

Boomer[®]Audio Power Amplifier Series

Audio Sub-System with Stereo DAC, Mono Class AB Loudspeaker Amplifier, OCL/SE Stereo Headphone Output and RF Suppression

General Description

The LM49321 is an integrated audio sub-system designed for mono voice, stereo music cell phones connecting to base band processors with mono differential analog voice paths. Operating on a 3.3V supply, it combines a mono speaker amplifier delivering 520mW into an 8Ω load, a stereo headphone amplifier delivering 36mW per channel into a 32Ω load, and a mono earpiece amplifier delivering 55mW into a 32Ω load. The headphone amplifier can be configured for output capacitor-less (OCL) or single-ended (SE) mode. It integrates the audio amplifiers, volume control, mixer, and power management control all into a single package. In addition, the LM49321 routes and mixes the single-ended stereo and differential mono inputs into multiple distinct output modes. The LM49321 features an I²S serial interface for full range audio and an I²C or SPI compatible interface for control. The full range music path features an SNR of 85dB with up to 192kHz playback.

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

Key Specifications

P _{OUT} LS, 8Ω, 3.3V, 1% THD+N	520mW (typ)
P _{OUT} HP, 32Ω, 3.3V, 1% THD+N	36mW (typ)
■ P _{OUT} Mono Earpiece, 32Ω 1% THD+N	55mW (typ)
Shutdown current	0.6µA (typ)
SNR (DAC + Amplifier)	85dB (typ)

August 31, 2009

Features

- 18-bit stereo DAC with up to 192kHz sampling rate
- Multiple distinct output modes
- Mono class AB speaker amplifier
- Stereo OCL/SE headphone amplifier
- Mono earpiece amplifier
- Differential mono analog input
- Single-ended analog inputs
- Independent loudspeaker, headphone and mono earpiece volume controls
- I²C/SPI (selectable) compatible interface
- Ultra low shutdown current
- Click and Pop Suppression circuit

Applications

- Cell Phones
- PDAs
- Laptop computers
- Portable devices

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FIGURE 1B: Typical Audio Amplifier Subsystem Application circuit with Cap-C`oupled single-ended (SE) Headphone configuration

LM49321

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Pin [Pin Descriptions 查询"LM49321"供应商					
Pin	Pin Name	Digital/ Analog	I/O, Power	Description		
A1	DGND	D	Р	DIGITAL GND		
A2	MCLK	D	I	MASTER CLOCK		
A3	I ² S_WS	D	I/O	I2S WORD SELECT		
A4	SDA/SDI	D	I/O	I ² C SDA OR SPI SDI		
A5	DV _{DD}	D	Р	DIGITAL SUPPLY VOLTAGE		
A6	I/O_V _{DD}	D	Р	I/O SUPPLY VOLTAGE		
B1	PLL_VDD	D	Р	PLL SUPPLY VOLTAGE		
B2	I ² S_SDATA	D	I	I ² S SERIAL DATA INPUT		
B3	I ² S_CLK	D	I/O	I ² S CLOCK SIGNAL		
B4	GPIO	D	0	TEST PIN (MUST BE LEFT FLOATING)		
B5	I ² C_V _{DD}	D	Р	I ² C SUPPLY VOLTAGE		
B6	SDL/SCK	D	I	I ² C_SCL OR SPI_SCK		
C1	PLL_GND	D	Р	PHASE LOCK LOOP GROUND		
C2	PLL_OUT	D	0	PHASE LOCK LOOP FILTER OUTPUT		
C3	PLL_IN	D	I	PLL FILTER INPUT		
C4	ADDR/ENB	D	I	I ² C ADDRESS OR SPI ENB DEPENDING ON MODE		
C5	BYPASS	А	I	HALF-SUPPLY BYPASS		
C6	AV _{DD}	А	Р	ANALOG SUPPLY VOLTAGE		
D1	AGND	А	Р	ANALOG GROUND		
D2	AGND	А	Р	ANALOG GROUND		
D3	NC			NO CONNECT (MUST BE LEFT FLOATING)		
D4	MODE	D	I	SELECTS BETWEEN I ² C OR SPI CONTROL		
D5	RHP	A	0	RIGHT HEADPHONE OUTPUT		
D6	СНР	A	0	HEADPHONE CENTER PIN OUTPUT (1/2 VDD or GND)		
E1	DIFF-	A	I	ANALOG NEGATIVE DIFFERENTIAL INPUT		
E2	LIN	A	I	ANALOG LEFT CHANNEL INPUT		
E3	RIN	A	I	ANALOG RIGHT CHANNEL INPUT		
E4	NC			NO CONNECT (MUST BE LEFT FLOATING)		
E5	LHP	A	0	LEFT HEADPHONE OUTPUT		
E6	AGND	A	Р	ANALOG GROUND		
F1	DIFF+	A	I	ANALOG POSITIVE DIFFERENTIAL INPUT		
F2	EP-	A	0	MONO EARPIECE- OUTPUT		
F3	EP+	A	0	MONO EARPIECE+ OUTPUT		
F4	LS-	A	0	LOUDSPEAKER OUTPUT-		
F5	AV _{DD}	A	Р	ANALOG SUPPLY VOLTAGE		
F6	LS+	A	0	LOUDSPEAKER OUTPUT+		

Absolute Maximum Ratings (Note 1, Note 資調"LM49321"供应商

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

Analog Supply Voltage (Note 1)	6.0V
Digital Supply Voltage (Note 1)	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 Ratings (Note 4)	2000V
ESD Ratings (Note 5)	200V
Junction Temperature (T _{JMAX})	150°C
Thermal Resistance	

θ_{JA} (RLA36) Soldering Information See AN-1279 "Microfill Wafer Level Underfilled Chip Scale package."

Operating Ratings (Note 1, Note 2)

Temperature Range	
$T_{MIN} \le T_A \le T_{MAX}$	–40°C ≤ T _A ≤ +85°C
Supply Voltage	
	$2.7V \le AV_{DD} \le 5.5V$
	$2.7 \text{V} \le \text{DV}_{\text{DD}} \le 4.0 \text{V}$
	$1.7V \le I^2C_V_{DD} \le 4.0V$

 $1.7V \le I/O_V_{DD} \le 4.0V$

100°C/W

Audio Amplifier Electrical Characteristics $AV_{DD} = 3.0V$, $DV_{DD} = 3.0V$ (*Note 1, Note 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 $T_A = 25^{\circ}$ C.

	Parameter		LM49321		Unito
Symbol		Conditions	Typical (<i>Note 6</i>)	Limits (<i>Note 7</i>)	(Limits)
		V _{IN} = 0, No Load All Amps On + DAC, OCL (Note 10)	13	18	mA (max)
		Headphone Mode Only, OCL, DAC off	4.6	6.25	mA (max)
		Headphone Mode Only, OCL, DAC Off STEREO_OUTPUT_ONLY = 1, STEREO_INPUT_ONLY = 1	4	5.5	mA
I _{DD}	Supply Current	Headphone Mode only OCL, DAC On, OSR = 64, DAC_INPUT_ONLY = 1 STEREO_OUTPUT_ONLY = 1	7.5	10	mA (max)
		Mono Loudspeaker Mode Only (Note 11)	6.5	11.5	mA (max)
		Mono Earpiece Speaker Mode Only MONO_ONLY = 1 (register 01h) MONO_ONLY = 0	3.7 3.3	5	mA (max) mA
		DAC Off, All Amps On (OCL) (Note 10)	10	13.5	mA (max)
I _{SD}	Shutdown Current	(Note 8)	0.6	1	μA (max)
	Output Power	Speaker; THD = 1%; f = 1kHz, 8Ω BTL	420	370	mW (min)
Po		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		V _{RMS}
		Speaker; P _O = 200mW; f = 1kHz, 8Ω BTL	0.04		%
THD+N	Total Harmonic Distortion+Noise	Headphone; P _O = 10mW; f = 1kHz, 32Ω SE	0.01		%
		Earpiece; P _O = 20mW; f = 1kHz, 32Ω BTL	0.04		%
		Speaker	10	55	mV (max)
V _{OS}	Offset Voltage	Earpiece	8	50	mV (max)
		Headphone (OCL)	8	15	mV (max)
∈ _O	Output Noise	A-weighted; 0dB gain	Table 1		
PSRR	Power Supply Rejection Ratio	$f = 217Hz; V_{RIPPLE} = 200mV_{P-P}$ $C_{B} = 2.2\mu F$	Table 2		

			LM49321		Unite
Symbo)"LM4932 Pa供加速商	Conditions	Typical (<i>Note 6</i>)	Limits (<i>Note 7</i>)	(Limits)
X _{TALK}	Crosstalk	Headphone; P _O = 10mW, f = 1kHz; OCL	-60		dB
T _{WU}	Wake-Up Time	$C_{B} = 2.2 \mu F, CD_{6} = 0$	35		ms
		$C_{B} = 2.2 \mu F, CD_{6} = 1$	85		ms
CMRR	Common-Mode Rejection Ratio	f = 217Hz, V_{RMS} = 200m V_{PP}	56		dB

Audio Amplifier Electrical Characteristics AV_{DD} = 5.0V, DV_{DD} = 3.3V (*Note 1, Note 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 $T_A = 25^{\circ}C$.

	Parameter		LM49321		Unite
Symbol		Conditions	Typical (<i>Note 6</i>)	Limits (<i>Note 7</i>)	(Limits)
		V _{IN} = 0, No Load All Amps On + DAC, OCL (Note 10)	17.5		mA (max)
		Headphone Mode Only, OCL, DAC Off	5.8		mA (max)
		Headphone Mode Only, OCL, DAC Off STEREO_OUTPUT_ONLY = 1, STEREO_INPUT_ONLY = 1	5.5		mA
DD	Supply Current	Headphone Mode Only, OCL, DAC On, OSR = 64, DAC_INPUT_ONLY = 1 STEREO_OUTPUT_ONLY = 1	9.5		mA
		Mono Loudspeaker Mode Only (Note 10)	11.6		mA
		Mono Earpiece Mode Only (Note 10)	5		mA
		DAC Off, All Amps On (OCL) (Note 10)	12.9		mA
I _{SD}	Shutdown Current	(Note 8)	1.6		μΑ
	Output Power	Speaker; THD = 1%; f = 1kHz, 8Ω BTL	1.25		mW
P _o		Headphone; THD = 1%; f = 1kHz, 32 Ω SE	80		mW
		Earpiece; THD = 1%; f = 1kHz, 32 Ω BTL	175		mW
V _{FS DAC}	Full Scale DAC Output		2.4		V _{RMS}
	Total Harmonic Distortion + Noise	Speaker; P _O = 500mW; f = 1kHz, 8Ω BTL	0.03		%
THD+N		Headphone; $P_0 = 30mW$; f = 1kHz, 32 Ω SE	0.01		%
		Earpiece; $P_0 = 40$ mW; f = 1kHz, 32 Ω BTL	0.04		%
		Speaker	10		mV
V _{OS}	Offset Voltage	Earpiece	8		mV
		HP (OCL)	8		mV
€o	Output Noise	A-weighted; 0dB gain;	Table 1		
PSRR	Power Supply Rejection Ratio	$f = 217Hz; V_{ripple} = 200mV_{P-P}$ $C_{B} = 2.2\mu F$	Table 3		
X _{TALK}	Crosstalk	Headphone; P _O = 15mW, f = 1kHz; OCL	-56		dB
		C _B = 2.2μF, CD_6 = 0	45		ms
I WU	I vvake-up Time	$C_{B} = 2.2 \mu F, CD_{6} = 1$	130		ms

Volume Control Electrical Characteristics (*Note 1*, *Note 2*) $T_A = 25^{\circ}C$. (*Note 1*, *Note 2*)

	Parameter		LM49321		Unito
Symbol		Conditions	Typical (<i>Note 6</i>)	Limits (<i>Note 7</i>)	(Limits)
		minimum goin cotting	6	-7	dB (min)
	Stereo Analog Inputs Pre-Amp Gain	Inininum gain setting	-0	-5	dB (max)
	Setting Range	movimum agin potting	15	15.5	dB (max)
PCP		maximum gain setting	15	14.5	dB (min)
FUN		minimum goin cotting	10	-13	dB (min)
	Differential Mono Analog Input Pre- Amp Gain Setting Range	minimum gain setting	-12	-11	dB (max)
		maximum gain setting	9	9.5	dB (max)
				8.5	dB (min)
	Output Volume Control for Loudspeaker, Headphone Output, or Earpiece Output	minimum gain setting	-56	-59	dB (min)
				-53	dB (max)
VCR		maximum gain setting		4.5	dB (min)
			+5	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, Headphone	-90		dB
	DIFF+, DIFF-, L _{IN} and R _{IN} Input			18	$k\Omega$ (min)
R _{INPUT}	Impedance		23	28	kΩ (max)

Digital Section Electrical Characteristics (Note 1, Note 2)

The following specifications apply for $3.0V \le AV_{DD} \le 5.0V$ and $2.7V \le DV_{DD} \le 4.0V$, unless otherwise specified. Limits apply for T_A = 25°C.

			LM49321		Unito	
Symbol	Symbol Parameter Conditions		Typical (<i>Note 6</i>)	Limits (<i>Note 7</i>)	(Limits)	
	Disitel Chutdows Current	Mode 0, DV _{DD} = 3.0V				
DI _{SD}	Digital Shutdown Current	No MCLK	0.01		μA	
DI _{DD}	Digital Power Supply Current	f _{MCLK} = 12MHz, DV _{DD} = 3.0V ALL MODES EXCEPT 0	5.3	6.5	mA (max)	
PLLI _{DD}	PLL Quiescent Current	$f_{MCLK} = 12MHz, DV_{DD} = 3.0V$	4.8	6	mA (max)	
Audio DAC (1	Typical numbers are with 6.144MHz	audio clock and 48kHz sampling frequency	у			
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	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	Table 4		dB	
SNR _{DAC}	Internal DAC SNR	A-weighted (Note 9)	95		dB	
PLL						
f	Input Fraguanay on MCLK pin		10	10	MLI-7	
IN	Input Frequency on MCLK pin		12	26		
SPI/I ² C (1.7V	$\leq I^2 C_V_{DD} \leq 2.2 V$					
f _{SPI}	Maximum SPI Frequency			1000	kHz (max)	
t _{SPISETD}	SPI Data Setup Time			250	ns (max)	
t _{SPISETENB}	SPI ENB Setup Time			250	ns (max)	

			LM4	LM49321	
Symboti	1"LM49321+"出版面面	Conditions	Typical	Typical Limits	
			(Note 6)	(Note 7)	(Linits)
t _{SPIHOLDD}	SPI Data Hold Time			250	ns (max)
t _{SPIHOLDENB}	SPI ENB Hold Time			250	ns (max)
t _{SPICL}	SPI Clock Low Time			500	ns (max)
t _{SPICH}	SPI Clock High Time			500	ns (max)
f _{CLKI2C}	I ² C_CLK Frequency			400	kHz (max)
t _{I2CHOLD}	I ² C_DATA Hold Time			250	ns (max)
t _{I2CSET}	I ² C_DATA Setup Time			250	ns (max)
V _{IH}	I2C/SPI Input High Voltage		I ² C_V _{DD}	0.7 x I²C_V _{DD}	V (min)
V _{IL}	I ² C/SPI Input Low Voltage		0	0.25 x I ² C_V _{DD}	V (max)
SPI/I2C (2.2V	$V \leq I^2 C_V_{DD} \leq 4.0 V$	•	•		
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 _{SPIHOLENB}	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 ² C_CLK Frequency			400	kHz (max)
t _{I2CHOLD}	I ² C_DATA Hold Time			100	ns (max)
t _{I2CSET}	I ² C_DATA Setup Time			100	ns (max)
V _{IH}	I ² C/SPI Input High Voltage		I ² C_V _{DD}	0.7 x I ² C_V _{DD}	V (min)
V _{IL}	I ² C/SPI Input Low Voltage		0	0.3 x I ² C_V _{DD}	V (max)
I²S (1.7V ≤ I/	O_V _{DD} ≤ 2.7V)	•	•		
		I ² S_RESOLUTION = 1	1536	6144	kHz (max)
f	I ² S_CLK Frequency	$I^2S_RESOLUTION = 0$	3072	12288	kHz (max)
'CLKI2S	I ² S_WS Duty Cycle		50	40 60	% (min) % (max)
V _{IH}	Digital Input High Voltage			0.75 x I/O_V _{DD}	V (min)
V _{IL}	Digital Input Low Voltage			0.25 x I/O_V _{DD}	V (max)
I²S (2.7V ≤ I/	 O_V _{DD} ≤ 4.0V)		I		
		12S BESOLUTION - 0	1536	6144	kHz (max)
faura			3072	12288	kHz (max)
	I ² S_WS Duty Cycle	I ² S_RESOLUTION = 1	50	40 60	% %
V _{IH}	Digital Input High Voltage			0.7 x I/O_V _{DD}	V (min)
V _{IL}	Digital Input Low Voltage			0.3 x I/O_V _{DD}	V (max)

Note 1: "Absolute Maximum Ratings" indicate limits beyond which damage to the device may occur, including inoperability and degradation of device reliability end/or performance, "Functional operation of the device and/or non-degradation at the Absolute Maximum Ratings Ratings or other conditions beyond those which the device in the Recommended Operating Conditions indicate conditions at which the device is functional and the device should not be operated beyond such conditions. All voltages are measured with respect to the ground pin, unless otherwise specified.

Note 2: The *Electrical Characteristics* tables list guaranteed specifications under the listed *Recommended Operating Conditions* except as otherwise modified or specified by the *Electrical Characteristics Conditions* and/or Notes. Typical specifications are estimations only and are not guaranteed.

Note 3: 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, applicable std. JESD22-A114C.

Note 5: Machine model, applicable std. JESD22-A115-A.

Note 6: Typical values represent most likely parametric norms at $T_A = +25^{\circ}C$, and at the *Recommended Operation Conditions* at the time of product characterization and are not guaranteed.

Note 7: Datasheet min/max specification limits are guaranteed by test or statistical analysis.

Note 8: Shutdown current is measured in a normal room environment.

Note 9: Internal DAC only with DAC modes 00 and 01.

Note 10: Enabling mono bit (MONO_ONLY in Output Control Register 01h) will save 400µA (typ) form specified current.

Units

μV

查询"LM49321"供应商 Output Noise AV_{DD} = 5.0V and AV_{DD} = 3.0V. All gains set to 0dB. Units in µV, A-weighted, Inputs terminated to ground. MODE EΡ LS 22 1 22 22 22 0

2	22	22	8	μV
3	22	22	8	μV
4	68	88	46	μV
5	38	48	24	μV
6	29	34	18	μV
7	38	48	24	μV

TABLE 1. Output Noise

HP OCL

8

TABLE 2. PSRR AV_{DD} = 3.0V

PSRR AV_{DD} = 3.0V, f_{RIPPLE} = 217Hz; $V_{RIPPLEe}$ = 200mV_{P-P}; C_B = 2.2µF; All gains set to 0dB.

			••• -			
MODE	EP(Typ)	LS (Typ)	LS (Limit)	НР (Тур)	HP (Limit)	Units
1	69	76		72		dB
2	69	76	67	72	68	dB
3	69	76		72		dB
4	63	62		55		dB
5	69	68		61		dB
6	69	70		64		dB
7	69	68		61		dB

TABLE 3. PSRR AV_{DD} = 5.0V

PSRR AV_{DD} = 5.0V, f_{RIPPLE} = 217Hz; V_{RIPPLE} = 200m V_{P-P} ; C_B = 2.2µF; All gains set to 0dB,

	···			
MODE	EP (Typ)	LS (Typ)	НР (Тур)	Units
1	68	72	71	dB
2	68	72	71	dB
3	68	72	71	dB
4	68	66	69	dB
5	68	69	70	dB
6	69	72	71	dB
7	68	69	70	dB

TABLE 4. Dynamic Range and SNR

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Dynamic Range and SNR. 3.0V \leq AV<sub>DD</sub> \leq 5.0V. All programmable gain set to 0dB. Units in dB.
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	DR (Typ)	SNR (Typ)	Units
LS	95	85	dB
HP	95	85	dB
EP	97	85	dB

System Control 函词"LM49321"供应商

The LM49321 is controlled via either a two wire I²C compatible interface or three wire SPI interface, selectable with the MODE pin. This interface is used to configure the operating mode, interfaces, data converters, mixers and amplifiers. The LM49321 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)



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)



When the part is configured as an I²C device then the LM49321 will respond to one of two addresses, according to the ADDR input. It is protocold by the address portion of the I²C 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

	A7	A6	A5	A4	A3	A2	A1	A0
Chip Address	0	EC	EC	1	0	0	0	0
ADR = 0	0	0	0	1	0	0	0	0
ADR = 1	0	1	1	1	0	0	0	0

EC — Externally configured by ADR pin

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	DO		MONO			0				7	FAST_ CLOCK						I²S_MASTEI SLAVE								
	5	ONTROL	LS_ OUTPUT	LS_ OUTPUT 0			ANA_L_GAIN	MONO_L_GA	PLL_INPUT				٩	DAC_MODE	I ² S_ RESOLUTION	-									
	D2		HP_L_ OUTPUT	EP_VOL	LS_VOL	0	HP_L_VOL	HP_R_VOL						PLL_N_MOD	PLI	MUTE_L	I2S_MODE	-							
	D3		HP_R_ OUTPUT		0				_GAIN	PLL ENABLE	PLL_M				MUTE_R	I2C_FAST	F0_LSB	Fo_MSB	F1_LSB	F1_MSB	F2_LSB	F2_MSB			
. Control Registers	D4	OCL	STEREO_INPUT_ DNLY			0			ANA_R_GAIN	DAC_L_	PLL_N		PLL_N	0	DITHER_OFF	0	ENSATION_C OEF	ENSATION_C OEF	ENSATION _C OEF	ENSATION _C OEF	ENSATION _C OEF	INSATION_C OEF			
TABLE 6	D5	0	DAC_INPUT	0	0	0	0	0		R_GAIN	R_DIV			DITHER_LEVEL	0	DITHER_ALW_ON	0	COMPI	COM	COMP	COMPE	COMPI	COM		
	D6	CD_6		0	0	0	0	0	0	DAC_F					DITHER_LEVEL	0	CUST_COMP	0	-						
	D7	0	STEREO_ OUT_ONLY	0	0	0	0	0	0	0		0		VCO_FAST	0	0	0							oower-up.	
	Register	MODE_CONTROL	OUTPUT_ CONTROL	EP_VOL	LS_VOL	RESERVED	HP_L_VOL	HP_R_VOL	ANALOG_INPUT GAIN	ANALOG_DAC _GAIN	CLOCKS	PLL_M		PLL_N_MOD	PLL_P	DAC_SET UP	INTERFACE							ters default to 0 on initial p	
	Address	400	01h	02h	03h	04h	05h	06h	07h	08h	460	0Ah	0Bh	oCh	0Dh	0Eh	0Fh	10h	11h	12h	13h	14h	15h	Note: All regis	

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14

Mixe查角qmmagy界eregion槽rs

TABLE 7. Mode Control Register (00h)

This register is used to control the different mixer modes that the LM49321 supports.

Bits	Field			Desc	ription			
3:0	MODE	This sets the different	nt mixer ou	itput modes.				
	_CONTROL	MODE_CONTROL	Mode	Mono Earpiece	Loudspeaker	Headphone Left	Headphone Right	
		0000	0	SD	SD	SD	SD	
		1001	1	М	М	М	М	
		1010	2	AL+AR	AL+AR	AL	AR	
		1011	3	M+AL+AR	M+AL+AR	M+AL	M+AR	
		1100	4	DL+DR	DL+DR	DL	DR	
		1101	5	DL+DR+AL+AR	DL+DR+AL+AR	DL+AL	DR+AR	
		1110	6	M+DL+DR+AL +AR	M+DL+DR+AL +AR	M+DL+AL	M+DR+AR	
		1111	7	M+DL+DR	M+DL+DR	M+DL	M+DR	
4	OCL	This sets the headpl	hone outpu	it to use output cap	acitor-less configu	ration.	и 	
			OCL		Headphone output configuration			
			0		Cap-coupled Single-ended Mode (SE)			
			1		Output capacitor-less (OCL)			

SD — Shutdown

M — Mono Differential Input

AL — Analog Left Channel AR — Analog Right Channel DL — I²S DAC Left Channel

DR — I²S DAC Right Channel

Note: Power-On Default Mode is Mode 0

TABLE 8. Output Control (01h)

This register is used to control the different output configurations.

Bits	Field		Description			
0	EP_OUTPUT	This enables the Mono Earp	iece output.			
		EP_OUTPUT	Status			
		0	Mono earpice output off			
		1	Mono earpice output on			
1	1 LS_OUTPUT This enables the M		lspeaker output.			
		LS_OUTPUT	Status			
		0	Loudspeaker output off			
		1	Loudspeaker output on			
2	HP_L_OUTPUT	This enables the Headphone left output.				
		HP_L_OUTPUT	Status			
		0	Headphone left output off. If OCL=1, output is in			
			mute.			
		1	Headphone left output on			
3	HP_R_OUTPUT	This enables the Headphone	e right output.			
		HP_R_OUTPUT	Status			
		0	Headphone right output off. If OCL=1, output is in			
			mute.			
		1	Headphone right output on			

Bits	Field		Description				
旬"LM ₄ 9321"1	共政的EO_INPUT_ONLY	This enables the analog left (AL) and analog right (AR) and disables all other inputs.					
		STEREO_INPUT_ONLY	Status				
		0	Normal				
		1	Enables AL and AR inputs only				
5	DAC_INPUT_ONLY	This enables the DAC left (DL	This enables the DAC left (DL) and analog right (DR) and disables all other inputs				
		DAC_INPUT_ONLY	Status				
		0	Normal				
		1	Enables DL and DR inputs only				
6	MONO_ONLY	This enables mono earpiece	(EP) and loudspeaker (LS) outputs MUX and				
		disables the headphone outputs MUX. Enabling this mode can save up to $400\mu A$					
		of current.	1				
		MONO_ONLY	Status				
		0	Normal				
		1	Enable mono earpiece and loudspeaker outputs MUX				
7	STEREO_OUTPUT_ONLY	This enables the headphone	output MUX only and disables all other output				
		MUX's. Enabling this mode of	can save up to 200µA of current.				
		STEREO_OUTPUT_ONLY	Status				
		0	Normal				
		1	Enables the headphone output MUX				

Volume Control Registers 查询"LM49321"供应商

TABLE 9. Volume Control Register EP_VOL (02h), LS_VOL (03h), HP_L_VOL (05h), HP_R_VOL (06h)

These registers are used to control output volume control levels for Earpiece, Loudspeaker and Headphone.

Bits	Field	Description			
4:0	EP_VOL LS VOL	This programs the Earpiece, Lo volume level.	oudspeaker and Headphone		
	HP_L_VOL	VOL	Level (dB)		
	HP_R_VOL	00000	MUTE		
		00001	-56		
		00010	-52		
		00011	-48		
		00100	-45		
		00101	-42		
		00110	-39		
		00111	-36		
		01000	-33		
		01001	-30		
		01010	-28		
		01011	-26		
		01100	-24		
		01101	-22		
		01110	-20		
		01111	-18		
		10000	–16		
		10001	-14		
		10010	-12		
		10011	-10		
		10100	-8		
		10101	-6		
		10110	-4		
		10111	-3		
		11000	-2		
		11001	-1		
		11010	0		
		11011	1		
		11100	2		
		11101	3		
		11110	4		
		11111	5		

TABLE 10. Analog Left and Right Input Control (07h) 查询考试验验 应读如trol input gain for left and right analog inputs. Bits Field Description 2:0 ANA_L_GAIN This program the analog left input gain. ANA_L_GAIN Level (dB) 000 -6 001 -3

		010	0					
		011	3					
		100	6					
		101	9					
		110	12					
		111	15					
5:3	ANA_R_GAIN	This program the analog Right input gain.						
		ANA_R_GAIN	Level (dB)					
		000	-6					
		001	-3					
		010	0					
		011	3					
		100	6					
		101	9					
		110	12					
		111	15					

TABLE 11. Mono and DAC Input Gain Control (08h)

This register is sued to control input gain for Mono, DAC left and right inputs.

Bits	Field	Description			
2:0	MONO_IN_GAIN	This program the mono inp	out gain.		
		MONO_IN_GAIN	Level (dB)		
		000	-12		
		001	-9		
		010	-6		
		011	-3		
		100	0		
		101	3		
		110	6		
		111	9		
4:3	DAC_L_GAIN	This program the DAC left input gain.			
		DAC_L_GAIN	Level (dB)		
		00	-3		
		01	0		
		10	3		
		11	6		
6:5	DAC_R_GAIN	This program the DAC Rig	ht input gain.		
		DAC_R_GAIN	Level (dB)		
		00	-3		
		01	0		
		10	3		
		11	6		

Clock Configuration Register 查询"LM49321"供应商 This register is used to control the multiplexers and clock R divider in the clock module.

TABLE 12. CLOCK (09h)

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 sel	ect:				
		PLL_INPUT	PLL Input Source				
		0	MCLK				
		1	I ² S 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				



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:

TABLE 13.	BLE 13.
------------------	---------

DAC MODE	Over sampling 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

TABLE 14.

Fs (kHz)	DAC Oversampling Ratio	Required CLock at DAC Clock Generator Input (MHz)
2	'LM49321"供应商	6
24	128	6.144
32	125	8
32	128	8.192
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></mhz>	Sample Rate Supported <khz></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

TABLE 15.

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

PLL Configuration Registers

PLL M DIVIDER CONFIGURATION REGISTER

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

TABLE 16. PLL_M (0Ah)

Bits	Register	Description	Description					
6:0	PLL_M	Programs the PLL input divider to select:						
		PLL_M	Divide Ratio					
		0000000	Divider Off					
		0000001	1					
		0000010	1.5					
		0000011	2					
		0000100	2.5					
		1111110	63.5					

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 division of the M divider is derived from PLL_M as such:

 $\mathsf{M} = (\mathsf{PLL}_\mathsf{M}+1) \ / \ 2$

PLL N DIVIDER CONFIGURATION REGISTER

TABLE 17. PLL_N (0Bh)

Bits	Register	Description					
7:0	PLL_N	Programs the PLL feedback divider:					
		PLL_N	Divide Ratio				
		00000000	Divider Off				
		00000001 →00001010	10				
		00001011	11				
		00001100	12				
		11111000	248				
		11111001	249				

NOTES:

The N 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.

TABLE 18. PLL_P

Bits	Register	Description				
3:0	PLL_P	Programs the PLL input divider to select:				
		0000	Divider Off			
		0001	1			
		0010	1.5			
		0011	2			
			-> 2.5			
		1101	7			
		1110	7.5			
		1111	8			

NOTES:

The output of this divider should be either 12 or 24MHz in USB mode or 11.2896MHz, 12.288MHz or 24.576MHz in non-USB modes.

The division of the P divider is derived from PLL_P as such:

 $P = (PLL_P+1) / 2$

PLL N MODULATOR AND DITHER SELECT CONFIGURATION REGISTER

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

TABLE 19. PLL_N_MOD (0Ch)

Bits	Register	Description						
4:0	PLL_N_MOD	This programs the PLL N Modulator's fractional component:						
		PLL_N_MOD Fractional Addition						
		00000 0/32						
		00001 1/32						
		00010 → 11110 2/32 → 30/32						

Bits	Register	Description						
6:5 🕯	6:5 章询员中Meleg3全V进应例llows 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)						
7	VCO_FAST	If set the VCO maximum and minim	um frequencies are raised:					
		VCO_FAST Maximum F _{VCO}						
		0	40–55MHz					

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 词"LM49321"供应商

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.



	Evample		Sottings	For	18Kh7	Sample	Rates
TADLE 20.	Example	OFLL	Settings	FUI -	40N112	Sample	nales

f_in (MHz)	fsamp (kHz)	М	N	Р	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

TABLE 21. Example PLL Settings For 44.1Khz Sample Rates

f_in (MHz)	fsamp (kHz)	М	Ν	Р	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 11.289曾昭明 12,000 明 13,289 14,000 明 14,000 H 14,0000 H 14,000 H 14,000 H 14,000 H 14,0000

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 $P = 4.0 \rightarrow 7.0$ sweep the M inputs from $2.5 \rightarrow 24$. 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 LM49321 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 LM49321 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.

DAC Setup Register

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

Bits	Register	Description			
1:0	DAC_MODE	The DAC used in the LM49321 can operate in one of 4 oversampling modes.			
		The modes are de	scribed as follows:		
		DAC_MODE	Oversampling Rate	Typical f _S	MCLK Required
		00	125	48KHz	12.000MHz (USB Mode)
		01	128	44.1KHz 48KHz	11.2896MHz 12.288MHz
		10	64	96KHz	12.288MHz
		11	32	192KHz	24.576MHz
2	MUTE_L	Mutes the left DAC channel on the next zero crossing.			
3	MUTE_R	Mutes the right DAC channel on the next zero crossing.			
4	DITHER_OFF	If set the dither in DAC is disabled.			
5	DITHER ALWAYS_ON	If set the dither in DAC is enabled all the time.			
6	CUST_COMP	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.			

TABLE 22. DAC_SETUP (0Eh)

_ **Interface Control Register** 查询"LM49321"供应商 ____This register is used to control the I²S and I²C compatible interface on the chip.

TABLE 23. INTERFACE (0Fh)

Bits	Field	Description	
0	I ² S_MASTER_SLAVE	This enables I ² S in master or slave mode.	
		I ² S_MASTER_SLAVE	Comments
			LM49321 acts as a slave
		0	where both I ² S clock and
			word select are configured as inputs.
			LM49321 acts as a master
		1	for I ² S, so both I ² S clock and
			I ² S word select are
			configured as outputs.
1	I ² S_RESOLUTION	This set the I ² S resolution an	d affects the I ² S Interface in
		master mode. In slave mode	the I ² S Interface can support
		any I ² S compatible resolution	n. In master mode the I2S
		below explains	le DAC mode as the hote
		I ² S RESOLUTION	Comments
		0	I ² S resolution is set to 16 bits.
		1	I ² S resolution is set to 32 bits.
2	I ² S_MODE	This set the I ² S mode timing.	
		I2S_MODE	Comments
		0	I ² S interface is configured in
		0	normal I ² S mode timing.
		1	I ² S is configured in left
		1	justified mode timing.
3	I ² C_FAST	This set the I ² C Clock speed.	
		I ² C_FAST	Comments
			I ² C speed gets its default
		0	value of a maximum of
			400kHz.
		1	This enables the I ² C to run in
			tast mode with an I ² C clock
			up to 3.4MHz.

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_RESOLUTION and the duty cycle is always 50-50. In slave mode it will decode any I²S compatible data stream.



FIR Compensation Filter 查询"LM49321"供应商 Configuration Registers

must be programmed via the I2C/SPI Interface in bytes as follows:

These registers are used to configure the DAC's FIR compensation filter. Three 16 bit coefficients are required and

TABLE 24. COMP_COEFF (10h → 15h)

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	434	-2291	26984
01, 10, 11	61	-371	25699

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 查询"LM49321"供应商



THD+N vs Frequency $AV_{DD} = 3.0V, LS Out, R_L = 8\Omega, P_O = 200mW$









THD+N vs Frequency $AV_{DD} = 5.0V$, HP Out, $R_L = 32\Omega$, $P_O = 30mW$



















100k

30061891

100k

30061893

100k

30061895







4.5

5

5.5

30061897

Application Information 查询"LM49321"供应商

3.072MHz (48kHz stereo, 32bit). The basic format is shown below:

The LM49321 supports both master and slave I^2S transmission at either 16 or 32 bits per word at clock rates up to



MONO ONLY SETTING

The LM49321 may be restricted to mono amplification only by setting MONO_ONLY in Output Control register 0x01h to 1. This may save an additional 400μ A from I_{DD}.

LM49321 DEMOBOARD OPERATION

BOARD LAYOUT

DIGITAL SUPPLIES

JP14 — Digital Power DVDD

- JP10 I/O Power IOVDD
- JP13 PLL Supply PLLVDD
- JP16 USB Board Supply BBVDD
- JP15 I²CVDD

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
- 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 Differential 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

JP5 — BTL Loudspeaker Output

JP1 — Left Headphone Output (Single-Ended or OCL)

- JP3 Right Headphone Output (Single-Ended or OCL)
- ${\rm P1}$ Stereo Headphone Jack (Same as JP1, JP2, Single-Ended or OCL)

JP12 — Mono BTL Earpiece 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

MISCELLANEOUS

I²S BUS SELECT

S23, S24, S26, S27 – I²S Bus select. Toggles between onboard 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 LM49321 for this feature. Jumpers S1 and S3 bypass the output DC blocking capacitors when OCL mode is required. S2 connects the center amplifier HPCOUT to the 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
- S2 = ONS3 = ON
- S4 = OFF
- Cap-Coupled
- S1 = OFF
- S2 = OFF
- S2 = OFFS3 = OFF
- 33 = OFF
- S4 = ON

PLL FILTER CONFIGURATION

The LM49321 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 LM49321 demo board allows quick demonstration of the capabilities of the LM49321 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 3.0V and analog supplies of less than 4V. This means low analog supply voltage testing of the LM49321 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.

LM49321 I²C/SPI INTERFACE SOFTWARE

Convenient graphical user interface software is available for demonstration purposes of the LM49321. It allows for either SPI or I²C 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 查询"LM49321"供应商

LM49321

查询"LM49321"供应商

Enlarged Board Schematic Part 1 of 2

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Rev	Date	Description
1.0	09/10/08	Initial release.
1.01	09/23/08	Text edits.
1.02	08/31/09	Edited the package drawing and the top markings.

查询"LM49321"供应商

Notes

Pr	oducts	Design Support		
Amplifiers	www.national.com/amplifiers	WEBENCH® Tools	www.national.com/webench	
Audio	www.national.com/audio	App Notes	www.national.com/appnotes	
Clock and Timing	www.national.com/timing	Reference Designs	www.national.com/refdesigns	
Data Converters	www.national.com/adc	Samples	www.national.com/samples	
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LVDS	www.national.com/lvds	Packaging	www.national.com/packaging	
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Switching Regulators	www.national.com/switchers	Distributors	www.national.com/contacts	
LDOs	www.national.com/ldo	Quality and Reliability	www.national.com/quality	
LED Lighting	www.national.com/led	Feedback/Support	www.national.com/feedback	
Voltage Reference	www.national.com/vref	Design Made Easy	www.national.com/easy	
PowerWise® Solutions	www.national.com/powerwise	Solutions	www.national.com/solutions	
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