

±1 220µF +250





Fig. 9. Actual-size etching and drilling guide for the pc board for the preamplifier is shown at left. A diagram for components placement is above.

bly techniques are strongly recommended. Although it is possible to employ perforated board and point-to-point wiring, the layout of the high-gain stages (phono and microphone preamplifiers) is critical. These circuits are very sensitive to ground loops, hum fields and stray feedback paths. Extensive power supply decoupling is called for because the IC's employed in this project have very high gain-bandwidth products and can break into oscillation very easily.

Suitable etching and drilling and parts placement guides for the preamp and mixer boards are shown in Figs. 9 and 10, respectively. If mixing facilities are not desired, the preamp board can be used on its own as a high-quality stereo preamplifier. You will notice that both the schematic diagrams and the pc guides contain numerous points labelled with letters of the alphabet. These have been included to ease interconnections between the boards, and the boards and the jacks, switches and potentiometers. Most of these connections can be made (Continued on page 71)

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Fig. 10. Etching and drilling guide (left) and components place diagram (above) for the mixer pc board.

#### (Continued from page 69)

with hookup wire, but two-conductor shielded cable should be used between the phono input jacks and the preamp circuit board. If the preamp-only version of the project is built, install jumper wires between points AA and BB.

If desired, IC sockets or Molex Soldercons can be used when mounting the integrated circuits. Be sure to observe polarity and pin basing when installing electrolytics and semiconductors. Use good soldering practices.

When circuit board wiring has been completed, mount the boards in a metallic enclosure using machine hardware and spacers. The power transformer should be mounted directly in the enclosure. Those assembling the project using perforated board should take care to keep the power supply away from the high-gain sections of the preamplifier.

Use. The Disco Mixer is straightforward in its use. Patch in the various program sources with suitable lengths of shielded cable and experiment with the different preamp and mixer controls and monitor modes. Then invite your friends over and have a disco party! ♦