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CIRRUS LOGIC®

CS4391A

24-Bit, 192 kHz Stereo DAC with Volume Control

Features

- Complete Stereo DAC System: Interpolation, D/A, Output Analog Filtering
- 108 dB Dynamic Range
- 94 dB THD+N
- Direct Stream Digital Mode
- Low Clock Jitter Sensitivity
- +5 V Power Supply
- ATAPI Mixing
- On-Chip Digital De-emphasis for 32, 44.1, and 48 kHz
- Volume Control with Soft Ramp
  - 119 dB Attenuation
  - 1 dB Step Size
  - Zero Crossing Click-Free Transitions
- Direct Interface with 5 V to 1.8 V Logic

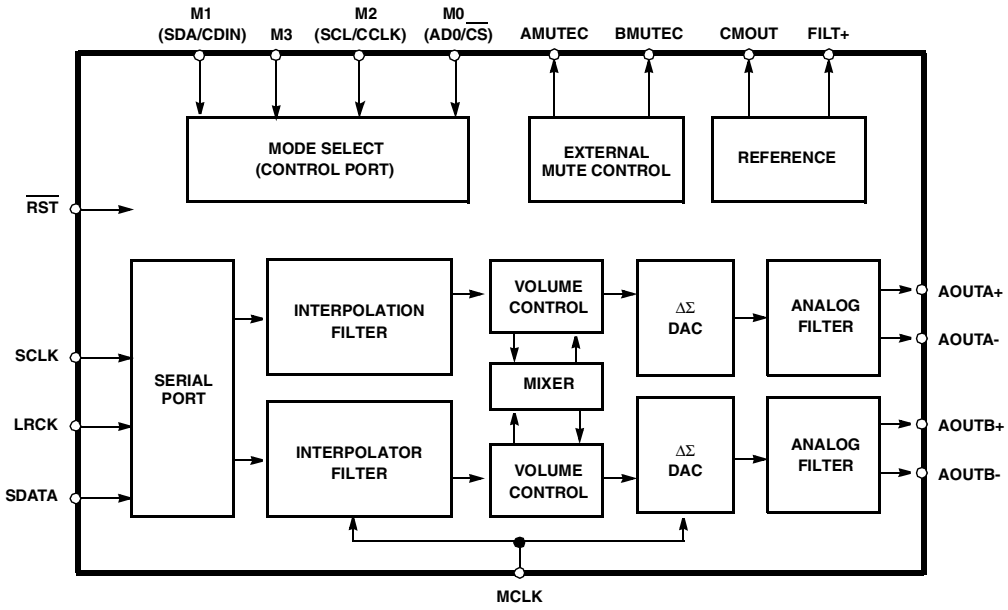
Description

The CS4391A is a complete stereo digital-to-analog system including digital interpolation, fourth-order delta-sigma digital-to-analog conversion, digital de-emphasis, volume control, channel mixing and analog filtering. The advantages of this architecture include: ideal differential linearity, no distortion mechanisms due to resistor matching errors, no linearity drift over time and temperature and a high tolerance to clock jitter.

The CS4391A accepts PCM data at sample rates from 4 kHz to 192 kHz, DSD audio data, consumes very little power and operates over a wide power supply range. These features are ideal for DVD, A/V receivers, CD and set-top box systems.

ORDERING INFORMATION

CS4391A-KS	20-pin SOIC	-10 to 70 °C
CS4391A-KZ	20-pin TSSOP	-10 to 70 °C
CS4391A-KZZ	20-pin TSSOP, Lead Free	-10 to 70 °C
CDB4391A	Evaluation Board	



Preliminary Product Information | This document contains information for a new product.
Cirrus Logic reserves the right to modify this product without notice.

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Contacting Cirrus Logic Support

For all product questions and inquiries contact a Cirrus Logic Sales Representative.
To find one nearest you go to **http://www.cirrus.com**

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1. CHARACTERISTICS/SPECIFICATIONS

(Min/Max performance characteristics and specifications are guaranteed over the *Specified Operating Conditions*. Typical performance characteristics are derived from measurements taken at T_A = 25 °C, VA = 5.0 V)

SPECIFIED OPERATING CONDITIONS (AGND = 0V; all voltages with respect to ground.)

Parameters	Symbol	Min	Typ	Max	Units
DC Power Supply	VA	4.75	5.0	5.25	V
	VL	1.8	-	VA	V
Specified Temperature Range -KS & -KZ	T_A	-10	-	70	°C

ABSOLUTE MAXIMUM RATINGS (AGND = 0 V; all voltages with respect to ground.)

Parameters	Symbol	Min	Max	Units
DC Power Supply	VA	-0.3	6.0	V
	VL	-0.3	VA	V
Input Current, Any Pin Except Supplies	I_{in}	-	±10	mA
Digital Input Voltage	V_{IND}	-0.3	VL+0.4	V
Ambient Operating Temperature (power applied)	T_A	-55	125	°C
Storage Temperature	T_{stg}	-65	150	°C

WARNING: Operation at or beyond these limits may result in permanent damage to the device. Normal operation is not guaranteed at these extremes.

ANALOG CHARACTERISTICS (Test conditions (unless otherwise specified): Input test signal is a 997 Hz sine wave at 0 dBFS; measurement bandwidth is 10 Hz to 20 kHz; Test load R_L = 5 kΩ, C_L = 10 pF)

Parameter		Symbol	VA = 5 V Min	Typ	Max	Unit
Dynamic Performance						
Dynamic Range	(Note 1)					
	unweighted		100	105	-	dB
	A-Weighted		103	108	-	dB
40 kHz Bandwidth	A-Weighted		-	102	-	dB
Total Harmonic Distortion + Noise	(Note 1,2)	THD+N				
	0 dB		-	-94	-89	dB
	-20 dB		-	-85	-	dB
	-60 dB		-	-45	-40	dB
Idle Channel Noise / Signal-to-Noise Ratio			-	108	-	dB
Interchannel Isolation	(1 kHz)		-	100	-	dB
Power Supplies						
Power Supply Current	normal operation	$I_A + I_L$	-	17	35	mA
	power-down state	$I_A + I_L$	-	60	-	µA
Power Dissipation						
	normal operation		-	85	175	mW
	power-down		-	0.3	-	mW
Power Supply Rejection Ratio (1 kHz)	(Note 3)	PSRR	-	60	-	dB
	(60 Hz)		-	40	-	dB

Parameter	Symbol	Min	Typ	Max	Units
Analog Output					
Full Scale Differential Output Voltage		1.05VA	1.1VA	1.15VA	Vpp
Common Mode Voltage	CMOUT	-	0.43VA	-	VDC
Interchannel Gain Mismatch		-	0.1	-	dB
Gain Drift		-	100	-	ppm/°C
AC-Load Resistance	R_L	5	-	-	kΩ
Load Capacitance	C_L	-	-	100	pF

ANALOG CHARACTERISTICS (continued)

Parameter		Symbol	Min	Typ	Max	Unit
Combined Digital and On-chip Analog Filter Response - Single Speed Mode						
Passband	(Note 3)					
	to -0.05 dB corner		0	-	.4535	Fs
	to -3 dB corner		0	-	.4998	Fs
Frequency Response 10 Hz to 20 kHz			-.02	-	+.035	dB
StopBand			.5465	-	-	Fs
StopBand Attenuation	(Note 5)		50	-	-	dB
Group Delay		tgd	-	9/Fs	-	s
Passband Group Delay Deviation	0 - 20 kHz		-	±0.36/Fs	-	s
De-emphasis Error (Relative to 1 kHz)						
Control Port Mode	Fs = 32 kHz		-	-	+.2/-.1	dB
	Fs = 44.1 kHz		-	-	+.05/-.14	dB
	Fs = 48 kHz		-	-	+0/.22	dB
Stand-Alone Mode	Fs = 44.1 kHz		-	-	+.05/-.14	dB
Combined Digital and On-chip Analog Filter Response - Double Speed Mode						
Passband	(Note 4)					
	to -0.1 dB corner		0	-	.4621	Fs
	to -3 dB corner		0	-	.4982	Fs
Frequency Response 10 Hz to 20 kHz			-0.1	-	0	dB
StopBand			.577	-	-	Fs
StopBand Attenuation	(Note 5)		55	-	-	dB
Group Delay		tgd	-	9/Fs	-	s
Passband Group Delay Deviation	0 - 20 kHz		-	±0.23/Fs	-	s
On-chip Analog Filter Response - Quad Speed Mode						
Passband	(Note 4)					
	to -3 dB corner		0	-	0.25	Fs
Frequency Response 10 Hz to 20 kHz			-0.7	-	0	dB
On-chip Analog Filter Response - DSD Mode						
Passband	(Note 4)					
	to -3 dB corner		0	-	1.0	Fs
Frequency Response 10 Hz to 20 kHz			-0.7	-	0	dB

Notes: 1. Triangular PDF dithered data.

2. THD+N specifications for 48 kHz sample rates are made over a 20 kHz Bandwidth.

3. Valid with the recommended capacitor values on FILT+ and CMOUT as shown in Figure 1. Increasing the capacitance will also increase the PSRR.

4. Response is clock dependent and will scale with Fs. Note that the response plots (Figures 18-25) have been normalized to Fs and can be de-normalized by multiplying the X-axis scale by Fs.

5. For Single-Speed Mode, the Measurement Bandwidth is 0.5465 Fs to 3 Fs.
For Double-Speed Mode, the Measurement Bandwidth is 0.577 Fs to 1.4 Fs.

DIGITAL CHARACTERISTICS (AGND = 0 V; all voltages with respect to ground.)

Parameters	Symbol	Min	Typ	Max	Units
High-Level Input Voltage	V_{IH}	70%	-	-	VL
Low-Level Input Voltage	V_{IL}		-	20%	VL
Input Leakage Current	I_{in}	-	-	±10	µA
Input Capacitance		-	8	-	pF
Maximum MUTEC Drive Current		-	3	-	mA

SWITCHING CHARACTERISTICS - PCM MODES (Inputs: Logic 0 = 0 V, Logic 1 = VL)

Parameters	Symbol	Min	Typ	Max	Units
Input Sample Rate	Fs	4	-	200	kHz
LRCK Duty Cycle		45	50	55	%
MCLK Duty Cycle		40	50	60	%
SCLK Frequency		-	-	MCLK/2	Hz
SCLK Frequency (Note 6)		-	-	MCLK/4	Hz
SCLK rising to LRCK edge delay	t_{slrd}	20	-	-	ns
SCLK rising to LRCK edge setup time	t_{slrs}	20	-	-	ns
SDATA valid to SCLK rising setup time	t_{sdlrs}	20	-	-	ns
SCLK rising to SDATA hold time	t_{sdh}	20	-	-	ns

Notes: 6. This serial clock is available only in Control Port Mode when the MCLK Divide bit is enabled.

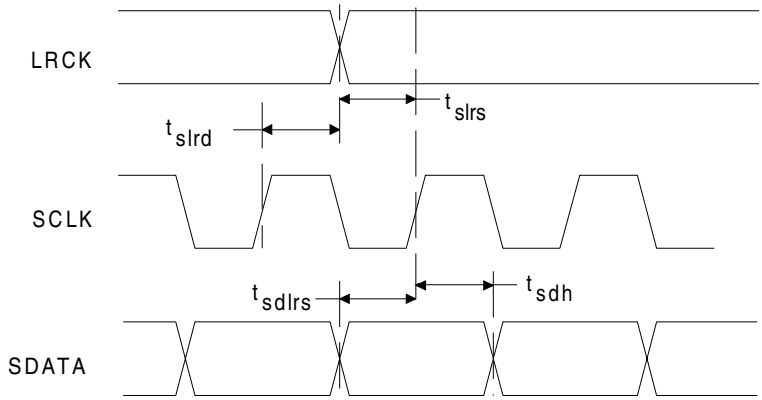


Figure 1. Serial Mode Input Timing

SWITCHING CHARACTERISTICS - DSD (Logic 0 = AGND = DGND; Logic 1 = VL)

Parameter	Symbol	Min	Typ	Max	Unit
MCLK Duty Cycle		40	50	60	%
SCLK Pulse Width Low	t_{sclkl}	20	-	-	ns
SCLK Pulse Width High	t_{sclkh}	20	-	-	ns
SCLK Period	t_{sclkw}	20	-	-	ns
SDIN valid to SCLK rising setup time	t_{sdlrs}	20	-	-	ns
SCLK rising to SDIN hold time	t_{sdh}	20	-	-	ns

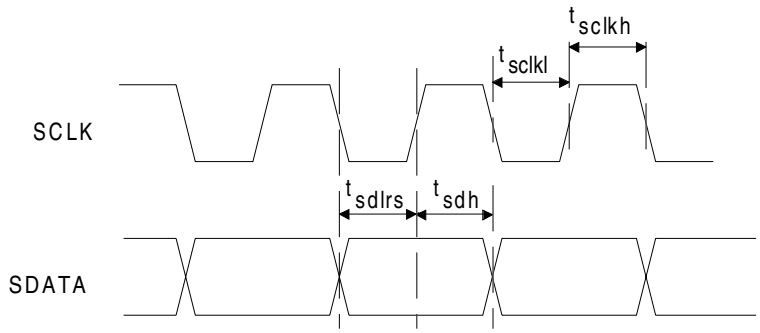


Figure 2. Direct Stream Digital - Serial Audio Input Timing

SWITCHING CHARACTERISTICS - I²C CONTROL PORT (Inputs: logic 0 = AGND, logic 1 = VL)

Parameter	Symbol	Min	Max	Unit
I²C® Mode				
SCL Clock Frequency	f_{scl}	-	100	KHz
$\overline{\text{RST}}$ Rising Edge to Start	t_{irs}	500	-	ns
Bus Free Time Between Transmissions	t_{buf}	4.7	-	µs
Start Condition Hold Time (prior to first clock pulse)	t_{hdst}	4.0	-	µs
Clock Low time	t_{low}	4.7	-	µs
Clock High Time	t_{high}	4.0	-	µs
Setup Time for Repeated Start Condition	t_{sust}	4.7	-	µs
SDA Hold Time from SCL Falling (Note 7)	t_{hdd}	0	-	µs
SDA Setup time to SCL Rising	t_{sud}	250	-	ns
Rise Time of Both SDA and SCL Lines	t_r	-	1	µs
Fall Time of Both SDA and SCL Lines	t_f	-	300	ns
Setup Time for Stop Condition	t_{susp}	4.7	-	µs

Notes: 7. Data must be held for sufficient time to bridge the 300 ns transition time of SCL.

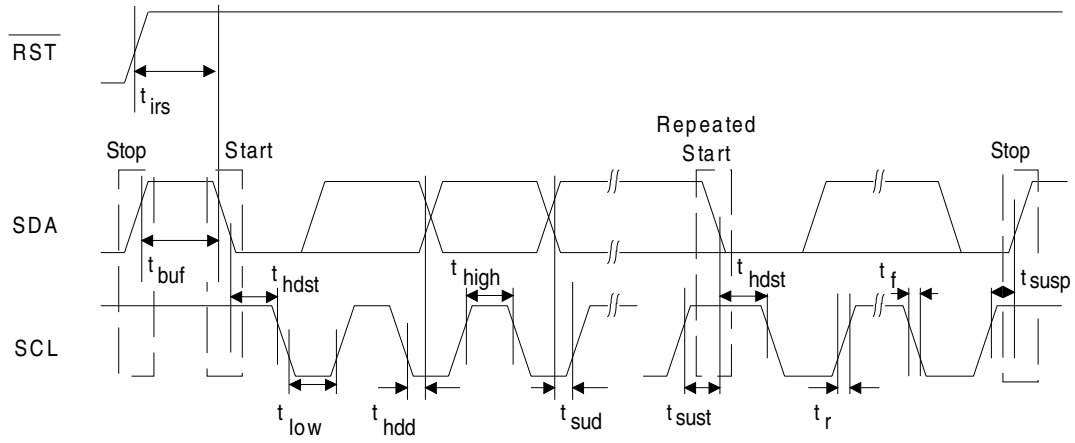


Figure 3. I²C Control Port Timing

SWITCHING CHARACTERISTICS - SPI CONTROL PORT (Inputs: logic 0 = AGND, logic 1 = VL)

Parameter		Symbol	Min	Max	Unit
SPI Mode					
CCLK Clock Frequency		f_{sclk}	-	6	MHz
$\overline{RST}$ Rising Edge to CS Falling		t_{srs}	500	-	ns
CCLK Edge to $\overline{CS}$ Falling	(Note 8)	t_{spi}	500	-	ns
$\overline{CS}$ High Time Between Transmissions		t_{csh}	1.0	-	µs
CS Falling to CCLK Edge		t_{css}	20	-	ns
CCLK Low Time		t_{scl}	66	-	ns
CCLK High Time		t_{sch}	66	-	ns
CDIN to CCLK Rising Setup Time		t_{dsu}	40	-	ns
CCLK Rising to DATA Hold Time	(Note 9)	t_{dh}	15	-	ns
Rise Time of CCLK and CDIN	(Note 10)	t_{r2}	-	100	ns
Fall Time of CCLK and CDIN	(Note 10)	t_{f2}	-	100	ns

Notes: 8. t_{spi} only needed before first falling edge of $\overline{CS}$ after $\overline{RST}$ rising edge. t_{spi} = 0 at all other times.

9. Data must be held for sufficient time to bridge the transition time of CCLK.

10. For F_{SCK} < 1 MHz

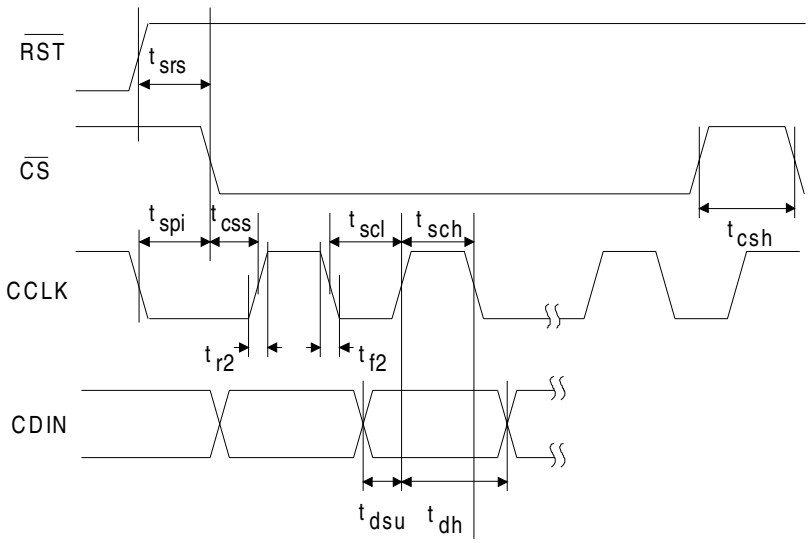


Figure 4. SPI Control Port Timing

2. TYPICAL CONNECTION DIAGRAMS

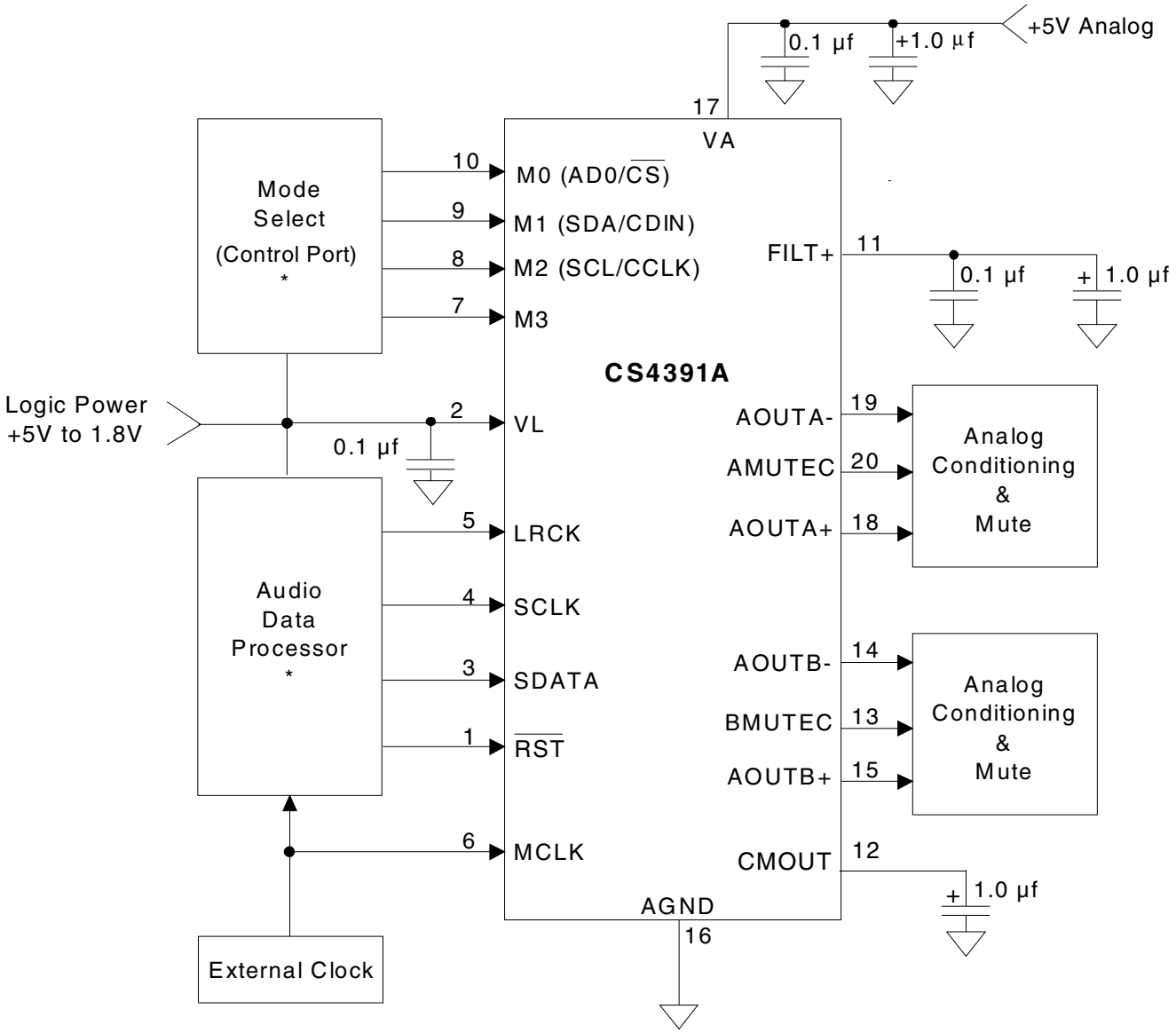


Figure 5. Typical Connection Diagram - PCM Mode

* A high logic level for all digital inputs should not exceed VL.

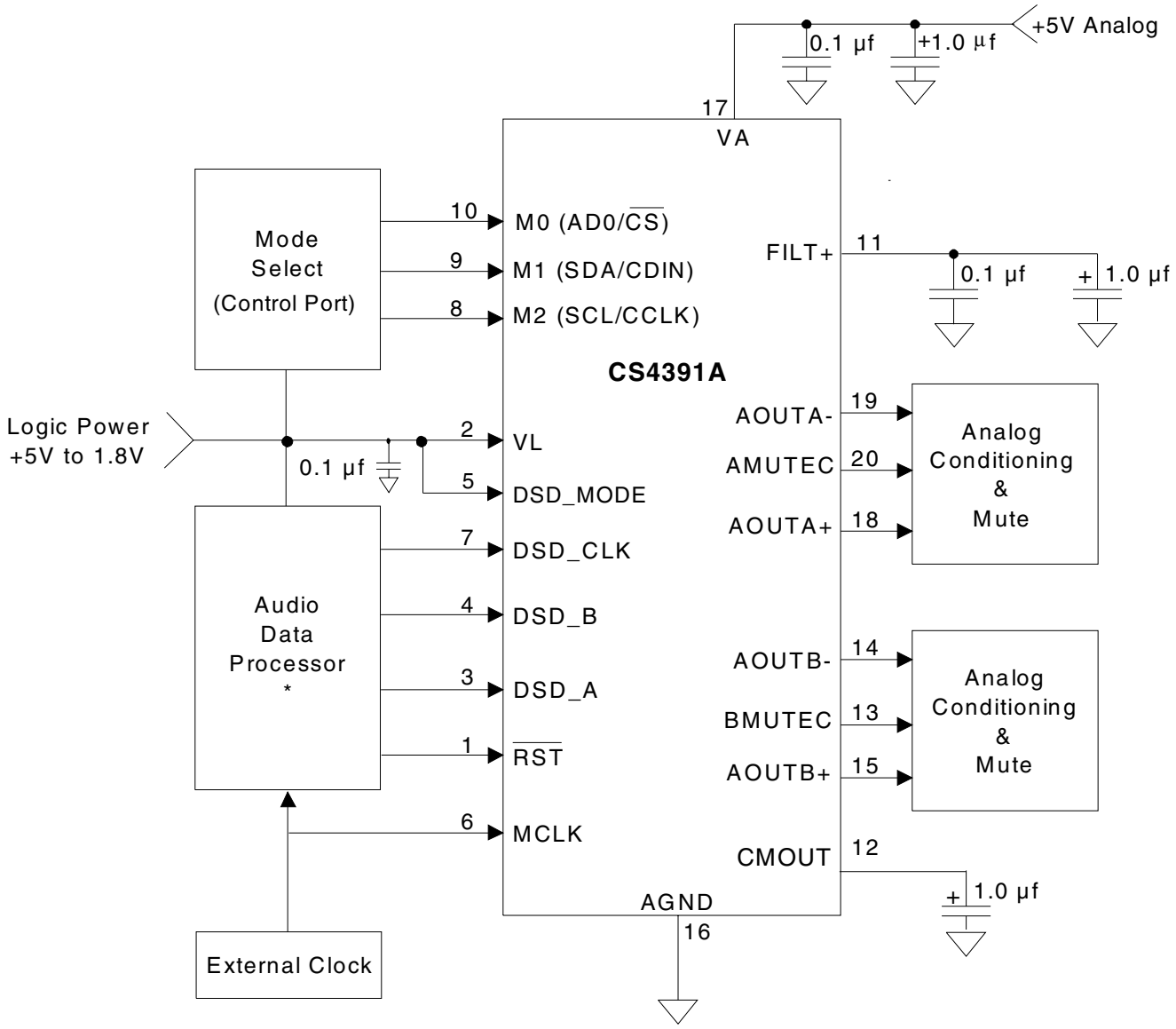


Figure 6. Typical Connection Diagram - DSD Mode

* A high logic level for all digital inputs should not exceed VL.

3. REGISTER QUICK REFERENCE

** “default” ==> bit status after power-up-sequence or reset**

3.1 MODE CONTROL 1 (ADDRESS 01H)

7	6	5	4	3	2	1	0
AMUTE	DIF2	DIF1	DIF0	DEM1	DEM0	FM1	FM0
1	0	0	0	0	0	0	0

AMUTE (Auto-mute)

Default = ‘1’.
0 - Disabled
1 - Enabled

DIF2, DIF1 and DIF0 (Digital Interface Format - PCM Modes). See Table 1

Default = ‘0’.
000 - Format 0, Left Justified, up to 24-bit data
001 - Format 1, I^2S, up to 24-bit data
010 - Format 2, Right Justified, 16-bit Data
011 - Format 3, Right Justified, 24-bit Data
100 - Format 4, Right Justified, 20-bit Data
101 - Format 5, Right Justified, 18-bit Data
110 - Reserved
111 - Reserved

DIF2, DIF1 and DIF0 (Digital Interface Format - DSD Mode Only). SeeTable 2

Default = ‘0’.
000 - Format 0, 64x oversampled DSD data with a 4x MCLK to DSD data rate
001 - Format 1, 64x oversampled DSD data with a 6x MCLK to DSD data rate
010 - Format 2, 64x oversampled DSD data with a 8x MCLK to DSD data rate
011 - Format 3, 64x oversampled DSD data with a 12x MCLK to DSD data rate
100 - Format 4, 128x oversampled DSD data with a 2x MCLK to DSD data rate
101 - Format 5, 128x oversampled DSD data with a 3x MCLK to DSD data rate
110 - Format 6, 128x oversampled DSD data with a 4x MCLK to DSD data rate
111 - Format 7, 128x oversampled DSD data with a 6x MCLK to DSD data rate

DEM1, DEM0 (De-Emphasis Mode). See Table 3

Default = ‘00’.
00 - No De-emphasis
01 - 44.1 kHz De-Emphasis
10 - 48 kHz De-Emphasis
11 - 32 kHz De-Emphasis

FM1, FM0 (Functional Mode). See Table 4

Default = ‘00’.
00 - Single-Speed Mode (4 to 50 kHz sample rates)
01 - Double-Speed Mode (50 to 100 kHz sample rates)
10 - Quad-Speed Mode (100 to 200 kHz sample rates)
11 - Direct Stream Digital Mode

3.2 VOLUME AND MIXING CONTROL (ADDRESS 02H)

7	6	5	4	3	2	1	0
A = B	Soft	Zero Cross	ATAPI4	ATAPI3	ATAPI2	ATAPI1	ATAPI0
0	1	0	0	1	0	0	1

A = B (Channel A Volume = Channel B Volume)

Default = '0'.
0 - AOUTA volume is determined by register 03h and AOUTB volume is determined by register 04h.
1 - AOUTA and AOUTB volumes are determined by register 03h and register 04h is ignored.

Soft & Zero Cross (Soft control and zero cross detection control)

Default = '10'.
SoftZero CrossMode
00 Changes take effect immediately
01 Changes take effect on zero crossings
10 Changes take effect with a soft ramp (default)
11 Changes take effect in 1/8 dB steps on each zero crossing

ATAPI 0-4 (Channel mixing and muting). SeeTable 6

Default = '01001', (Stereo)
AOUTA = Left Channel
AOUTB = Right Channel

3.3 CHANNEL A VOLUME CONTROL (ADDRESS 03H)

See *Channel B Volume Control (address 04h)*

3.4 CHANNEL B VOLUME CONTROL (ADDRESS 04H)

7	6	5	4	3	2	1	0
MUTE	VOL6	VOL5	VOL4	VOL3	VOL2	VOL1	VOL0
0	0	0	0	0	0	0	0

MUTE

Default = '0'
0 - Disabled
1 - Enabled

Volume

Default = '0'
(Refer to Table 7)

3.5 MODE CONTROL 2 (ADDRESS 05H)

7	6	5	4	3	2	1	0
INVERT_A	INVERT_B	CPEN	PDN	MUTEC A = B	FREEZE	MCLK Divide	Reserved
0	0	1	1	0	0	0	0

INVERT_A (Invert Channel A)

Default = '0'.
0 - Disabled
1 - Enabled

INVERT_B (Invert Channel B)

Default = '0'.
0 - Disabled
1 - Enabled

CPEN (Control Port Enable)

Default = '0'
0 - Disabled (Stand-Alone Mode)
1 - Enabled (Control Port Mode)

PDN (Power-Down)

Default ='1'.
0 - Disabled
1 - Enabled

MUTEC A=B

Default = '0'.
0 - Disabled
1 - Enabled

FREEZE

Default = 0.
0 - Disabled
1 - Enabled

MCLK Divide

Default = 0.
0 - Disabled
1 - Enabled

4. REGISTER DESCRIPTION

** All register access is R/W in I^2C mode and write only in SPI mode **

4.1 MODE CONTROL 1 - ADDRESS 01H

7	6	5	4	3	2	1	0
AMUTE	DIF2	DIF1	DIF0	DEM1	DEM0	FM1	FM0

4.1.1 Auto-Mute (Bit 7)

Function:

The Digital-to-Analog converter output will mute following the reception of 8192 consecutive audio samples of static 0 or -1. A single sample of non-static data will release the mute. Detection and muting is done independently for each channel. (However, Auto-Mute detection and muting can become dependent on either channel if the Mute A = B function is enabled.) The common mode on the output will be retained and the Mute Control pin for that channel will go active during the mute period. The muting function is effected, similar to volume control changes, by the Soft and Zero Cross bits in the Volume and Mixing Control register.

4.1.2 Digital Interface Formats (Bits 6:4)

Function:

PCM Mode - The required relationship between the Left/Right clock, serial clock and serial data is defined by the Digital Interface Format and the options are detailed in Table 2 and Figures 7-24.

DSD Mode - The relationship between the oversampling ratio of the DSD audio data and the required Master clock to DSD data rate is defined by the Digital interface Format pins. Note that the Functional Mode registers must be set to DSD Mode.

See Table 1 (PCM Modes)

See Table 2 (DSD Mode)

4.1.3 De-Emphasis Control (Bits 3:2)

Function:

Implementation of the standard 15 μs/50 μs digital de-emphasis filter response, Figure 13, requires reconfiguration of the digital filter to maintain the proper filter response for 32, 44.1 or 48 kHz sample rates. NOTE: De-emphasis is available only in Single-Speed Mode.

See Table 3

4.1.4 Functional Mode (Bits 1:0)

Function:

Selects the required range of input sample rates or DSD Mode.

See Table 4

4.2 VOLUME AND MIXING CONTROL (ADDRESS 02H)

7	6	5	4	3	2	1	0
A = B	Soft	Zero Cross	ATAPI4	ATAPI3	ATAPI2	ATAPI1	ATAPI0

4.2.1 Channel A Volume = Channel B Volume (Bit 7)

Function:

The AOUTA and AOUTB volume levels are independently controlled by the A and the B Channel Volume Control Bytes when this function is disabled. The volume on both AOUTA and AOUTB are determined by the A Channel Volume Control Byte and the B Channel Byte is ignored when this function is enabled.

4.2.2 Soft Ramp or Zero Cross Enable (Bits 6:5)

Function:

Soft Ramp Enable

Soft Ramp allows level changes, both muting and attenuation, to be implemented by incrementally ramping, in 1/8 dB steps, from the current level to the new level at a rate of 1dB per 8 left/right clock periods.

Zero Cross Enable

Zero Cross Enable dictates that signal level changes, either by attenuation changes or muting, will occur on a signal zero crossing to minimize audible artifacts. The requested level change will occur after a timeout period between 512 and 1024 sample periods (10.7 ms to 21.3 ms at 48 kHz sample rate) if the signal does not encounter a zero crossing. The zero cross function is independently monitored and implemented for each channel.

Soft Ramp and Zero Cross Enable

Soft Ramp and Zero Cross Enable dictates that signal level changes, either by attenuation changes or muting, will occur in 1/8 dB steps and be implemented on a signal zero crossing. The 1/8 dB level change will occur after a timeout period between 512 and 1024 sample periods (10.7 ms to 21.3 ms at 48 kHz sample rate) if the signal does not encounter a zero crossing. The zero cross function is independently monitored and implemented for each channel.

See Table 5

4.2.3 ATAPI Channel Mixing and Muting (Bits 4:0)

Function:

The CS4391A implements the channel mixing functions of the ATAPI CD-ROM specification.

See Table 6

4.3 CHANNEL A VOLUME CONTROL - ADDRESS 03H

See Section 4.4 *Channel B Volume Control - Address 04h*

4.4 CHANNEL B VOLUME CONTROL - ADDRESS 04H

7	6	5	4	3	2	1	0
MUTE	VOL6	VOL5	VOL4	VOL3	VOL2	VOL1	VOL0

4.4.1 Mute (Bit 7)

Function:

The Digital-to-Analog converter output will mute when enabled. The common mode voltage on the output will be retained. The muting function is effected, similiar to attenuation changes, by the Soft and Zero Cross bits in the Volume and Mixing Control register. The MUTEC pin for that channel will go active during the mute period if the Mute function is enabled. Both the AMUTEC and BMUTEC will go active if either MUTE register is enabled and the MUTEC A = B bit (register 5) is enabled.

4.4.2 Volume Control (Bits 6:0)

Function:

The digital volume control allows the user to attenuate the signal in 1 dB increments from 0 to -119 dB. Volume settings are decoded as shown in Table 7. The volume changes are implemented as dictated by the Soft and Zero Cross bits in the Volume and Mixing Control register. All volume settings less than -119 dB are equivalent to enabling the Mute bit.

4.5 MODE CONTROL 2 - ADDRESS 05H

7	6	5	4	3	2	1	0
INVERT_A	INVERT_B	CPEN	PDN	MUTEC A = B	FREEZE	MCLK Divide	Reserved

4.5.1 Invert Signal Polarity (Bits 7:6)

Function:

When set, this bit inverts the signal polarity.

4.5.2 Control Port Enable (Bit 5)

Function:

This bit defaults to 0, allowing the device to power-up in Stand-Alone mode. The Control port mode can be accessed by setting this bit to 1. This will allow the operation of the device to be controlled by the registers and the pin definitions will conform to Control Port Mode. To accomplish a clean and click free power-up, the user should write 30h to register 5 within 10 ms following the release of Reset.

4.5.3 Power Down (Bit 4)

Function:

The device will enter a low-power state whenever this function is activated. The power-down bit defaults to 'enabled' on power-up and must be disabled before normal operation will begin. The contents of the control registers are retained when the device is in power-down.

4.5.4 AMUTEC = BMUTEC (Bit 3)

Function:

When this function is enabled, the individual controls for AMUTEC and BMUTEC are internally connected through a AND gate prior to the output pins. Therefore, the external AMUTEC and BMUTEC pins will go active only when the requirements for both AMUTEC and BMUTEC are valid.

4.5.5 Freeze (Bit 2)

Function:

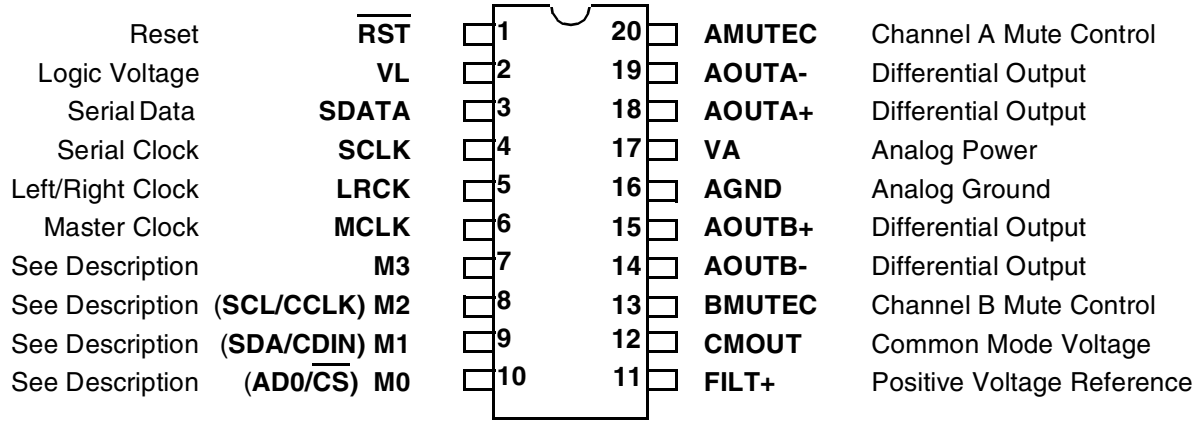
This function allows modifications to the registers without the changes being taking effect until Freeze is disabled. To make multiple changes in the Control port registers take effect simultaneously, set the Freeze Bit, make all register changes, then Disable the Freeze bit.

4.5.6 Master Clock Divide (Bit 1)

Function:

This function allows the user to select an internal divide by 2 of the Master Clock. This selection is required to access the higher Master Clock rates as shown in Table 9.

5. PIN DESCRIPTION - PCM DATA MODE



Reset - $\overline{\text{RST}}$

Pin 1, Input

Function:

Hardware Mode: The device enters a low power mode and the internal state machine is reset to the default setting when low. When high, the device becomes operational.

Control Port Mode: The device enters a low power mode and all internal registers are reset to the default settings, including the control port, when low. When high, the control port becomes operational and the PDN bit must be cleared before normal operation will occur. The control port can not be accessed when reset is low. The Control Port Enable Bit must also be enabled after a device reset.

$\overline{\text{RST}}$ is required to remain low until the power supplies and clocks are applied and stable.

Interface Power - VL

Pin 2, Input

Function:

Digital interface power supply. The voltage on this pin determines the logic level high threshold for the digital inputs.

Serial Audio Data - SDATA

Pin 3, Input

Function:

Two's complement MSB-first serial data is input on this pin. The data is clocked into SDATA via the serial clock and the channel is determined by the Left/Right clock. The required relationship between the Left/Right clock, serial clock and serial data is defined by the Mode Control Byte in Control Port Mode or the Mode Pins in Hardware Mode. The options are detailed in Figures 7-24.

Serial Clock - SCLK

Pin 4, Input

Function:

Clocks the individual bits of the serial data into the SDATA pin. The required relationship between the Left/Right clock, serial clock and serial data is defined by the Mode Control Byte in Control Port Mode or the Mode pins in Hardware Mode. The options are detailed in Figures 7-24.

Left / Right Clock - LRCK

Pin 5, Input

Function:

The Left / Right clock determines which channel is currently being input on the serial audio data input, SDATA. The frequency of the Left/Right clock must be at the input sample rate. Audio samples in Left/Right sample pairs will be simultaneously output from the digital-to-analog converter whereas Right/Left pairs will exhibit a one sample period difference. The required relationship between the Left/Right clock, serial clock and serial data is defined by the Mode Control Byte in Control Port Mode or the Mode pins in Stand-alone Mode. The options are detailed in Figures 7-24.

Master Clock - MCLK

Pin 6, Input

Function:

The master clock frequency must be either 256x, 384x, 512x, 768x or 1024x the input sample rate in Single Speed Mode; either 128x, 192x 256x, 384x or 512x the input sample rate in Double Speed Mode; or 64x, 96x 128x, 192x or 256 x the input sample rate in Quad Speed Mode. Tables 8-10 illustrate the standard audio sample rates and the required master clock frequencies.

Note: These clocking ratios are only available in Control Port Mode when the MCLK Divide bit is enabled.

Mode Select - M3, M2, M1 and M0 (Stand-alone Mode)

Pins 7, 8, 9 and 10 Inputs

Function:

The Mode Select Pins, M0-M3, select the operational mode of the device as detailed in Tables 11-15.

Mode Select - M3 (Control Port Mode)

Pin 7, Input

Function:

The Mode Select Pin, M3, is not used in PCM Control Port mode and should be terminated to ground.

Serial Control Interface Clock - SCL/CCLK (Control Port Mode)

Pin 8, Input

Function:

Clocks the serial control data into or from SDA/CDIN.

Serial Control Data I/O - SDA/CDIN (Control Port Mode)

Pin 9, Input/Output

Function:

In I^2C mode, SDA is a data I/O line. CDIN is the input data line for the control port interface in SPI mode.

Address Bit / Chip Select - AD0 / $\overline{CS}$ (Control Port Mode)

Pin 10, Input

Function:

In I^2C mode, AD0 is a chip address bit. $\overline{CS}$ is used to enable the control port interface in SPI mode. The device will enter the SPI mode at anytime a high to low transition is detected on this pin. Once the device has entered the SPI mode, it will remain until either the part is reset or undergoes a power-down cycle.

Positive Voltage Reference - FILT+

Pin 11, Output

Function:

Positive reference for internal sampling circuits. External capacitors are required from FILT+ to analog ground, as shown in Figures 5 and 6. The recommended values will typically provide 60 dB of PSRR at 1 kHz and 40 dB of PSRR at 60 Hz. FILT+ is not intended to supply external current. FILT+ has a typical source impedance of 250 kΩ and any current drawn from this pin will alter device performance.

Common Mode Voltage - CMOUT

Pin 12, Output

Function:

Filter connection for internal common mode reference voltage, typically 50% of VA. Capacitors must be connected from CMOUT to analog ground, as shown in Figures 5 and 6. CMOUT is not intended to supply external current. CMOUT has a typical source impedance of 250 kΩ and any current drawn from this pin will alter device performance.

Channel A and Channel B Mute Control - AMUTEC and BMUTEC

Pins 13 and 20, Outputs

Function:

The Mute Control pins go high during power-up initialization, reset, muting, master clock to left/right clock frequency ratio is incorrect or power-down. These pins are intended to be used as a control for an external mute circuit to prevent the clicks and pops that can occur in any single supply system. Use of Mute Control is not mandatory but recommended for designs requiring the absolute minimum in extraneous clicks and pops.

Differential Analog Output - AOUTB+, AOUTB- and AOUTA+, AOUTA-

Pins 14, 15 and 18, 19, Outputs

Function:

The full scale differential analog output level is specified in the Analog Characteristics specifications table.

Analog Ground - AGND

Pin 16, Input

Function:

Analog ground reference.

Analog Power - VA

Pin 17, Input

Function:

Analog power supply.