Owner's Manual

PCM 70

DIGITAL EFFECTS PROCESSOR

With Rev. 3.00 Software



Unpacking and Inspection

After unpacking the PCM 70, save all packing materials in case you ever need to ship the unit. Thoroughly inspect the PCM 70 and its packing materials for signs of damage in shipment. Reprt any damage to the carrier at once.

Precautions

The Lexicon PCM 70 is a rugged device with extensive electronic protection. However, you should observe the same reasonable precautions that apply to any piece of audio equipment.

- Always use the correct line voltage. Refer to Chapter 1 of this manual for power requirements.
- Don't install the PCM 70 in an unventilated rack, or directly above heat-producing equipment such as power amplifiers. Maximum ambient operating temperature is 35°C (95°F).
- · Never attach audio power amplifier outputs directly to any of the PCM 70's connectors.
- Before turning the PCM 70 on or off, mute your monitor speakers to avoid possible damage from transients.
- To prevent fire or shock hazard, do not expose the PCM 70 to rain or moisture.

Notice

Lexicon reserves the right to make improvements in this manual and the product it describes at any time and without notice or obligation.

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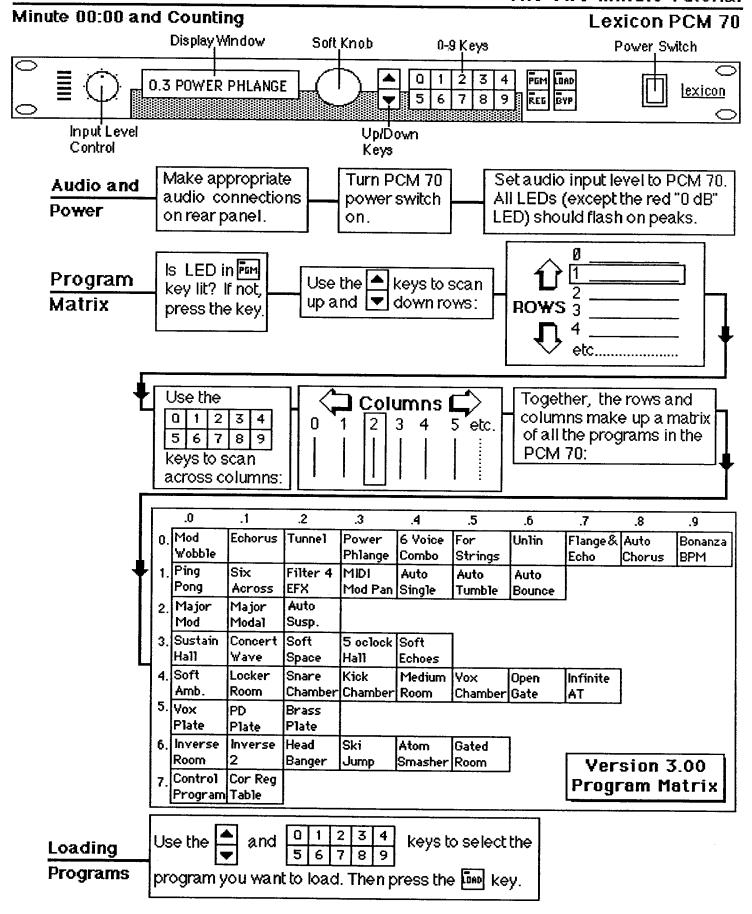
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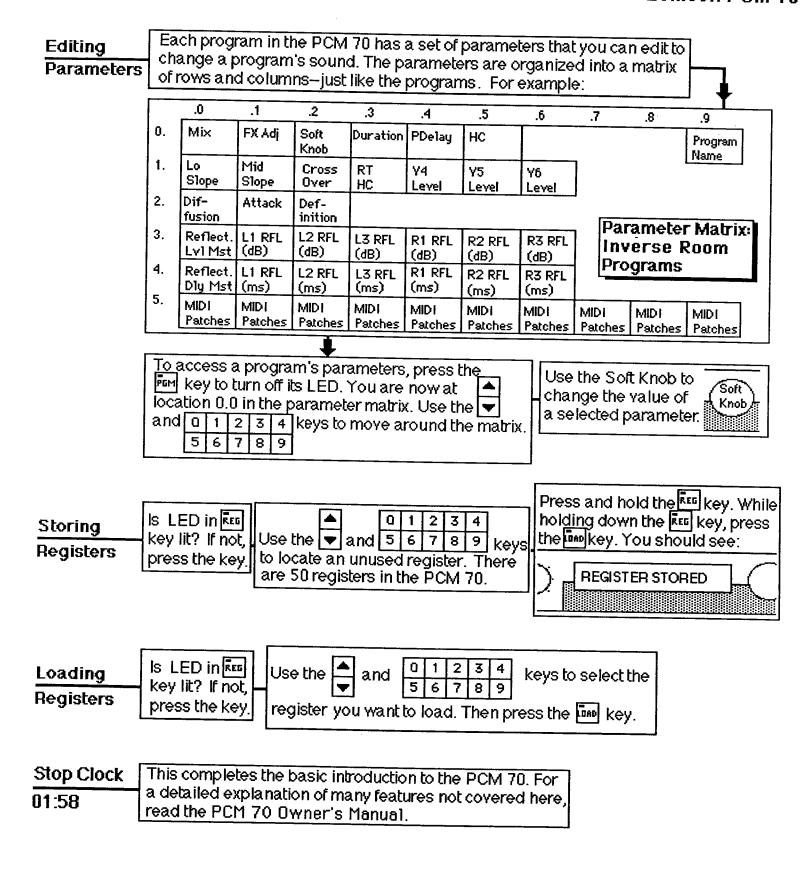
Table of Contents

	Unpacking and Inspection Two-Minute Tutorial New Features in Software Version 3.00 Introduction	ii iv vi vii
1.	Installing the PCM 70 Front Panel Controls Rear Panel Controls and Connections Mounting AC Power Requirements Voltage Changeover Instructions Making the Right Connections	1-1 1-2 1-3 1-4 1-4 1-4
2.	Getting Started with the PCM 70 The PCM 70 Operating System Using the Program Mode Using the Parameter Mode Using the Register Mode Setting Audio Levels A Good Start	2-1 2-2 2-2 2-3 2-5 2-6 2-6
3.	What the Parameters Do Effects Parameters Reverb Parameters Control Parameters	3-1 3-2 3-5 3-8
4.	Using the Preset Programs Row 0 - Echo and Chorus Programs Row 1 - Multiband Delay Programs Row 2 - Resonant Chord Programs Row 3 - Concert Hall Programs Row 4 - Rich Chamber Programs Row 5 - Rich Plate Programs Row 6 - Inverse Room Programs	4-1 4-2 4-4 4-6 4-8 4-10 4-12
5.	MIDI and the PCM 70 MIDI Connections and Applications Linking Two PCM 70s Moving Registers Between Two PCM 70s Using a Keyboard to Control a PCM 70 Controlling a PCM 70 from a Sequencer Register Editing and Storage with a Microcomputer Recording MIDI Program Changes	5-1 5-2 5-2 5-2 5-3 5-6 5-6 5-7
6.	MIDI Implementation Data MIDI Implementation Chart System Exclusive Message Protocols	6-1 6-2 6-4
7.	Solving Problems	7-1
В.	Specifications	8-1



Minute 01:00 and Counting

Lexicon PCM 70



Installing the PCM

This chapter describes how to install the PCM 70, and gives a brief introduction to its controls and connectors.

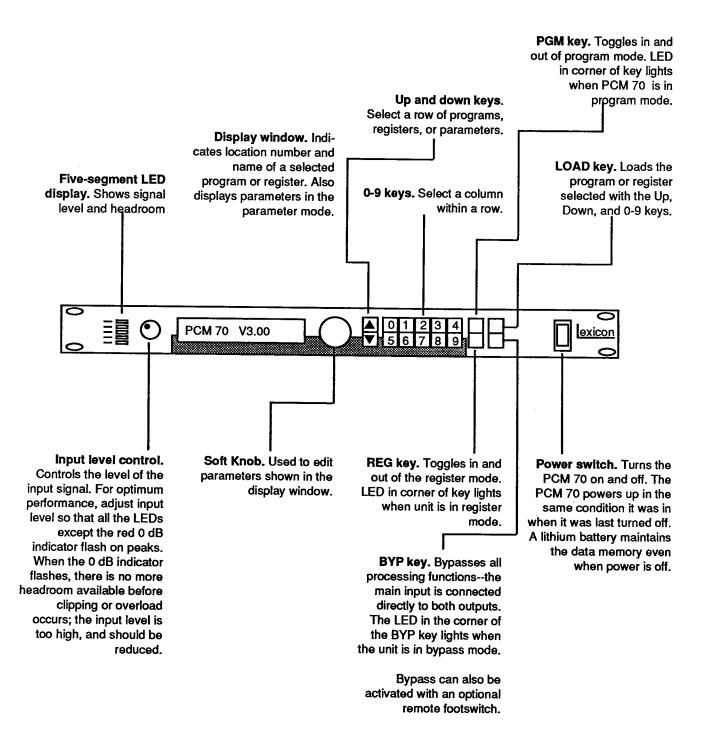


Figure 1.1. PCM 70 Front Panel.

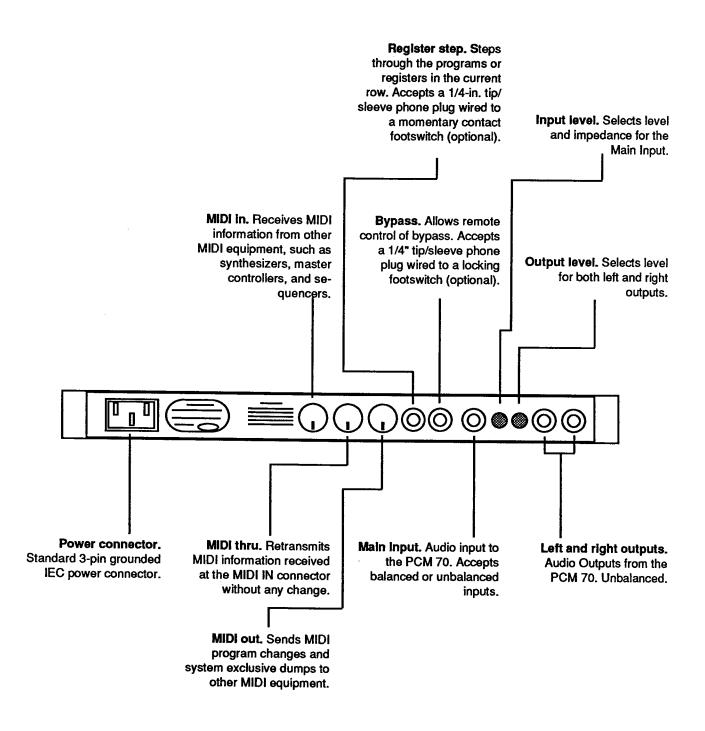


Figure 1.2. PCM 70 Rear Panel.

Mounting

The PCM 70 measures 19" wide x 1.75" high x 13.5" deep (483 x 45 x 344 mm. It uses one rack space and can be mounted on any level surface or in a standard 19 inch (483 mm) rack. Maximum ambient operating temperature is 35°C (95°F). Do not mount the PCM 70 directly above heat producing equipment like power amplifiers. Be sure to provide adequate ventilation if it will be mounted in a closed rack with other equipment such as effects units, synthesizer modules or power amplifiers.

If the PCM 70 is mounted in a rack or road case, support the rear of the chassis to prevent possible damage from mechanical shock.

AC Power Requirements

The PCM 70 is equipped with a three-pin IEC power connector and detachable cord, providing chassis grounding to the AC mains line. Plug the female end of the power cord into the PCM 70, and the male end into a wall outlet.

The PCM 70 operates at either 100/120 VAC or 220/240 VAC, depending on the setting of an internal voltage changeover switch and jumper and the rating of the internal mains fuse. The operating voltage set at the factory is marked on a label attached to the rear panel. Check the label *before* applying power to the unit.

AC Line Voltage	Jumper Position	F1 Fuse Rating
100V	E24-E23	1/4A 3AG Slo-Blo
120V	E24-E22	1/4A 3AG Slo-Blo
220V	E24-E23	1/8A 3AG Slo-Blo
240V	E24-E22	1/8A 3AG Slo-Blo

Table 1.3. Jumper Positions and Fuse Types.

Voltage Changeover

To change the operating voltage, have a qualified technician carry out the following procedure:

- 1. Set the voltage option switch (SW3) to the proper line voltage.
- 2. Set the power supply jumper (below the voltage option switch) to the proper line voltage, as shown in Table 1.3.
- 3. Replace fuse F1 with the correct fuse shown in Table 1.3.

Making the Right Connections

This section outlines some wiring possibilities for the PCM 70. Every system is unique, so we recommend that you experiment to arrive at the best configuration for you. Always check connections for proper impedance, polarity, and levels. This section discusses audio connections; for MIDI connections, refer to Chapter 5, MIDI and the PCM 70.

Input

The main input connector accepts a 1/4" tip/ring/sleeve phone plug for balanced inputs, or a 1/4" tip/sleeve phone plug for unbalanced inputs.

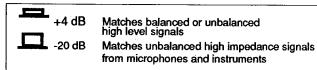


Figure 1.4. Input Level Switch Positions.

Input sensitivity and impedance are determined by the input level pushbutton.

Outputs

The left and right output connectors accept 1/4" tip/sleeve phone plugs for unbalanced outputs.

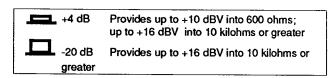


Figure 1.5. Output Level Switch Positions.

Output level is determined by the output level pushbutton, and also by the program the PCM 70 is running. Output impedance is 600 ohms, regardless of level setting.

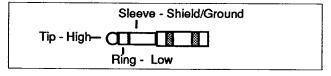


Figure 1.6. Wiring for Balanced Input.

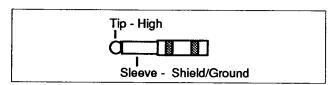


Figure 1.7. Wiring for Unbalanced Input or Outputs.

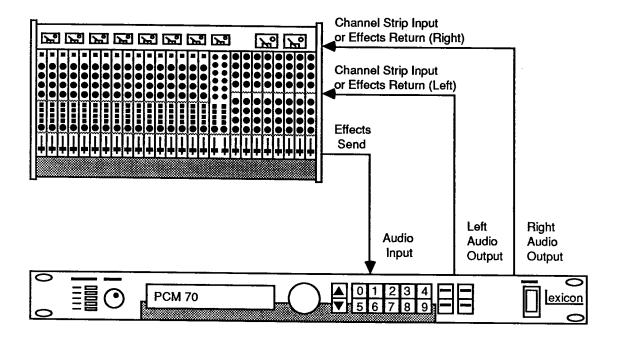


Figure 1.8. Audio Connections to Mixing Console.

Connection to a Mixing Console

The PCM 70 is a mono input, stereo output device. If your system includes a mixing console with one or more auxiliary (effects) sends, connect it as shown in Figure 1.8. In most applications, it is preferable to connect the PCM 70's outputs to two of the console's input channels rather than the effects returns. This allows greatest flexibility in routing, panning, and equalization. In either case, the outputs should generally be panned full left and right.

Using the PCM 70 in Mono

The PCM 70 produces wonderful stereo effects from a mono signal source. We recommend that you use the PCM 70 outputs in stereo whenever possible. However, if mono output is required, you can use either of the two output connectors--the left and right signals are summed internally when only one connector is used.

Getting Started with the PCM 70

This chapter describes how to use the PCM 70 controls and features.

The PCM 70 Operating System

The PCM 70's front panel is deceptively empty. A small number of knobs and switches together with the display window perform hundreds of functions, depending upon which operating mode is currently active. There are three operating modes:

- Program Mode
- Register Mode
- · Parameter Mode

The program mode is used to select and load preset programs--sets of instructions that tell the PCM 70 how to process the input signal. One program may produce chorusing, another echoes, another reverberation, and so on. The preset programs are permanently stored on ROM chips inside the PCM 70, and they cannot be accidentally erased or altered.

The parameter mode is used to adjust (edit) the controls (parameters) within a program, altering the sound it produces.

Once a program is edited, the new parameter settings can be stored in a register. The *register mode* is used to select, load, and store these registers.

A Synthesizer-Like Interface

If you are familiar with synthesizers, you will discover that the PCM 70's programs, registers and parameters are organized and accessed like the programs and parameters on many popular synthesizers.

Using the Program Mode

The PCM 70 is shipped with approximately 40 preset programs (the exact number may vary, depending upon the version of the software supplied with your unit.) This section describes how to select and load programs. Detailed information about the programs themselves is found in Chapter 4, Using the Preset Programs.

To allow quick and efficient program selection, the PCM 70 has a directory containing the name of each program. The directory is organized in a matrix of rows and columns, as shown in Figure 2.1. (If you're familiar with computer spreadsheets, you'll feel right at home with this system.)

The Display Window

If the PCM 70 had a very large display screen, you could view the entire program, register, and parameter directory on the screen at the same time. However, to keep things compact we have supplied a movable display window that lets you view a single item at a time. When an item appears in the display window, we say that it is *selected*.

The Up, Down, and 0-9 keys are used to move the display window over the matrix. If you press the Down key once, the display window moves down to the next row. Press the Up key once, and it returns to the previous row.

Each of the 0-9 keys corresponds to a column within the selected row. Pressing one of these keys moves you directly to that column.

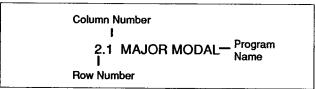
Some columns within a row may not be in use. If you try to move to one of them, a "NOT AVAILABLE" message appears in the display window.

A Unique Name for Every Program

How do you know where the display window is located within the matrix? Each program has a unique name and a location number which identifies the row and column of the display window. If you see:

2.1 MAJOR MODAL

you know that the Major Modal program is selected. The two-digit number to the left of the program name is the location number.



0.0 MOD WOBBLE	0.1 ECHORUS	0.2 TUNNEL	0.3 POWER PHLANGE	0.4 6 VOICE COMBO	0.5 FOR STRINGS	0.6 UNLIN	0.7 FLANGE ECHO	0.8 AUTO CHORUS	0.9 BONANZA BPM
1.0 PING PONG MOD	1.1 SIX ACROSS	1.2 FILTER 4 EFX	1.3 MIDI MOD PAN	1.4 AUTO SINGLE	1.5 AUTO TUMBLE	1.6 AUTO BOUNCE			
2.0 MAJOR MOD	2.1 MAJOR MODAL	2.2 AUTO SUSPENSE					1		
3.0 SUSTAIN HALL	3.1 CONCERT WAVE	3.2 SOFT SPACE	3.3 5 OCLOCK HALL	3.4 SOFT ECHOES					
4.0 SOFT AMBIENCE	4.1 LOCKER ROOM	4.2 SNARE CHAMBER	4.3 KICK CHAMBER	4.4 MEDIUM ROOM	4.5 VOX CHAMBER	4.6 OPEN GATE	4.7 INFINITE A T		
5.0 VOX	5.1 PD	5.2 BRASS				·		I	
PLATE	PLATE	PLATE							
6.0 INVERSE ROOM	6.1 INVERSE 2	6.2 HEAD BANGER	6.3 SKI JUMP	6.4 ATOM SMASHER	6.5 GATED ROOM				

Figure 2.1. Program Matrix.

With only a couple of exceptions, programs in the same row are related to each other—they use the same basic program to create their sound. For example, row 0 contains a variety of Chorus and Echo programs, and row 4 contains Rich Chamber reverberation programs. To display the type of programs that are stored in the current row, press and *hold* either of the Up or Down keys.

How to Select a Program

- 1. Enter the program mode by pressing the PGM key briefly. The LED in the PGM key lights when the program mode is active. If the LED doesn't light, the PCM 70 is still in the parameter mode--press the PGM key again.
- 2. To move the display window from row to row, (up and down) use the Up and Down keys.
- 3. To move the display window from column to column (side to side) use the 0-9 keys.
- 4. To select a specific program for use; 4.2 SNARE CHAMBER, for example, use the Up and Down keys to move to row 4, and then press the 2 key to select column 2.
- 5. Press the LOAD key to activate the selected program. The display window reads

LOADING PROGRAM

and then displays the program name. The program that was active previously will halt and the new program will begin processing audio after a brief pause.

Note: When you select a new program, the previously-loaded program will continue to process audio until you actually load the new program by pressing the LOAD key.

Using The Parameter Mode

The sounds of the preset programs supplied with the PCM 70 cover an astounding range of possibilities, but sooner or later you will want to alter the sounds of the programs to more perfectly fit your requirements. Each program in the PCM 70 contains a set of parameters that can be edited to create a sound uniquely your own.

The parameters are laid out in an arrangement of rows and columns, exactly like the program directory shown above. See Chapter 4 for a discussion of the parameters available for each of the programs in the PCM 70.

Entering the Parameter Mode

The PCM 70 is in the parameter mode whenever it is not in the program or register mode; that is, whenever both the PGM and REG LEDs are not lit.

To enter the parameter mode, look at the front panel. If the PGM LED is lit, press the PGM key briefly.

If the REG LED is lit, press the REG key briefly. The LED key goes out, indicating that the PCM 70 is now in the parameter mode, ready for you to edit the current program's parameters.

Editing Parameters

To get you started with parameters, let's load program 4.2 SNARE CHAMBER and edit two parameters that are frequently adjusted. The first parameter we'll edit is 0.0 MIX. MIX sets the ratio of dry (direct) sound to wet (processed) sound at the PCM 70 main outputs.

- 1. Select and Load 4.2 SNARE CHAMBER.
- 2. Put the PCM 70 in the parameter mode by pressing the PGM key briefly. The LEDs on the PGM and REG keys will both be unlit.
- 3. Experiment with the Up, Down and 0-9 keys. Notice that you can select parameters in the same way that you selected programs.
- 4. Select parameter 0.0. The display window reads

0.0 MIX 100% WET

- 5. Turn the Soft Knob counterclockwise, then clockwise. Notice that the MIX value in the display window varies as you turn the Soft Knob.
- 6. If the PCM 70 is connected to a mixing console, you'll probably want to use the console's controls to set the effects level in the mix, so MIX should be set to 100% wet.
- 7. If a musical instrument is connected directly to the input of the PCM 70, use MIX to balance the dry and wet sound. Adjust the Soft Knob until the balance sounds right to you.

Another frequently-adjusted parameter is 0.3 SIZE. SIZE sets the apparent size of the acoustic space that the PCM 70 creates.

1. Press the 3 key. The display window now reads:

0.3 SIZE 24.8 M

2. Turn the Soft Knob counterclockwise, then clockwise. Note that the SIZE parameter ranges from 5.6 to 32.6 meters.

The techniques you have just learned can be used to edit any PCM 70 parameter.

Using the Master Parameters

Many of the PCM 70 programs contain MASTER parameters that simultaneously affect an entire row of parameters. For example, in the Chorus and Echo programs, 2.0 DLY MST (delay master) simultaneously changes the individual delay settings for all six voices. MASTER parameters are always located in column 0.

The LOAD key has a special function when MASTER parameters are selected. Pressing the LOAD key allows you to view each of the parameters controlled by the MASTER. This is useful because you can quickly see the results of the overall changes you are making with the MASTER. Let's experiment.

- Load program 0.3 POWER PHLANGE
- 2. Enter the parameter mode and select 2.0 DLY MST.
- 3. Notice that the indicator on the LOAD key lights.
- 4. Press the LOAD key. The V1 DLY parameter appears in the display. Note the amount of delay currently set.
- 5. Press the LOAD key again. The V2 DLY parameter now appears in the display window.
- 6. Adjust the Soft Knob. Note that the V2 DLY value changes.
- 7. Press the LOAD key repeatedly until the V1 DLY parameter reappears. Note that the V1 DLY amount has also changed.

Using a MASTER parameter in combination with the LOAD key is an efficient way to adjust several related parameters.

For even more creative control, you can create a custom MASTER to control any combination of up to ten parameters simultaneously. To learn more about this advanced application, read Chapter 5, *MIDI* and the *PCM 70*.

Using the Register Mode

The ability to edit programs would be of little use if there was no way to store the edited programs. Not to worry-the PCM 70 has 50 registers available to store edited versions of the preset programs. Registers are organized into a matrix, selected, and loaded exactly like the preset programs. Once stored in a register, an edited program will be retained even if power is disconnected for long periods of time.

To enter the register mode, press the REG key. The LED on the REG key lights, indicating you are in the register mode. When you first receive your PCM 70, the registers will probably be empty and the display window will read

UNUSED

If you try to load an unused register, the display window reads

REGISTER UNUSED

Storing Edited Programs in Registers

To store an edited version of a preset program:

- 1. Press the REG key once. The LED in the corner of the key lights. This indicates that the PCM 70 is in the register selection mode. Names of stored registers appear in the display window.
- 2. Use the Up, Down, and 0-9 keys to locate an unused register.
- 3. Press and hold the REG key. While holding the REG key, press the LOAD key. The display window reads

REGISTER STORED

Note: Before you store a program, go into the parameter mode and select the parameter you adjust most often. Later, when you enter the parameter mode after loading the register, that parameter will be automatically selected and appear in the display window.

Loading Registers

- 1. Use the Up, Down, and 0-9 keys to select a register to load.
- 2. Press the LOAD key. The Display Window reads

LOADING REGISTER

Naming Registers

After it is stored, the edited program will still have the same name as the original program. To avoid confusion, the PCM 70 allows you to rename programs. The name of a program or register is always stored as parameter 0.9. Like any other parameter, the names of the programs cannot be permanently changed, but names of registers can be changed as often as you like. To change the name of a register:

- 1. Press the REG key briefly. The LED in the corner of the REG key should go out, indicating that the PCM 70 is ready to edit parameters.
- 2. Select parameter 0.9.
- 3. Press the LOAD key. The first character of the program name should begin to blink.
- 4. Adjust the Soft Knob until the first character of the new name appears in the display. In addition to the characters A-Z, you can select a numeral from 0-9, or a blank space.
- 5. When you have selected a character, press the LOAD key once. The display will advance to the next character. Continue entering characters until the display is full (enter blank characters at the end of the name if it is shorter than 13 characters).
- 6. When you are finished editing the name, press the REG key. The LED in the upper left-hand corner of the REG key lights.
- 7. To permanently store the new name, press and hold the REG key. While holding the REG key, press the LOAD key. The register is stored and the new name appears in the display window.

Note: In this example, we asked you to first store an edited program, change its name, and then store it again. Actually, there is no reason why you can't change the program's name before you store it in a register.

Register Protection

Setting up registers to meet your personal requirements can represent a considerable investment in time and effort. To reduce the possibility of accidental loss of these registers, the PCM 70 includes a memory protection feature. When this feature is turned on, the PCM 70 will not allow you to erase the contents of a register by overwriting it with a new program. However, unused registers remain available for storage of new programs. This allows you to protect registers from accidental loss of data without unnecessary inconvenience.

PCM 70 Owner's Manual - Software Version 3.00

To activate memory protection:

- 1. Load 7.0 CONTROL PGM.
- 2. Enter the parameter mode and select 0.1 M PROTECT (memory protection).
- 3. Rotate the Soft Knob until the display reads

M PROTECT ON

The memory protection will remain in effect until this parameter is changed back to

M PROTECT OFF

Setting Audio Levels

To obtain the best possible performance from the PCM 70, you should set the input and output audio levels with care.

- 1. Set the Input and Output Level pushbuttons on the rear panel to the position that best matches the other components in your system.
- 2. Apply an input signal at a level that you typically use.

- 3. While watching the HEADROOM display, adjust the front panel INPUT control so that all the LEDs except the red 0 dB indicator flash on peaks. A flashing red LED indicates that there is no more headroom available before clipping or overload occurs. The input signal is too high, and its level should be reduced.
- 4. Next, enter the parameter mode and set 0.0 MIX as described on page 24.

A Good Start

You should now know how to set audio levels, load a program, edit parameters, store the edited program in a register, and recall the register. This is all the information you need to begin using the PCM 70 effectively. Feel free to experiment--you can't hurt anything!

When you encounter a parameter you don't understand, refer to Chapter 3, What the Parameters Do for a description. Of course, not everything can be discovered by experimentation. When you are ready to find out more about how to use the programs stored in the PCM 70, read Chapter 4, Using the PCM 70 Preset Programs.

What the Parameters Do

This chapter describes (in alphabetical order) each of the parameters available in the PCM 70. Effects parameters are listed first, then reverb parameters, and finally control parameters. No single program has all the parameters described in this section; see the tables in Chapter 4, *Using the Preset Programs* for the location and availability of parameters for each program.

Effects Parameters

CHORUSING

CHORUSING modulates the time delays in both effects and reverb programs. It controls the rate of the chorus effect (depth is a preset value). At moderate settings it adds richness to the sound of nearly any instrument and creates the illusion of added voices. In larger amounts it produces flanging and pitch-twisting effects.

CHORUS (MODE)

This parameter is only available for the Row 0 Chorus and Echo programs which feature multiple voices (delays). It determines if CHORUSING is active, how many voices are chorused, and whether a triangle or sine wave LFO is used.

With the Soft Knob fully counterclockwise, the display window reads

CHORUS OFF

If the Soft Knob is turned one click in the clockwise direction, the display window reads

CHORUS 1 VC S

This indicates that chorusing is on for the first voice only, using a sine wave LFO. As you continue turning the Soft Knob in the clockwise direction, chorusing is added to voices 2, 3, 4, 5, and 6. Note that by only chorusing voices you are actually using, you increase the effect. When you pass voice 6, the display window returns to voice 1, but notice that the LFO changes from a sine wave to a triangle wave:

CHORUS 1 VC T

DIFFUSION

DIFFUSION controls the density of echoes. High levels of diffusion thicken or smear the echoes. This is most noticeable on material with sharp transients.

DLY MST (Delay Master)

DLY MST provides simultaneous changes in the delay times for all voices. To see how adjusting the master affects the individual voices, press the LOAD key to sequentially view each voice.

FDBK

See V1 FDBK.

FDBK MST (Feedback Master)

FDBK MST is available for programs which have a feedback level control for each voice. It allows simultaneous changes in feedback level for all voices. To see how adjusting the master affects the individual voices, press the LOAD key to sequentially view each voice.

FX ADJ (Effects Level Adjust)

FX ADJ controls the level of the processed (wet) signal before it is mixed with the dry signal (if any) and sent to the audio outputs. The range of this parameter is between -90 dB and +12 dB. FX ADJ is inactive when the PCM 70 is in the bypass mode, so it's a good idea to compare the levels with the bypass on and off while editing this parameter. Adjust it until the audio level sounds the same with bypass on and off.

HC (High-Frequency Cutoff)

HC sets the high-frequency cutoff (frequency) of a lowpass filter. Some programs offer separate left and right HC parameters, and others have a single HC parameter for both channels.

HC MST (High-Frequency Cutoff Master)

HC MST allows simultaneous adjustment of all HC parameters in multi-voice programs.

LC (Low-Frequency Cutoff)

LC sets the low-frequency cutoff of a high-pass filter.

LC MST (Low-Frequency Cutoff Master)

The LC MST allows simultaneous adjustment of the low-frequency cutoff (frequency) for multi-voice programs.

LVL

See V1 LVL.

LVL MST (Level Master)

The LVL MST controls the overall level in programs which have more than one voice. This permits simultaneous changes in all voice levels without altering the balance between individual voices.

MIX

MIX controls the ratio of dry vs. wet signal present at the PCM 70's outputs. When the PCM70 is patched into a

console or an instrument amplifier through an auxiliary or effects loop, this control should always be set to 100% wet. When an instrument is plugged directly into the PCM 70, a setting between 45 and 65% is a good starting point.

PAN MST (Pan Master)

PAN MST provides simultaneous control over the panning of all voices in a program, and shifts the stereo image from left to right.

PATCH

One of the PCM 70's most powerful features is its MIDI patching capability (Dynamic MIDI™). Each program has ten different patches which allow controls and events on keyboards, synths, and other MIDI devices to dynamically control the settings of up to ten different PCM 70 parameters--individually or together, and in real time. You can use performance events (such as last note played, after touch, or velocity) or controls (such as volume, pitch bender, mod control, footpedal, and other controls and switches). For a complete discussion of the PATCH parameter, see Chapter 5, MIDI and the PCM 70.

PCH MST (Pitch Master)

PCH MST provides simultaneous control over the pitches of all voices in the row 2 resonant chord programs. This allows a change of key without altering the relationship between individual voices.

PDL MST (Predelay Master)

PDL MST allows the predelay settings for all voices to be changed simultaneously.

RATE BPM (Rate in Beats Per Minute)

In programs which include BPM in their program name, the DLY MST parameter has been replaced with a RATE BPM parameter. This makes it very easy to match the tempo of rhythmic effects like echoes and resonant chords to the tempo of the music being processed by the PCM 70. For a different effect, try setting RATE BPM to twice or half the actual tempo of the music.

RESN

See V1 RESN.

RESN MST (Resonance Master)

RESN MST allows the resonance setting for all voices to be changed simultaneously.

SOFT KNOB

The PATCH parameter (described above) can assign one or more parameters to be controllable from the Soft Knob, instead of a MIDI device or event. While you probably won't want to use this capability to control a single parameter, it is very useful when you need to alter several parameters simultaneously. For a complete discussion of the SOFT KNOB and PATCH parameters, read Chapter 5, MIDI and the PCM 70.

V1 DLY

In programs with multiple delay voices, this parameter sets the delay time (in milleseconds) for the first voice. In BPM programs, DLY is set as a fraction of a single beat. The smallest fraction is 1/24th of a beat.

V1 FDBK (Voice 1 Feedback)

V1 FDBK controls the level and polarity of signals recirculated back to the input for the first voice in a program. Some programs have feedback controls for all voices, others only have themfor two of the available voices. A FDBK MST (Feedback Master) control (described above) allows simultaneous changes in all feedback parameters.

Increasing the amount of feedback can create interesting resonant effects, but be careful--if feedback is set too high, runaway feedback (oscillation) can occur. This does not indicate a problem with the PCM 70. Reduce the level of feedback and the oscillation will stop.

V1 HC (Voice 1 High-Frequency Cutoff)

V1 HC sets the high-frequency cutoff frequency of a low-pass filter for the first voice. There are as many HC controls as there are voices. The HC MST control (described above) allows simultaneous changes in all high-frequency cutoff frequencies, without altering the relationship between the voices.

V1 LC (Voice 1 Low-Frequency Cutoff)

V1 LC sets the low-frequency cutoff frequency of a 6 dB/octave low-pass filter. Most programs have a low-frequency cutoff control for each voice. An LC MST parameter (described above) allows simultaneous changes in all low-frequency cutoffs without altering the relationship between individual voices.

V1 LVL (Voice 1 Level)

V1 LVL adjusts the level for the first voice in a program. The voice can be turned completely off, full on, or

PCM 70 Owner's Manual - Software Version 3.00

anywhere in between, in 1/2 dB increments. There are as many voice level controls as there are voices. The LVL MST control (described above) allows simultaneous changes in all voice levels without altering the level balance between individual voices.

V1 PAN (Voice 1 Pan)

V1 PAN positions the output for voice one full left, full right, centered, or anywhere in between. Each voice can be panned independently, so some rather startling stereo effects are possible. A PAN MST control (described above) allows simultaneous changes in overall image position.

V1 PDL (Voice 1 Predelay Time)

V1 PDL adjusts predelay time for the first voice, in milliseconds. Each voice has its own voice delay time control, and a PDL MST (Predelay Master) control (described above) permits simultaneous control over all voices.

V1 PITCH (Voice 1 Pitch)

V1 PITCH tunes the first voice to a specific note and octave. Each voice has its own pitch control. A PCH MST control allows the program to be tuned to a new key, without altering the relationship between individual voices.

V1 RESN (Voice 1 Resonance)

V1 RESN sets the amount and polarity of feedback for the first voice. Each voice has its own resonance control.

Reverb Parameters

ATTACK

ATTACK sets the sharpness of the initial response to an input signal. High settings cause an explosive sound, while low settings cause the sound to build up more slowly with time. ATTACK only affects the level of sound within the first 50 milliseconds.

CHORUSING

CHORUSING randomizes delay times and introduces modulation to make reverberation sound less metallic. Increasing CHORUSING increases the rate of modulation. Because CHORUSING can cause pitch variation, this parameter should be set with care when using sources with very little pitch wobble (like guitar and piano). A good practice is to increase the setting until pitch wobble becomes noticeable, and then lower it slightly.

DECAY OPT (Decay Optimization)

DECAY OPT alters program characteristics in response to changes in input level to make reverb decay sound more natural. This parameter should normally be set to ON. However, with certain types of source material (e.g., soft low-frequency tones from a synthesizer) audible clicks may occur during level changes. If clicks are heard, set DECAY OPT to OFF.

DEFINITION

DEFINITION affects the echo density buildup rate during the latter part the decay period. With DEFINITION set to 0, the rate is determined by the program. Raising DEFINITION causes the sound to become choppier--the decrease in density of the echoes creates increasingly distinct, repetitive echo trails.

DIFFUSION

DIFFUSION controls the degree to which echo density increases over time. High DIFFUSION settings result in high initial buildup of echo density, and low settings cause low initial buildup. After the initial period (in which echo buildup is controlled by DIFFUSION), density continues to rise at a preset rate determined by the program.

DLY MST (Delay Master)

DLY MST allows simultaneous adjustment of the early reflection delay times.

DURATION

Duration determines the length of time which passes before cutoff in the Inverse Room programs. The value of this parameter is displayed in ms.

FX ADJ (Effects Level Adjust)

FX ADJ controls the level of the processed (wet) signal before it is mixed with the dry signal (if any) and sent to the audio outputs. The range of this parameter is between -90 dB and +12 dB. FX ADJ is inactive when the PCM 70 is in the bypass mode, so it's a good idea to compare the levels with the bypass both on and off while editing this parameter. Adjust it until the audio level sounds the same with bypass on and off. Some adjustment of the Input Level Control may also be necessary.

GATE

GATE sets the time delay in switching between running reverb time and stopped reverb time.

HC (High-Frequency Cutoff)

HC sets the frequency above which a 6 dB/octave lowpass filter attenuates the processed signal. It attenuates both preechoes and reverberant sound. High frequencies are often rolled off with this parameter, resulting in more natural-sounding reverberation.

LO SLOPE (Low Frequency Slope)

LO SLOPE determines the shape of the reverb envelope for low frequencies in the Inverse Room progams. When LO SLOPE is set to 0, the level of low reverb remains unchanged over its duration (see DURATION), and then cuts off abruptly (depending on how much DIFFUSION is in use).

Setting LO SLOPE above 0 causes the level of low frequency reverb to rise smoothly from soft to loud until the sound is cutoff. The greater the slope, the softer the initial reverberation and the more pronounced its rise. With negative values of LO SLOPE, the low frequency reverberation level drops from its initial level to a quieter one before cutoff. The lower the slope, the more pronounced the dropoff.

L RFL DB (Left Early Reflection Levels)

For each of the L RFL MS parameters, there is a corresponding L RFL DB parameters. L RFL DB sets the level (amplitude) for one of the left channel early reflections.

Early reflections can best be understood by visualizing a stage where the early reflections are the sounds emanating from the rear and side stage walls directly after the sound from the stage. Usually the rear stage wall reflection is earlier and louder than those from the two side walls. The early reflection parameters change the perceived reflecting surfaces surrounding the source.

L RFL MS (Left Early Reflection Delay Time)

L RFL MS sets the delay time in ms for the one of the left channel early reflections. The L RFL MS delay time parameters can be set independently of each other.

LVL MST (Level Master)

LVL MST allows simultaneous changes in the early reflection levels without altering the relationship between individual reflections.

MD SLOPE (Mid Frequency Slope)

Same as LO SLOPE, but applies to middle and high frequencies. The actual frequencies affected are determined by XOVER.

MIX

MIX controls the ratio of dry vs. wet signal present at the PCM 70's outputs. When the PCM 70 is patched into a console through an auxiliary or effects loop, this control should always be set to 100% wet. When an instrument is plugged directly into the PCM 70, a setting between 45 and 65% is a good starting point for experimentation with this parameter.

PATCH

One of the PCM 70's most powerful features is its MIDI patch capability. Each program has ten different patches which allow controls and events on keyboards, synths, and other MIDI devices to dynamically control the settings of up to ten different PCM 70 parameters--in real time. You can use performance events (such as last note played, after touch, or velocity) or controls (such as volume, pitch bender, mod control, footpedal, and other controls). For a complete discussion of the PATCH parameter, see Chapter 5, MIDI and the PCM 70.

PDELAY (Predelay)

PDELAY sets the amount of time that elapses between input and the onset of reverberation. Under natural conditions, the amount of predelay depends on the size and construction of the acoustic space and the relative 3-6

position of the sound source and the listener(s). PDE-LAY duplicates this phenomenon and is used to create a sense of distance and volume within an acoustic space. Relatively long PDELAY settings place the reverberant field behind rather than on top of the input material. Extremely long PDELAY settings produce unnatural sounds that often prove quite interesting.

A sense of continuity between source and reverb is maintained up to around 40 ms of predelay, after which the sound begins to separate into distinct patterns; however, large values of predelay can effectively give the impression of large size if early reflections are used to fill in the spaces between input and the delayed reverberation.

R RFL DB (Right Early Reflection Level)

The R RFL level parameter is identical to the L RFL level parameter discussed above, except that it controls early reflections for the right channel instead of the left.

R RFL MS (Right Early Reflection Delay Time)

The R RFL MS delay parameter is identical to the L RFL MS delay parameter discussed above, except that it controls early reflections for the right channel instead of the left.

INFINITE AT

Unlike the other reverb programs in the PCM 70, 4.7 INFINITE AT has a single reverb decay time control-REV TIME. This simplifies use of the infinite reverb feature. To activate infinite reverb, simply set REV TIME to its highest level. Whatever signal was present during the transition from normal reverb to infinite reverb will be sustained indefinitely.

RT HC (Reverberation Time High-Frequency Cutoff)

RT HC sets the frequency above which sounds decay at a progressively faster rate.

RT LOW (Low-Frequency Running Reverberation Time)

RT LOW sets the reverberation time for low-frequency signals.

RTL STOP (Low-frequency Stopped Reverberation Time)

RTL Stop sets the reverberation time for low-frequency signals in the absence of input--i.e., during pauses between signals input into the PCM 70.

RT Mid (Mid-Frequency Running Reverberation Time)

RT MID sets the reverberation time for mid-frequency signals.

RTM STOP (Mid-Frequency Stopped Reverberation Time)

RTM STOP sets the reverberation time for mid-frequency signals in the absence of input--i.e., during pauses between signals input into the PCM 70.

SIZE (Room Size)

SIZE sets the apparent size of the acoustic space produced by the PCM 70 (essentially the longest dimension, measured in meters). Note: Changing SIZE while a signal is present will result in a momentary muting of the reverb signal.

SOFT KNOB

The PATCH parameter (described above) can assign one or more parameters to be controllable from the Soft Knob, instead of a MIDI device or event. While there is no logical reason to use this capability to control a single parameter, it is very useful when you wish to alter several parameters simultaneously, without resorting to an external MIDI controller. For a complete discussion of the SOFT KNOB and PATCH parameters, read Chapter 5, MIDI and the PCM 70.

XOVER (Crossover Frequency)

XOVER sets the frequency at which the transition from low-frequency reverb time to mid-frequency reverb time takes place.

Control Parameters

AUTOLOAD

When AUTOLOAD is ON, every time a program is selected with the 0-9 keys, it is automatically loaded and used. When AUTOLOAD is OFF, programs must be manually loaded after selection. When shipped from Lexicon, AUTOLOAD is set to OFF. Be cautious about editing programs in the AUTOLOAD mode. It is very easy to erase your changes by accident.

M PROTECT (Memory Protect)

When M PROTECT is ON, the PCM 70 will not allow you to erase the contents of a register by overwriting it with a new program. Unused registers can still be written to. This allows you to protect registers from accidental loss of data while still creating new registers.

MIDI CHNL (MIDI Channel Select)

MIDI CHNL sets the MIDI Receive and Transmit Channel for the PCM 70. When OMNI MODE is OFF, only MIDI information received on this channel will be processed. For more information about MIDI applications, read Chapter 5, *MIDI and the PCM 70*.

OMNI MODE

When OMNI MODE is ON, information received on any of the 16 MIDI channels will be processed. When OMNI MODE is OFF, MIDI information will be processed only if it is received on the channel set with the MIDI CHNL parameter. In all but the simplest systems, OMNI MODE should usually be set to OFF to avoid undesired program changes and other unexpected results. For more information about MIDI applications, read Chapter 5, MIDI and the PCM 70.

PGM CHNG (Program Change Mode)

This parameter has three settings--OFF, FIX, and TABLE. When PGM CHNG is set to OFF, program change messages received from another MIDI device will be ignored.

When PGM CHNG is set to FIX, program change messages received from another MIDI device will be followed literally. Program change messages from 0 to 49 will load user registers 0.0 to 4.9. Program change messages 50 and up will select and load preset programs 0.0 and up. So, for example, if you select program 23 on a synthesizer, register 2.3 will be selected and loaded on the PCM 70. If you select program 64 on a synthesizer, preset program 1.4 will be selected and loaded on the PCM 70.

When PGM CHNG is set to TABLE, program change messages received from another MIDI device will be looked up on the Corresponding Register Table. The corresponding register or preset will then be loaded. This makes optimal use of the PCM 70's registers by allowing several synthesizer presets to be assigned to a single PCM 70 register. For more information on the Corresponding Register Table read Chapter 5, MIDI and the PCM 70.

REG CLEAR (Register Clear)

REG CLEAR allows the complete erasure of a user register. Select the register to be cleared with the Soft Knob, and press the LOAD key to erase its contents.

RESET MIDI

RESET MIDI restores the PCM 70's record of the values of all the MIDI controllers to their power-on condition. Select RESET MIDI and press the LOAD key. This parameter is useful when a series of parameter changes through Dynamic MIDI™ have left the PCM 70 in an unknown condition.

Using the Preset Programs

This chapter describes the effects and reverberation programs and presets supplied with Version 3.00 software. The programs are organized as shown in Table 2.1 in Chapter 2.

In general, all the presets in a row are closely related. They use the same algorithm, and share a common set of variable parameters. What makes the preset programs within a row sound different from one another are the *values* assigned to the parameters.

Row 0 - Chorus and Echo Programs

R0 Controls	0.0 MIX	0.1 FX ADJ	0.2 SOFT KNOB	0.3 CHORUSING	0.4 CHORUS MODE
R1 Voice Levels	1.0 LVL MST	1.1 V1 LVL	1.2 V2 LVL	1.3 V3 LVL	1.4 V4 LVL
R2 Delay Times	2.0 DLY MST	2.1 V1 DLY	2.2 V2 DLY	2.3 V3 DLY	2.4 V4 DLY
R3 Feedback	3.0 FDBK MST	3.1 V1 FDBK	3.2 V2 FDBK	3.3 V3 FDBK	3.4 V4 FDBK
R4 Panning	4.0 PAN MST	4.1 V1 PAN	4.2 V2 PAN	4.3 V3 PAN	4.4 V4 PAN
R5 MIDI Patches	5.0 MIDI PATCH	5.1 MIDI PATCH	5.2 MIDI PATCH	5.3 MIDI PATCH	5.4 MIDI PATCH

Table 4.1. Parameters--Row 0 Chorus and Echo.

In contemporary production, much attention is given to achieving a quality of fullness in the sound. The Chorus and Echo programs are extremely useful for increasing the fullness or richness of musical parts. It is very easy to start with a simple chorus effect and add repeats by increasing the delay time of one of the voices.

The Chorus and Echo programs in row 0 have six separately adjustable voices--allowing the PCM 70 to sound like a rack of six digital delay boxes. Each voice has its own independently-adjustable level control, delay time, feedback, and panning control. Sine Wave or Triangle wave chorusing can be selected for one or more of the voices.

Notes on the Parameters

All the parameters (and their locations) available for the Chorus and Echo programs are shown in Table 4.1.

The key to the sound of the Chorus and Echo programs is 0.3 CHORUSING. CHORUSING affects all six voices at once, and it varies the intensity of the random delay variation (pitch shifting). CHORUSING should be set with care to avoid pitch shifting, detuning, and unpleasant artifacts (unless these are desired!). A good approach is to increase CHORUSING until detuning or noise becomes audible, and then reduce it slightly.

The maximum delay variation for the CHORUSING parameter is 7.5 ms. If you use CHORUSING to produce interference effects between voices, their delays should be set to within 7 ms of each other.

The maximum delay time available in the Chorus and Echo programs is 432 ms. When the DIFFUSION parameter is in use, it adds between 4 and 20 ms of delay to the delay times shown in the display window. The exact amount of extra delay depends on the setting of DIFFUSION. This extra delay can be completely removed by setting DIFFUSION to 0. Delay times shown in the display window will then be accurate.

As with a traditional digital delay line, the delay times you select have a profound effect on the sound produced. Very short delays often affect the tonal quality of

the sound through phase cancellation. Longer delays are heard as discrete repeats or slaps. This is useful for either doubling or true slapback echoes. For doubling, try delay times between 15 and 50 ms, with slightly different delay times on several voices. Pan the voices to different locations for a thicker, more interesting sound.

For slapback echoes, try delays between 50 and 150 ms. Percussive material accentuates the discreteness of longer delays, but this can be smoothed out with 0.6 DIFFUSION. Vocals and other nonpercussive material can use longer delay times and lower diffusion.

The number of voices used has a marked effect on the sound produced by the Chorus and Echo programs. Generally, the more voices used, the thicker the sound. Phase cancellations between the voices can be used to color the sound, and panning the voices to different positions can create a dramatic sense of dimensionality.

Recirculation of the delays with the FDBK (feedback) parameters in row three forms a resonant comb filter and colors the sound in a pronounced manner. The result can be both interesting and bizarre.

Technical Background

Chorus and Echo is structured as a mono input feeding two distinct 0.6 DIFFUSERS. Each diffuser output is 0.5 HICUT filtered and feeds its own 432 ms delay line. When the diffuser is on, it adds an additional minimum delay of between 4 and 20 ms to the delay time shown on the display. This additional delay can be completely removed by setting DIFFUSION to 0. In applications which require precise delay time settings, or relatively short delay times (as in chorusing and flanging) DIFFUSION should be set to 0.

Voices 1 to 3 are taps from the first DELAY [1.1, 1.2, 1.3]. Voices 4 to 6 are taps from the second [1.4, 1.5, 1.6]. Each voice tap is individually CHORUSED [0.3, 0.4]. The chorused voices are fed to separate FEED-BACK [3.n] and output LEVEL [1.n] controls.

0.5 HC	0.6 DIFFUSION	0.7	0.8	0.9 NAME	R0 Controls
1.5 V5 LVL	1.6 V6 LVL	1.7	1.8	1.9	R1 Voice Lvls
2.5 V5 DLY	2.6 V6 DLY	2.7	2.8	2.9	R2 Delay Times
3.5 V5 FDBK	3.6 V6 FDBK	3.7	3.8	3.9	R3 Feedback
4.5 V5 PAN	4.6 V6 PAN	4.7	4.8	4.9	R4 Panning
5.5 MIDI PATCH	5.6 MIDI PATCH	5.7 MIDI PATCH	5.8 MIDI PATCH	5.9 MIDI PATCH	R5 MIDI Patches

The feedback for a voice is fed back to the beginning of its delay line, after the diffuser, but before the filter. Therefore the recirculation is always filtered, but only diffused once. Each voice may be individually PANNED [4.n] to either left or right regardless of its diffuser/delay source.

Notes on the Presets

0.0 Mod Wobble

Mod Wobble is a basic chorus effect with the mod wheel patched to several individual parameters: 2.0 DLY MST, 3.0 FDBK MST, 2.2 V2 DLY, 2.5 V5 DLY, 1.2 V2 LVL, and 1.5 V5 LVL. This allows you to go from a lush chorus to a full six voice echo by advancing the mod wheel.

0.1Echorus

Echorus is a combination program. Voices 1 and 2 are chorused and delay is quite short. Voices 4 and 5 are not chorused but have long delay times with moderate amounts of feedback. When patched to a MIDI controller, mod wheel and/or the last note played controls the panning of voices 4 and 5. This effect is subtle but elegant.

0.2 Tunnel

Tunnel is identical to the old Swarble preset, except that mod wheel is now patched as a controller over the delay times of voices 1, 2 and 6. Voices 3, 4, and 5 are not affected.

0.3 Power Phlange

Power Phlange uses six voices which are all set to very short delay times. Recirculation is fairly high and varies between positive and negative values. The negative values give the deep resonances.

0.4 6 Voice Combo (Six Voice Combination)

In 6 Voice Combo, chorusing appears at one output and echoes appear at the other. The chorused voices are panned independently from the unchorused echoes. Voices 1, 2, and 3 are short delays (chorused), all panned left. Voices 4, 5, and 6 have longer delay times, but are not chorused. They are panned right.

0.5 For Strings

For Strings is a heavily diffused effect designed specifically for use with strings. Nothing else is likely to sound good with it. Aftertouch is patched to 0.3 CHORUS.

0.6 UNLIN

Unlin is a six-voice delay effect, in which each voice gets progressively louder over time. Last note is patched to 2.0 DLY MST and 3.0 FDBK MST.

0.7 Flange O Echo

Flange O Echo is another combination program. Voices 1, 2, and 3 are extremely short delays with feedback. Voices 4, 5, and 6 are much longer delays, panned right, with voice 6 fed back to the input. Mod wheel is patched to control 2.1 V1 DLY, 3.1 V1 FDBK, and 3.6 V6 FDBK. Advancing the mod wheel gives the effect more balance between left and right.

0.8 Auto Chorus

Auto Chorus is a two voice delay. The voices have independent feedback paths. MIDI clock is patched to control 2.0 DLY MST (Tempo).

0.9 Bonanza BPM

Remember the TV show? Bonanza BPM does!. It is a six voice panned delay with a catchy rhythm. It operates best at tempos faster than 114 BPM.

Row 1 - Multiband Delay Programs

R0	Controls	0.0	MIX	0.1	FX ADJ	0.2	SOFT KNOB	0.3	CHORUSING	0.4	CHORUS MODE
R1	Voice Lvls	1.0	LVL MST	1.1	V1 LVL	1.2	V2 LVL	1.3	V3 LVL	1.4	V4 LVL
R2	Delay Times	2.0	DLY MST	2.1	V1 DLY	2.2	V2 DLY	2.3	V3 DLY	2.4	V4 DLY
R3	Low Freq. Cut	3.0	LC MST	3.1	V1 LC	3.2	V2 LC	3.3	V3 LC	3.4	V4 LC
R4	Hi Freq. Cut	4.0	HC MST	4.1	V1 HC	4.2	V2 HC	4.3	V3 HC	4.4	V4 HC
	Panning										V4 PAN
R6	MIDI Patches	6.0	MIDI PATCH	6.1	MIDI PATCH	6.2	MIDI PATCH	6.3	MIDI PATCH	6.4	MIDI PATCH

Table 4.2. Parameters - Row 1 Multiband Delays.

The family of seven programs in row one are called Multiband Delays. They feature six separately adjustable voices, each with its own level control, delay time, low and high frequency filters, and pan controls. Voices 1 and 2 have independent feedback controls, and a master diffusion parameter affects all six voices.

The maximum delay time available is 936 ms--more than twice that of the Chorus and Echo programs. There are individual LC (low-cut) and HC (high-cut) parameters for each voice.

Notes on the Parameters

When the DIFFUSION parameter is in use, it adds between 4 and 20 ms of delay to the delay times shown in the display window. The exact amount of extra delay depends on the setting of DIFFUSION. If desired, this extra delay can be completely removed by setting DIFFUSION to 0. Delay times shown in the display window will then be accurate.

The parameters (and their locations) available for the Multiband Delay programs are shown in Table 4.2.

Technical Background

Multiband Delay has a single DIFFUSER [0.3] feeding a 936 ms delay line. When the diffuser is on, it adds an additional minimum delay of between 4 and 20 ms to the delay time shown on the display. This additional delay can be completely removed by setting DIFFUSION to 0. In applications which require precise delay time settings, or relatively short delay times (as in chorusing and flanging) DIFFUSION should always be set to 0.

Each DELAY [2.n] tap feeds individual sets of LOCUT [3.n] and HICUT [4.n] filters for each voice. Each voice also has individual LEVEL [1.n] and PAN [5.n] controls. The filters for voices 1, 2, and 3 have the most settings and a 12 dB/octave rolloff. The filters for voices 4 and 5 have 12 dB slopes, but have fewer settings. Voice 6 has a more gentle 6 dB per octave slope and a limited number of values.

Pre-fade FEEDBACK [0.4, 0.5] is available on voices

1 and 2. It is mixed with the input prior to the diffuser, so that the recirculated sound is repeatedly diffused and filtered.

Notes on the Presets

1.0 Ping Pong Mod

Ping Pong Mod is a straight forward two voice delay effect with mod wheel patched to 2.0 DLY MST. Negative scaling is used so that the delay master decreases as the mod wheel is advanced.

1.1 Six Across

Six Across uses six voices--all filtered to different bandwidths and panned to different locations. V1 is the longest and is recirculated back to the input for another "trip across".

1.2 Filter 4 EFX

This six voice delay program loads with all voices fairly closely spaced with no recirculation. The voices are moderatley diffused, filtered, and tight, producing an "inverse" delay effect. Turning on a port switch on a MIDI controller feeds voice 1 back to the input where it is rediffused. Duane Eddy never had it so good!

Mod wheel controls 4.0 HC MST. As the mod wheel is pushed forward, each voice gets brighter, up to a maximum of seven increments. Mod wheel is also assigned to 2.0 DLY MST--twice, to give the mod wheel more effective range.

1.3 MIDI Mod Pan

MIDI Mod Pan is a two voice delay program with MIDI clock patched to 2.0 MST TEMPO. The sound source should be mixed through the PCM 70, with 0.0 MIX set to 100% WET. Voice 1 is set for a 0 ms delay and panned left. Voice 2 is set to 3/24 (eighth note delay). Advancing the Mod Wheel changes the pan of each voice to the opposite direction. MIDI clock maintains the BPM master parameter. Great for guitar solos!

0.5 V2 FDBK	0.6	0.7	0.8	0.9 NAME	R0 Controls
1.5 V5 LVL	1.6 V6 LVL	1.7	1.8	1.9	R1 Voice Lvls
2.5 V5 DLY	2.6 V6 DLY	2.7	2:8	2.9	R2 Delay Times
3.5 V5 LC	3.6 V6 LC	3.7	3.8	3.9	R3 Low Freq. Cut
4.5 V5 HC	4.6 V6 HC	4.7	4.8	4.9	R4 Hi Freq. Cut
5.5 V5 PAN	5.6 V6 PAN	5.7	5.8	5.9	R5 Panning
6.5 MIDI PATCH	1 6.6 MIDI PATCH	6.7 MIDI PATCH	6.8 MIDI PATCH	6.9 MIDI PATCH	R6 MIDI Patches

1.4 Auto Single

Auto Single is a single voice delay that syncs with drum machines, sequencers, or any controller that transmits a MIDI clock. If MIDI clock isn't available, simply adjust 2.0 RATE manually. Voice 1 is adjusted at 24/24 = 1 beat.

1.5 Auto Tumble

Auto Tumble is a four voice tumbling delay program

with MIDI Clock controlling the tempo (2.0 RATE) of the tumbling delays.

1.6 Auto Bounce

Auto Bounce is similar to the old Bouncing BPM program, except that the mod wheel controls the spread or panning of the first two voices. The Soft Knob is also patched to the same two voices. Go to parameter 0.2 SOFTKNOB to access this local custom master control.

Row 2 - Resonant Chord Programs

1						
	Controls		0.1 FX ADJ	0.2 SOFT KNOB	0.3 V3 FDBK	0.4 V6 FDBK
		1.0 LVL MST		1.2 V2 LVL		1.4 V4 LVL
		2.0 PITCH MST		2.2 V2 PITCH	2.3 V3 PITCH	2.4 V4 PITCH
R3	Resonances	3.0 RESN MST	3.1 V1 RESN	3.2 V2 RESN	3.3 V3 RESN	3.4 V4 RESN
		4.0 PDL MST				4.4 V4 DELAY
				5.2 V2 PAN	5.3 V3 PAN	5.4 V4 PAN
R6	MIDI Patches	6.0 MIDI PATCH	6.1 MIDI PATCH	6.2 MIDI PATCH	6.3 MIDI PATCH	6.4 MIDI PATCH

Table 4.3. Parameters - Row 2 Resonant Chords.

The family of four programs in row two are called Resonant Chords. These programs are unlike anything that has ever appeared before in an effects processor. Impulsive energy at the input is used to excite six resonant voices (notes). The level, pitch, duration, and high-frequency cutoff of the overtones for each voice are separately controllable. Each voice may be panned independently.

The notes resonate to some degree with any input, but the most effective excitation contains all frequencies, like percussion. Experimentation with drums or a drum machine can produce interesting and original sounds. Other instruments (like synthesizers and guitars) may give a quality of tonal ambience in which certain notes rise ethereally out of the background.

Notes on the Parameters

The parameters (and their locations) available for the Resonant Chord programs are shown in Table 4.3.

Row one contains level parameters for each voice. Use these level controls to balance the volume levels of each voice.

If fewer than six voices are required, reduce the level control for one or more voices until the display reads "OFF."

Row two contains pitch controls for each voice. Once a chord is created with the individual pitch controls, PCH MST (pitch master) can quickly change the chord's key.

Row three contains RESN (resonance) parameters for each voice. RESN determines how long the note resonates. These parameters have a zero center, producing increasing amounts of positive resonance if turned in one direction and increasing amounts of negative resonance if turned in the other.

Positive feedback generates all the harmonics of each note, and negative feedback produces only the odd harmonics.

Row four contains DLY (delay) parameters, which set the timing for each voice. Small amounts of delay create a chord. By gradually increasing delay with each note, the pulse becomes a strum. A large amount of delay produces a sequence of individual notes.

Row five contains PAN parameters for each voice.

Technical Background

Resonant Chords has a single 773 ms delay line. Each DLY [4.n] tap feeds individual HICUT filters grouped for control as HICUT LEFT [0.5] and HICUT RIGHT [0.6] for voices 1 to 3, and 4 to 6 respectively. Each filter feeds a delay line resonator whose length determines its PITCH [2.n]. The amount of feedback around the pitch delay line determines the amount of RESONANCE [3.n]. The sign of the resonance determines whether the note has the pitch shown and all its harmonics (positive) or a pitch one octave lower and odd harmonics (negative). The feedback is to the filter input so that the recirculated sound becomes progressively smoother.

The output of each resonator has individual LEVEL [1.n] and PANNING [5.n] controls. Each voice can be panned left or right independently of which hicut filter group it is in.

There are two separate overall FEEDBACK [0.3, 0.4] paths that are linked to the delay tap settings of voices 3 and 6. The sound is picked off the delay line before the filter and resonator. The sound from tap 3 has negative feedback; tap 6 has positive feedback.

Notes on the Presets

2.0 Major Mod

Major Mod incorporates six resonating voices tuned to an F Major chord. the highest note resonates first, with the other five voices following with a 75 ms delay between each voice. Mod wheel is patched with negative scaling to 4.0 DLY MST, which collapses the voices towards 0.0 ms. Pitch wheel is patched to control the relative pitches of voices 3, 2, and 5.

2.1 Major Modal

Major Modal is designed for use with a keyboard. Last note played is patched to control the pitch of all six voices through 2.0 PCH MST. Each note on the keyboard corresponds to the associated modal chord.

2.2 Auto Suspense

MIDI clock is patched to control 2.0 PCH MST and 4.0 RATE MST. MIDI clock is also patched with negative scaling to 0.5 HI CUT L and 0.6 HI CUT R. Great for daytime soaps.

Row 3 - Concert Hall Programs

1			
R0 Reverb Ctrls 0.0 MIX	0.1 FX ADJ 0.2	SOFT KNOB =0.3 SIZE	0.4 GATE
R1 ReverbTime 1.0 RT LOW	1.1 RT MID 1.2	XOVER - 1.3 RT HC	1.4 RT L STOP
R2 Reverb Design 2.0 DIFFUSION	2.1 ATTACK 2.2	DEFINITION 2.3	2.4
R3 Reflection LvI 3.0 LVL MST	3.1 L1 RFL (DB) 3.2	L2 RFL (DB) 3.3 R1 RFL (DB)	3.4 R2 RFL (DB)
R4 Reflection Dly 4.0 DLY MST	4.1 L1 RFL (MS) 4.2	L2 RFL (MS) 4.3 R1 RFL (MS)	4.4 B2 BFL (MS)
LR5 MIDI Patches 5.0 MIDI PATCH	5.1 MIDI PATCH 5.2	MIDLPATCH 5.3 MIDLPATCH	5.4 MIDI PATCH

Table 4.4. Parameters - Row 3 Concert Halls.

The Concert Hall Programs in Row 3 are reverberation programs which emulate real concert halls. The reverberation is very clean, and designed to remain behind the direct sound--adding ambience, but leaving the source unchanged. These programs have a relatively low initial echo density which builds up gradually over time.

The Concert Hall programs are especially good with classical music. For popular music, they can give separately recorded tracks the sense of belonging to the same performance by putting the whole mix in the context of a real-sounding acoustic space.

Notes on the Parameters

The parameters (and their locations) available for the Concert Hall programs are shown in Table 4.4.

0.8 CHORUSING is unique to the Concert Hall program and may be adjusted to give a more random decay characteristic.

Technical Background

All of the reverb programs share the same basic structure. Sound is sent through a HICUT [0.6] filter to a delay line. The PREDELAY [0.5] tap feeds two separate diffusers with a common control, DIFFUSION [2.0]. The diffusers feed the reverberators which also have common controls. The static parameters, ATTACK [2.1] and DEFINITION [2.2], control the quality of the reverberation.

The reverb time parameters, REVERB TIME LOW [1.0], REVERB TIME MID [1.1], CROSSOVER [1.2], and REVERB TIME HICUT [1.3] control the frequency response and length of the decay.

The SIZE [0.3] control changes the lengths of the loops used in the reverberator structure and simulates changing the physical dimensions (in meters) of the acoustic space being recreated. Large sizes may also limit the amount of memory available for predelays and early reflections. This is most noticeable for very large sizes of Rich Chamber. The PCM 70 will tell you if you request a predelay too long for the size, and will give the maximum available.

For example, the display

0.5 PDELAY > 116MS

results from requesting more than 116 ms of predelay for a Rich Chamber with a SIZE of 29.1 meters.

Early Reflections

Early Reflections are unprocessed sound tapped directly from the delay line and sent unpanned to the left and right outputs. They may be used to simulate strong reflections off hard surfaces in real spaces, or for special effects. The Concert Hall program has four, and the Rich Chamber and Rich Plate have six. Their LEVEL [3.n] and DELAYS [4.n] are independent of each other and the reverberator. It is possible to have the early reflections arrive after the onset and decay of the reverberation.

Decay Optimization

This feature is available on all reverb programs and gives a more realistic timbre to long decays. If it produces artifacts on pure sounds that are varying slowly around its threshold, DECAY OPT [0.7] may be switched off.

1					
0.5 PDELAY 1.5 RT M STOP	0.6 HC		0.8 CHORUSING	0.9 NAME	R0 Reverb Ctrls
	1.6	1.7	1.8	1.9	R1 Reverb Time
2.5	2.6	2.7	2.8	2.9	R2 Reverb Design
3.5	3.6	3.7	3.8	3.9	
4.5	4.6	4.7	4.8		R3 Reflection Lvl
5.5 MIDI DATCH			4.0	4.9	R4 Reflection Dly
D.D. WIDLI ATOLI	S.O. MIDI PATCH	5.7 MIDI PATCH	5.8 MIDLPATCH	5.9 MIDI PATCH	R5 MIDI Patches

3.0 Sustain Hall

Sustain Hall is a bright concert hall with no preechoes. The Soft Knob is patched to 1.0 RT LOW and 1.1 RT MID with identical scaling. Note that the Soft Knob comes up with a value of 10; turning it counterclockwise shortens the running reverb time, while turning it clockwise lengthens the reverb times. Pressing the MIDI sustain pedal increases the setting of 1.0 RT LOW by 10 increments to 14 seconds, increases the 1.1 RT MID value by 6 increments to 3.8 second, and increases the 0.8 CHORUSING value by 8 increments to 59.

3.1 Concert Wave

Concert Wave is excellent for clean, bright sources like compressed Stratocaster guitars, FM synthesizers, or even a Rhodes. The reverb attack is set fairly far back, and room size is set to maximum. The lush wave effect comes from the long (5.0 second) setting of 1.0 RT LOW and high (15.0 kHz) setting of 1.2 XOVER. Some chorusing is added to the reverb wave, and even more can be added with aftertouch. Velocity has been patched to the preecho levels.

3.2 Soft Space

Soft Space can take you from a small, confined space to a 747 hangar with a twist of the Soft Knob. The Soft Knob is patched to both 1.0 RT LOW and 1.1 RT MID. To use the Softknob, load the program, go into the parameter mode and select location 0.2 SOFT KNOB. Mod wheel is patched to 0.8 CHORUS to add a little wobble to the space.

3.3 5 o Clock Halli

5 o Clock Hall is an empty hall. It has no MIDI patches.

3.4 Soft Echoes

Soft Echoes is a reverb effect which starts off with four discrete predelay echoes. Turning 0.2 SOFT KNOB lengthens the RT times and raises 1.3 RT HC, making the ambience brighter.

Row 4 - Rich Chamber Programs

	s 0.0 MIX	0.1 FX ADJ	0.2 SOFT KNOB 20.3 SIZE	0.4 GATE
	1.0 RT LOW		1.2 XOVER 1.3 RT HC	1.4 RT L STOP
			2.2 DEFINITION 2.3	2.4
R3 Reflection L	vl 3.0 LVL MST	3.1 L1 RFL (DB)	3.2 L2 RFL (DB) 3.3 L3 RFL (DB)	3.4 R1 RFL (DB)
R4 Reflection D	ly 4.0 DLY MST	4.1 L1 RFL (MS)	4.2 L2 RFL (MS) 4.3 L3 RFL (MS)	4.4 R1 RFL (MS)
R5 MIDI Patche	s_5.0 MIDLPATC	H 5.1 MIDLPATCH	5.2 MIDI PATCH 5.3 MIDI PATCH	5.4 MIDLPATCH

Figure 4.5. Parameters - Rich Chamber Programs.

The Rich Chamber Programs in row 4 produce an even, relatively dimensionless reverberation, with little change in color as the sound decays. The initial diffusion is similar to the Concert Hall programs, but the sense of space and size is much less obvious. This characteristic, along with the low color in the decay tail make the Rich Chamber programs useful on a wide range of material. They are especially useful on spoken voice, giving a noticable increase in loudness with very low color.

Notes on the Parameters

When 2.0 DIFFUSION is set to a low or moderate level, the Rich Chamber programs are useful for classical music, especially piano (with short reverberation times) and organ (with long reverberation times). High diffusion settings produce the sound of a large acoustic chamber and are useful on all types of popular music.

Use 0.3 SIZE to maximize the versatility of the Rich Chamber programs. When set to sizes of 16 meters or less, Rich Chamber is tight and articulate--very useful for dialogue and music. Larger sizes (around 60 meters) give an excellent concert hall sound, especially if 2.0 DIFFUSION is set to about 50.

The parameters (and their locations) available for the Rich Chamber programs are shown in Table 4.5.

Notes on Gated Reverb

Each of the basic reverb programs can be gated, but generally the Rich Chamber or Rich Plate will sound better. The gate works by changing the reverberator from the running RT [1.0, 1.1] parameters to REVERB TIME LOW STOP [1.4] and REVERB TIME MID STOP [1.5] when the current input level drops below an internally calculated threshold, for longer than the GATE [0.4] time. Artifacts of this parameter change may be audible on some program material, but can be minimized by using PREDELAY [0.5], typically around 24 ms.

An effect similar to gated reverb can be obtained with the Inverse Room program, described later in this chapter. If you are encountering problems with the gate we recommend that you try Inverse Room.

Notes on the Presets

4.0 Soft Ambience

Soft Ambience starts off fairly tight with 1.0 RT LOW at 1.6 seconds and 1.1 RT MID at 0.36 seconds. Turning 0.2 SOFT KNOB lengthens the reverb times and also raises 1.3 RT HC, making the ambience brighter.

4.1 Locker Room

Locker Room is a medium sized (17.0 meter) room with very short reverb times. 2.2 DIFFUSION is set very high. Four preechoes bring the sound close in.

4.2 Snare Chamber

Snare Chamber is a medium-large room with moderate density. The running reverb times are .51 seconds for 1.0 RT LOW and .31 seconds for 1.1 RT MID. 2.2 DIFFUSION is set medium to low, with a sharp 2.1 ATTACK. 2.2 DEFINITION is set very high, which produces a fair amount of "room chatter." Adjust 0.3 SIZE to best reinforce the drum resonance.

4.3 Kick Chamber

Kick Chamber is designed specifically to complement kick drum. It has some strong short reflections which bring the sound up front.

4.4 Medium Room

Medium Room is useful for adding density or sense of space without adding a lot of reverb. Velocity is patched to 3.0 LVL MST. The harder one plays the keyboard, the louder the reflections are.

4.5 Vox Chamber (Vocal Chamber)

Vox Chamber is a room with a large area with low amplitude and short preechoes. 1.2 XOVER is set to maximum, 1.0 RT LOW is set to .46 seconds, and 1.1 RT MID is set to 29 seconds. The resulting sound has a unique combination of echo and reverb, excellent for lofty vocals. 0.2 SOFTKNOB is set to 127; turning it counterclockwise lowers 1.0, 1.1 and 1.2 to yield a more compact reverb.

0.5 PDELAY	0.6 HC	0.7 DECAY OPT	0.8	0.9 NAME	R0 Reverb Ctrls
1.5 RT M STOP	1.6	1.7	1.8	1.9	R1 Reverb Time
2.5	2.6	2.7	2.8	2.9	R2 Reverb Design
	3.6 R3 RFL (DB)		3.8	3.9	R3 Reflection Lvl
	4.6 R3 RFL (MS)		4.8	4.9	R4 Reflection Dly
5.5 MIDI PATCH	5.6 MIDI PATCH	5.7 MIDI PATCH	5.8 MIDI PATCH	5.9 MIDLPATCH	R5 MIDI Patches

4.6 Open Gate

Open Gate is exactly the opposite of a gated chamberthe reverb time goes from short to long. 0.4 GATE is set for 126 ms. This means that 126 ms after signal is no longer present at the input, the reverb time switches to the values at 1.4 RTL STOP and 1.5 RTM STOP.

4.7 Infinite A T

Infinite AT uses a slightly different algorithm than the other Rich Chamber programs. We have provided a single reverb time control (0.4 REV TIME) to simplify use of the infinite reverb effect. Just turn this control all the way up to INFINITE to activate infinite reverb. To add another sound to the loop, lower the control one or more steps, input the sound, and then immediately return the control to INFINITE. For use with MIDI controlers, aftertouch and mod wheel have been patched to 0.4 REV TIME. Play hard to "capture" a sound into infinite reverb. Advance the mod wheel to release the sound and set a lower reverb time.

Early Reflections are also available, allowing you to have extra delay lines running with the infinite loop.

Row 5 - Rich Plate Programs

R0	Reverb Ctrls	0.0	MIX	0.1	FX ADJ	0.2 SQFT KNOB	0.3	SIZE	0.4	GATE
	ReverbTime			1.1	RT MID	1.2 XOVER	1.3	RT HC	1.4	RT L STOP
	Reverb Design					2.2 DEFINITION			2.4	
	Reflection LvI			3.1	L1 RFL (DB)	3.2 L2 RFL (DB)	3.3	L3 RFL (DB)	3.4	R1 RFL (DB)
	Reflection Dly		DLY MST	4.1	L1 RFL (MS)	4.2 L2 RFL (MS)	4.3	L3 RFL (MS)	4.4	R1 RFL (MS)
R5	MIDI Patches	5.0	MIDI PATCH	<u>5.1</u>	MIDI PATCH	5.2 MIDI PATCH	5.3	MIDI PATCH	5.4	MIDI PATCH

Figure 4.6. Parameters - Rich Plate Programs.

The Rich Plate programs in row 5 mimic the sounds of metal plates, with high initial diffusion and a relatively bright sound. This makes them a good choice for enhancing any type of percussion.

Notes on the Parameters

The parameters (and their locations) available for the Rich Plate programs are shown in Table 4.6.

If 2.0 DIFFUSION is set low, the Rich Plates have a very clear sound that is excellent on vocals, and can be used with the early reflection parameters in rows two and three to create a wide variety of acoustic environments. If 2.0 DIFFUSION is set high, the result is a smooth, dense sound with useful applications in all types of popular music.

When 0.3 SIZE is set at around 16 meters, the sound is dense and tight--ideal for percussion. Larger sizes and longer reverb times are more suitable for vocals and brass.

	PDELAY	0.6 HC	0.7 DECAY OPT	0.8	0.9 NAME	R0 Reverb Ctrls
	RT M STOP	1.6	1.7	1.8	1.9	R1 Reverb Time
2.5		2.6	2.7	2.8	2.9	R2 Reverb Design
		3.6 R3 RFL (DB)		3.8	3.9	R3 Reflection Lvl
4.5	R2 RFL (MS)	4.6 R3 RFL (MS)	4.7	4.8	4.9	R4 Reflection Dly
5.5	MIDI PATCH	5.6 MIDI PATCH	5.7 MIDI PATCH	5.8 MIDI PATCH	5.9 MIDI PATCH	R5 MIDI Patches

Notes on the Presets

5.0 Vox Plate

Vox Plate is a bright, diffused plate program designed for vocal enhancement. There are no preechoes. 0.2 SOFT KNOB is linked to both reverb times.

5.1 PD Plate (Predelay Plate)

PD Plate is a medium size plate with greater density than 5.0 Vox Plate. It has 60 ms of predelay as well as two discrete preechoes. 0.2 SOFTKNOB is patched to both reverb times, although 1.1 RT MID is scaled twice as high as 1.0 RT LOW.

5.2 Brass Plate

Brass Plate is a heftier, denser plate.

Row 6 - Inverse Room Programs

Figure 4.7. Parameters - Inverse Room Programs.

Inverse Room is similar to the other reverb programs, but with some important differences. Inverse Room allows you to vary the slope of the initial portion of the reverb envelope. It can decay, remain level, or rise over a variable time interval. When the time interval is up, the reverberation cuts off abruptly. The resulting effect is similar to a gate, but is not at all dependent on the level or complexity of the input signal. The time interval is set in ms with 0.3 DURATION.

We mentioned slope above; there are actually two controls which affect slope. 1.0 LO SLOPE determines the shape of the reverb envelope for low frequencies. 1.1 MID SLOPE determines the shape of the envelope for mid and high frequencies. The exact frequencies affected by 1.0 LO SLOPE and 1.1 MID SLOPE are determined by 1.2 CROSSOVER.

High slope settings cause the sound to *increase* over time (until cutoff), producing what is often referred to as inverse reverb. The sound has great impact and interest, particularly when used on percussion. Used on vocals, it can give the impression of the sound being backwards.

At medium slope settings, the sound does not decay at all until it cuts off; its amplitude remains constant. this is sometimes referred to as non-decaying reverb.

Lower slope settings produce sound that decays more or less normally until cutoff. This is useful for creating gated reverb effects.

The lowest slope settings create ambience and upfront sound—body and space without the effect being too obvious.

				· - · · · - · ·	
0.5 HC	0.6	0.7	0.8	0.9 NAME	R0 Reverb Ctrls
1.5	1.6	1.7	1.8	1.9	R1 Reverb Time
2.5	2.6	2.7	2.8	2.9	R2 Reverb Design
	DB) 3.6 R3 RFL (DB)		3.8	3.9	R3 Reflection Lvl
			4.8	4.9	R4 Reflection Dly
∟5.5_MIDI PAT	CH 5.6 MIDLPATCH	5.7 MIDLPATCH	5.8 MIDI PATCH	5.9 MIDI PATCH	R5 MIDI Patches

Notes on the Presets

6.0 Inverse Room

Inverse Reverb is a good meat-and-potatoes inverse reverb sound. 0.3 DURATION is set to 385 ms. This preset has no MIDI patches.

6.1 Inverse 2

Inverse 2 is another inverse room variation.

6.2 Head Banger

Head Banger starts as a fairly radical sound. Duration is set to 562 ms with no MIDI clock present. If MIDI clock is sent to the PCM 70 the duration will decrease as the tempo increases. 0.2 SOFT KNOB and the mod wheel is patched to the two slopes, gives you the option of changing the "complexion" of the reverb tails.

6.3 Ski Jump

Ski Jump is an extreme variation on an inverse room. Mod wheel is patched to the two slope controls. they are scaled negatively.

6.4 Atom Smasher

This preset is a highly diffused preset designed for maximum impact. It works well with any source, but is ideal for sounds with a solid bottom. Use 0.3 DURATION to fit the effect to the beat.

6.5 Gated Room

A gated room without the artifacts sometimes heard when using the gate parameter in the other reverb programs.

MIDI and the PCM 70

The PCM 70 introduces innovative MIDI features and control capabilities that were previously not available in effects processors. While you may discover an application that we haven't thought of yet, most uses of MIDI will fall into one of six basic categories:

- Connection of two or more PCM 70 units for automatic linked program and register changes
- Automatic selection of a PCM 70 program or register when a program is selected on a synthesizer
- Real time control of up to ten PCM 70 parameters at a time from a remote keyboard or controller, using the PCM 70's Dynamic MIDI™
- Automatic program selection and parameter control from a MIDI sequencer
- Transfer of registers from one PCM 70 to another
- Use of a computer and patch editing software to edit PCM 70 registers via MIDI System Exclusive messages; can also store registers in computer

We'll discuss each of these applications in this chapter.

MIDI Connections and Applications

It's important to realize that the MIDI connections and applications described on the following pages are only intended as general guidelines. You will probably need to modify them to meet your personal requirements.

Linking Two PCM 70s

Two PCM 70s can be linked via MIDI for simultaneous program and register selection and loading.

- 1. Connect the MIDI OUT from one PCM 70 to the MIDI IN of another PCM 70, as shown in Figure 5.1. As with any MIDI connection, use only standard MIDI cables and keep them as short as possible to avoid data errors. 15 meters is generally accepted as the longest length that should be used.
- 2. Load 7.0 CONTROL PROGRAM on both units.
- 3. Enter the parameter mode and select parameter 1.0 MIDI CHNL on both units. Set both units to the same channel.
- 4. Select parameter 1.1 OMNI MODE. Set it to OFF.
- 5. Select parameter 1.2 PGM CHANGE. Set it to FIX.
- 6. Select and load an effects or reverb program or register on the first PCM 70. The second unit will load the same program or register at the same time.

3. Now turn it on again. While the software version number is displayed, press the 7 key. The display window will read

SEND REGISTERS

4. Press the LOAD key. The registers from the first machine will be transferred to the second, and in a few moments the display window will read

DONE

Sending Single Registers

The currently-running program or register can be sent (via MIDI System Exclusive) to another MIDI device at any time by pressing the Up key and keeping it pressed while pressing LOAD. Some MIDI sequencers can record this data, but others (like the Roland MC-500) limit the size of system exclusive messages that they can receive--these will ignore the PCM 70 dump.

Sending a Single Register to Another PCM 70

The currently running program or register can be sent (via MIDI System Exclusive) to another PCM 70 at any time. To do this, Press and hold the UP key. While holding the UP key, press LOAD. The program or register will automatically be loaded into the receiving PCM 70. The receiving PCM 70 will

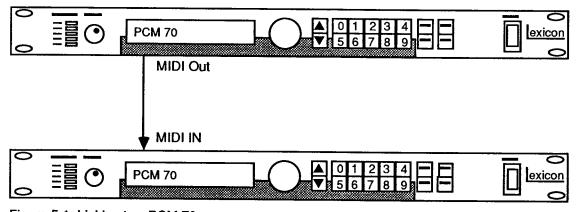


Figure 5.1. Linking two PCM 70s.

Moving Registers Between Two PCM 70s

It is possible to transfer register data between two PCM 70s, via the MIDI port, using System Exclusive messages.

- 1. Connect the two units as shown in Figure 5.1.
- 2. Turn off the unit you want to transfer register data from.

automatically go into the register mode. The display will read

5.0 NAME OF REGISTER

To store this program you must go to an unused register in register bank 1-4 and store just as you would any other register. You cannot store in register 5.0. You cannot send any Inverse Room programs into a V2.00 PCM 70; this will cause the machine to malfunction and you will have to reset the system.

Using a Keyboard to Control a PCM 70

You can select PCM 70 registers or programs, and alter up to ten parameters simultaneously using the controllers found on MIDI-equipped instruments. Nearly any MIDI-equipped keyboard or synthesizer can be used to select registers and programs. However, if you plan to make extensive use of Dynamic MIDITM, consider purchasing a keyboard (such as the Kurzweil MIDIBoard, Yamaha KX76 or KX88) which is specifically designed to function as a MIDI system controller. These keyboards have a large number of controllers, allowing you to control the PCM 70 without sacrificing control over your synthesizers and sound modules.

Remote Register and Program Selection

With parameter 1.2 PGM CHANGE set to FIX, when the PCM 70 and a keyboard are both in the OMNI mode or on the same channel, program changes made on the keyboard will cause an identical program change in the PCM 70. This is useful since it allows you to tailor specific effects for each synthesizer program. Because the PCM 70 registers change at the same time as the synthesizer programs, you are free to concentrate on your musical performance, without thinking about

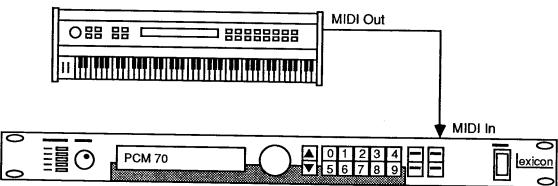


Figure 5.2. Configuration for Master Keyboard or Synthesizer and PCM 70.

Choosing a MIDI Channel

For most applications, we recommend that you use OMNI OFF, and select a specific channel for the PCM 70. Many users assign synthesizers to the lower channels, and then jump to the higher channels (14, 15 and 16) for effects like the PCM 70. If you operate with OMNI ON, be careful not to send the PCM 70 commands you don't want it to act on.

To choose a MIDI channel:

- 1. Select and Load 7.0 CONTROL PGM.
- 2. Enter the parameter mode and select parameter 1.0 MIDI CHNL. Set it to the channel you plan to use for remote control over the PCM 70. For now, just set it to channel 1, since many synthesizers only output MIDI data on channel 1.
- Select parameter 1.1 OMNI MODE. Set it to OFF.
- Select parameter 1.2 PGM CHANGE. Set it to FIX.
 We'll discuss the TABLE mode later in this chapter.

selecting programs or registers on the PCM 70.

To perform remote register selection:

- 1. Make sure you have set up the PCM 70 MIDI channel and mode as described in steps 1-4 above.
- 2. Put the PCM 70 back in the program mode by briefly pressing the PGM key.
- 3. Select some programs on the synthesizer. The PCM 70 should select and automatically load a new program each time you select a program on the synthesizer. Note that when MIDI program change messages are received, the PCM 70 automatically loads the requested program, even when AUTOLOAD is off.

You may be confused about the relationship between the program number you select on the synth and the program number that is loaded on the PCM 70. When PGM CHNG is set to FIX, program change messages received from another MIDI device will be followed literally. Program change messages from 0 to 49 will load registers 0.0 to 4.9. Program change messages 50 and up will load programs 0.0 and up. So, for

example, if you select program 23 on a synthesizer, register 2.3 will be loaded on the PCM 70. If you select program 64 on a synthesizer, program 1.4 will be loaded on the PCM 70.

Synthesizer manufacturers use different methods for numbering their presets--some (like Yamaha) number from 1 to 32, and others (like Korg) are arranged in banks and number from 1.1 to 1.8, 2.1 to 2.8, and so forth. Still others (like Oberheim) start at 00 and go to 100. Whatever numbering method is used, selecting the first program in the synthesizer sends a MIDI program change message 0, the second a MIDI program change message 1, and so on. If you experiment a bit, the relationship between your synthesizer's preset numbers and the programs and registers on the PCM 70 will become clear.

Using Corresponding Registers

You will quickly discover that a fixed relationship between synth program changes and PCM 70 program changes is not very convenient. It demands a very rigid relationship between programs in the PCM 70 and programs in your synths. Changing the program contents of one or more of your synths can require a complete rearrangement of the registers in the PCM 70.

To solve this problem the PCM 70 has a corresponding register table which lets you assign any PCM 70 program or register to any MIDI program change number. This lets several synthesizer programs share the same PCM 70 register or program, making much better use of the PCM 70's registers. Most importantly, the corresponding register table saves you the work of having to rearrange the programs in the PCM 70 every time you change the programs in your MIDI instruments.

To use the corresponding register program:

- 1. Load program 7.1 COR REG TABLE.
- 2. Enter the parameter mode.
- 3. Use the up and down and 0-9 keys to move around the table. Notice that each location in the table corresponds to a MIDI program number--0.0 is program one, 0.1 is program two, etc.
- 4. Now turn the Soft Knob. Notice that any register or preset program can be assigned to a particular MIDI program number. Also, one or more MIDI program numbers can be set to IGNORE. No action will be taken when these MIDI program change numbers are received.

Using Dynamic MIDI™

Some-extremely useful effects can be created when one or more parameters are controlled remotely in real time. Many of the controllers found on a MIDI keyboard (pitch benders, mod wheels, breath controllers, sliders, and switches) can be used to adjust PCM 70 parameters. MIDI Events like last note played, velocity, and aftertouch can also be used. Even MIDI clock is available as a patch source!

To use Dynamic MIDITM, you "patch" a parameter to a MIDI controller or event, using the PATCH parameters found in the last row of the parameter matrix. There are ten patch locations for each register, allowing you to control up to 10 parameters remotely at the same time. Because each register has its own set of patches, each register can respond to MIDI controllers in a different way.

To get an idea of what Dynamic MIDI™ can do for you, try the following example with any synthesizer:

- 1. Load program 0.2 TUNNEL.
- 2. Send some audio to the PCM 70. While listening to it, move the mod wheel. Notice that the delay times change as the mod wheel is set higher.
- 3. Enter the parameter mode and select parameter 2.1 V1 DLY. Move the mod wheel. Notice that the delay time changes as you move the mod wheel.

How does the PCM 70 do this? If you look at 5.1 PATCH of program 0.2 TUNNEL, you will notice that it patches the mod wheel to 2.1 V1 delay. Many of the programs in the PCM 70 already have MIDI patches, but you can create your own MIDI patches too.

Creating Your Own Dynamic MIDI™ Patches

- 1. With the PATCH parameter selected, rotate the Soft Knob. You will see a display of all the different MIDI controls and events that the PCM 70 will respond to.
- 2. When the controller you wish to use appears in the display window, press the LOAD key.
- 3. Now, use the Soft Knob to select the location number of the parameter you wish to remotely control. It may help to refer to the parameter tables found in Chapter 4 while doing this.

- 4. When the parameter you wish to control appears in the display window, press the LOAD key again.
- 5. Next, set the SCALING for the controller. SCAL-ING sets the range of effectiveness for a MIDI controller, and it can be set in a range from 1 to 128. With SCALING set to 128, one step of a MIDI controller is equal to one click of the Soft Knob. With SCALING at 64, two steps of a MIDI controller are equal to one click of the SOFT KNOB. Use the Soft Knob to set the SCALING you wish to use and press the LOAD key.

Note: It is possible to set negative amounts of scaling. This allows you to have the PCM 70 reduce the setting of a parameter as the controller increases.

6.The parameter you chose in step 3 now appears in the display window. Use the Soft Knob to set the parameter value to the starting point you wish to use for MIDI control. If the MIDI controller attempts to set a parameter beyond its normal operating range, < or > appears in the display next to the highest or lowest possible value.

Creating Custom Master Controls

Control over a single parameter at a time is useful, but things really begin to get exciting as you experiment with controlling several parameters simultaneously from a single MIDI controller. For example, you might patch the mod wheel on a synth to control feedback and panning at the same time in the Chorus and Echo program. In effect, you are creating a custom master control for a unique set of parameters. Using this custom master in real time can produce stunning effects never heard before. Several of the preset programs in the V3.00 software include very useful custom masters.

The ability to choose different settings of SCALING for two or more parameters controlled by the same event or controller also raises some interesting possibilities. Don't forget that using negative SCALING for one parameter and positive SCALING for another will cause the first parameter to decrease while the other increases.

A word of caution: not all parameters respond well to real-time control. Due to the current limitations of digital technology, it is simply impossible to alter certain parameters in real time without audible artifacts. This is the case whether you are controlling the parameter remotely via MIDI, or from the unit's front panel. We considered locking out these parameters, but after careful thought we included them, since what is not acceptable in one application may not be a problem in another.

Experimentation is the key to success here. The following list contains some patches that we have tried with good results. You will undoubtedly find many more.

- Aftertouch or velocity to REVERB TIME
- Any event or control to REVERB TIME
- Any event or control to FEEDBACK
- Switches to 1/24th of a beat in BPM programs change from one chord to another)
- Pitch Wheel to pitches in resonant chords
- Any event or control to MIX
- Volume to FX ADJ (in MIX = 100% applications)
- Note # to PITCH MST in resonant chords.

Some Notes On Controllers

Many MIDI synths and keyboards have a very limited number of controllers. Sometimes the pitch and modulation wheels or levers are the only options available for remotely controlling the PCM 70. However, you may not wish to produce modulation or pitch bending at the same time that you are controlling the PCM 70. All is not lost. Most synths allow you to shut off the effect of these controllers. So, for example, moving the pitch bender doesn't actually bend pitch.

This is where things get interesting. Usually, when the synth is set to ignore its controllers, controller data is still sent out the MIDI port. We have found this to be the case with a variety of different brands and models of synthesizers. Just set the synth to ignore its mod wheel and pitch bender, and then use them to control the PCM 70. As long as you don't wish to control the PCM 70 and bend pitch or add modulation at the same time, these controllers can easily do double duty. Synthesizers which memorize ranges for the mod wheels and pitch benders for each preset program are the best choices for use with the PCM 70. By using the corresponding register table, you could have some programs use the pitch bender to bend pitch, and not affect the PCM 70, and other programs could control the PCM 70, but not bend pitch.

The Yamaha DX7 II and DX5, the Oberheim Matrix 6, and the Korg DW-8000 are examples of synthesizers that can be used in this manner.

The Soft Knob as Controller

When setting up the PATCH parameters, you may have noticed that one of the options listed as potential MIDI controllers is the PCM 70's own Soft Knob. The Soft Knob isn't a true MIDI controller, but it was included to allow non-MIDI users to create their own custom MASTER controls. Many of the presets in software version 3.00 have Patches assigned to the Soft Knob.

Everything we have said about using MIDI controllers with the PCM 70 applies equally to the Soft Knob.

Because it is not a true MIDI controller, changes made with the Soft Knob cannot be stored and repeated with a sequencer.

When the Soft Knob is specified as a controller with the PATCH parameter, you must select parameter 0.2 SOFT KNOB before it can be used.

Controlling a PCM 70 from a Sequencer

Since you can control the PCM 70 in real time with MIDI controllers, it stands to reason that you could record your manipulation of those controllers with a MIDI sequencer, and then repeat the performance automatically. In fact, this works perfectly, and this capability gives the PCM 70 a fairly sophisticated level of automation. If your sequencer can sync to tape, you can even use it to provide automated effects for non-MIDI instruments. For example, you can control the PCM 70 from a keyboard, recording commands onto a sequencer, but the audio the PCM 70 processes might be percussion, guitar, vocals, or even the whole mix. If you perform live with sequencers, there is no reason why you can't sequence several PCM 70s along with everything else.

When working with sequencers, it is always a good idea to put the PCM 70 on a different MIDI channel than other devices in the system. This avoids the possibility of the PCM 70 responding to commands that aren't really intended for it.

Keep in mind that as of this writing, no sequencers offer "chase" mode for MIDI controllers. This means that if you attempt to punch in to the middle of a sequence, the PCM 70's parameters will be in an unknown condition. To avoid problems, always start the sequence at the very beginning when overdubbing or adding new parts. Also, it is a good idea to use the first measure of the sequence to reset all the controllers to 0.

If a series of MIDI commands leaves the PCM 70 in an undesirable, unknown state, use the following procedure to reset all controllers to 0:

- 1. Load the Control Program 7.0.
- 2. Go to location 1.3 RESET MIDI.
- 3. Press LOAD.

The PCM 70 also will respond to a MIDI reset command.

Register Editing and Storage with a Microcomputer

It is possible to use a microcomputer to edit PCM 70 registers and preset programs, and also to store sets of registers. Several software houses are offering software compatible with the PCM 70. Because software is such a fast-changing market, we will not list any specific manufacturers here. Check the pages of the popular keyboard and music magazines for more information.

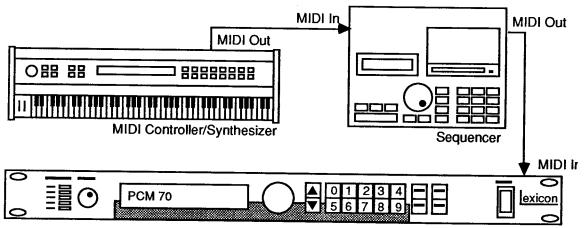


Figure 5.3. Connections for Keyboard, Sequencer, and PCM 70.

Recording MIDI Program Changes from the PCM 70

Because the PCM 70 sends MIDI program change messages as well as receiving them, it is possible to configure your system for sequenced preset program changes without using another MIDI controller, as shown in Figure 5.4.

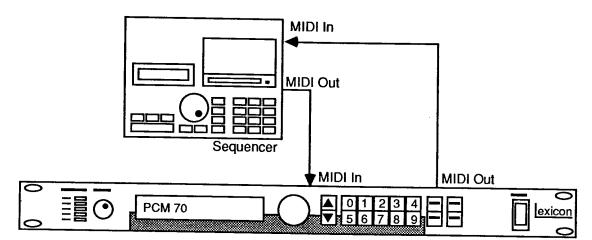


Figure 5.4. Connections for Sequencer and PCM 70.



MIDI Implementation Data

This chapter contains the MIDI implementation chart and information about the PCM 70's System Exclusive message protocols

PCM 70 MIDI Implementation Chart

Function	. Transmitted	Recognized	Remarks	
Basic Channel	Default Changed	1-16 1-16	1-16 1-16	Saved in non volatile memory
Mode	Default Messages Altered	X X *******	Mode 1,3 OMNI ON/OFF	Saved in non volatile memory
Note Number	True Voice	X ********	0-127 x	See note 1
Velocity	Note On Note Off	x x	o x	See note 1
After Touch	Key's Ch's	x x	x o	See note 1
Pitch Bende	r	х	0	Default value = 40h
Control Change		x x	00-31 64-95	See note 1 See note 1
Prog Change	True #	0-119 see note 3	0-127 See note 2	See note 2
System Exclusive		o (bulk data only)	0	
System Common	Song Pos Song Sel Tune	x x x	x x x	
System Real Time	Clock Commands	x x	0	See note 1 Start only
Aux Mes sages	Local On/Off All Notes Off Active Sense Reset	x x x x	x x x o	Resets MIDI patch sources to default values

Notes

Note 1. Used as patch source, power-on default value = 0

Note 2. Two Modes:

Fixed: 00 - 49 = Registers 0.0 + 4.9

50 - 119 = Programs 0.0 6.9

Table: 00 -127 = Programmed to correspond to

any program or register

Note 3. Program changes are always transmitted in Fixed format

Mode 1: OMNI ON, POLY

Mode 2 : OMNI ON, MONO

o:yes

Mode 3: OMNI OFF, POLY

Mode 4: OMNI OFF, MONO

x:no

1. Transmitted Data

Status	Second	Third	Description
1100nnn	0ppp pppp		Program Change
			pppppppp = 0 - 49 = Register 0.0 - 4.9 pppppppp = 50 - 115 = Program 0.0 -6.5

2. Recognized Receive Data

Status	Second	Third	Description
1001 nnnn	Okkk kkkk	0vvv vvvv	NOTE ON kkkkkkk = 0 - 127* Velocity vvvvvvv = 1 - 127*
1011 nnnn	Occc cccc	0vvv vvvv	cccccc = 0-31, continuous controller cccccc = 64-95, switch controllers
			0vvvvvv = 0 - 127*
1100 nnnn	Оррррррр		Program Change Fixed Mode: ppppppp = 0 - 49 = Register 0.0 - 4.9 ppppppp = 50 - 119 = Program 0.0 - 6.9 pppppppp = 120 - 117 = ignored
			Table Mode: ppppppp = 0-127 TBL (ppppppp) = any Program or Register
1101 nnnn	0vvvvvv		Channel Aftertouch vvvvvvv = 0 - 127*
1110 nnnn	00000000	0vvvvvv	Pitch Bend vvvvvvv = 0 - 127* (default = 64)
1011 nnnn	01111100	00000000	OMNI OFF
1011 nnnn	01111101	00000000	OMNI ON
1111 1111			RESET - resets MIDI patch source values to default (power-up) condition

^{*}Used as a MIDI patch source value

System Exclusive Message Protocols

I) Receive

A.	Active/Stored	Bulk	Data	(Single	Register))
----	---------------	------	------	---------	-----------	---

F0 (hex) System Exclusive ID 06 Lexicon ID 00 **PCM 70 ID** (bin) oooc nnnn c = 0 for active data (running program) c = 1 for stored data (register) n = MIDI channel 0-15 (0 = channel 1, etc.) oopp pppp p = register # = 0-49 or 50 for the active data buffer oooo oobb (02)b = byte count of all data bytes = 334 obbb bbbb (most significant bits sent first) (4E) oooo dddd (hi) d = data 167 bytes, sent in hi nybble/lo nybble pairs oooo dddd (low) for a total of 334 bytes (see tables 2-9 for description) osss ssss sumcheck of nybblized data bytes (high bit = 0) (hex) F7 End of Exclusive Message

Important Note: The PCM 70 requires 30 ms. between stored bulk data messages.

System Exclusive ID

B. Parameter Data

F0

(hex)

	06	Lexicon ID
	00	PCM 70 ID
(bin)	oo1o nnnn	n = MIDI channel 0 - 15 (0 = channel 1)
	oggg gggg	g = parameter # 0 - 89 (see table 1)
	0000 0۷۷۷	v = parameter value - 10 bits; most significant bits
	ovvv vvvv	sent first (see tables for parameter limits)
(hex)	F7	End of Exclusive Message
Request	ts	
(hex)	F0	System Exclusive ID
	06	Lexicon ID
	00	PCM 70 ID
(bin)	oo11 nnnn	n = MIDI channel 0 - 15 (0 = channel 1)
	oeee eeee	e = event code: (hex) 60 = send active bulk data
		(hex) 61 = send stored bulk data
	оррр рррр	p = register # = 0 - 49 or 50 for active data
(hex)	F7	End of Exclusive Message

II) Transmitted

C.

A. Active/Stored Bulk Data (single register)

Sent on receipt of system exclusive request or via front panel "Send Registers"

diagnostic.

B. Parameter Changes

Not sent from the PCM 70 via MIDI.

TABLE 1: Parameter Numbering for System Exclusive Messages

Parameter #	Parameter
0 - 59	Number corresponds to (10* row) + column
60 - 69	Patch Sources (10)
70 - 79	Patch Destinations (10) (by parameter #)
80 - 89	Patch Scale Factors (10)
90 - 103	Name (14)

TABLE 2: General Bulk Data Format

Byte #	Data			
Program Type (see Table 3) Stored Parameter Matrix Position: Row Stored Parameter Matrix Position: Column 3 - 16 Name (14 characters; last character must be 0) 17 - 26 MIDI Patch Sources 27 - 36 MIDI Patch Destinations 37 - 46 MIDI Scale Factors 47 - 167 60 2-byte words, different for each program type. Se				
	Byte 0 dddd dddd Low 8 bits 1 oooo oodd High 2 bits			

TABLE 3: Program Types

0	Unused
1-3	Not Available
4	Chorus and Echo
5	Multiband
6	Resonant Chords
7	Concert Hall
8	Rich Chamber
9	Rich Plate
10	Infinite Reverb
11	Multiband Rhythm
12	Chorus and Rhythm
13	Rhythmic Chords
14	Inverse Room

TABLE 4: Chorus and Echo; Chorus and Echo BPM

Row#	Byte #	Description	Param #	Min.	Max	Display
0	47,48	MIX	0	462	562	0% - 100%
	49,50	FX ADJ	1	461	563	-90 - +12dB
	51,52	SOFT KNOB	2	448	575	0 - 127
	53,54	CHORUSING	3	462	561	0 - 99
	59,60	CHORUS	4	506	518	OFF, I VCS - 6 VCT
	57,58	HC	5	496	527	0Hz - 15.0KHz
	55,56	DIFFUSION	6	462	561	0 - 99
	61,62	(Not Available)				
	63,64	11 11				
	65,66	"				
1	67,68	LVL MST	10	477	547	-35 - +35
	69,70	V1 LVL	11	495	530	OFF - FULL
	71,72	V2 LVL	12	"	**	" "
	73,74	V3 LVL	13	**	"	8P 99
	75,76	V4 LVL	14	"	**	** **
	77,78	V5 LVL	15	**		P9 P7
	79,80	V6 LVL	16	**	**	11 11
2	81,82	DLY MST	20	260*	764*	-252 - +252ms
	83,84	V1 DLY*	21	386*	638*	0 - 432ms
	85,86	V2 DLY	22	"	**	17 91
	87,88	V3 DLY	23	**	"	** **
	89,90	V4 DLY	24	**	"	11 91
	91,92	V5 DLY	25	**	**	11 11
	93,94	V6 DLY	26	"	"	11 11
3	95,96	FDBK MST	30	318	706	-194 - +194
	97,98	V1 FDBK	31	415	609	-97% - +97%
	99,100	V2 FDBK	32	**	••	19 19
	101,102	V3 FDBK	33	••	"	** **
	103,104	V4 FDBK	34	**	11	** **
	105,106	V5 FDBK	35	••	**	
	107,108	V6 FDBK	36	••	1)	11 11
4	109,110	PAN MST	40	412	612	-100 - +100
	111,112	V1 PAN	41	462	562	50L - 50R
	113,114	V2 PAN	42	**	**	" "
	115,116	V3 PAN	43	**	••	** **
	117,118	V4 PAN	44	••	H	11 11
	119,120	V5 PAN	45	11	**	17 11
	121,122	V6 PAN	46	**	**	
	123-167	(Not Available)				

^{*}Limits for BPM version: master parameter: 448-575; voice parameters: 500-524

TABLE 5: Multiband Delay; Multiband Delay BPM

Row#	Byte #	Description	Param #	Min.	Max	Display
0	47,48	MIX	0	460	F00	- •
Ū	49,50	FX ADJ		462	562	0% - 100%
	51,52	SOFT KNOB	1	461	563	-90 - +12dB
			2	448	575	0 - 127
	53,54	DIFFUSION	3	462	561	0 - 99
	55,56	V1 FDBK	4	***	**	11 11
	57,58	V2 FDBK	5	"	**	11 11
	59,60	(Not Available)				
	61,62	17 11				
	63,64	11 11				
	65,66	11 11				•
1	67,68	LVL MST	10	477	547	-35 - +35
	69,70	V1 LVL	11	495	530	OFF - FULL
	71,72	V2 LVL	12	"	"	" "
	73,74	V3 LVL	13	**		67 68
	75,76	V4 LVL	14	••	**	11 11
	77,78	V5 LVL	15	**	**	H H
	79,80	V6 LVL	16	"	11	11 11
	75,00	VOLVE	10			
2	81,82	DLY MST*	20	292	732	-315 - +315
	83,84	V1 DLY*	21	354	669	0 - 936ms
	85,86	V2 DLY	22	11	,,	" "
	87,88	V3 DLY	23	11	**	11 11
	89,90	V4 DLY	24	11	**	11 H
	91,92	V5 DLY	25	11	71	11 11
	93,94	V6 DLY	26	**	**	11 11
3	95,96	LC MST	30	481	543	-30 - +30
	97,98	V1 LC	31	496	527	0Hz - 13.6KHz
	99,100	V2 LC	32	"	"	# #
	101,102	V3 LC	33	**	**	11 11
	103,104	V4 LC	34	••	**	. 17 17
	105,106	V5 LC	35	,,	**	
	107,108	V6 LC		**	 H	n n
	107,100	VOLC	36		"	
4	109,110	HC MST	40	481	543	-30 - +30°
	111,112	V1 HC	41	496	527	170Hz - 15.0KHz
	113,114	V2 HC	42	**	,	" "
	115,116	V3 HC	43	50	••	10 10
	117,118	V4 HC	44	**	**	** **
	119,120	V5 HC	45	,,	**	11 /1
	121,122	V6 HC	46	**	**	11 11
5	123,124	PAN MST	50	412	612	-100 - +100
	125,126	V1 PAN	51	462	562	50L - 50R
	127,128	V2 PAN	52	"	302	30L - 30H
	129,130	V3 PAN	53	**	••	H H
	131,132	V4 PAN	53 54	,,	11	
	133,134	V5 PAN		,,	"	
	135,134	V6 PAN	55 50	"	"	** **
	•		56	••	••	11 11
	137-166	(Not Available)				

TABLE 6: Resonant Chords

Row#	Byte #	Description	Param #	Min.	Max	Display
0	47,48	MIX	0			
Ü			0	462	562	0% - 100%
	49,50	FX ADJ	1	461	563	-90 - +12dB
	51,52	SOFT KNOB	2	448	575	0 - 127
	53,54	V3 FDBK	3	462	561	099
	55,56	V6 FDBK	4	462	561	0 - 99
	57,58	HFC LEFT	5	496	527	170Hz - 15.0KHz
	59,60	HFC RIGHT	6	496	527	170Hz - 15.0KHz
	61,62	(Not Available)	Ū	400	327	170112 - 13.01(12
	63,64	" "				
	65,66	11 11				
	05,00					
1	67,68	LVL MST	10	477	547	-35 - +35
	69,70	V1 LVL	11	495	530	OFF - FULL
	71,72	V2 LVL	12		**	** 11
	73,74	V3 LVL	13	••	11	11 11
	75,76	V4 LVL	14	**	**	0 B
	77,78	V5 LVL	15	11	**	H H
	79,80	V6 LVL	16	**	,,	11 17
	70,00	1021 2	10			
2	81,82	PCH MST	20	438	586	-74 - +74
	83,84	V1 PITCH	21	475	549	Db1 - Eb7
	85,86	V2 PITCH	22	"	"	# "
	87,88	V3 PITCH	23	••	**	11 11
	89,90	V4 PITCH	24	11	**	
	91,92	V5 PITCH	25	11		H II
	93,94	V6 PITCH	25 26	71	"	Pt 11
	00,0 .	10111011	20			
3	95,96	RESN MST	30	318	706	-194 - +194
	97,98	V1 RESN	31	415	609	-97% - +97%
	99,100	V2 RESN	32	"	"	11 11
	101,102	V3 RESN	33	**	**	** **
	103,104	V4 RESN	34	••	.,	
	105,106	V5 RESN	35	**	**	
	107,108	V6 RESN		**	.,	
	107,100	VO NEON	36		**	17 19
4	109,110	PDL MST*	40	308	716	-204 - +204
	111,112	V1 PDL*	41	410	614	0 - 773ms
	113,114	V2 PDL	42	**	11	11 11
	115,116	V3 PDL	43	**	••	H H
	117,118	V4 PDL	44	**	••	11 11
	119,120	V5 PDL	45		**	11 11
	121,122	V6 PDL	46	**	**	** **
5	123,124	PAN MST	50	410	040	100 100
_	125,126	V1 PAN	51	412	612	-100 - +100
	127,128	V11 AN V2 PAN		462 "	562	50R - 50L
			52 50		"	17 11
	129,130	V3 PAN	53		**	11 11
	131,132	V4 PAN	54	••	•••	11 11
	133,134	V5 PAN	55	••	**	11 11
	135,136	V6 PAN	56	11	"	" "
	137-166	(Not Available)				

^{*}Limits for BPM version are the same as for Chorus and Echo 6-8

TABLE 7: Concert Hall

Row#	Byte #	Description	Param #	Min.	Max	Display
0	47,48	MIX	0	462	562	0% - 100%
	49,50	FX ADJ	1	461	563	-80 - +12dB
	51,52	SOFT KNOB	2	448	575	0 - 127
	53,54	SIZE (49)	3	488	537	3.5 - 38.3m
	55,56	GATE TIME	. 4	384	639	0.5 - 36.311 0ms - 4.57s/OFF
	57,58	PDELAY	5	385	638	0 - 506ms**
	59,60	HC	6	497	527	170Hz - 15.0KHz
	61,62	CHORUSING	8	462	561	0 - 99
	63,64	DCY OPT	7	512	513	OFF - ON
	65,66	11 11		0.2	310	OIT - OIN
1	67,68	RT LOW	10	496	527	depends on size
	69,70	RT MID	11	**	"	depends on size
	71,72	XOVER	12	497	**	170Hz - 15.0 KHz
	73,74	RT HC	13	496	"	OHz - 15.0 KHz
	75,76	RTL STOP	14		**	depends on size
	77,78	RTM STOP	15	**	**	depends on size
2	79,80	DIFFUSION	20	462	561	0 - 99
	81,82	ATTACK	21	**	**	11 11
	83,84	DEFINITION	22	11	**	11 11
3	85,86	REFL LVL MSTR	30	472	547	-35 - +35
	87,88	REFL L1	31	495	530	OFF- FULL
	89,90	REFL L2	32	II .	**	11 11
	91,92	REFL R1	33	9.8	••	н н
	93,94	REFL R2	34	**	"	11 11
4	99,100	REFL DLY MSTR	40	206	040	000 000
	101,102	REFL L1	41	400	818 624	-306 - +306
	103,104	REFL L2	42	400	624 "	0 - 864 ms**
	105,106	REFL L3	43	••		". " H H
	107,108	REFL R1	44	,,	**	41 91
	109,110	REFL R2	45	••		" "
	111,112 113-167	REFL R3 (Not Available)	46		 H	
		(and the state of				

^{**}Limits depend on size

PCM 70 Owner's Manual - Software Version 3.00

TABLE 8: Rich Chamber and Rich Plate

Row#	Byte #	Description	Param #	Min.	Max	Display
0	47,48	MIX	0	462	562	0% - 100%
	49,50	FX ADJ	1	461	563	-80 - +12dB
	51,52	SOFT KNOB	2	448	575	0 - 127
chamber		SIZE (38)	3	493	531	5.6 - 32.6m
plate	53,54	SIZE (41)	3	491	532	5.6 - 34.7m
	55,56	GATE TIME	4	384	639	0ms - 4.57s/OFF
	57,58	PDELAY	5	385	638	0 - 506ms**
	59,60	HC	6	497	527	170Hz - 15.0KHz
	61,62	DCY OPT	7	512	513	OFF - ON
	63,64	(Not Available)				
	65,66	W H				
1	67,68	RT LOW	10	496	527	depends on size
	69,70	RT MID	11	**	**	depends on size
	71,72	XOVER	12	497	11	170Hz - 15.0 KHz
	73,74	RT HC	13	496	**	OHz - 15.0 KHz
	75,76	RTL STOP	14	**		depends on size
	77,78	RTM STOP	15	**	**	depends on size
2	79,80	DIFFUSION	20	462	561	0 - 99
	81,82	ATTACK	21	**	**	11 11
	83,84	DEFINITION	22	••	11	19 99
3	85,86	REFL LVL MSTR	30	472	547	-35 - +35
	87,88	REFL L1	31	495	530	OFF - FULL
	89,90	REFL L2	32	••	••	11 11
	91,92	REFL L3	33	••	**	11 11
	93,94	REFL R1	34	**	71	и и
	95,96	REFL R2	35		**	** **
	97,98	REFL R3	36	**	••	tr tr
	99,100	REFL DLY MSTR	40	206	818	-306 - +306
	101,102	REFL L1	41	400	624	0 - 776 ms**
	103,104	REFL L2	42	**	**	(728 for Chamber)
	105,106	REFL L3	43	**	**	н н
	107,108	REFL R1	44	•	11	
	109,110	REFL R2	45	••		91 19
	111,112	REFL R3	46			" "
	113-167	(Not Available)	70			-

^{**}Limits depend on size

TABLE 9: Infinite Reverb

Row#	Byte #	Description	Param #	Min.	Max	Display
0	47,48 49,50 51,52 53,54 55,56 57,58 59,60 61,62 63,64	MIX FX ADJ SOFT KNOB SIZE (38) REV TIME PDELAY HC (Not Available)	0 1 2 3 4 5 6	462 461 448 493 496 385 497	562 563 575 531 528 638 527	0% - 100% -80 - +12dB 0 - 127 5.6 - 32.6m .04 - INF* 0 - 506** 170Hz - 15.0KHz
1	65,66 67,68 69,70 71,72	DIFFUSION ATTACK DEFINITION	10 11 12	462 "	561 "	0 - 99 " "- " "
2	73,74 75,76 77,78 79,80 81,82 83,84 85,86	REFL LVL MSTR REFL L1 REFL L2 REFL L3 REFL R1 REFL R2 REFL R3	20 21 22 23 24 25 26	472 495 " " "	547 530 "	-35 - +35 OFF- FULL " " " " " "
3	87,88 89,90 91,92 93,94 95,96 97,98 99,100 101-167	REFL DLY MSTR REFL L1 REFL L2 REFL L3 REFL R1 REFL R2 REFL R3 (Not Available)	30 31 32 33 34 35 36	206 400 " " "	818 624 "	-306 - +306 0 - 506ms** " " " "

^{*}Display depends on size
**Limits depend on size

TABLE 9A: Inverse Room

Row#	Byte #	Description	Param #	Min.	Max	Display
0	47,48	MIX	0	462	562	0% - 100%
	49,50	FX ADJ	1	461	563	-80 - +12dB
	51,52	SOFT KNOB	2	448	575	0 - 127
	53,54	DURATION	3	493	531	102 - 600 (ms)
	57,58	PDELAY	4	385	638	0 - 506ms**
	59,60	HC	5	497	527	170Hz - 15.0KHz
	63,64 65,66	(Not Available)				
1	67,68	LO SLOPE	10	496	527	-64% - 60%
	69,70	MD SLOPE	11	"	**	-64% - 60%
	71,72	XOVER	12	**	**	OHz - 15.0 KHz
	73,74	RT HC	13	"	. 11	OHz - 15.0 KHz
	75,76	(Not Available)				
2	79,80	DIFFUSION	20	462	561	0 - 99
	81,82	ATTACK	21	**	**	11 11
	83,84	DEFINITION	22	**	10	11 11
3	85,86	REFL LVL MSTR	30	472	547	-35 - +35
	87,88	REFL L1	31	495	530	OFF - FULL
	89,90	REFL L2	32	••	••	11 11
	91,92	REFL L3	33	11	**	11 11
	93,94	REFL R1	34	***	**	" "
	95,96	REFL R2	35	**	υ	" "
	97,98	REFL R3	36	**	**	19 - 10
4	99,100	REFL DLY MSTR	40	206	818	-306 - +306
	101,102	REFL L1	41	400	624	0 - 728ms**
	103,104	REFL L2	42	**	**	11 11
	105,106	REFL L3	43	••	**	11 11
	107,108	REFL R1	44	**	**	11 11
	109,110	REFL R2	45	· H	11	11 11
	111,112	REFL R3	46	.01	**	f1 f4
	113-167	(Not Available)				

^{**}Limits depend on duration

TABLE 10: Reverberation Minimum Sizes and Time Constants

Program Type	Min. Size	Time Constant
Rich Chamber	8	388
Rich Plate	8	471
Concert Hali	5	444
Infinite Reverb	8	388

Relating Stored Parameter Values to Numbers Displayed

1. Size Display

Program	Min. Size	Time Const.	Size Const.	Size Base
Concert Hall	5	444	164	31362
Rich Chamber	8	388	511	30151
Rich Plate	8	471	424	30892
Inf. Reverb8	388	511	30151	

Calculating displayed size:

```
displayed size = (size value + minimum size)*71/100
where minimum size is the value found in the table above, and size value is found as follows:
size value = "size" parameter value - "size" parameter low limit
```

Param. values and low limits are found in tables 4 - 9. The first three significant figures of the final result are displayed.

2. Reverb Time Display

Calculate size value as above. Then define size factor, time factor and time value by the following, each rounded to the nearest integer:

```
size factor = [(size value * 10 )/minimum size] + 10 (see table 10)
time factor = (size factor * time constant)/1000 (see table 10)
time value = "time" parameter value - "time" parameter low limit
```

(See table 10 for time constants and tables 4 - 9 for parameter values and low limits) Then displayed time in seconds (rounded to the nearest integer) is given by:

displayed time = (time factor * table11[time value])/500

3. Level Parameter Display

Define level value analogously to size value and time value. Then

displayed level = table12[level value]

4. Reverb Predelay Display

```
displayed predelay = value * 2 (milliseconds) where value is defined analogously to level value in item 3.
```

5. Interaction Between Displayed Size & Predelay Parameters

The Size parameter affects the "max" limit for predelay as stated in tables 7 and 8. This does not change the internal parameter value itself, so that if the "max" falls below the internal value due to a Size change, the display informs that by stating that the internal value is now out of range. Size affects the "max" limit as follows:

Let size base, size factor and time constant be as given in Item 1. Then define samples by:

```
samples = size base - (size factor)*(size constant) (samples is limited from below by 0)

Now define range by:
range = (samples/68) -1
```

The value of range is limited from above by 254. Its effect on the "max" value is given by:
"max" value = "min" value + range - 1

6. Interaction Between Displayed Size & Refl. Delay Parameters

Size also affects the "max" values of reflection delay parameters. Let samples be defined as in Item 5. Then let x be the index of the first entry in table 14 which is at least as big as samples. Then "max" value is given by:

PCM 70 Owner's Manual - Software Version 3.00

7. Gate Time Display

displayed gate time = value * 18 (milliseconds) where value is defined analogously to level value in item 3.

8. Frequency Display

displayed frequency = table13[value] where value is defined analogously to level value in item 3.

9. Delay Time Display (Not for Resonant Chords)

Parameter Value Displayed Delay Time

0 - 97	see below*
98 - 147	value - 48
148 - 197	(value - 98) * 2
198 - 247	(value - 148) * 4
248 - up	(value - 198) * 8
*get samples = table14[value];	then calculate:

displayed delay time = (samples * 148)/5000 (the first three significant figures of the result are displayed)

10. Chorus Display

Parameter Value Displayed Chorus Type

0	OFF
1 - 6	1VC S - 6VC S
7 - 12	1VC T - 6VC T

11. Patch Source Display

Parameter Value Displayed Patch Source

1 - 32 33 - 64 Switches 64 - 95 65 Pitch Wheel 66 After Touch 67 Last Note 68 Last Velocity 69 Soft Knob 70 Midi Clock (not implemented in version 2.0)	0	OFF
65 Pitch Wheel 66 After Touch 67 Last Note 68 Last Velocity 69 Soft Knob	1 - 32	Controls 0 - 31
66 After Touch 67 Last Note 68 Last Velocity 69 Soft Knob	33 - 64	Switches 64 - 95
67 Last Note 68 Last Velocity 69 Soft Knob	65	Pitch Wheel
68 Last Velocity 69 Soft Knob	66	After Touch
69 Soft Knob	67	Last Note
69 Soft Knob	68	Last Velocity
	69	
	70	

12. Patch Destination Display

displayed value = (row * 10) + column

13. Patch Scale Factor Display

Parameter Value Displayed Patch Scale Factor

0-12/	-1281
128-255	+1 - +128

14. Examples

Program Concert Hall	Min. Size 5	Time Const 444	Size Const 164	Size Base 31362	Sam- ples 29722	Range 254	Size Factor 10	Size Value 0	Pdly H. Lim* 506	Dly Tbi Index 308	Ref. H. Lim* 864
	5	444	164	31362	28738	254	16	3	506	305	840
	5	444	164	31362	17258	252	86	38	502	262	496
	5	444	164	31362	13650	199	108	49	396	249	396
Rich Chamber	8	388	511	30151	25041	254	10	0	506	291	728
	8	388	511	30151	513	6	58	38	10	52	14.5
Rich Plate	8	471	424	30892	26652	254	10	0	506	297	776
	8	471	424	30892	5028	72	61	41	142	173	146

^{*}Actual displayed parameter value

TABLE 11: Reverberation Times

5	6	8	9	11	12	13	14	15	17
18	20	22	24	26	28	30	34	38	42
46	52	57	65	75	85	100	120	160	220
350	700						0	100	LLU

Note: table11[value] is found by starting at the top left and counting out, row by row "value" entries. Thus, the row and column of the desired entry will be such that value = (row * 10) + column.

TABLE 12: Levels

	OFF	-30	-27	-24	-22	-21	-19	-18	-17	-16	
	-15	-14	-13	-12	-11	-10	-9.5	-9.0	-8.5	-8.0	-75
-7.0	-6.5	-6.0	-5.5	-5.0	-4.5	-4.0	-3.5	-3.0	-2.5	-2.0	-1.5
-1.0	-0.5	FULL								2.0	1.0

Note: to find table12[value] see table 11 above.

TABLE 13: Frequencies

0	170	350	530	720	920	1120	1330	1550	1780
2020	2270						4080		4850
5280	5750						10000		12300
13600	15000								

Note: to find table13[value] see table 11 above. Results are in Hertz.

TABLE 14: Sample Table for Delay Times

0	1	2	4	6	8	10	13	16	20
24	28	33	38	43	49	55	61	67	74
81	89	97	105	114	123	132	142	152	162
173	184	195	207	219	231	244	257	270	284
298	312	327	342	358	374	390	406	423	440
457	475	493	512	531	550	569	589	609	629
650	671	693	715	737	759	782	805	829	853
877	902	927	952	977	1003	1029	1056	1083	1110
1138	1166	1194	1222	1251	1280	1310	1340	1370	1401
1432	1463	1495	1527	1559	1592	1625	1659	1692	1726
1760	1794	1827	1861	1895	1929	1963	1997	2031	2064
2098	2132	2166	2200	2234	2268	2302	2336	2370	2404
2437	2471	2505	2539	2573	2607	2641	2674	2708	2742
2776	2810	2844	2877	2911	2945	2979	3013	3047	3081
3114	3148	3182	3216	3250	3284	3318	3351	3385	3453
3521	3588	3656	3724	3792	3859	3927	3995	4062	4096
4198	4266	4333	4401	4469	4536	4604	4672	4739	4807
4875	4943	5010	5078	5146	5213	5281	5349	5417	5484
5552	5620	5687	5755	5823	5890	5958	6026	6094	6161
6229	6297	6364	6432	6500	6568	6635	6703	6771	6906
7042	7177	7312	7448	7583	7719	7854	7989	8125	8260
8396	8531	8667	8802	8937	9073	9208	9344	9479	9614
9750	9885	10021	10156	10292	10427	10562	10698	10833	10969
11104	11239	11375	11510	11646	11781	11917	12052	12187	12323
12458	12594	12729	12864	13000	13135	13271	13406	13542	13812
14083	14354	14625	14896	15167	15437	15708	15979	16250	16521
16792	17062	17333	17604	17875	18146	18417	18687	18958	19229
19500	19770	20041	20312	20583	20854	21125	21396	21666	21937
22208	22479	22750	23021	23291	23562	23833	24104	24375	24646
24916	25187	25458	25729	26000	26271	26541	26812	27083	27354
27625	27896	28166	28437	28708	28979	29250	29521	29792	30062
30333	30604	30874	31145	31416	31687				

Solving Problems

Read this chapter for basic troubleshooting information.

PCM 70 occasionally mutes audio and displays POWER LOW

Your power line voltage is too low for the PCM 70 to function reliably. It monitors line voltage and automatically shuts down when the voltage level falls 20% below the nominal line voltage that the unit is set for. Solution? You may have too many devices plugged into a single circuit--drawing too much current and pulling down the line voltage. Try turning off several devices and see if the problem goes away. If it does, the circuit is probably overloaded. Either permanently reduce the load on the circuit or have an electrician install a heftier circuit. If you must temporarily continue operations with the overloaded circuit, try plugging the PCM 70 into a VARIAC. This should only be done as a last resort in an emergency, since an overloaded electrical circuit represents a very real safety hazard to both you and your equipment.

PCM 70 seems confused--gibberish appears in the display window

The PCM 70 has extensive protection against power line noise, spikes, glitches, and hash, but like any computer-based device, its memory can become scrambled by power line noise. If you think that the registers might still be intact, try clearing memory without erasing the registers. To do this:

Turn off the power switch for a few moments, and then turn it back on. When the software version number appears on the display, press the 8 key.

After a few moments you will be asked

CLR ALL BUT REGS

Press the LOAD key. The display flashes

CLEARING

If this fails to clear up the problem, you will have to clear the entire memory, including registers. All user registers will be erased. However, this procedure will not erase the factory preset programs.

To completely reset the PCM 70, turn off the power switch for a few moments, and then turn it back on. When the software version number appears on the display, press the 0 key.

After a few moments you will be asked

CLEAR MEMORY?

Press the LOAD key. You will be asked

ARE YOU SURE?

Press the 1 key. The PCM 70 will briefly flash

CLEARING MEMORY

and then load program 0.0. All registers will be erased and all control parameters will be returned to their factory preset settings.

When I plug the PCM 70 into my guitar amp, it sounds like I'm playing in a cavern

Check parameter 0.0 MIX on the programs and registers that you use. If it is set to 100% WET, reduce it to somewhere between 40 and 75%. 100% WET is primarily intended for use with consoles and amplifiers with an effects loop.

When connected to the effects loop on my console, turning up the console's effects send and returns just makes the dry signal louder

Check parameter 0.0 MIX on all the programs and registers you use. It should be set to 100% wet for proper operation in an effects loop.

PCM 70 displays "BATTERY LOW" when I turn it on

The PCM 70 has a lithium battery which maintains data memory (parameter settings, user registers, etc.) even when the unit is disconnected from power mains. The battery has a service life of up to five years.

When the battery is nearly exhausted, a "BATTERY LOW" message will appear in the display window each time you turn on the PCM 70. Press the LOAD key to remove the message and continue on. Contact Lexicon or a Lexicon dealer as soon as possible to have the battery replaced.

PCM 70 won't allow me to overwrite registers

The register protection option is on. Read the section entitled *Register Protection* in Chapter 2 to learn how to turn it off.



Specifications

PCM 70 Owner's Manual - Software Version 3.00

Specifications

The following specifications are subject to change without notice.

Frequency Response

Processed Signal Direct Signal

20 Hz to 15 kHz, ±1 dB 20 Hz to 20 kHz, ±0.25 dB

Dynamic Range

Processed Signal

80 dB, 20 Hz to 20 kHz noise bandwidth

Total Harmonic Distortion and Noise

Processed Signal Direct Signal

<0.05% @ 1 kHz and full level <0.025% @ 1 kHz @ 3V out

Audio Input

Levels

+4 dB; -8 to +18 dBV balanced -20 dB; -23 to +3 dBV unbalanced

Impedance

+4 dB; 40 kilohms, paralleled with 150 pF (balanced)

-20 dB; >500 kilohms, paralleled with 150 pF (unbalanced)

Connector

1/4" tip/ring/sleeve phone jack

Audio Outputs (Two)

Levels

+4 dB; +10 dBV into 600 ohms

+16 dBV into >10 kilohms -20 dB; -8 dBV into >10 kilohms

Impedance

600 ohm unbalanced

Connectors

1/4" tip/sleeve phone jack

Remote Bypass

1/4" tip/sleeve phone jack for latching footswitch (Lexicon A-FS-41)

Remote Register Select

1/4" tip/sleeve phone jack for momentary contact footswitch (Lexicon 750-02834)

Displays

FIP

16 digit, 14 segment alphanumeric fluorescent display

LEDs

5 segment headroom indicator with 24 dB range

Bypass, Program, Register, and Load buttonindicators

Power

Nominal

100, 120, 220, 240 Vac (-10%, +5%)

Switch-Selectable; 50-60 Hz, 25 W maximum

RFI Shielding

Meets FCC Class A computer equipment requirements

Protection

Mains fused; internal voltage and current limiting

Environment

Operating

0 to 35°C (32 to 95°F)

Storage

-30 to 75°C (-22 to 167°F)

Humidity

95% maximum without condensation

Dimensions

Standard 19" rack mount; 19"w x 1 3/4"h x 13.5"d (483 x 45 x 344 mm)

Weight

10.7 lb (4.9 kg)

Shipping weight 12.5 lb (5.7 kg)

PCM 70 Register Contents Record

Register Name							Number					
Original Program Name							Number					
Notes	<u> </u>											
					<u>_</u>		1.					
0.0	0.1	0.2	0.3	0.4	0.5	0.6	0.7	0.8	0.9			
1.0	1.1	1.2	1.3	1.4	1.5	1.6	1.7	1.8	1.9			
2.0	2.1	2.2	2.3	2.4	2.5	2.6	2.7	2.8	2.9			
3.0	3.1	3.2	3.3	3.4	3.5	3.6	3.7	3.8	3.9			
4.0	4.1	4.2	4.3	4.4	4.5	4.6	4.7	4.8	4.9			
5.0	5.1	5.2	5.3	5.4	5.5	5.6	5.7	5.8	5.9			
	<u> </u>								Parame			
<u> </u>	1	2	3	4	5	6	7	8	9			
									0411			
	T		<u> </u>		T				Controlle			
_					ŀ							
								Patch D	estinatio			
									Scalin			