National Semiconductor

February 2006

LM4934 Boomer® Audio Power Amplifier Series
3D Audio Sub-System with Stereo Speaker, OCL/SE Stereo Headphone, Earpiece and Mono Line Level Outputs

General Description

The LM4934 is an integrated audio sub-system designed for stereo cell phone applications. Operating on a 3.3V supply, it combines a stereo speaker amplifier delivering 520mW per channel into an 8Ω load, a stereo headphone amplifier delivering 36mW per channel into a 32Ω load, a mono earpiece amplifier delivering 55mW into a 32Ω load, and a line output for an external powered handsfree speaker. It integrates the audio amplifiers, volume control, mixer, power management control, and National 3D enhancement all into a single package. In addition, the LM4934 routes and mixes the stereo and mono inputs into multiple distinct output modes. The LM4934 features an I2S serial interface for full range audio and an I2C/SPI compatible interface for control.

Boomer audio power amplifiers are designed specifically to provide high quality output power with a minimal amount of external components.

Key Specifications

- P_{OUT}, Stereo BTL, 8Ω, 3.3V, 1% THD+N — 520mW (typ)
- P_{OUT} HP, 32Ω, 3.3V, 1% THD+N — 36mW (typ)
- P_{OUT} Mono Earpiece, 32Ω, 3.3V, 1% THD+N — 55mW (typ)
- Shutdown current — 0.6µA (typ)
- DAC SNR — 95dB (typ)

Features

- 18-bit stereo DAC
- Multiple distinct output modes
- Stereo speaker amplifier
- Stereo headphone amplifier
- Mono earpiece amplifier
- Mono Line Output for external handsfree carkit
- Independent Left, Right, headphone and Mono speaker volume controls
- National 3D enhancement with programmable effect level
- I^2C/SPI (selectable) compatible interface
- Ultra low shutdown current
- Click and Pop Suppression circuit

Applications

- Cell Phones
- PDAs

LM4934 3D Audio Sub-System with Stereo Speaker, OCL/SE Stereo Headphone, Earpiece and Mono Line Level Outputs

Block Diagram

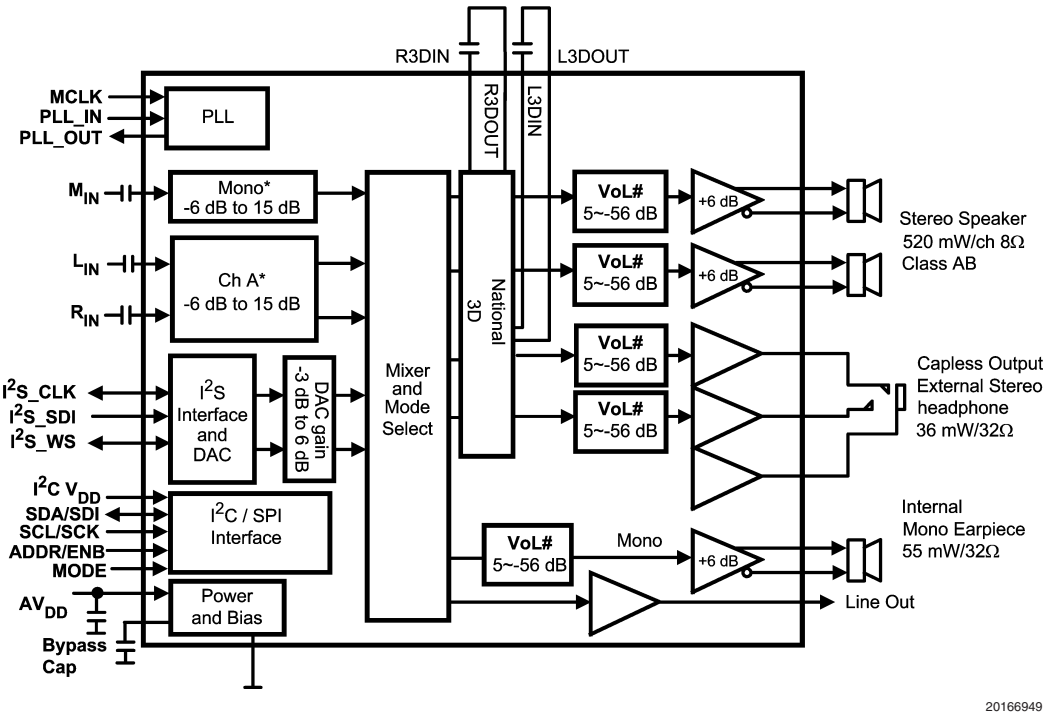


FIGURE 1. Audio Sub-System Block Diagram with OCL HP Outputs

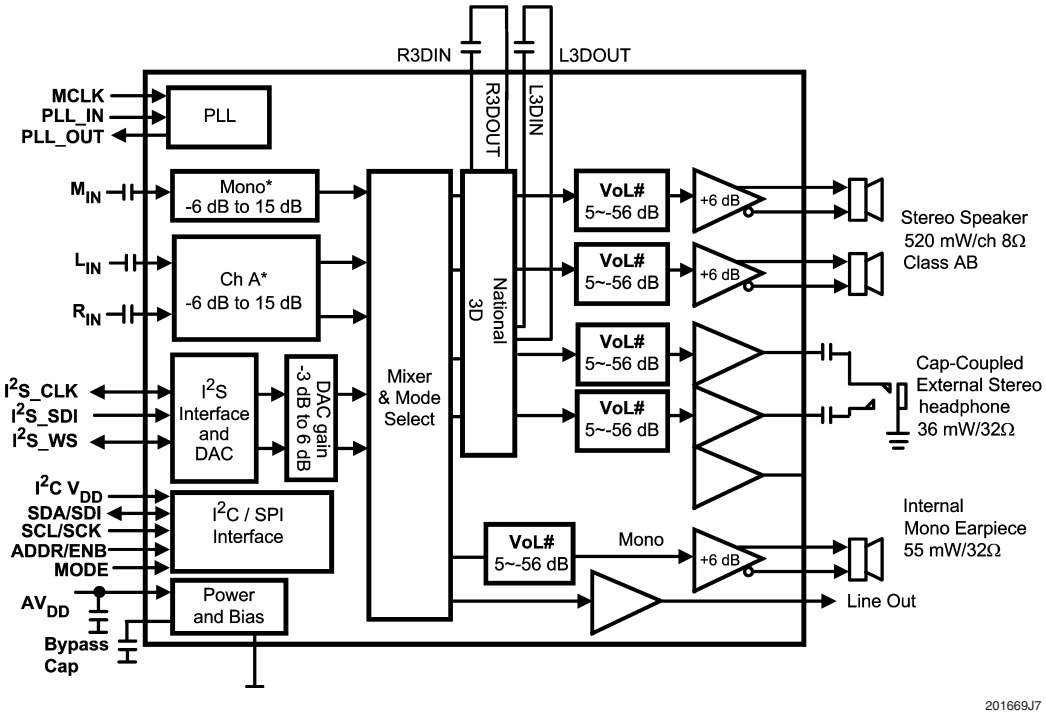
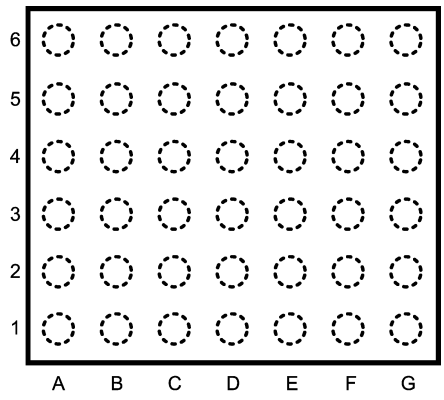


FIGURE 2. Audio Sub-System with SE HP Outputs

Connection Diagrams

42-Bump Micro SMDxt



Top View (Bump Side Down)
Order Number LM4934RL
See NS Package Number RLA42

Top Marking Drawing

Top View
XY — 2 Digit Date Code
TT — Traceability
G — Boomer Family
G9 — LM4934RL
I — Pin 1 Marking

Pin Descriptions

PIN	PIN NAME	D/A	I/O	DESCRIPTION
A1	DGND	D	P	DIGITAL GROUND
A2	MCLK	D	I	MASTER CLOCK
A3	I2S_WS	D	I/O	I2S WORD SELECT
A4	GPIO	D	O	TEST PIN (MUST BE LEFT FLOATING)
A5	ADDR/ENB	D	I	I2C_ADDR OR SPI_ENB DEPENDING ON I2C or SPI MODE SELECT
A6	DVDD	D	P	DIGITAL SUPPLY VOLTAGE
B1	PLLVDD	D	P	PLL SUPPLY VOLTAGE
B2	I2S_SDI	D	I	I2S SERIAL DATA INPUT
B3	I2S_CLK	D	I/O	I2S CLOCK SIGNAL
B4	MODE	D	I	SELECTS BETWEEN SPI AND I2C CONTROL INTERFACE
B5	I2C_VDD	D	P	I2C SUPPLY VOLTAGE
B6	VDDIO	D	P	I/O SUPPLY VOLTAGE
C1	PLL_IN	D	I	PLL FILTER INPUT
C2	PLL_OUT	D	O	PLL FILTER OUTPUT
C3	PLLGND	D	P	PLL GND
C4	SDA/SDI	D	I/O	I2C SDA OR SPI SDI
C5	SCL/SCK	D	I	I2C_SCL OR SPI_SCK
C6	AVDD	A	P	ANALOG SUPPLY VOLTAGE

Pin Descriptions (Continued)

D1	AGND	A	P	ANALOG GROUND
D2	R_{IN}	A	I	RIGHT ANALOG IN
D3	NC	A		NO CONNECT
D4	BYPASS	A	I	HALF-SUPPLY BYPASS
D5	LINEOUT	A	O	MONO LINE OUT
D6	RHP	A	O	RIGHT HEADPHONE OUTPUT
E1	EP-	A	O	MONO EARPIECE OUT-
E2	M_{IN}	A	I	MONO ANALOG IN
E3	L_{IN}	A	I	LEFT ANALOG IN
E4	R3DOUT	A	I	RIGHT CHANNEL 3D OUTPUT
E5	LHP	A	O	LEFT HEADPHONE OUTPUT
E6	CHP	A	O	HEADPHONE CENTER PIN OUTPUT (1/2 VDD)
F1	AGND	A	P	ANALOG GND
F2	EP+	A	O	MONO EARPIECE OUT+
F3	L3DIN	A	I	LEFT CHANNEL 3D INPUT
F4	L3DOUT	A	I	LEFT CHANNEL 3D OUTPUT
F5	R3DIN	A	I	RIGHT CHANNEL 3D INPUT
F6	AGND	A	P	ANALOG GND
G1	LLS-	A	O	LEFT SPEAKER OUT-
G2	AVDD	A	P	ANALOG SUPPLY VOLTAGE
G3	LLS+	A	O	LEFT SPEAKER OUT+
G4	RLS-	A	O	RIGHT SPEAKER OUT-
G5	AVDD	A	P	ANALOG SUPPLY VOLTAGE
G6	RLS+	A	O	RIGHT SPEAKER OUT+

Absolute Maximum Ratings (Notes 1, 2)

If Military/Aerospace specified devices are required, please contact the National Semiconductor Sales Office/ Distributors for availability and specifications.

Analog Supply Voltage	6.0V
Digital Supply Voltage	6.0V
Storage Temperature	-65˚C to +150˚C
Input Voltage	-0.3V to V_{DD} +0.3V
Power Dissipation (Note 3)	Internally Limited
ESD Susceptibility (Note 4)	2000V
ESD Susceptibility(Note 5)	200V
Junction Temperature	150˚C
Thermal Resistance	
θ_{JA} (RLA42)	61˚C/W
See AN-1279	

Operating Ratings

Temperature Range	
$T_{MIN} \le T_A \le T_{MAX}$	$-40˚C \le T_A \le +85˚C$
Supply Voltage	
	$2.7V \le AV_{DD} \le 5.5V$
	$2.7V \le DV_{DD} \le 4.0V$
	$2.4V \le I^2CV_{DD} \le 4.0V$

Audio Amplifier Electrical Characteristics AV_{DD} = 3.0V, DV_{DD} = 3.0V (Notes 1, 2)

The following specifications apply for the circuit shown in Figure 1 with all programmable gain set at 0dB, unless otherwise specified. Limits apply for T_A = 25˚C.

Symbol	Parameter	Conditions	LM4934		Units (Limits)
			Typical (Note 6)	Limits (Notes 7, 8)	
I_{DD}	Supply Current	V_{IN} = 0, No Load All Amps On + DAC, OCL	18.5	26.5	mA (max)
		Headphone Mode Only, OCL	5.6	8	mA (max)
		Stereo Speaker Mode Only	12	19.5	mA (max)
		Mono Speaker Mode Only	5.9	8	mA (max)
		DAC Off, All Amps On, OCL	14.6	22	mA (max)
I_{SD}	Shutdown Current		0.6	2	µA (max)
P_O	Output Power	Speaker; THD = 1%; f = 1kHz, 8Ω BTL	420	370	mW (min)
		Headphone; THD = 1%; f = 1kHz, 32Ω SE	27	24	mW (min)
		Earpiece; THD = 1%; f = 1kHz, 32Ω BTL	45	40	mW (min)
$V_{FS\ DAC}$	Full Scale DAC Output		2.4		Vpp
THD+N	Total Harmonic Distortion	Speaker; P_O = 200mW; f = 1kHz, 8Ω BTL	0.04		%
		Headphone; P_O = 10mW; f = 1kHz, 32Ω SE	0.01		%
		Earpiece; P_O = 20mW; f = 1kHz, 32Ω BTL	0.04		%
		Line Out; V_O = 1Vrms; f = 1kHz, 10kΩ SE	0.004		%
V_{OS}	Offset Voltage	Speaker	8	55	mV (max)
		Earpiece	8	50	mV (max)
		HP (OCL)	8	40	mV
$\in_O$	Output Noise	A = weighted; 0dB gain; See Table 1	Table 1		
PSRR	Power Supply Rejection Ratio	f = 217Hz; V_{ripple} = 200mV_{P-P} C_B = 2.2µF; See Table 2	Table 2		

Audio Amplifier Electrical Characteristics AV_{DD} = 3.0V, DV_{DD} = 3.0V (Notes 1, 2) (Continued)

The following specifications apply for the circuit shown in Figure 1 with all programmable gain set at 0dB, unless otherwise specified. Limits apply for T_A = 25˚C.

Symbol	Parameter	Conditions	LM4934		Units (Limits)
			Typical (Note 6)	Limits (Notes 7, 8)	
Xtalk	Crosstalk	Loudspeaker; P_O= 200mW f = 1kHz	−84		dB
		Headphone; P_O= 10mW f = 1kHz; SE	−85		dB
		Headphone; P_O= 10mW f = 1kHz; OCL	−60		dB
T_{WU}	Wake-Up Time	C_B = 2.2µF, CD6 = 0	35		ms
		C_B = 2.2µF, CD6 = 1	85		ms

Audio Amplifier Electrical Characteristics AV_{DD} = 5.0V, DV_{DD} = 3.3V (Notes 1, 2)

The following specifications apply for the circuit shown in Figure 1 with all programmable gain set at 0dB, unless otherwise specified. Limits apply for T_A = 25˚C.

Symbol	Parameter	Conditions	LM4934		Units (Limits)
			Typical (Note 6)	Limits (Notes 7, 8)	
I_{DD}	Supply Current	V_{IN} = 0, No Load All Amps On - DAC	24		mA (max)
		Headphone Mode Only	5.8		mA
		Stereo Speaker Mode Only	17		mA
		Mono Speaker Mode Only	7		mA
		DAC Off, All Amps On	19		mA
I_{SD}	Shutdown Current		1.6		µA
P_O	Output Power	Speaker; THD = 1%; f = 1kHz, 8Ω BTL	1.2		W
		Headphone; THD = 1%; f = 1kHz, 32Ω SE	80		mW
		Earpiece; THD = 1%; f = 1kHz, 32Ω BTL	175		mW
$V_{FS\ DAC}$	Full Scale DAC Output		2.4		Vpp
THD+N	Total Harmonic Distortion	Speaker; P_O = 500mW; f = 1kHz, 8Ω BTL	0.03		%
		Headphone; P_O = 30mW; f = 1kHz, 32Ω SE	0.01		%
		Earpiece; P_O = 40mW; f = 1kHz, 32Ω BTL; CD4 = 0	0.04		%
		Line Out; V_O = 1Vrms; f = 1kHz, 10kΩ SE	0.003		%
V_{OS}	Offset Voltage	Speaker	8		mV
		Earpiece	8		mV
		HP (OCL)	8		mV
$\in_O$	Output Noise	A = weighted; 0dB gain; See Table 1	Table 1		

Audio Amplifier Electrical Characteristics AV_{DD} = 5.0V, DV_{DD} = 3.3V (Notes 1, 2) (Continued)

The following specifications apply for the circuit shown in Figure 1 with all programmable gain set at 0dB, unless otherwise specified. Limits apply for T_A = 25˚C.

Symbol	Parameter	Conditions	LM4934		Units (Limits)
			Typical (Note 6)	Limits (Notes 7, 8)	
PSRR	Power Supply Rejection Ratio	f = 217Hz; V_{ripple} = 200mV_{P-P} C_B = 2.2µF; See Table 3	Table 3		
Xtalk	Crosstalk	Loudspeaker; P_O= 400mW f = 1kHz	−86		dB
		Headphone; P_O= 15mW f = 1kHz; OCL	−56		dB
		Headphone; P_O= 15mW f = 1kHz, SE	−80		dB
T_{WU}	Wake-Up Time	C_B = 2.2µF, CD6 = 0	45		ms
		C_B = 2.2µF, CD6 = 1	130		ms

Volume Control Electrical Characteristics (Notes 1, 2)

The following specifications apply for 3V ≤ AV_{DD} ≤ 5V and 2.7V ≤ DV_{DD} ≤ 4.0V, unless otherwise specified. Limits apply for T_A = 25˚C.

Symbol	Parameter	Conditions	LM4934		Units (Limits)
			Typical (Note 6)	Limits (Notes 7, 8)	
PGR	Stereo or Mono Analog Inputs PreAmp Gain Setting Range	minimum gain setting	−6	−6.5	dB (min)
				−5.5	dB (max)
		maximum gain setting	15	15.5	dB (max)
				14.5	dB (min)
VCR	Output Volume Control for Stereo Speakers, Headphone Output, or Mono Output Range	minimum gain setting, Vol = 00001	−56	−56.5	dB (min)
				−55.5	dB (max)
		maximum gain setting	5	4.5	dB (min)
				5.5	dB (max)
ΔA_{CH-CH}	Stereo Channel to Channel Gain Mismatch		0.3		dB
A_{MUTE}	Mute Attenuation	V_{in} = 1V_{rms}, Gain = 0dB with load, Vol = 00000			
		Headphone	<-90		dB
		Line Out	<-90		dB
R_{INPUT}	M_{IN}, L_{IN} and R_{IN} Input Impedance		23	18	kΩ (min)
				28	kΩ (max)

Digital Section Electrical Characteristics (Notes 1, 2)

The following specifications apply for 3V ≤ AV_{DD} ≤ 5V and 2.7V ≤ DV_{DD} ≤ 4.0V, unless otherwise specified. Limits apply for T_A = 25˚C.

Symbol	Parameter	Conditions	LM4934		Units (Limits)
			Typical (Note 6)	Limits (Notes 7, 8)	
DI_{SD}	Digital Shutdown Current	Mode 0, DV_{DD} = 3.0V			
		No MCLK	0.01	1	µA
DI_{DD}	Digital Power Supply Current	f_{MCLK} = 12MHz, DV_{DD} = 3.0V			
		ALL MODES EXCEPT 0	5.3	8	mA
$PLLI_{DD}$	PLL Quiescent Current	f_{MCLK} = 12MHz, DV_{DD} = 3.0V	4.8	6	mA
Audio DAC (Typical numbers are with 6.144MHz audio clock and 48kHz sampling frequency					
R_{DAC}	Audio DAC Ripple	20Hz - 20kHz through headphone output	+/-0.1		dB
PB_{DAC}	Audio DAC Passband width	-3dB point	22.6		kHz
SBA_{DAC}	Audio DAC Stop band Attenuation	Above 24kHz	76		dB
DR_{DAC}	Audio DAC Dynamic Range	DC - 20kHz, –60dBFS; AES17 Standard See Table 4	Table 4		dB
SNR	Audio DAC-AMP Signal to Noise Ratio	A-Weighted, Signal = V_O at 0dBFS, f = 1kHz Noise = digital zero, A-weighted, See Table 4	Table 4		dB
SNR_{DAC}	Internal DAC SNR	A-weighted (Note 10)	95		dB
PLL					
f_{IN}	Input Frequency on MCLK pin		12	10	MHz
				26	
SPI/I^2C					
f_{SPI}	Maximum SPI Frequency			4000	kHz (max)
$t_{SPISETD}$	SPI Data Setup Time			100	ns (max)
$t_{SPISETENB}$	SPI ENB Setup Time			100	ns (max)
$t_{SPIHOLDD}$	SPI Data Hold Time			100	ns (max)
$t_{SPIHOLDENB}$	SPI ENB Hold Time			100	ns (max)
t_{SPICL}	SPI Clock Low Time			125	ns (max)
t_{SPICH}	SPI Clock High Time			125	ns (max)
f_{CLKI2C}	I^2C_CLK Frequency			400	kHz (max)
$t_{I2CHOLD}$	I^2C_DATA Hold Time			100	ns (max)
t_{I2CSET}	I^2C_DATA Setup Time			100	ns (max)
V_{IH}	I^2C/SPI Input High Voltage		I^2CV_{DD}	0.7 x I^2CV_{DD}	V (min)
V_{IL}	I^2C/SPI Input Low Voltage		0	0.3 x I^2CV_{DD}	V (max)
I^2S					
f_{CLKI^2S}	I^2S_CLK Frequency	I^2S_RES = 0	1536	6144	kHz (max)
		I^2S_RES = 1	3072	12288	
	I^2S_WS Duty Cycle		50	40	%
V_{IH}	Digital Input High Voltage			0.7 x DV_{DD}	V (min)
V_{IL}	Digital Input Low Voltage			0.3 x DV_{DD}	V (max)

Note 1: All voltages are measured with respect to the GND pin unless otherwise specified.

Note 2: Absolute Maximum Ratings indicate limits beyond which damage to the device may occur. Operating Ratings indicate conditions for which the device is functional but do not guarantee specific performance limits. Electrical Characteristics state DC and AC electrical specifications under particular test conditions which guarantee specific performance limits. This assumes that the device is within the Operating Ratings. Specifications are not guaranteed for parameters where no limit is given, however, the typical value is a good indication of device performance.

Note 3: The maximum power dissipation must be derated at elevated temperatures and is dictated by T_{JMAX} ,θ_{JA}, and the ambient temperature, T_A. The maximum allowable power dissipation is $P_{DMAX} = (T_{JMAX} - T_A) / \theta_{JA}$ or the number given in Absolute Maximum Ratings, whichever is lower.

Note 4: Human body model: 100pF discharged through a 1.5kΩ resistor.

Note 5: Machine model: 220pF - 240pF discharged through all pins.

Note 6: Typicals are measured at 25˚C and represent the parametric norm.

Note 7: Limits are guaranteed to National's AOQL (Average Outgoing Quality Level).

Note 8: Datasheet min/max specification limits are guaranteed by design, test, or statistical analysis.

Note 9: Shutdown current is measured in a normal room environment.

Note 10: Internal DAC only with DAC modes 00 and 01.

TABLE 1. Output Noise

Output Noise AV_{DD} = 5V and AV_{DD} = 3V. All gains set to 0dB. Units in µV. A - weighted

MODE	EP	LS	HP OCL or SE	Lineout	Units
1	22	22	11	9	µV
2	22	22	11	9	µV
3	22	22	11	9	µV
4	68	88	46	35	µV
5	38	48	24	20	µV
6	29	34	18	15	µV
7	38	48	24	20	µV

TABLE 2. PSRR AV_{DD} = 3V

PSRR AV_{DD} = 3V. f = 217Hz; V_{ripple} = 200mVp-p; C_B = 2.2µF.

MODE	EP (Typ)	LS (Typ)	LS (Limit)	HP OCL or SE (Typ)	HP OCL or SE (Limit)	Lineout (Typ)	Units
1	69	71		72		70	dB
2	69	71	67	72	68	70	dB
3	69	71		72		70	dB
4	63	62		55		68	dB
5	69	68		61		69	dB
6	69	70		64		70	dB
7	69	68		61		69	dB

TABLE 3. PSRR AV_{DD} = 5V

PSRR AV_{DD} = 5V. All gains set to 0dB. f = 217Hz; V_{ripple} = 200mVp-p; C_B = 2.2µF

MODE	EP (Typ)	LS (Typ)	HP OCL or SE (Typ)	Lineout (Typ)	Units
1	68	72	71	70	dB
2	68	72	71	70	dB
3	68	72	71	70	dB
4	68	66	69	70	dB
5	68	69	70	70	dB
6	69	72	71	71	dB
7	68	69	70	70	dB

TABLE 4. Dynamic Range and SNR

Dynamic Range and SNR. 3V ≤ AV_{DD} ≤ 5V. All programmable gain set to 0dB. Units in dB.

	DR (Typ)	SNR (Typ)	Units
LS	95	85	dB
Lineout	100	87	dB

TABLE 4. Dynamic Range and SNR (Continued)

Dynamic Range and SNR. 3V ≤ AV_{DD} ≤ 5V. All programmable gain set to 0dB. Units in dB.

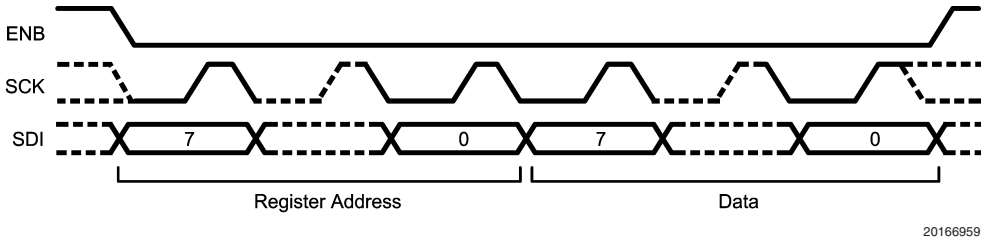
	DR (Typ)	SNR (Typ)	Units
HP	95	85	dB
EP	97	87	dB

System Control

The LM4934 is controlled via either a three wire SPI or a two wire I²C compatible interface, selectable with the MODE pin. When MODE is cleared the device is in I²C mode, when MODE is set the device is in SPI mode. This interface is used to configure the operating mode, interfaces, data converters, mixers and amplifiers. The LM4934 is controlled by writing 8 bit data into a series of write-only registers, the device is always a slave for both type of interfaces.

THREE WIRE, SPI INTERFACE (MODE = 1)

Three Wire Mode Write Bus Transaction



Three Wire Mode Write Bus Timing

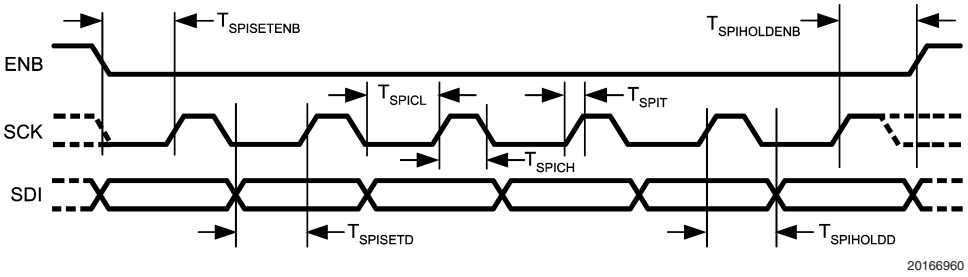
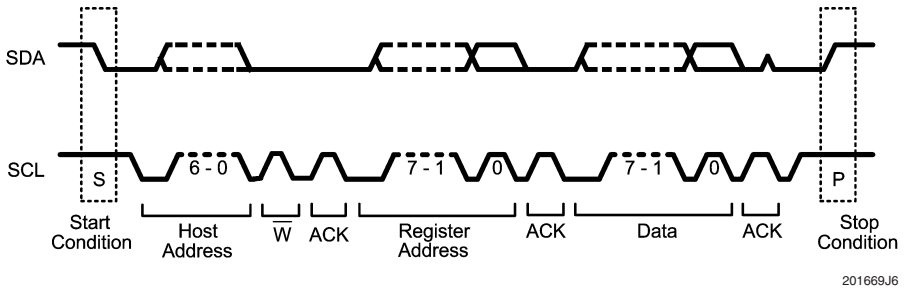


FIGURE 3. Three Wire Mode Write Bus

When the part is configured as an SPI device and the enable (ENB) line is lowered the serial data on SDI is clocked in on the rising edge of the SCK line. The protocol used is 16bit, MSB first. The upper 8 bits (15:8) are used to select an address within the device, the lower 8 bits (7:0) contain the updated data for this register.

TWO WIRE I²C COMPATIBLE INTERFACE (MODE = 0)

Two Wire Mode Write Bus Transaction



Two Wire Mode Write Bus Timing

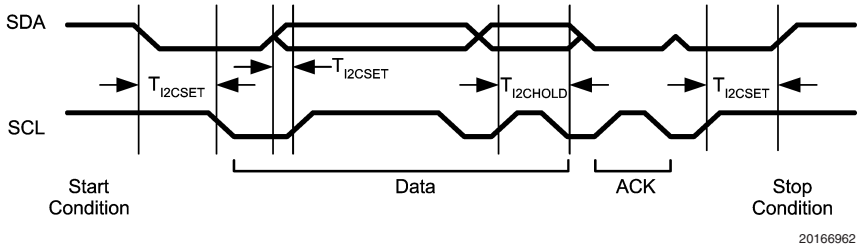


FIGURE 4. Two Wire Mode Write Bus

System Control (Continued)

When the part is configured as an I^2C device then the LM4934 will respond to one of two addresses, according to the ADDR input. If ADDR is low then the address portion of the I^2C transaction should be set to write to 0010000. When ADDR is high then the address input should be set to write to 1110000.

TABLE 5. Chip Address

	A6	A5	A4	A3	A2	A1	A0
Chip Address	EC	EC	1	0	0	0	0
ADR = 0	0	0	1	0	0	0	0
ADR = 1	1	1	1	0	0	0	0

EC — Externally configured by ADR pin

System Control (Continued)

TABLE 6. Control Registers

Address	Register	D7	D6	D5	D4	D3	D2	D1	D0
00h	Mode Control	0	CD_6	0	OCL	CD_3	CD_2	CD_1	CD_0
01h	Output Control	0	0	HP_R_ OUTPUT	HP_L_ OUTPUT	LS_R_ OUTPUT	LS_L_ OUTPUT	MONO_ OUTPUT	LINEOUT_ OUTPUT
02h	Mono Volume Control	0	0	0	MONO_VOL_4	MONO_VOL_3	MONO_VOL_2	MONO_VOL_1	MONO_VOL_0
03h	Loud Speaker LeftVolume and 3D Gain	0	3D_LEVEL_1	3D_LEVEL_0	LS_L_VOL_4	LS_L_VOL_3	LS_L_VOL_2	LS_L_VOL_1	LS_L_VOL_0
04h	Loud Speaker RightVolume and 3D Control	0	3D_MODE	3D_ENABLE	LS_R_VOL_4	LS_R_VOL_3	LS_R_VOL_2	LS_R_VOL_1	LS_R_VOL_0
05h	Headphone Left Volume Control	0	0	0	HP_L_VOL_4	HP_L_VOL_3	HP_L_VOL_2	HP_L_VOL_1	HP_L_VOL_0
06h	Headphone Right Volume Control	0	0	0	HP_R_VOL_4	HP_R_VOL_3	HP_R_VOL_2	HP_R_VOL_1	HP_R_VOL_0
07h	Analog R & L Input Gain Control	0	0	ANA_R_ GAIN_2	ANA_R_ GAIN_1	ANA_R_ GAIN_0	ANA_L_ GAIN_2	ANA_L _GAIN_1	ANA_L _GAIN_0
08h	Analog Mono & DAC Input Gain Control	0	DIG_R_ GAIN_1	DIG_R_ GAIN_0	DIG_L_ GAIN_1	DIG_L_ GAIN_0	MONO_IN_ GAIN_2	MONO_IN_ GAIN_1	MONO_IN_ GAIN_0
09h	Clock Configu ration	R_DIV_3	R_DIV_2	R_DIV_1	R_DIV_0	PLL_ ENABLE	AUDIO _CLK_SEL	PLL_INPUT	FAST_ CLOCK
0Ah	PLL M Divider	0	PLL_M_6	PLL_M_5	PLL_M_4	PLL_M_3	PLL_M_2	PLL_M_1	PLL_M_0
0Bh	PLL N Divider	PLL_N_7	PLL_N_6	PLL_N_5	PLL_N_4	PLL_N_3	PLL_N_2	PLL_N_1	PLL_N_0
0Ch	PLL N_MOD Divider and Dither Level	VCO_FAST	PLL_DITH_LEV_1	PLL_DITH_LEV_0	PLL_N_MOD_4	PLL_N_MOD_3	PLL_N_MOD_2	PLL_N_MOD_1	PLL_N_MOD_0
0Dh	PLL_P Divider	0	0	0	0	PLL_P_3	PLL_P_2	PLL_P_1	PLL_P_0
0Eh	DAC Setup	0	CUST_COMP	DITHER_ALW_ON	DITHER_OFF	MUTE_R	MUTE_L	DAC_MODE_1	DAC_MODE_0
0Fh	Interface	0	0	0	0	I2C_FAST	I2S_MODE	I2S_RESOL	I2S_M/S
10h	COMPENSATION _C OEFF0_LSB	COMP0_7	COMP0_6	COMP0_5	COMP0_4	COMP0_3	COMP0_2	COMP0_1	COMP0_0
11h	COMPENSATION _C OEFF0_MSB	COMP0_15	COMP0_14	COMP0_13	COMP0_12	COMP0_11	COMP0_10	COMP0_9	COMP0_8
12h	COMPENSATION _C OEFF1_LSB	COMP1_7	COMP1_6	COMP1_5	COMP1_4	COMP1_3	COMP1_2	COMP1_1	COMP1_0

System Control (Continued)

TABLE 6. Control Registers (Continued)

Address	Register	D7	D6	D5	D4	D3	D2	D1	D0
13h	COMPENSATION _COEFF1_MSB	COMP1_15	COMP1_14	COMP1_13	COMP1_12	COMP1_11	COMP1_10	COMP1_9	COMP1_8
14h	COMPENSATION _COEFF2_LSB	COMP2_7	COMP2_6	COMP2_5	COMP2_4	COMP2_3	COMP2_2	COMP2_1	COMP2_0
15h	COMPENSATION _COEFF2_MSB	COMP2_15	COMP2_14	COMP2_13	COMP2_12	COMP2_11	COMP2_10	COMP2_9	COMP2_8
16h	TEST_ REGISTER	RESERVED	RESERVED	RESERVED	RESERVED	RESERVED	RESERVED	RESERVED	RESERVED

Note: All registers default to 0 on initial power-up.

System Controls

TABLE 7. Stereo or Mono, Left or Right Volume Control

MONO_VOL_4, LS_L_VOL_4, LS_R_VOL_4, HP_L_VOL_4, HP_R_VOL_4	MONO_VOL_3, LS_L_VOL_3, LS_R_VOL_3, HP_L_VOL_3, HP_R_VOL_3	MONO_VOL_2, LS_L_VOL_2, LS_R_VOL_2, HP_L_VOL_2, HP_R_VOL_2	MONO_VOL_1, LS_L_VOL_1, LS_R_VOL_1, HP_L_VOL_1, HP_R_VOL_1	MONO_VOL_0, LS_L_VOL_0, LS_R_VOL_0, HP_L_VOL_0, HP_R_VOL_0	Gain (dB)
0	0	0	0	0	Mute
0	0	0	0	1	−56
0	0	0	1	0	−52
0	0	0	1	1	−48
0	0	1	0	0	−45
0	0	1	0	1	−42
0	0	1	1	0	−39
0	0	1	1	1	−36
0	1	0	0	0	−33
0	1	0	0	1	−30
0	1	0	1	0	−28
0	1	0	1	1	−26
0	1	1	0	0	−24
0	1	1	0	1	−22
0	1	1	1	0	−20
0	1	1	1	1	−18
1	0	0	0	0	−16
1	0	0	0	1	−14
1	0	0	1	0	−12
1	0	0	1	1	−10
1	0	1	0	0	−8
1	0	1	0	1	−6
1	0	1	1	0	−4
1	0	1	1	1	−3
1	1	0	0	0	−2
1	1	0	0	1	−1
1	1	0	1	0	0
1	1	0	1	1	+1
1	1	1	0	0	+2
1	1	1	0	1	+3
1	1	1	1	0	+4
1	1	1	1	1	+5

System Controls (Continued)

TABLE 8. Mixer Code Control

Mode	CD3	CD2	CD1	CD0	Mono Lineout	Mono Earpiece	Loud-Speaker L	Loud-Speaker R	Headphone L	Headphone R
0	0	0	0	0	SD	SD	SD	SD	SD	SD
1	1	0	0	1	M	M	M	M	M	M
2	1	0	1	0	AL+AR	AL+AR	AL	AR	AL	AR
3	1	0	1	1	M+AL+AR	M+AL+AR	M+AL	M+AR	M+AL	M+AR
4	1	1	0	0	DL+DR	DL+DR	DL	DR	DL	DR
5	1	1	0	1	DL+DR+ AL+AR	DL+DR+ AL+AR	DL+AL	DR+AR	DL+AL	DR+AR
6	1	1	1	0	M+DL+AL+ DR+AR	M+DL+AL+ DR+AR	M+DL+AL	M+DR+AR	M+DL+AL	M+DR+AR
7	1	1	1	1	M+DL+DR	M+DL+DR	M+DL	M+DR	M+DL	M+DR

SD — Shutdown
M — Mono Input
AL — Analog Left Channel
AR — Analog Right Channel
DL — I2S DAC Left Channel
DR — I2S DAC Right Channel
MUTE — Mute
Note: Power-On Default Mode is Mode 0

System Controls (Continued)

TABLE 9. Output Control (01h)

Loudspeaker Left Channel	LS_L_OUTPUT = 1	LS_L_OUTPUT = 0	
	Output On	Output Off	
Loudspeaker Right Channel	LS_R_OUTPUT = 1	LS_R_OUTPUT = 0	
	Output On	Output Off	
Headphone Left Channel	HP_L_OUTPUT = 1	HP_L_OUTPUT = 0	
	Output On	OCL = 1, Output Mute	OCL = 0, Output Mute
Headphone Right Channel	HP_R_OUTPUT = 1	HP_R_OUTPUT = 0	
	Output On	OCL = 1, Output Mute	OCL = 0, Output Mute
Mono Speaker Output	MONO_OUTPUT = 1	MONO_OUTPUT = 0	
	Output On	Output Off	
Lineout	LINEOUT_OUTPUT = 1	LINEOUT_OUTPUT = 0	
	Output On	Output Mute	
Headphone Output Mode	OCL = 1	OCL = 0	
	Headphone Output set to Capless ($CHP = 1/2\ AV_{DD}$)	Headphone Output Set to Cap-coupled	
All Outputs	CD3 = 1	CD3 = 0	
	Outputs Toggled Via Register Control	All Outputs Off	

TABLE 10. National 3D Enhancement Level Select (03h)

3D_LEVEL_1	3D_LEVEL_0	MIX RATIO
0	0	25%
0	1	40%
1	0	55%
1	1	70%

TABLE 11. National 3D Mode Control (04h)

3D_MODE	MODE
0	3D type 1
1	3D type 2

3D type 1: $R_{OUT} = R_i - G * L_{OUT3D}$, $L_{OUT} = L_i - G * R_{OUT3D}$
3D type 2: $R_{OUT} = -R_i - G * L_{OUT3D}$, $L_{OUT} = L_i + G * R_{OUT3D}$
R_i = Right Input
L_i = Left Input
G = 3D gain level (Mix Ratio)
$R_{OUT3D} = R_i$ through the high-pass filter R_{3D} and C_{3D}
$L_{OUT3D} = L_i$ through the high-pass filter R_{3D} and C_{3D}

System Controls (Continued)

TABLE 12. Analog Input Amplifier Gain Select

MONO_IN_GAIN_2 ANA_L_GAIN_2 ANA_R_GAIN_2	MONO_IN_GAIN_1 ANA_L_GAIN_1 ANA_R_GAIN_1	MONO_IN_GAIN_0 ANA_L_GAIN_0 ANA_R_GAIN_0	Input Gain Setting
0	0	0	−6dB
0	0	1	−3dB
0	1	0	0dB
0	1	1	3dB
1	0	0	6dB
1	0	1	9dB
1	1	0	12dB
1	1	1	15dB

TABLE 13. DAC Gain Select

DIG_L_GAIN_1 DIG_R_GAIN_1	DIG_L_GAIN_1 DIG_R_GAIN_1	Input Gain Setting
0	0	−3dB
0	1	0dB
1	0	3dB
1	1	6dB

PLL Configuration Registers

PLL M DIVIDER CONFIGURATION REGISTER

This register is used to control the input divider of the PLL.

PLL_M (0Ah) (Set = logic 1, Clear = logic 0)

Bits	Register	Description	
6:0	PLL_M	Programs the PLL input divider to select:	
		PLL_M	Divide Ratio
		0	Divider Off
		1	1
		2	1.5
		3	2
		4	2.5
		...	3→
		126	63.5
		127	64

NOTES:
The M divider should be set such that the output of the divider is between 0.5 and 5MHz. See the PLL setup section for details.
The divider of the M divider is derived from PLL_M as such:
M = (PLL_M+1) / 2

PLL Configuration Registers (Continued)

PLL N DIVIDER CONFIGURATION REGISTER

This register is used to control PLL N divider.

PLL_N (0Bh) (Set = logic 1, Clear = logic 0)

Bits	Register	Description	
7:0	PLL_N	Programs the PLL feedback divider:	
		PLL_N	Divide Ratio
		0	Divider Off
		1 → 10	10
		11	11
		12	12
		...	...
		248	248
		249	249
		250 → 255	250

NOTES:
The divider should be set such that the output of the divider is between 0.5 and 5MHz. See the PLL setup section for details. The N divider should never be set so that (Fin/M) * N > 55MHz (or 80MHz if FAST_VCO is set in the PLL_N_MOD register).
The non-sigma-delta division of the N divider is derived from the PLL_N as such:
N = PLL_N
Fin /M is often referred to as F_{comp} (Frequency of Comparison) or F_{ref} (Reference Frequency). In this document, F_{comp} is used.

PLL P DIVIDER CONFIGURATION REGISTER

This register is used to control the PLL's P divider.

PLL_P (0Dh) (Set = logic 1, Clear = logic 0)

Bits	Register	Description	
3:0	**PLL_P**	**Programs the PLL input divider to select:**	
		PLL_P	Divide Ratio
		0	Divider Off
		1	1
		2	1.5
		3	2
		...	–> 2.5
		13	7
		14	7.5
		15	8

NOTES:
The division of the P divider is derived from PLL_P as such:
P = (PLL_P+1) / 2

PLL Configuration Registers (Continued)

PLL N MODULATOR AND DITHER SELECT CONFIGURATION REGISTER

This register is used to control the Fractional component of the PLL.

PLL_N_MOD (0Ch) (Set = logic 1, Clear = logic 0)

Bits	Register	Description	
4:0	PLL_N_MOD	This programs the PLL N Modulator's fractional component:	
		PLL_N_MOD	Fractional Addition
		0	0/32
		1	1/32
		2 → 30	2/32 → 30/32
		31	31/32
6:5	DITHER_LEVEL	Allows control over the dither used by the N Modulator	
		DITHER_LEVEL	DAC Sub-system Input Source
		00	Medium (32)
		01	Small (16)
		10	Large (48)
		11	Off
7	FAST_VCO	If set the VCO maximum and minimum frequencies are raised:	
		FAST_VCO	Maximum F_{VCO}
		0	40–55MHz
		1	55–80MHz

NOTES:

The complete N divider is a fractional divider as such:

N = PLL_N + (PLL_N_MOD/32)

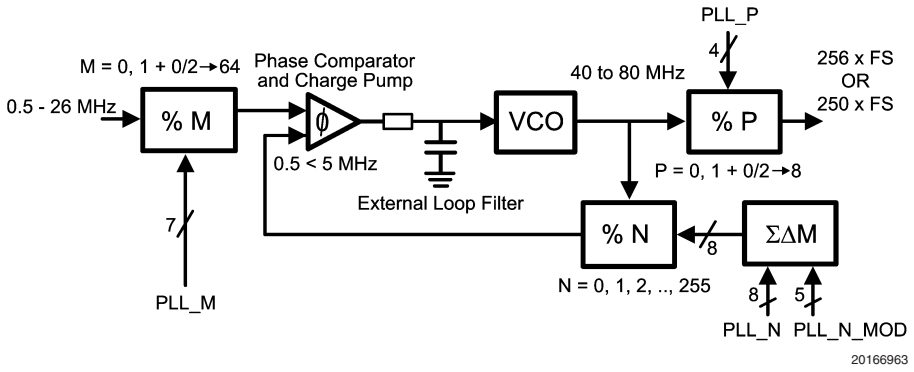
If the modulus input is zero, then the N divider is simply an integer N divider. The output from the PLL is determined by the following formula:

Fout = (Fin * N) / (M * P)

Please see over for more details on the PLL and common settings.

Further Notes on PLL Programming

The sigma-delta PLL is designed to drive audio circuits requiring accurate clock frequencies of up to 25MHz with frequency errors noise-shaped away from the audio band. The 5 bits of modulus control provide exact synchronization of 48kHz and 44.1kHz sample rates from any common clock source when the oversampling rate of the audio system is 125fs. In systems where 128x oversampling must be used (for example with an isochronous I2S data stream) a clock synchronous to the sample rate should be used as input to the PLL (typically the I2S clock). If no isochronous source is available then the PLL can be used to obtain a clock that is accurate to within typical crystal tolerances of the real sample rate.



Example Of PII Settings For 48Khz Sample Rates

f_in (MHz)	fsamp (kHz)	M	N	P	PLL_M	PLL_N	PLL_N_MOD	PLL_P	f_out (MHz)
11	48	11	60	5	21	60	0	9	12
12	48	5	25	5	9	25	0	9	12
12.288	48	4	19.53125	5	7	19	17	9	12
13	48	13	60	5	25	60	0	9	12
14.4	48	9	37.5	5	17	37	16	9	12
16.2	48	27	100	5	53	100	0	9	12
16.8	48	14	50	5	27	50	0	9	12
19.2	48	13	40.625	5	25	40	20	9	12
19.44	48	27	100	6	53	100	0	11	12
19.68	48	20.5	62.5	5	40	62	16	9	12
19.8	48	16.5	50	5	32	50	0	9	12

Example PII Settings For 44.1Khz Sample Rates

f_in (MHz)	fsamp (kHz)	M	N	P	PLL_M	PLL_N	PLL_N_MOD	PLL_P	f_out (MHz)
11	44.1	11	55.125	5	21	55	4	9	11.025000
11.2896	44.1	8	39.0625	5	15	39	2	9	11.025000
12	44.1	5	22.96875	5	9	22	31	9	11.025000
13	44.1	13	55.125	5	25	55	4	9	11.025000
14.4	44.1	12	45.9375	5	23	45	30	9	11.025000
16.2	44.1	9	30.625	5	17	30	20	9	11.025000
16.8	44.1	17	55.78125	5	33	55	25	9	11.025000
19.2	44.1	16	45.9375	5	31	45	30	9	11.025000
19.44	44.1	13.5	38.28125	5	26	38	9	9	11.025000
19.68	44.1	20.5	45.9375	4	40	45	30	7	11.025000
19.8	44.1	11	30.625	5	21	30	20	9	11.025000

These tables cover the most common applications, obtaining clocks for sample rates such as 22.05kHz and 192kHz should be done by changing the P divider value or the R divider in the clock configuration diagram.

If the user needs to obtain a clock unrelated to those described above, the following method is advised. An example of obtaining 11.2896 from 12.000MHz is shown below.

Further Notes on PLL Programming (Continued)

Choose a small range of P so that the VCO frequency is swept between 45 and 55MHz (or 60-80MHz if VCOFAST is used). Remembering that the P divider can divide by half integers. So for P = 4.0 → 7.0 sweep the M inputs from 2.5 → 24. The most accurate N and N_MOD can be calculated by:

N = FLOOR(((Fout/Fin)*(P*M)),1)

N_MOD = ROUND(32*((((Fout)/Fin)*(P*M)-N),0)

This shows that setting M = 11.5, N = 75 N_MOD = 47 P = 7 gives a comparison frequency of just over 1MHz, a VCO frequency of just under 80MHz (so VCO_FAST must be set) and an output frequency of 11.289596 which gives a sample rate of 44.099985443kHz, or accurate to 0.33 ppm.

Care must be taken when synchronization of isochronous data is not possible, i.e. when the PLL has to be used in the above mode. The I2S should be master on the LM4934 so that the data source can support appropriate SRC as required. This method should only be used with data being read on demand to eliminate sample rate mismatch problems.

Where a system clock exists at an integer multiple of the required DAC clock rate it is preferable to use this rather than the PLL. The LM4934 is designed to work in 8,12,16,24,32, and 48kHz modes from a 12MHz clock without the use of the PLL. This saves power and reduces clock jitter.

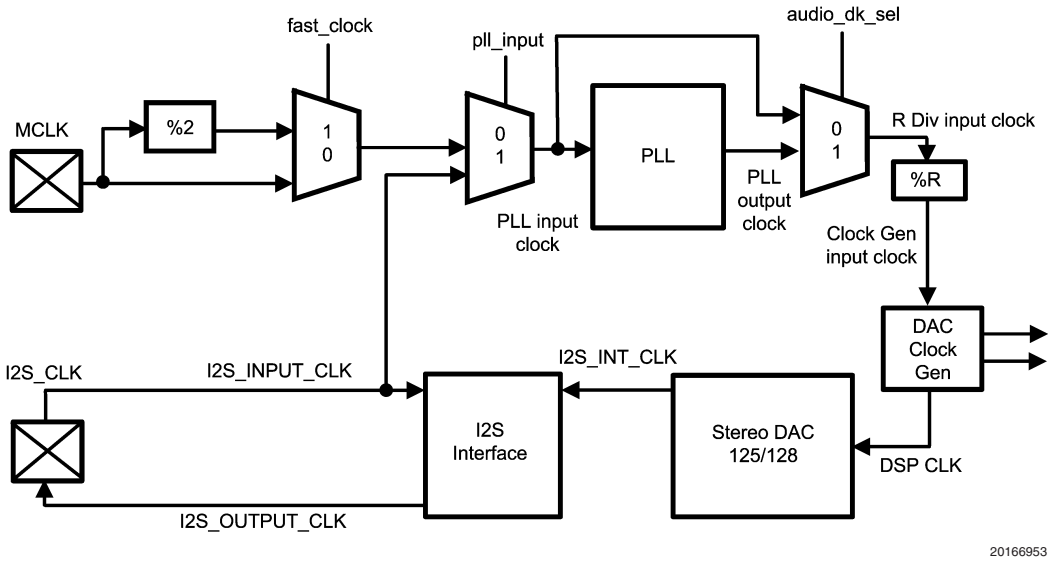
Clock Configuration Register

This register is used to control the multiplexers and clock R divider in the clock module.

CLOCK (09h) (Set = logic 1, Clear = logic 0)

Bits	Register	Description	
0	FAST_CLOCK	If set master clock is divided by two.	
		FAST_CLOCK	MCLK Frequency
		0	Normal
		1	Divided by 2
1	PLL_INPUT	Programs the PLL input multiplexer to select:	
		PLL_INPUT	PLL Input Source
		0	MCLK
		1	I2S Input Clock
2	AUDIO_CLK_SEL	Selects which clock is passed to the audio sub-system	
		DAC_CLK_SEL	DAC Sub-system Input Source
		0	PLL Input
		1	PLL Output
3	PLL_ENABLE	If set enables the PLL. (MODES 4–7 only)	
7:4	R_DIV	Programs the R divider	
		R_DIV	Divide Value
		0000	1
		0001	1
		0010	1.5
		0011	2
		0100	2.5
		0101	3
		0110	3.5
		0111	4
		1000	4.5
		1001	5
		1010	5.5
		1011	6
		1100	6.5
		1101	7
		1110	7.5
		1111	8

Clock Configuration Register (Continued)



By default the stereo DAC operates at 250*fs, i.e. 12.000MHz (at the clock generator input clock) for 48kHz data. It is expected that the PLL be used to drive the audio system unless a 12.000MHz master clock is supplied. The PLL can also use the I2S clock input as a source. In this case, the audio DAC uses the clock from the output of the PLL.

Common Clock Settings for the DAC

The DAC can work in 4 modes, each with different oversampling rates, 125,128,64 & 32. In normal operation 125x oversampling provides for the simplest clocking solution as it will work from 12.000MHz (common in most systems with Bluetooth or USB) at 48kHz exactly. The other modes are useful if data is being provided to the DAC from an uncontrollable isochronous source (such as a CD player, DAB, or other external digital source) rather than being decoded from memory. In this case the PLL can be used to derive a clock for the DAC from the I2S clock.

The DAC oversampling rate can be changed to allow simpler clocking strategies, this is controlled in the DAC SETUP register but the oversampling rates are as follows:

DAC MODE	Oversampling Ratio Used
00	125
01	128
10	64
11	32

The following table describes the clock required at the clock generator input for various clock sample rates in the different DAC modes:

Fs (kHz)	DAC Oversampling Ratio	Required CLock at DAC Clock Generator Input (MHz)
8	125	2
8	128	2.048
11.025	125	2.75625
11.025	128	2.8224
12	125	3
12	128	3.072
16	125	4
16	128	4.096
22.05	125	5.5125
22.05	128	5.6448
24	125	6
24	128	6.144
32	125	8
32	128	8.192

Common Clock Settings for the DAC (Continued)

Fs (kHz)	DAC Oversampling Ratio	Required CLock at DAC Clock Generator Input (MHz)
44.1	125	11.025
44.1	128	11.2896
48	125	12
48	128	12.288
88.2	64	11.2896
96	64	12.288
176.4	32	22.5792
192	32	24.576

Methods for producing these clock frequencies are described in the PLL section.

The R divider can be used when the master clock is exactly 12.00 MHz in order to generate different sample rates. The Table below shows different sample rates supported from 12.00MHz by using only the R divider and disabling the PLL. In this way we can save power and the clock jitter will be low.

R_DIV	Divide Value	DAC Clock Generator Input Frequency <MHz>	Sample Rate Supported <KHz>
11	6	2	8
9	5	2.4	9.6
7	4	3	12
5	3	4	16
4	2.5	4.8	19.2
3	2	6	24
2	1.5	8	32
0	1	12	48

The R divider can also be used along with the P divider in order to create the clock needed to support low sample rates.

DAC Setup Register

This register is used to configure the basic operation of the stereo DAC.

DAC_SETUP (0Eh) (Set = logic 1, Clear = logic 0)

<table>
<tr><th>Bits</th><th>Register</th><th colspan="4">Description</th></tr>
<tr><td rowspan="6">1:0</td><td rowspan="6">DAC_MODE</td><td colspan="4">The DAC used in the LM4934 can operate in one of 4 oversampling modes. The modes are described as follows:</td></tr>
<tr><td>DAC_MODE</td><td>Oversampling Rate</td><td>Typical FS</td><td>Clock Required</td></tr>
<tr><td>00</td><td>125</td><td>48KHz</td><td>12.000MHz (USB Mode)</td></tr>
<tr><td>01</td><td>128</td><td>44.1KHz
48KHz</td><td>11.2896MHz
12.288MHz</td></tr>
<tr><td>10</td><td>64</td><td>96KHz</td><td>12.288MHz</td></tr>
<tr><td>11</td><td>32</td><td>192KHz</td><td>24.576MHz</td></tr>
<tr><td>2</td><td>MUTE_L</td><td colspan="4">Mutes the left DAC channel on the next zero crossing.</td></tr>
<tr><td>3</td><td>MUTE_R</td><td colspan="4">Mutes the right DAC channel on the next zero crossing.</td></tr>
<tr><td>4</td><td>DITHER_OFF</td><td colspan="4">If set the dither in DAC is disabled.</td></tr>
<tr><td>5</td><td>DITHER ALWAYS_ON</td><td colspan="4">If set the dither in DAC is enabled all the time.</td></tr>
<tr><td>6</td><td>CUST_COMP</td><td colspan="4">If set the DAC frequency response can be programmed manually via a 5 tap FIR “compensation” filter. This can be used to enhance the frequency response of small loudspeakers or provide a crude tone control. The compensation Coefficients can be set by using registers 10h to 15h.</td></tr>
</table>

Interface Control Register

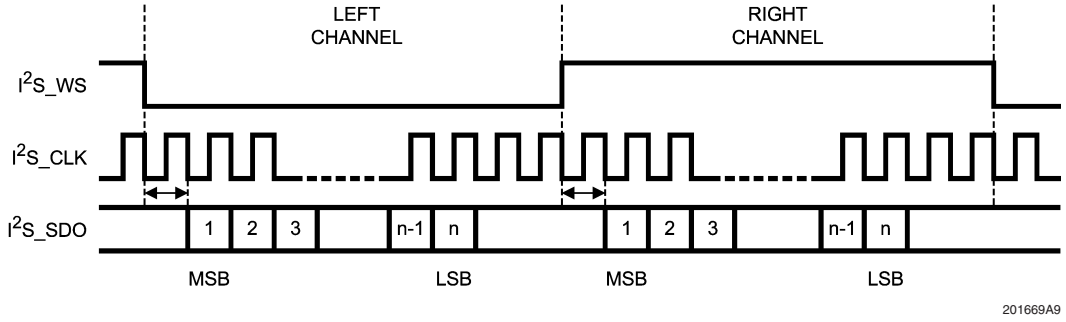
This register is used to control the I2S and I^2C compatible interface on the chip.

INTERFACE (0Fh) (Set = logic 1, Clear = logic 0)

Bits	Register	Description
0	I2S_MASTER_SLAVE	If set the LM4934 acts as a master for I2S, so both I2S clock and I2S word select are configured as outputs. If cleared the LM4934 acts as a slave where both I2S clock and word select are configured as inputs.
1	I2S_RESOLUTION	If set the I2S resolution is set to 32 bits. If clear, resolution is set to 16 bits. This bit only affects the I2S Interface in master mode. In slave mode the I2S Interface can support any I2S compatible resolution. In master mode the I2S resolution also depends on the DAC mode as the note below explains.
2	I2S_MODE	If set the I2S is configured in left justified mode timing. If clear, the I2S interface is configured in normal I2S mode timing.
3	I2C_FAST	If set enables the I2C to run in fast mode with an I2C clock up to 3.4MHz. If clear the I2C speed gets its default value of a maximum of 400kHz

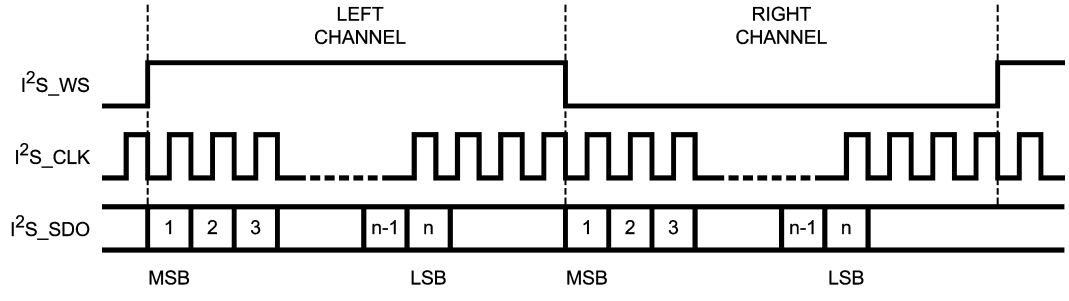
NOTES:

The master I2S format depends on the DAC mode. In USB mode the number of bits per word is 25 (i.e. 2.4MHz for a 48kHz sample rate). The duty cycle is 40/60. In non-USB modes the format is 32 or 16 bits per word, depending on I2S_RESOLTION and the duty cycle is always 50-50. In slave mode it will decode any I2S compatible data stream.



I2S Mode Timing

Interface Control Register (Continued)



Left Justified Mode Timing

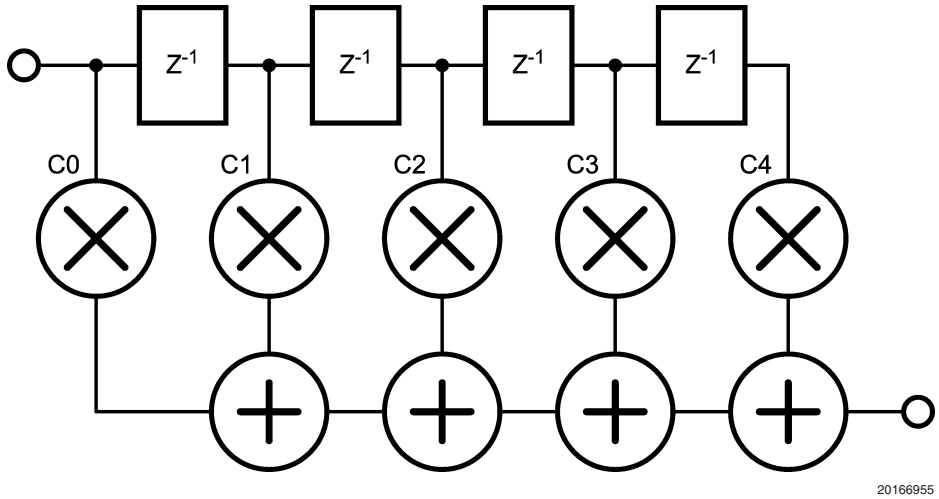
FIR Compensation Filter Configuration Registers

These registers are used to configure the DAC's FIR compensation filter. Three 16 bit coefficients are required and must be programmed via the I2C/SPI Interface in bytes as follows:

COMP_COEFF (10h → 15h) (Set = logic 1, Clear = logic 0)

Address	Register	Description
10h	COMP_COEFF0_LSB	Bits [7:0] of the 1st and 5th FIR tap (C0 and C4)
11h	COMP_COEFF0_MSB	Bits [15:8] of the 1st and 5th FIR tap (C0 and C4)
12h	COMP_COEFF1_LSB	Bits [7:0] of the 2nd and 4th FIR tap (C1 and C3)
13h	COMP_COEFF1_MSB	Bits [15:8] of the 2nd and 4th FIR tap (C1 and C3)
14h	COMP_COEFF2_LSB	Bits [7:0] of the 3rd FIR tap (C2)
15h	COMP_COEFF2_MSB	Bits [15:8] of the 3rd FIR tap (C2)

NOTES:
The filter must be phase linear to ensure the data keeps the correct stereo imaging so the second half of the FIR filter must be the reverse of the 1st half.



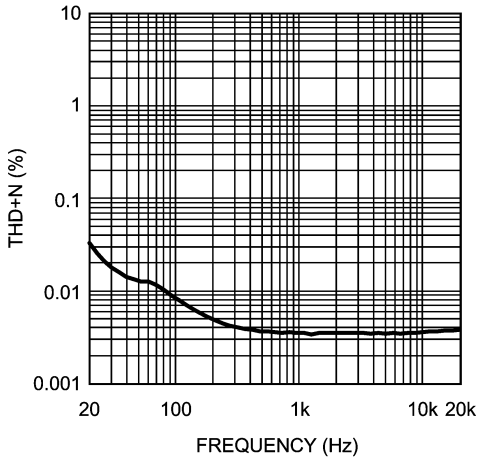
If the CUST_COMP option in register 0Eh is not set the FIR filter will use its default values for a linear response from the DAC into the analog mixer, these values are:

DAC_OSR	C0, C4	C1, C3	C2
00	68	–412	28526
01, 10, 11	112	–580	27551

If using 96 or 192kHz data then the custom compensation may be required to obtain flat frequency responses above 24kHz. The total power of any custom filter must not exceed that of the above examples or the filters within the DAC will clip. The coefficient must be programmed in 2's complement.

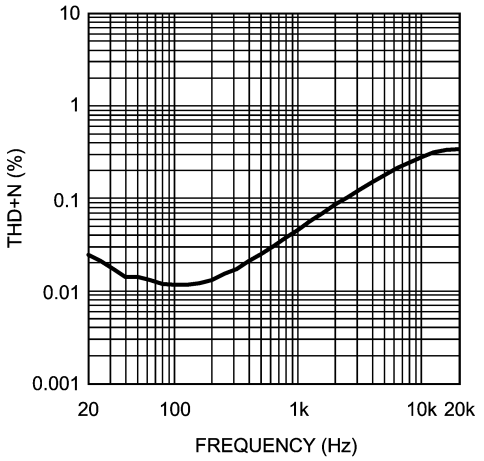
Typical Performance Characteristics

THD+N vs Frequency
3V Lineout, R_L = 10kΩ, V_O = 850mV



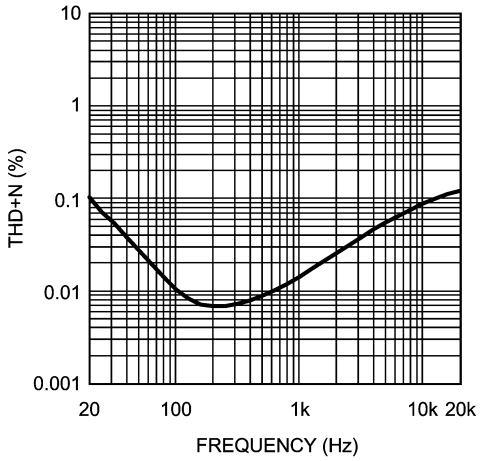
20166920

THD+N vs Frequency
3V EP Out, R_L = 32Ω, P_O = 20mW



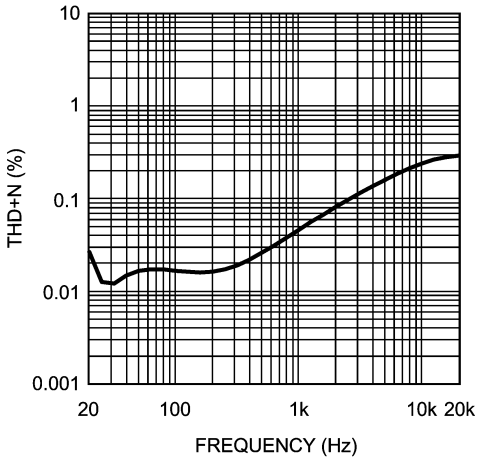
20166921

THD+N vs Frequency
3V HP Out, R_L = 16Ω, P_O = 20mW



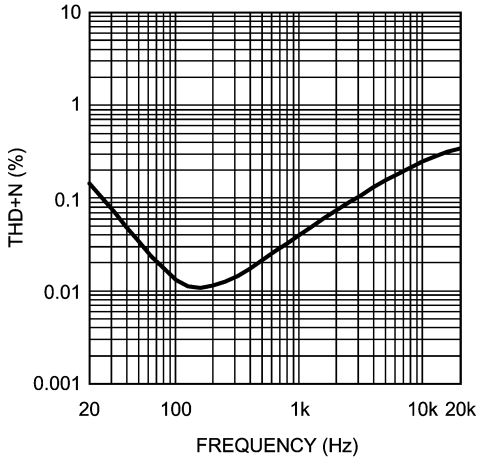
20166922

THD+N vs Frequency
3V LS Out, R_L = 8Ω, P_O = 200mW



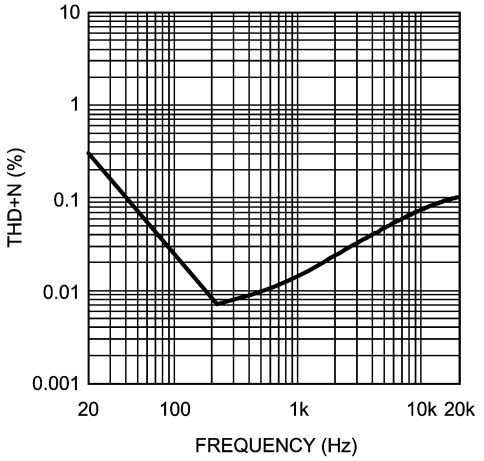
20166923

THD+N vs Frequency
5V EP, R_L = 32Ω, P_O = 40mW



20166924

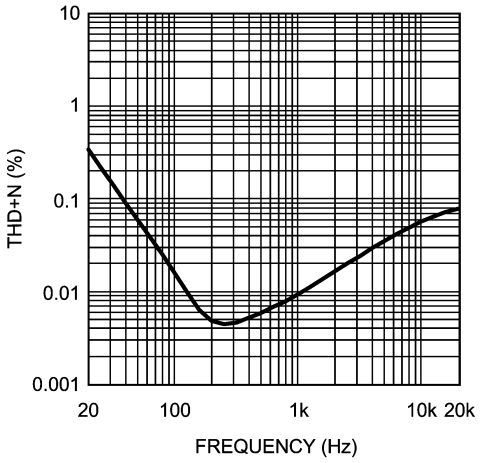
THD+N vs Frequency
5V HP Out, R_L = 16Ω, P_O = 60mW



20166926

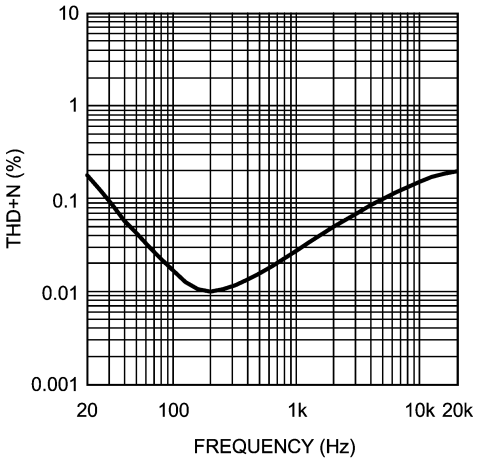
Typical Performance Characteristics (Continued)

THD+N vs Frequency
5V HP Out, R_L = 32Ω, P_O = 30mW



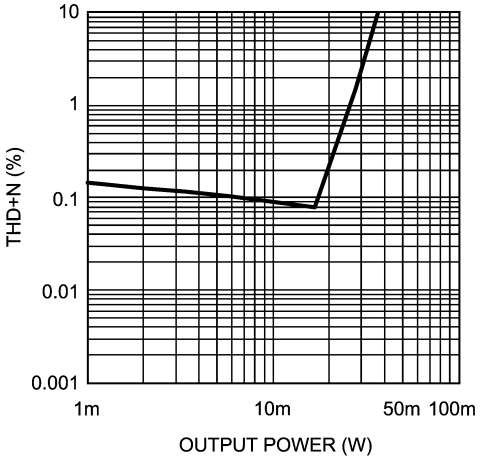
20166929

THD+N vs Frequency
5V LS Out, R_L = 8Ω, P_O = 500mW



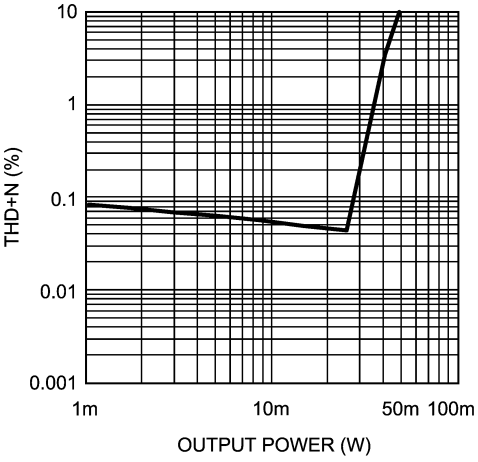
20166930

THD+N vs Output Power
3V EP Out, R_L = 16Ω, f = 1kHz



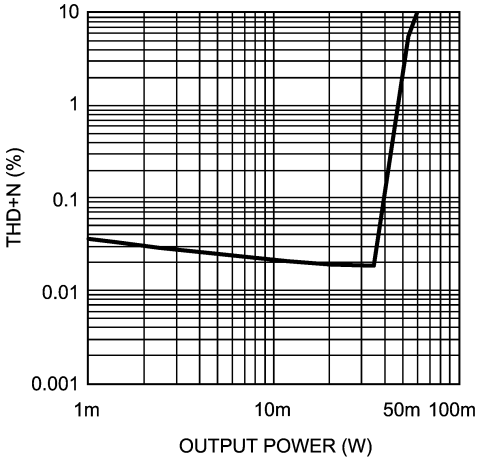
201669J8

THD+N vs Output Power
3V EP Out, R_L = 32Ω, f = 1kHz



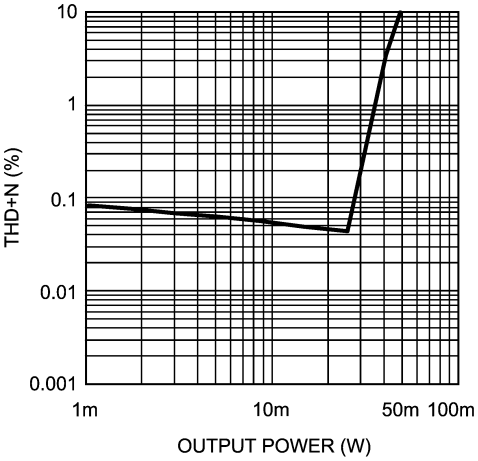
201669J9

THD+N vs Output Power
3V HP Out, R_L = 16Ω, f = 1kHz



201669K0

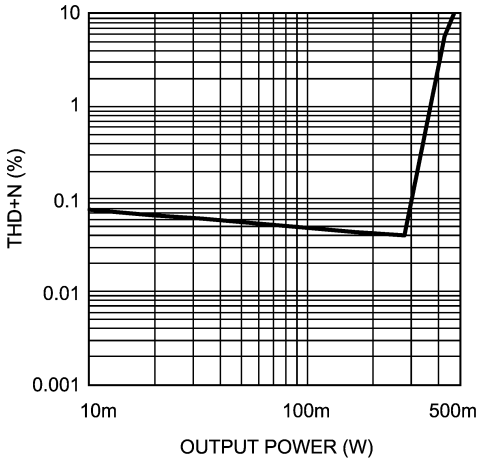
THD+N vs Output Power
3V HP Out, R_L = 32Ω, f = 1kHz



201669K1

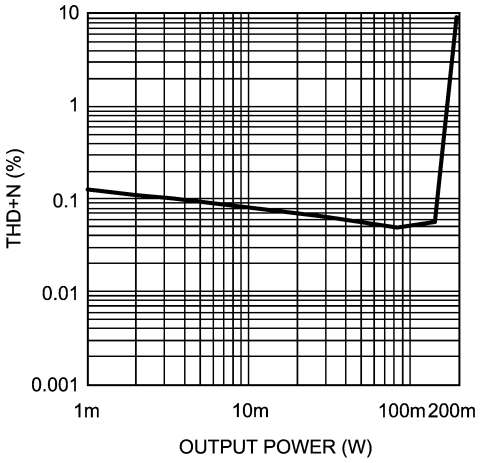
Typical Performance Characteristics (Continued)

THD+N vs Output Power
3V LS Out, R_L = 8Ω, f = 1kHz



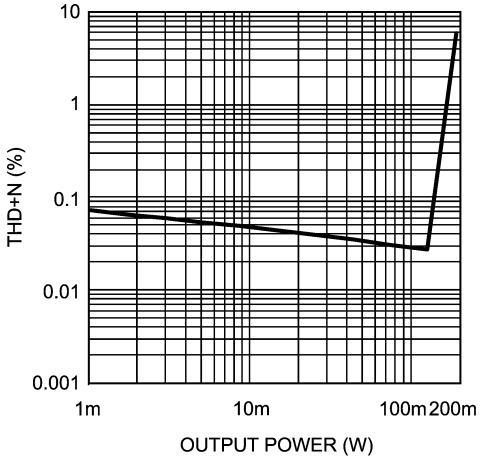
20166975

THD+N vs Output Power
5V EP Out, R_L = 16Ω, f = 1kHz



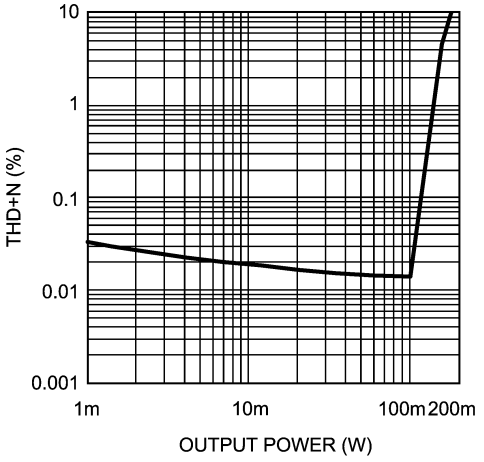
201669K2

THD+N vs Output Power
5V EP Out, R_L = 32Ω, f = 1kHz



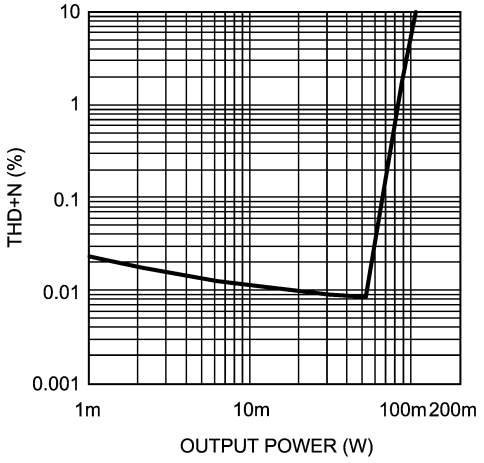
201669K3

THD+N vs Output Power
5V HP Out, R_L = 16Ω, f = 1kHz



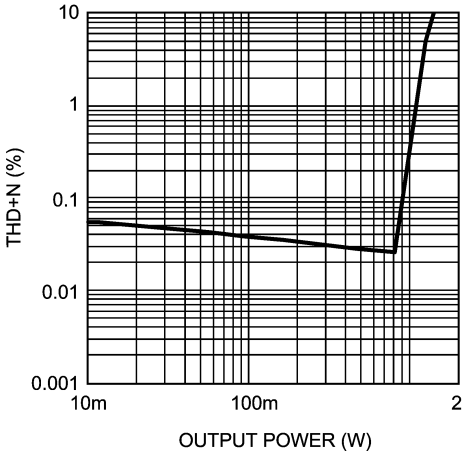
201669K4

THD+N vs Output Power
5V HP Out, R_L = 32Ω, f = 1kHz



201669K5

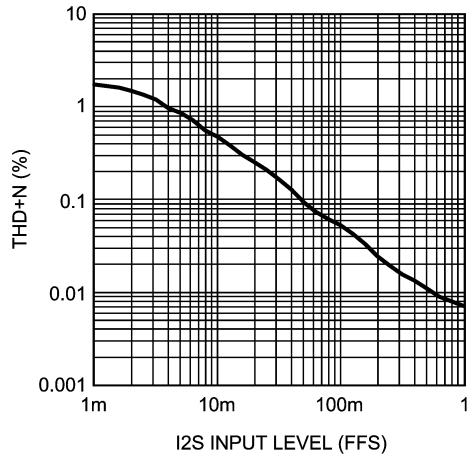
THD+N vs Output Power
5V LS Out, R_L = 8Ω, f = 1kHz



201669I9

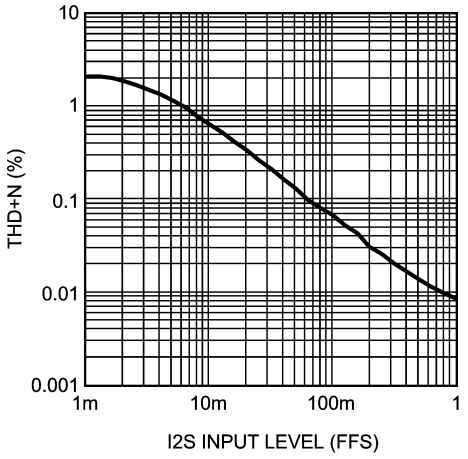
Typical Performance Characteristics (Continued)

THD+N vs I2S Level
EP Out



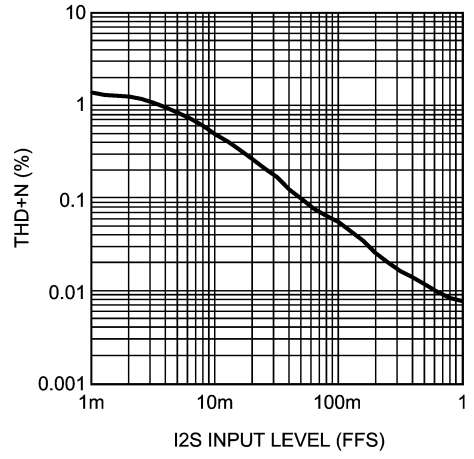
201669B1

THD+N vs I2S Level
HP Out



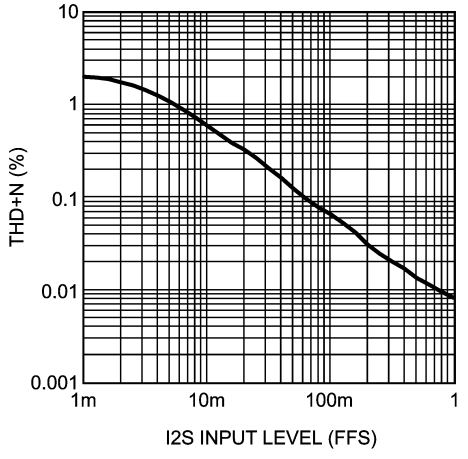
201669B2

THD+N vs I2S Level
Line Out



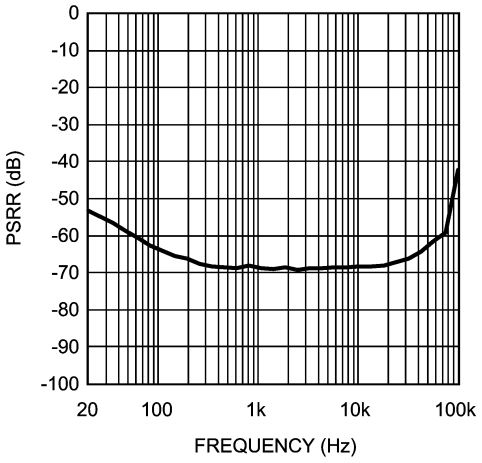
201669B3

THD+N vs I2S Level
LS Out



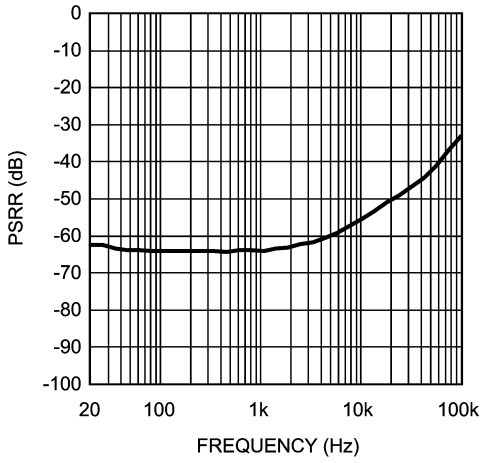
201669B4

PSRR vs Frequency
3V EP Out Mode 1



20166904

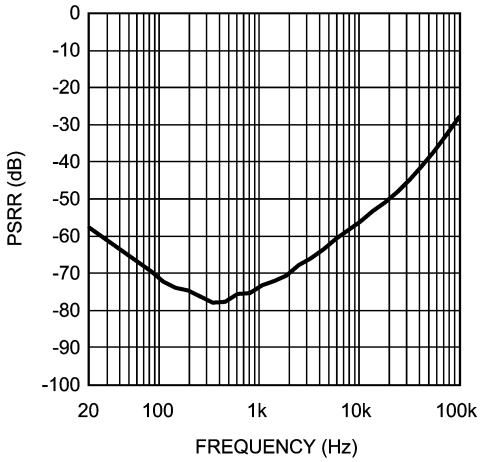
PSRR vs Frequency
3V EP Out Mode 4



20166905

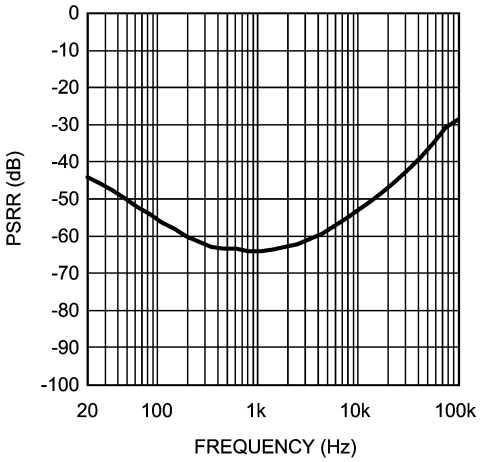
Typical Performance Characteristics (Continued)

PSRR vs Frequency
3V HP Out Mode 2



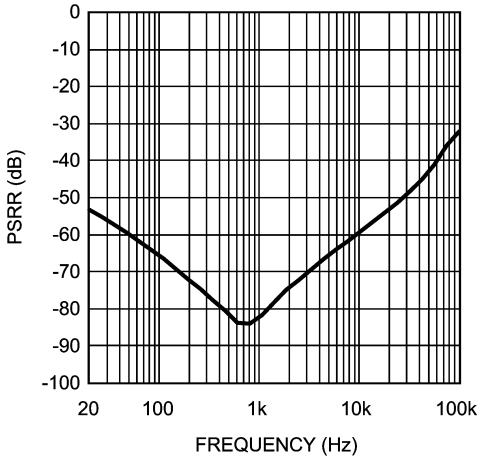
20166906

PSRR vs Frequency
3V HP Out Mode 4



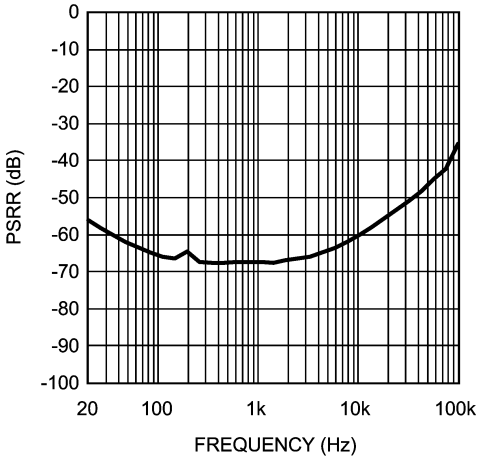
20166908

PSRR vs Frequency
3V Line Out Mode 1



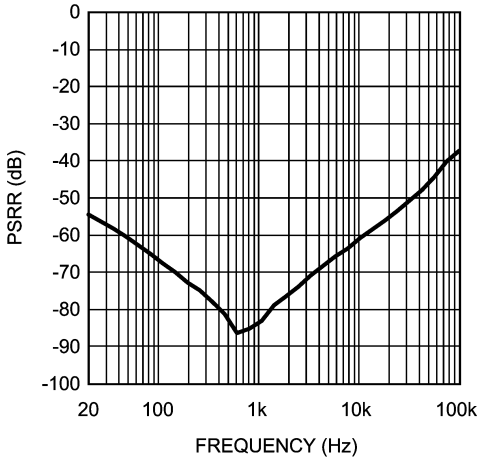
20166909

PSRR vs Frequency
3V Line Out Mode 4



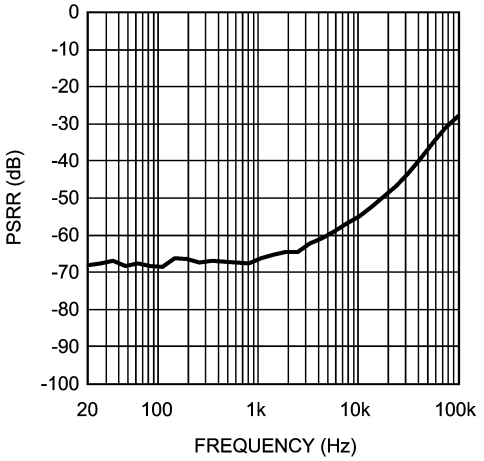
20166910

PSRR vs Frequency
3V LS Out Mode 2



20166911

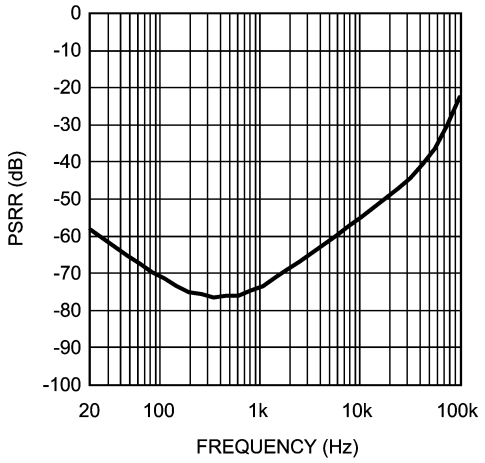
PSRR vs Frequency
3V LS Out Mode 4



20166912

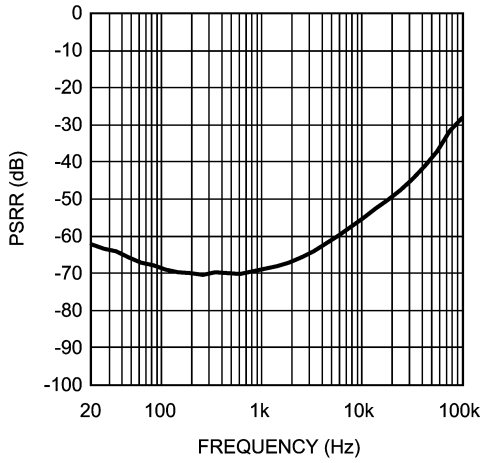
Typical Performance Characteristics (Continued)

PSRR vs Frequency
5V HP Out Mode 2



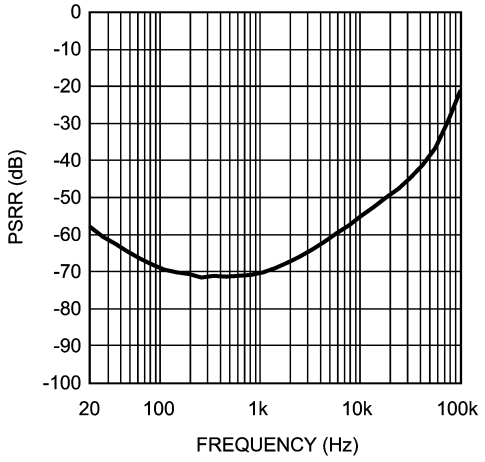
20166913

PSRR vs Frequency
5V HP Out Mode 4



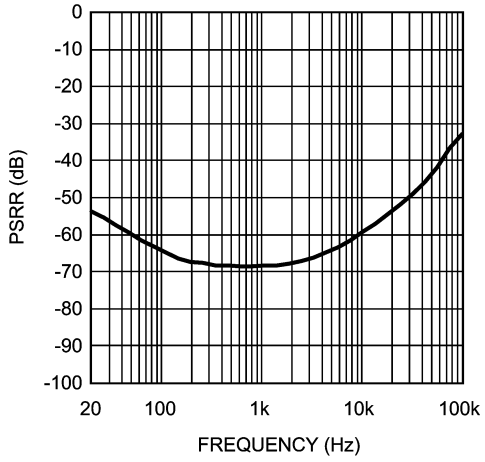
20166914

PSRR vs Frequency
5V Line Out Mode 1



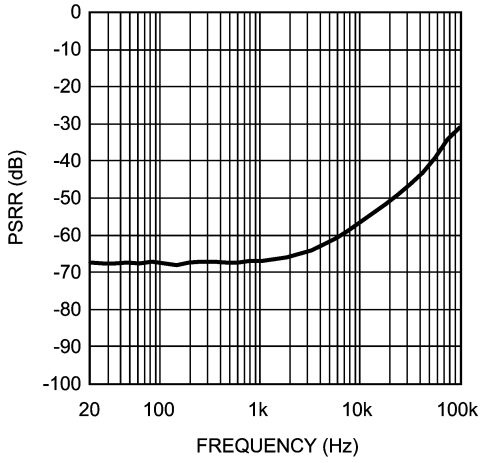
20166915

PSRR vs Frequency
5V Line Out Mode 4



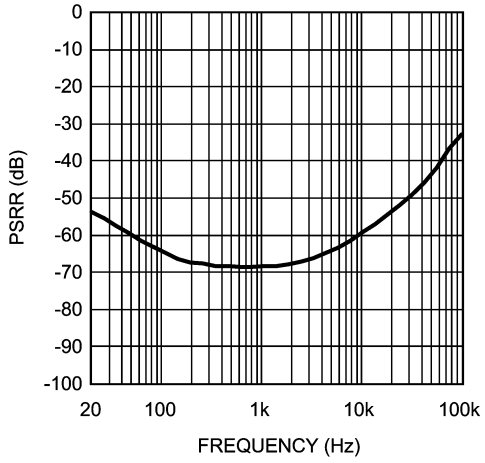
20166916

PSRR vs Frequency
5V LS Out Mode 4



20166917

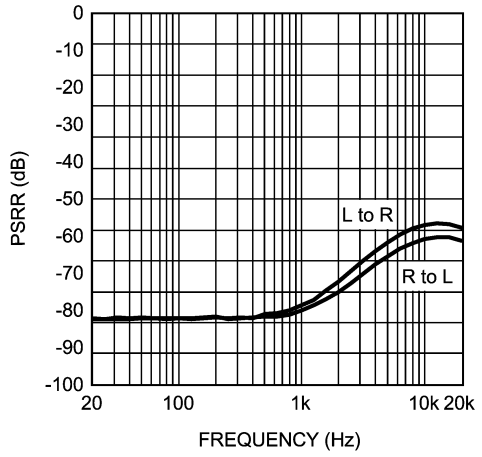
PSRR vs Frequency
5V LS Out Mode 2



20166918

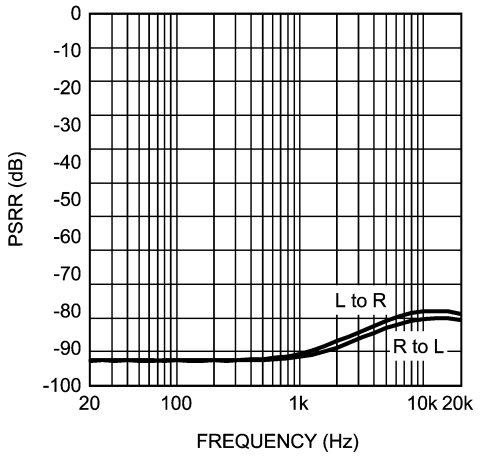
Typical Performance Characteristics (Continued)

XTalk vs Frequency
5V HP Out Mode 2, R_L = 32Ω, 1V_{RMS}, SE



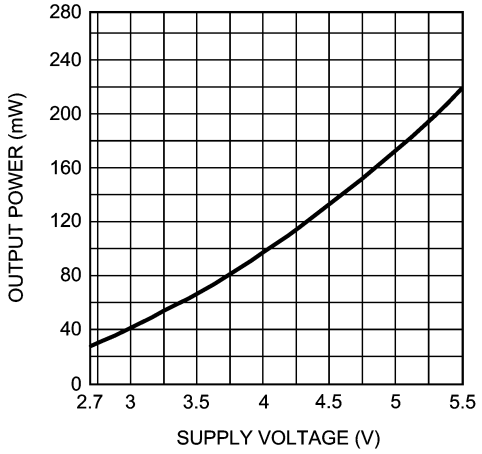
201669B5

XTalk vs Frequency
5V LS Out Mode 2, R_L = 8Ω, 1V_{RMS}



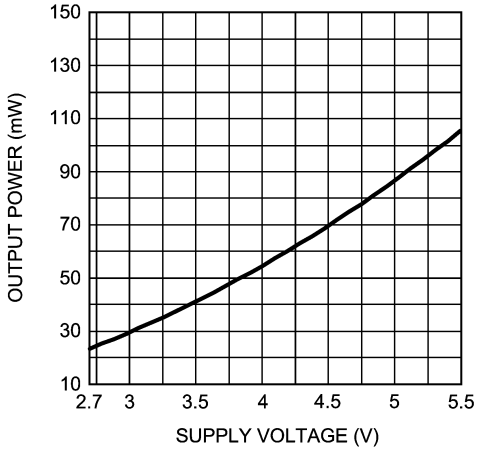
201669B6

Output Power vs Supply Voltage
EP Out, R_L = 32Ω, 1% THD+N



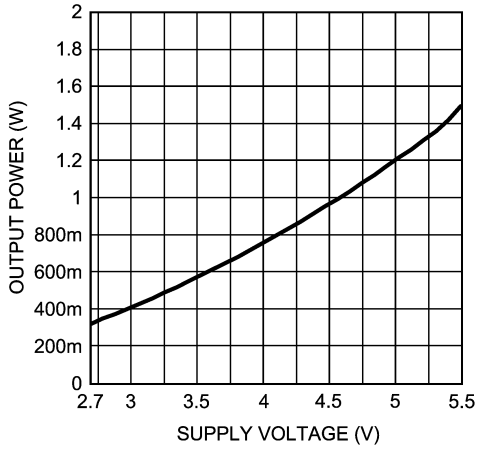
201669J0

Output Power vs Supply Voltage
HP Out, R_L = 32Ω, 1% THD+N



201669J1

Output Power vs Supply Voltage
LS Out, R_L = 8Ω, 1% THD+N



201669J2

Application Information

I²S

The LM4934 supports both master and slave I2S transmission at either 16 or 32 bits per word at clock rates up to 3.072MHz (48kHz stereo, 32bit). The basic format is shown below:

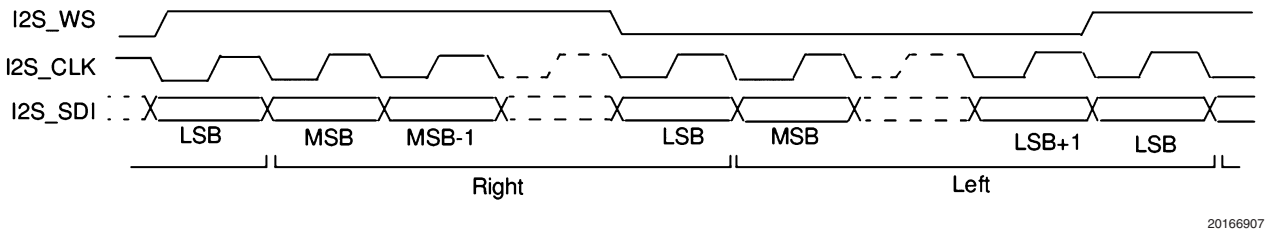


FIGURE 5.

NATIONAL SEMICONDUCTOR 3D AUDIO ENHANCEMENT

The LM4934 utilizes a programmable gain version of National Semiconductor's 3D audio enhancement circuit. This allows 3D gain only (not frequency response) to be controlled via I2C/SPI in the National 3D Enhancement Level Select registers (3D1 and 3D0). Also, this circuit uses the same 3D path for both the headphone and stereo loudspeaker outputs, so the 3D effect remains constant when switching from headphone to stereo loudspeaker outputs unless changed in the registers. An added benefit of this is that the gain of the original signal is unaffected when 3D is turned on/off.

3D gain is established internally with R_{3D} (approximately 30kΩ) and externally with C_{3D}. Typical values for C_{3D} are around 0.22µF, but may varied for altered 3D response.

Gain Considerations

When using the mixer and 2,3,4, or 5 channels are summed into the stereo output (headphone or speaker), the gain of each individual input is automatically reduced by 1/N, where N is the number of channels being summed. This has the effect of maintaining the total signal output level for different modes (i.e.; when LIN and RIN are summed for a mono output, gain for RIN and LIN will each be reduced by 6dB). This is not true for mono output modes, like EP and lineout. For these cases, stereo inputs are treated as one input with a –6dB gain for each input before summing this with a mono input. An example of relative output levels for each mode is given below:

Mode	Mono Out	Stereo R Out	Stereo L Out
1	M	M	M
2	(AL/2)+(AR+2)	AR	AL
3	[M+(AL/2)+(AR/2)]/2	(M+AR)/2	(M+AL)/2
4	(DL/2)+(DR/2)	DR	DL
5	[(AL/2)+(AR/2)+(DL/2)+(DR/2)]/2	(AR+DR)/2	(AL+DL)/2
6	[M+(AL/2)+(AR/2)+(DL/2)+(DR/2)]/2	(M+AR+DR)/3	(M+AL+DL)/3
7	[M+(DL/2)+(DR/2)]/2	(M+DR)/2	(M+DL)/2

LM4934 DEMOBOARD OPERATION

BOARD LAYOUT

DIGITAL SUPPLIES

JP14 — Digital Power DVDD

JP14 — I/O Power IOVDD

JP14 — PLL Supply PLLVDD

JP14 — USB Board Supply BBVDD

JP14 — I2C VDD

All supplies may be set independently. All digital ground is common. Jumpers may be used to connect all the digital supplies together.

S9 – connects VDD_PLL to VDD_D

S10 – connects VDD_D to VDD_IO

S11 – connects VDD_IO to VDD_I2C

Application Information (Continued)

S12 – connects VDD_I2C to Analog VDD
S17 – connects BB_VDD to USB3.3V (from USB board)
S19 – connects VDD_D to USB3.3V (from USB board)
S20 – connects VDD_D to SPDIF receiver chip

ANALOG SUPPLY

JP11 — Analog Supply
S12 — connects Analog VDD with Digital VDD (I2C_VDD)
S16 — connects Analog Ground with Digital Ground
S21 — connects Analog VDD to SPDIF receiver chip

INPUTS

Analog Inputs

JP2 — Mono Input
JP6 — Left Input
JP7 — Right Input

Digital Inputs

JP19 — Digital Interface
Pin 1 — MCLK
Pin 2 — I2S_CLK
Pin 3 — I2S_SDI
Pin 4 — I2S_WS
JP20 — Toslink SPDIF Input
JP21 — Coaxial SPDIF Input

Coaxial and Toslink inputs may be toggled between by use of S25. Only one may be used at a time. Must be used in conjunction with on-board SPDIF receiver chip.

OUTPUTS

JP4 — Right BTL Loudspeaker Output
JP5 — Left BTL Loudspeaker Output
JP1 — Left Headphone Output (Single-Ended or OCL)
JP3 — Right Headphone Output (Single-Ended or OCL)
P1 — Stereo Headphone Jack (Same as JP1, JP2, Single-Ended or OCL)
JP12 — Mono BTL Earpiece Output
JP8 — Single-Ended Line Level Output

CONTROL INTERFACE

X1, X2 – USB Control Bus for I2C/SPI

X1
Pin 9 – Mode Select (SPI or I2C)

X2
Pin 1 – SDA
Pin 3 – SCL
Pin 15 – ADDR/END
Pin 14 – USB5V
Pin 16 – USB3.3V
Pin 16 – USB GND

Application Information (Continued)

MISCELLANEOUS

I2S BUS SELECT

S23, S24, S26, S27 – I2S Bus select. Toggles between on-board and external I2S (whether on-board SPDIF receiver is used). All jumpers must be set the same. Jumpers on top two pins selects external bus (JP19). Jumpers on bottom two pins selects on-board SPDIF receiver output.

HEADPHONE OUTPUT CONFIGURATION

Jumpers S1, S2, S3, and S4 are used to configure the headphone outputs for either cap-coupled outputs or output capacitorless (OCL) mode in addition to the register control internal to the LM4934 for this feature. Jumpers S1 and S3 bypass the output DC blocking capacitors when OCL mode is required. S2 connects the center amplifer HPCOUT to the headphone ring when in OCL mode. S4 connects the center ring to GND when cap-coupled mode is desired. S4 must be removed for OCL mode to function properly. Jumper settings for each mode:

OCL

S1 = ON

S2 = ON

S3 = ON

S4 = OFF

Cap-Coupled

S1 = OFF

S2 = OFF

S3 = OFF

S4 = ON

PLL FILTER CONFIGURATION

The LM4934 demo board comes with a simple filter setup by connecting jumpers S5 and S6. Removing these and connecting jumpers S7 and S8 will allow for an alternate PLL filter configuration to be used at R2 and C23.

ON-BOARD SPDIF RECEIVER

The SPDIF receiver present on the LM4934 demo board allows quick demonstration of the capabilities of the LM4934 by using the common SPDIF output found on most CD/DVD players today. There are some limitations in its useage, as the receiver will not work with digital supplies of less than 3V and analog supplies of less than 4V. This means low analog supply voltage testing of the LM4934 must be done on the external digital bus.

The choice of using on-board or external digital bus is made usign jumpers S23, S24, S26, and S27 as described above.

S25 selects whether the Toslink or Coaxial SPDIF input is used. The top two pins connects the toslink, the bottom two connect the coaxial input.

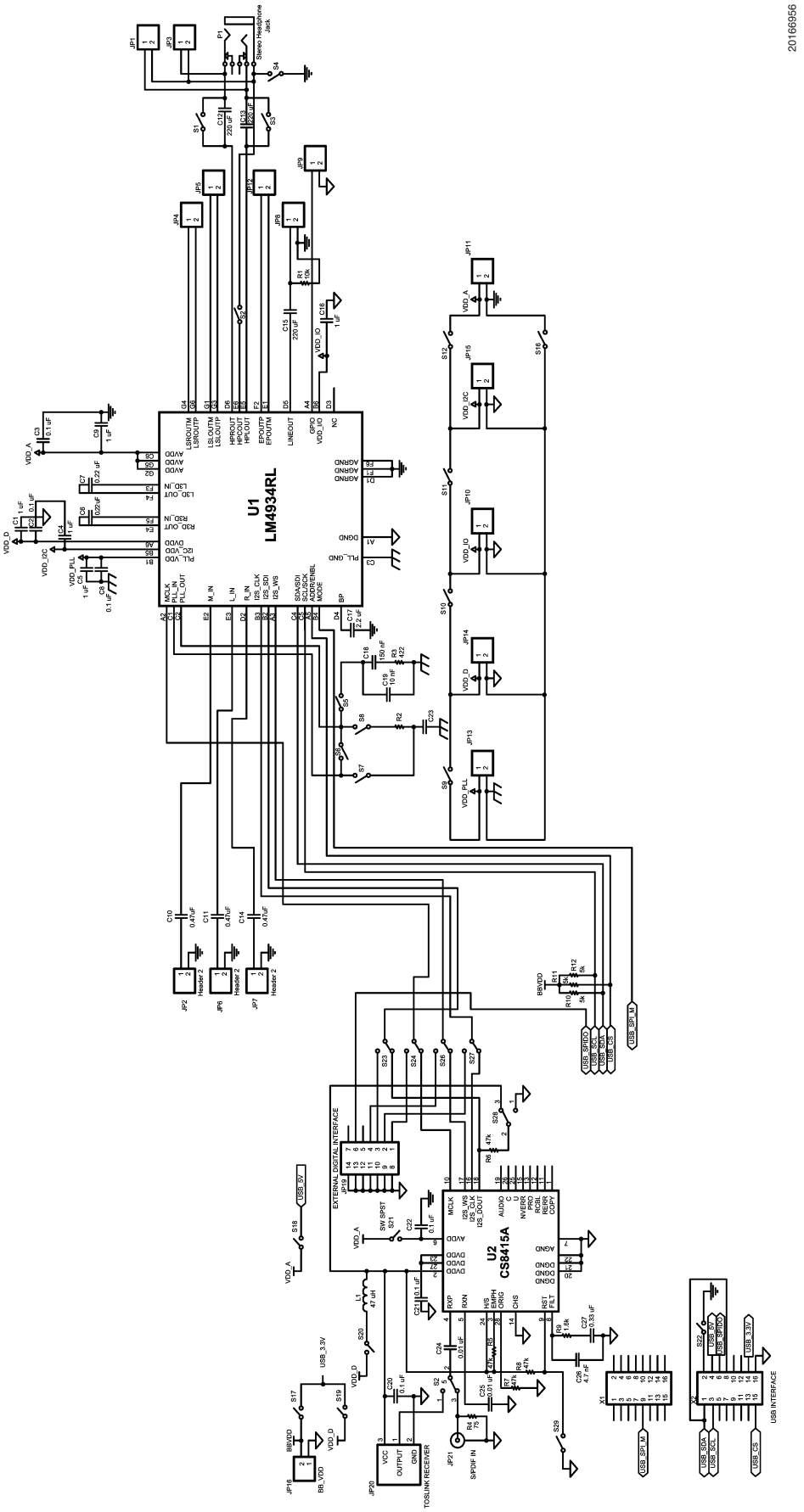
Power on the digital side is routed through S20 (connecting to the other digital supplies), while on the analog side it is interrupted by S21. Both jumpers must be in place for the receiver to function. The part is already configured for I2S standard outputs. Jumper S28 allows the DATA output to be pulled either high or low. Default is high (jumper on right two pins).

It may be necessary to quickly toggle S29 to reset the receiver and start it working upon initial power up.. A quick short across S29 should clear this condition.

LM4934 I^2C/SPI INTERFACE SOFTWARE

Convenient graphical user interface software is available for demonstration purposes of the LM4934. It allows for either SPI or I^2C control via either USB or parallel port connections to a Windows computer. Control options include all mode and output settings, volume controls, PLL and DAC setup, FIR setting and on-the-fly adjustment by an easy to use graphical interface. An advanced option is also present to allow direct, register-level commands. Software is available from www.national.com and is compatible with Windows operating systems of Windows 98 or more (with USB support) with the latest .NET updates from Microsoft.

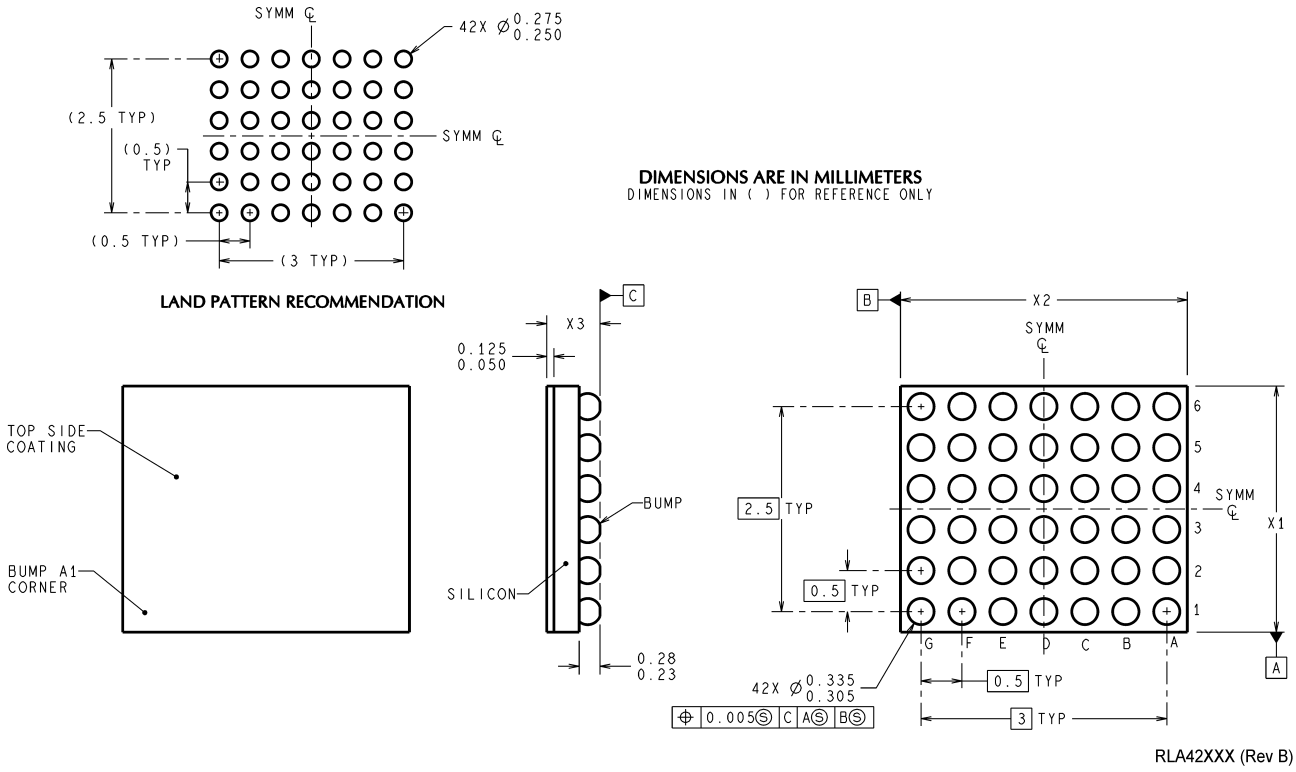
Demonstration Board Schematic



Revision History

Rev	Date	Description
1.0	9/22/05	Started D/S by copying LM4931 (DS201009). Did major edits.
1.1	9/27/05	Input some text/Typical/Limits on the EC tables.
1.2	10/6/05	Added table 1, 2, and 3. Input some text edits also.
1.3	10/11/05	Input more edits.
1.4	10/12/05	First WEB release of the D/S.
1.5	10/13/05	D/S was taken out of the WEB per Daniel.
1.6	10/19/05	Text edits and curves.
1.7	10/21/05	Added K6 (by Diane T.), will release to the per Daniel.
1.8	10/24/06	Fixed typos, then released to the WEB.
1.9	11/10/05	Added the internal DAC SNR (with 95dB typ) under Key Spec and into the Digital EC table.
2.0	11/15/05	Added the SNR DAC, then re-webd per Daniel.
2.1	12/14/05	Removed the WL package and replaced it with the RL.
2.2	12/19/05	Removed the WL package and replaced it with the RL package (per Veronica and Daniel A.), then released D/S to the WEB.
2.3	1/19/06	Edited 20166956 (board schem, changed WL to RL), X1, X2,and X3 values.
2.4	2/13/06	Switched the labels of B3 and B2 on the Demo Board Schematic (pg 37) per Daniel.

Physical Dimensions inches (millimeters) unless otherwise noted



42-Bump micro SMDxt
Order Number LM4934RL
NS Package Number RLA42MPA
X_1 = 3.245±0.030mm, X_2 = 3.796±0.030mm, X_3 = 0.650±0.075mm

National Semiconductor
Americas Customer
Support Center
Email: new.feedback@nsc.com
Tel: 1-800-272-9959

www.national.com

National Semiconductor
Europe Customer Support Center
Fax: +49 (0) 180-530 85 86
Email: europe.support@nsc.com
Deutsch Tel: +49 (0) 69 9508 6208
English Tel: +44 (0) 870 24 0 2171
Français Tel: +33 (0) 1 41 91 8790

National Semiconductor
Asia Pacific Customer
Support Center
Email: ap.support@nsc.com

National Semiconductor
Japan Customer Support Center
Fax: 81-3-5639-7507
Email: jpn.feedback@nsc.com
Tel: 81-3-5639-7560