

PCM1744

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## SoundPlus™ 24 Bits, 96kHz, Sampling Stereo Audio DIGITAL-TO-ANALOG CONVERTER

### FEATURES

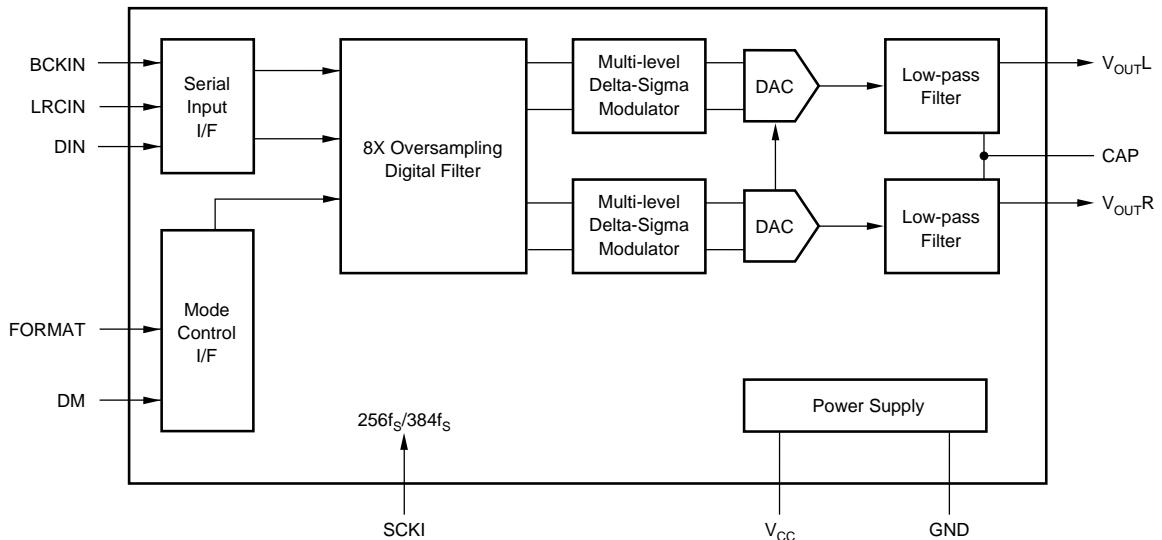
- COMPLETE STEREO DAC: Includes Digital Filter and Output Amp
- DYNAMIC RANGE: 95dB
- MULTIPLE SAMPLING FREQUENCIES: Up to 96kHz
- 8x OVERSAMPLING DIGITAL FILTER
- SYSTEM CLOCK:  $256f_s/384f_s$
- 24-BIT I<sup>2</sup>S DATA INPUT FORMAT
- SMALL 14-PIN SOIC PACKAGE

### DESCRIPTION

The PCM1744 is a complete low cost stereo audio digital-to-analog converter (DAC), operating off of a  $256f_s$  or  $384f_s$  system clock. The DAC contains a 3rd-order  $\Delta\Sigma$  modulator, a digital interpolation filter, and an analog output amplifier. The PCM1744 accepts 24-bit input data in a I<sup>2</sup>S format.

The digital filter performs an 8x interpolation function and includes de-emphasis at 44.1kHz. The PCM1744 can accept digital audio sampling frequencies from 16kHz to 96kHz, always at 8X oversampling.

The PCM1744 is ideal for low-cost, CD-quality consumer audio applications.



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Internet: <http://www.burr-brown.com/> • FAXLine: (800) 548-6133 (US/Canada Only) • Cable: BBRCORP • Telex: 066-6491 • FAX: (520) 889-1510 • Immediate Product Info: (800) 548-6132

# SPECIFICATIONS

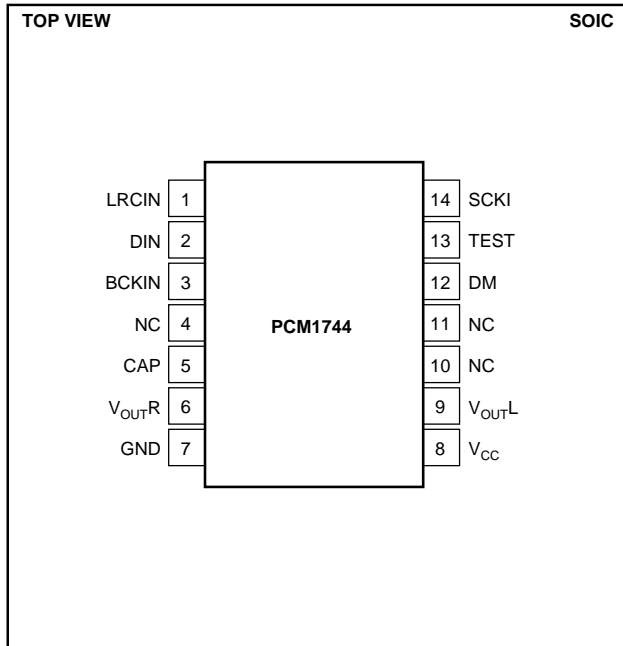
All specifications at +25°C, +V<sub>CC</sub> = +5V, f<sub>S</sub> = 44.1kHz, and 18-bit input data, SYSCLK = 384f<sub>S</sub>, unless otherwise noted.

PARAMETER	CONDITIONS	PCM1744			UNITS
		MIN	TYP	MAX	
<b>RESOLUTION</b>		24			Bits
<b>DATA FORMAT</b> Audio Data Interface Format Audio Data Format Sampling Frequency (f <sub>S</sub> ) Internal System Clock Frequency			I <sup>2</sup> S Two's Binary Complement		
		16		96	kHz
<b>DIGITAL INPUT/OUTPUT</b> Logic Level Input Logic Level V <sub>IH</sub> <sup>(1)</sup> V <sub>IL</sub> <sup>(1)</sup> Input Logic Current: I <sub>IN</sub> <sup>(1)</sup>		2.0	TTL	0.8 ±0.8	VDC VDC μA
<b>DYNAMIC PERFORMANCE</b> <sup>(2)</sup> THD+N at FS (0dB) THD+N at -60dB Dynamic Range Signal-to-Noise Ratio Channel Separation	f = 991kHz  EIAJ, A-weighted EIAJ, A-weighted		-83 -32 90 90 95	-79	dB dB dB dB dB
<b>DC ACCURACY</b> Gain Error Gain Mismatch, Channel-to-Channel Bipolar Zero Error	V <sub>OUT</sub> = V <sub>CC</sub> /2 at BPZ		±1.0 ±1.0 ±20	±10.0 ±5.0 ±50	% of FSR % of FSR mV
<b>ANALOG OUTPUT</b> Output Voltage Center Voltage Load Impedance	Full Scale (0dB)  AC Load		0.62 x V <sub>CC</sub> V <sub>CC</sub> /2		Vp-p VDC kΩ
<b>DIGITAL FILTER PERFORMANCE</b> Passband Stopband Passband Ripple Stopband Attenuation Delay Time		0.555		0.445 ±0.17	f <sub>S</sub> f <sub>S</sub> dB dB sec
<b>INTERNAL ANALOG FILTER</b> -3dB Bandwidth Passband Response	f = 20kHz		100 -0.16		kHz dB
<b>POWER SUPPLY REQUIREMENTS</b> Voltage Range Supply Current Power Dissipation		4.5	5 13 65	5.5 18 90	VDC mA mW
<b>TEMPERATURE RANGE</b> Operation Storage		-25 -55		+85 +125	°C °C

NOTES: (1) Pins 1, 2, 3, 12, 13, 14: LRCIN, DIN, BCKIN, DM, FORMAT, SCKI. (2) Dynamic performance specs are tested with 20kHz low pass filter and THD+N specs are tested with 30kHz LPF, 400Hz HPF, Average-Mode.

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## PIN CONFIGURATION



## PIN ASSIGNMENTS

PIN	NAME	I/O	FUNCTION
1 <sup>(1)</sup>	LRCIN	IN	Sample Rate Clock Input
2 <sup>(1)</sup>	DIN	IN	Audio Data Input
3 <sup>(1)</sup>	BCKIN	IN	Bit Clock Input for Audio Data.
4	NC	—	No Connection
5	CAP	—	Common Pin of Analog Output Amp
6	V <sub>OUTR</sub>	OUT	Right-Channel Analog Output
7	GND	—	Ground
8	V <sub>CC</sub>	—	Power Supply
9	V <sub>OUTL</sub>	OUT	Left-Channel Analog Output
10	NC	—	No Connection
11	NC	—	No Connection
12 <sup>(2)</sup>	DM	IN	De-Emphasis Control HIGH: De-emphasis ON LOW: De-emphasis OFF
13 <sup>(2)</sup>	TEST	—	Test Pin. Must be left open.
14 <sup>(1)</sup>	SCKI	IN	System Clock Input (256f <sub>s</sub> or 384f <sub>s</sub> )

NOTE: (1) Schmitt-Trigger input. (2) Schmitt-Trigger input with internal pull-up.

## ABSOLUTE MAXIMUM RATINGS

Power Supply Voltage	+6.5V
+V <sub>CC</sub> to +V <sub>DD</sub> Difference	±0.1V
Input Logic Voltage	-0.3V to (V <sub>DD</sub> + 0.3V)
Power Dissipation	290mW
Operating Temperature Range	-25°C to +85°C
Storage Temperature	-55°C to +125°C
Lead Temperature (soldering, 5s)	+260°C
Thermal Resistance, $\theta_{JA}$	+90°C/W



## ELECTROSTATIC DISCHARGE SENSITIVITY

This integrated circuit can be damaged by ESD. Burr-Brown recommends that all integrated circuits be handled with appropriate precautions. Failure to observe proper handling and installation procedures can cause damage.

ESD damage can range from subtle performance degradation to complete device failure. Precision integrated circuits may be more susceptible to damage because very small parametric changes could cause the device not to meet its published specifications.

## PACKAGE/ORDERING INFORMATION

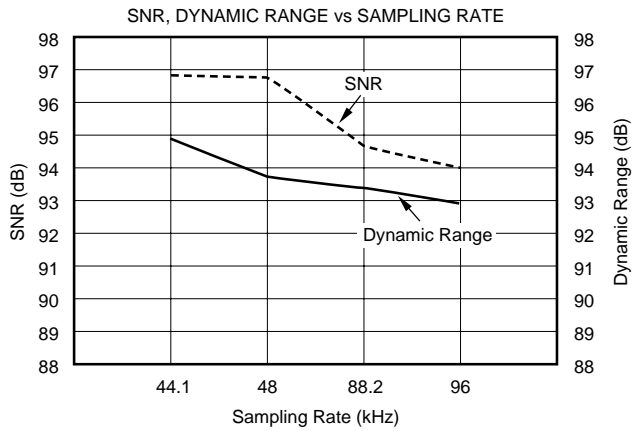
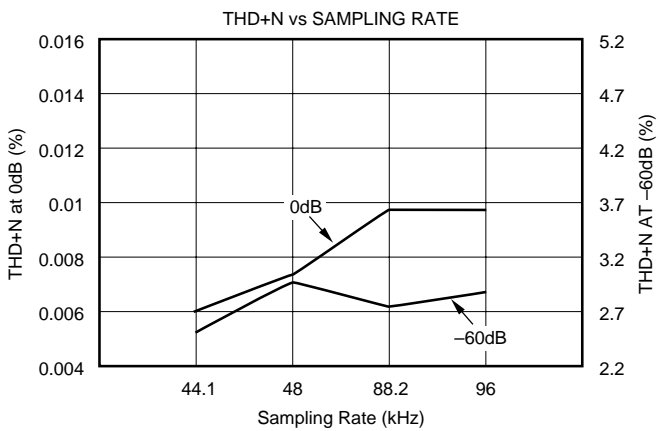
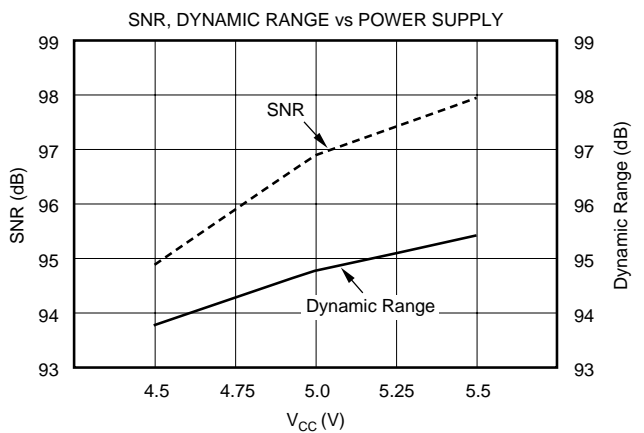
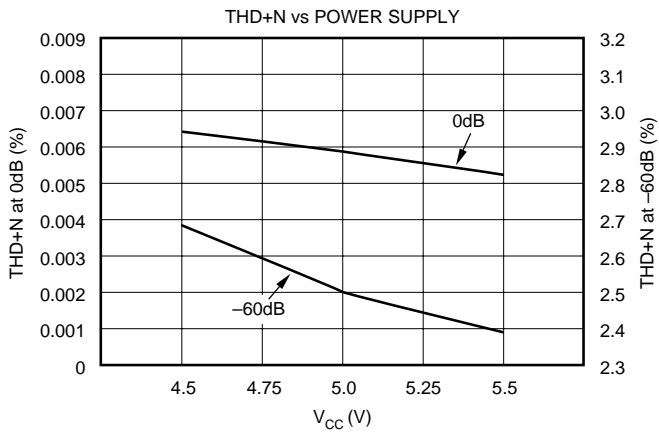
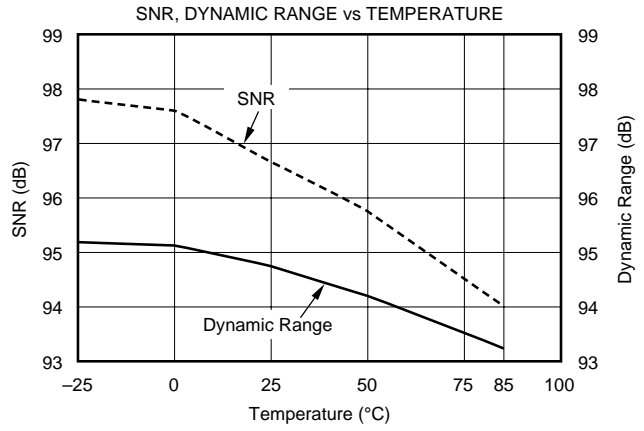
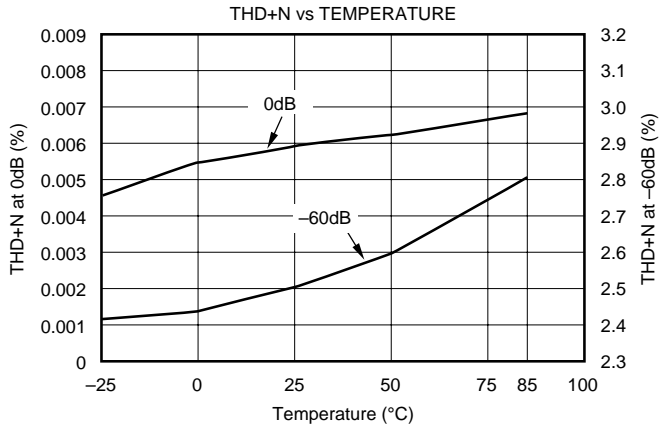
PRODUCT	PACKAGE	PACKAGE DRAWING NUMBER <sup>(1)</sup>	SPECIFIED TEMPERATURE RANGE	PACKAGE MARKING	ORDERING NUMBER <sup>(2)</sup>	TRANSPORT MEDIA
PCM1744	SO-14	235	-25°C to +85°C	PCM1744U	PCM1744U	Rails
"	"	"	"	PCM1744U	PCM1744U/2K	Tape and Reel

NOTES: (1) For detailed drawing and dimension table, please see end of data sheet, or Appendix C of Burr-Brown IC Data Book, or visit the Burr-Brown web site at [www.burr-brown.com](http://www.burr-brown.com). (2) Models with a slash (/) are available only in Tape and Reel in the quantities indicated (e.g., /2K indicates 2000 devices per reel). Ordering 2000 pieces of "PCM1744U/2K" will get a single 2500-piece Tape and Reel. For detailed Tape and Reel mechanical information, refer to Appendix B of Burr-Brown IC Data Book.

# TYPICAL PERFORMANCE CURVES

At  $T_A = +25^\circ\text{C}$ ,  $V_{CC} = +5\text{V}$ ,  $f_s = 44.1\text{kHz}$ ,  $\text{SYSCLK} = 256f_s$ , unless otherwise noted.

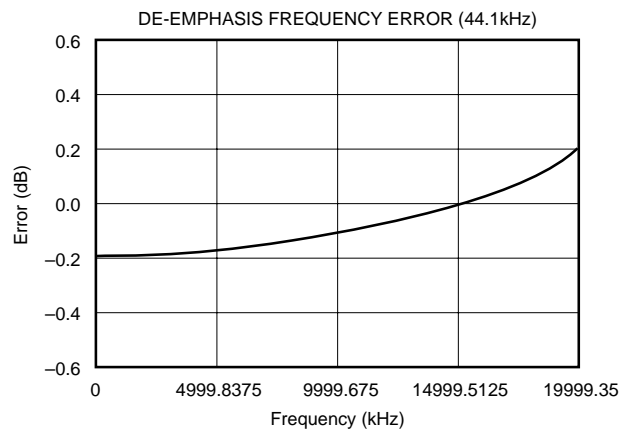
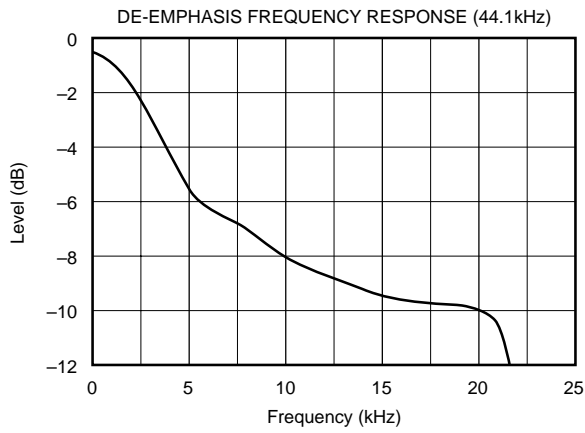
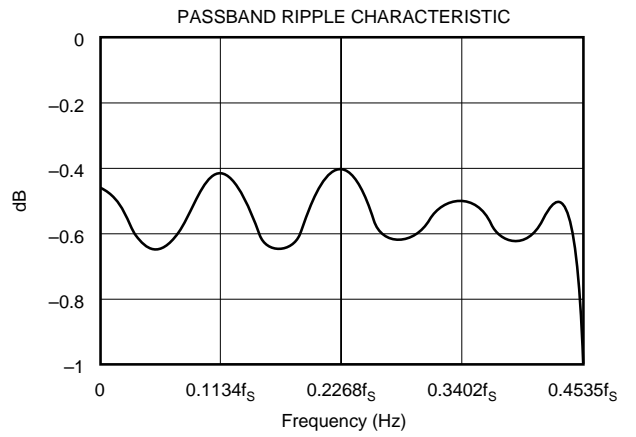
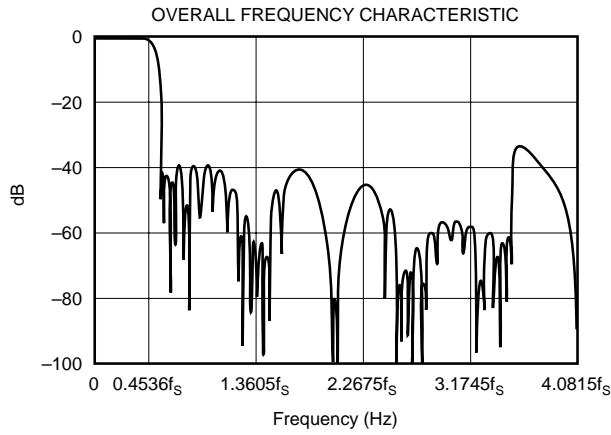
## DYNAMIC PERFORMANCE



# TYPICAL PERFORMANCE CURVES

At  $T_A = +25^\circ\text{C}$ ,  $+V_{CC} = +V_{DD} = +5\text{V}$ ,  $f_s = 44.1\text{kHz}$ , and 18-bit input data,  $\text{SYSCLK} = 384f_s$ , unless otherwise noted.

## DIGITAL FILTER



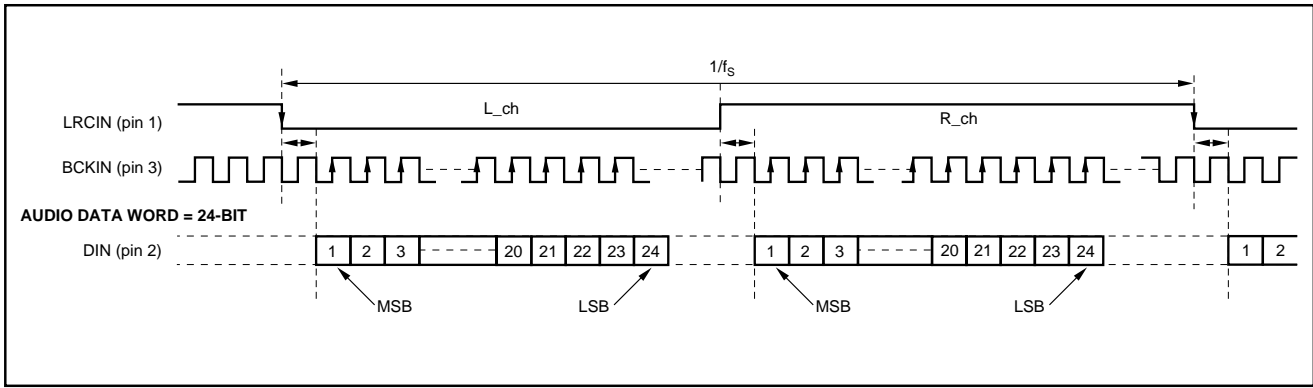


FIGURE 1. I<sup>2</sup>S Data Input Timing.

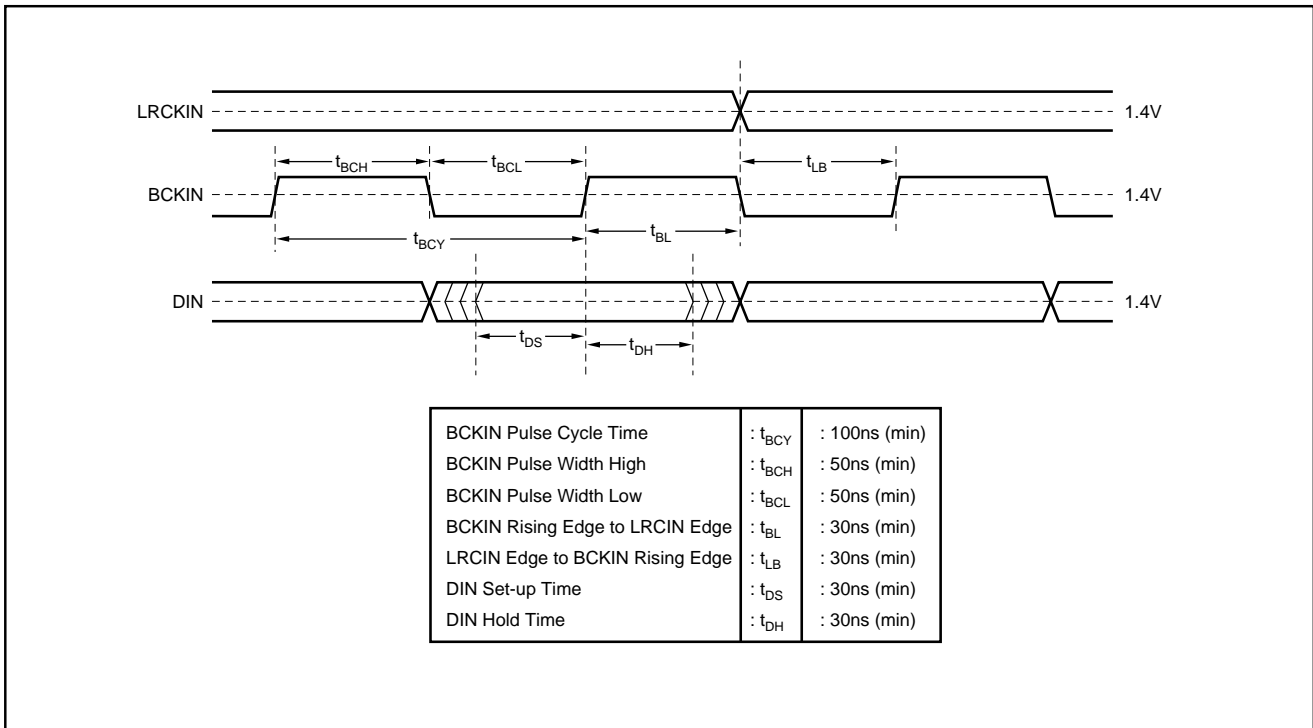


FIGURE 2. Audio Data Input Timing.

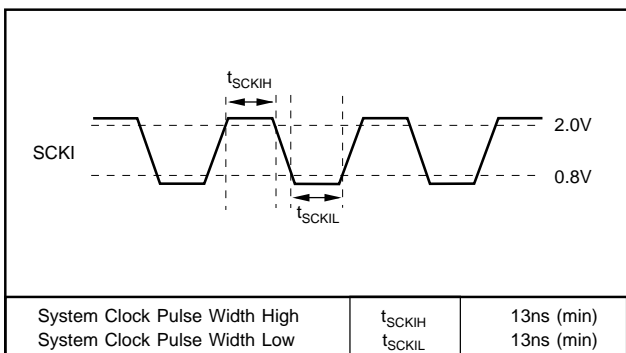


FIGURE 3. System Clock Timing Requirements.

## SYSTEM CLOCK

The system clock for PCM1744 must be either  $256f_s$  or  $384f_s$ , where  $f_s$  is the audio sampling frequency (LRCIN), typically 32kHz, 44.1kHz, 48kHz, 88.2kHz or 96kHz. The system clock is used to operate the digital filter and the noise shaper. The system clock input (SCKI) is at pin 14. Timing conditions for SCKI are shown in Figure 3.

PCM1744 has a system clock detection circuit which automatically detects the frequency, either  $256f_s$  or  $384f_s$ . The system clock should be synchronized with LRCIN (pin 1), but PCM1744 can compensate for phase differences. If the phase difference between LRCIN and system clock is greater

than  $\pm 6$  bit clocks (BCKIN), the synchronization is performed automatically. The analog outputs are forced to a bipolar zero state ( $V_{CC}/2$ ) during the synchronization function. Table I shows the typical system clock frequency inputs for the PCM1744.

SAMPLING RATE (LRCIN)	SYSTEM CLOCK FREQUENCY (MHz)	
	$256f_s$	$384f_s$
32kHz	8.192	12.288
44.1kHz	11.2896	16.9340
48kHz	12.288	18.432
88.2kHz	22.5792	33.868
96kHz	24.576	36.864

TABLE I. System Clock Frequencies vs Sampling Rate.

### TYPICAL CONNECTION DIAGRAM

Figure 4 illustrates the typical connection diagram for PCM1744 used in a stand-alone application.

### INPUT DATA FORMAT

PCM1744 can accept input data a 24-bit I<sup>2</sup>S format, as shown in Figure 1.

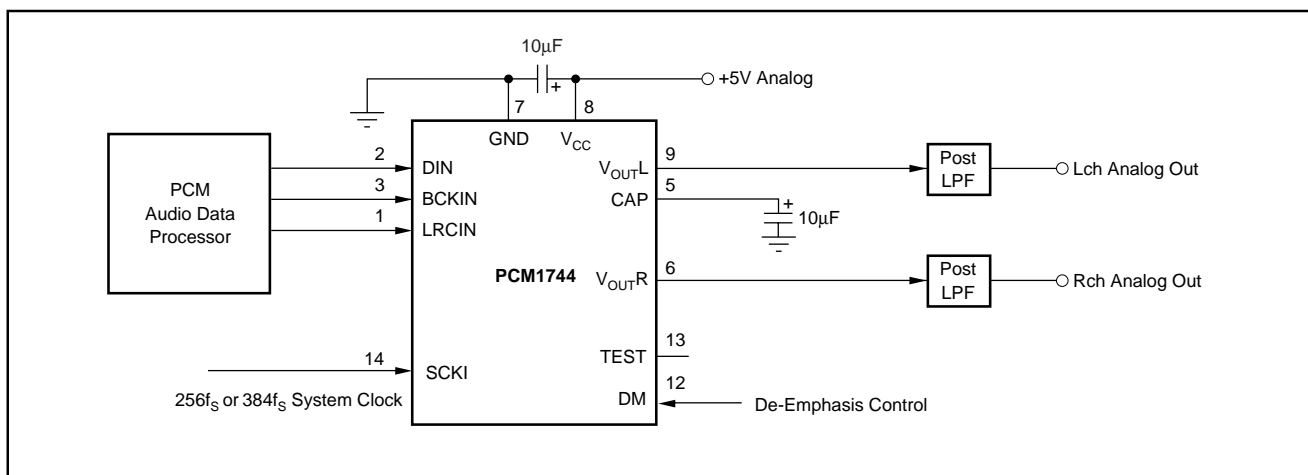


FIGURE 4. Typical Connection Diagram.

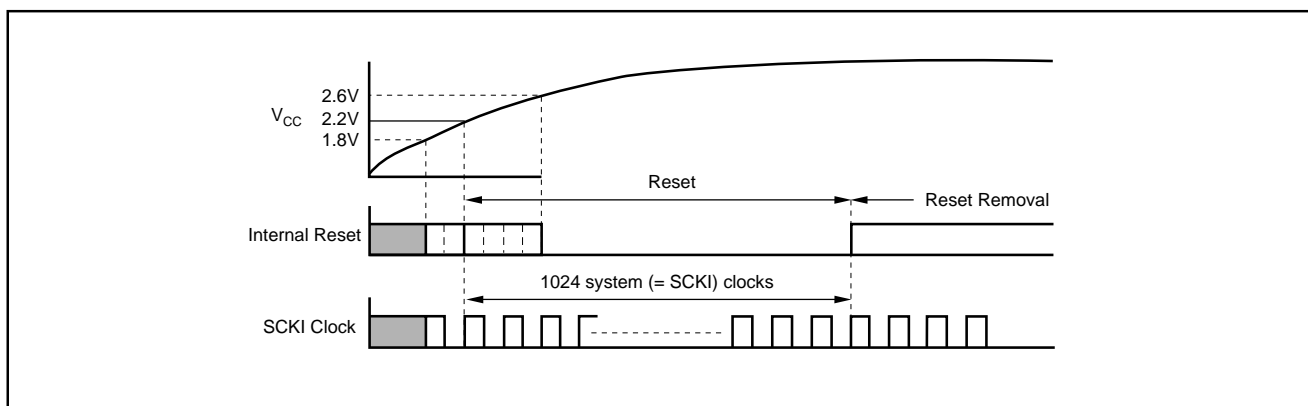


FIGURE 5. Internal Power-On Reset Timing.

### RESET

PCM1744 has an internal power-on reset circuit. The internal power-on reset initializes (resets) when the supply voltage  $V_{CC} > 2.2V$  (typ). The power-on reset has an initialization period equal to 1024 system clock periods after  $V_{CC} > 2.2V$ . During the initialization period, the outputs of the DAC are invalid, and the analog outputs are forced to  $V_{CC}/2$ . Figure 5 illustrates the power-on reset and reset-pin reset timing.

### DE-EMPHASIS CONTROL

Pin 12 (DM) enables PCM1744's de-emphasis function. De-emphasis operates only at 44.1kHz.

DM	
0	De-emphasis OFF
1	De-emphasis ON (44.1kHz)

TABLE II. De-emphasis Control Selection.

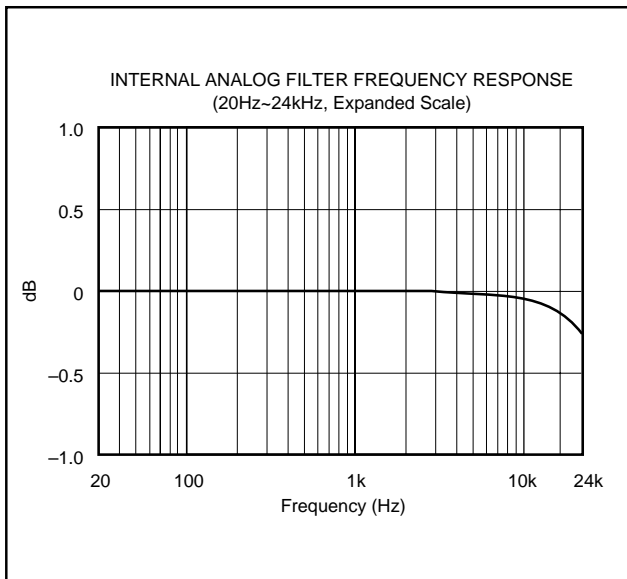


FIGURE 6. Low-Pass Filter Frequency Response.

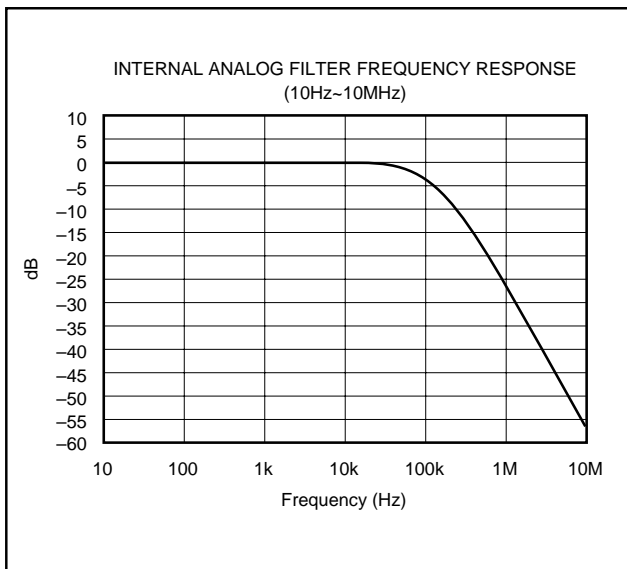


FIGURE 7. Low-Pass Filter Wideband Frequency Response.

## APPLICATION CONSIDERATIONS

### DELAY TIME

There is a finite delay time in delta-sigma converters. In A/D converters, this is commonly referred to as latency. For a delta-sigma D/A converter, delay time is determined by the order number of the FIR filter stage, and the chosen sampling rate. The following equation expresses the delay time of PCM1744:

$$T_D = 11.125 \times 1/f_S$$

For  $f_S = 44.1\text{kHz}$ ,  $T_D = 11.125/44.1\text{kHz} = 251.4\mu\text{s}$

Applications using data from a disc or tape source, such as CD audio, CD-Interactive, Video CD, DAT, Minidisc, etc., generally are not affected by delay time. For some professional applications such as broadcast audio for studios, it is important for total delay time to be less than 2ms.

### OUTPUT FILTERING

For testing purposes all dynamic tests are done on the PCM1744 using a 20kHz low-pass filter. This filter limits the measured bandwidth for THD+N, etc. to 20kHz. Failure to use such a filter will result in higher THD+N and lower SNR and dynamic range readings than are found in the specifications. The low-pass filter removes out-of-band noise. Although it is not audible, it may affect dynamic specification numbers.

The performance of the internal low pass filter from DC to 24kHz is shown in Figure 6. The higher frequency roll-off of the filter is shown in Figure 7. If the user's application has the PCM1744 driving a wideband amplifier, it is recommended to use an external low-pass filter. A simple 3rd-order filter is shown in Figure 8. For some applications, a passive RC filter or 2nd-order filter may be adequate.

### BYPASSING POWER SUPPLIES

The power supplies should be bypassed as close as possible to the unit. It is also recommended to include a 0.1 $\mu\text{F}$  ceramic capacitor in parallel with the 10 $\mu\text{F}$  tantalum bypass capacitor.

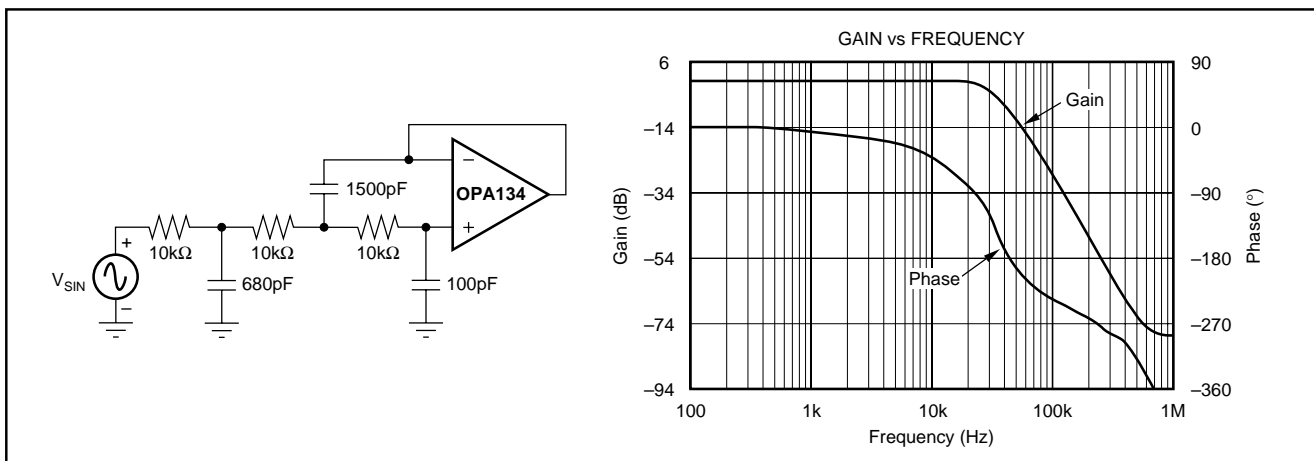


FIGURE 8. 3rd-Order LPF.



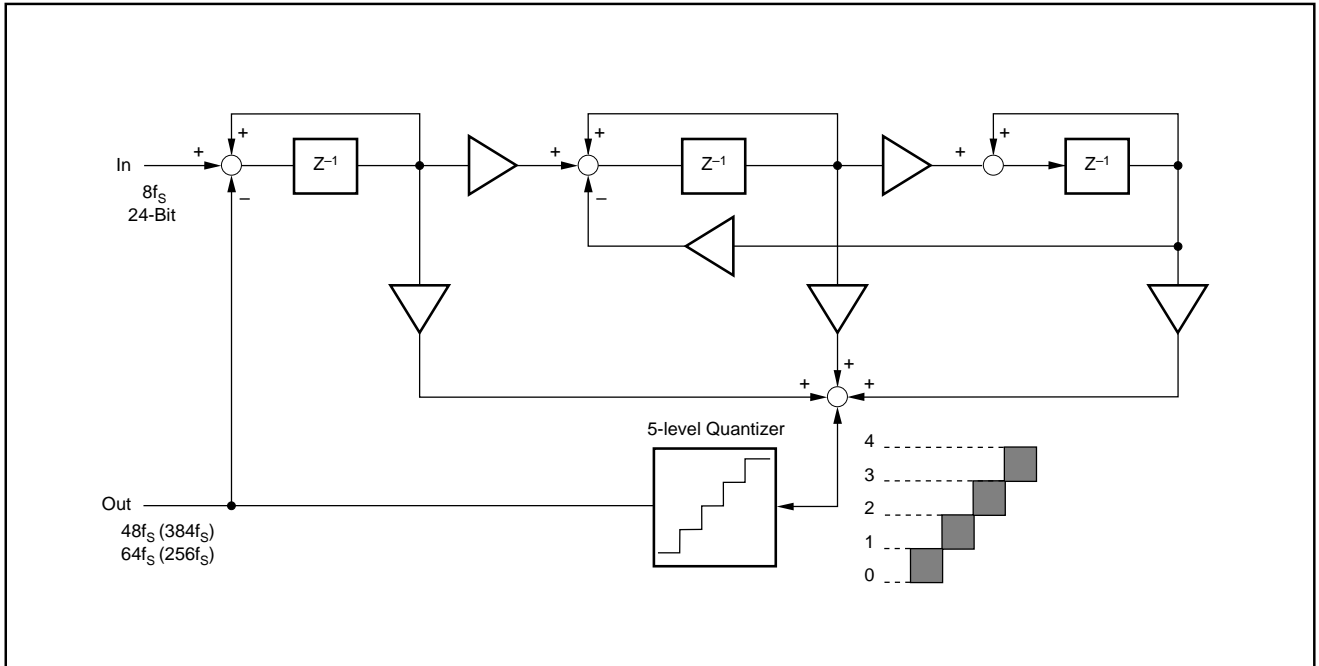


FIGURE 9. 5-Level  $\Delta\Sigma$  Modulator Block Diagram.

## THEORY OF OPERATION

The delta-sigma section of PCM1744 is based on a 5-level amplitude quantizer and a 3rd-order noise shaper. This section converts the oversampled input data to 5-level delta-sigma format. A block diagram of the 5-level delta-sigma modulator is shown in Figure 9. This 5-level delta-sigma modulator has the advantage of stability and clock jitter over the typical one-bit (2-level) delta-sigma modulator.

The combined oversampling rate of the delta-sigma modulator and the internal 8x interpolation filter is  $96f_s$  for a  $384f_s$  system clock, and  $64f_s$  for a  $256f_s$  system clock. The theoretical quantization noise performance of the 5-level delta-sigma modulator is shown in Figure 10.

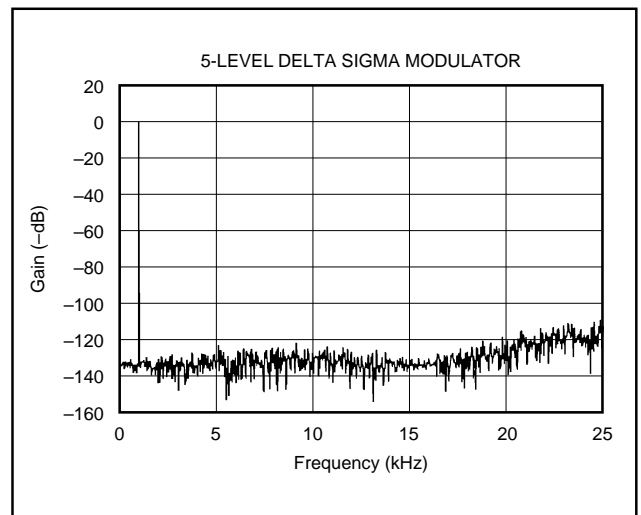


FIGURE 10. Quantization Noise Spectrum.