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## Features

- **Stereo Audio DAC**
  - 2.7V to 3.3V Analog Supply Operation
  - 2.4V to 3.3V Digital Supply Operation
  - 20-bit Stereo Audio DAC
  - 93 dB SNR Playback Stereo Channels
  - 32 Ohm/20 mW Stereo Headset Drivers with Master Volume and Mute Controls
  - Stereo Line Level Input with Volume Control/mute and Playback through the Headset Drivers
  - Differential Monaural Auxiliary Input, with Volume Control/mute and Playback through the Headset Drivers
  - Accepts Mixed Signals from All Signal Paths (Line Inputs, External Mono and DAC Output)
  - 8, 11.024, 16, 22.05, 24, 32, 44.1 and 48 kHz Sampling Rates
  - 256x or 384xFs Master Clock Frequency
  - I2S Serial Audio Interface
- **Mono Audio Power Amplifier**
  - Supply Input from Main Li-Ion Battery
  - 440mW on 8 Ohm Load
  - Low Power Mode for Earphone
  - Programmable Volume Control (-22 to +20 dB)
  - Fully Differential Structure, Input and Output
  - 8 mA Drain Current in Full Power Mode
  - Power-down mode (Consumption Less than 2  $\mu$ A)
  - Minimum External Components (Direct Connection of the Loudspeaker)
- **Applications: Mobile Phones, Digital Cameras, PDAs, SmartPhones, DECT Phones, Music Players**

## 1. Description

The AT73C213 is a fully integrated, low-cost, combined stereo audio DAC and audio power amplifier circuit targeted for Li-Ion or Ni-Mh battery powered devices such as mobile phones, smartphones, PDA, DECT phones, digital still cameras, music players or any other type of handheld device where an audio interface is needed.

The stereo DAC section is a complete high performance, stereo audio digital-to-analog converter delivering a 93 dB dynamic range. It comprises a multibit sigma-delta modulator with dither, continuous time analog filters and analog output drive circuitry. This architecture provides a high insensitivity to clock jitter. The digital interpolation filter increases the sample rate by a factor of 8 using 3 linear phase half-band filters cascaded, followed by a first order SINC interpolator with a factor of 8. This filter eliminates the images of baseband audio, retaining only the image at 64x the input sample rate, which is eliminated by the analog post filter. Optionally, a dither signal can be added that reduces possible noise tones at the output. However, the use of a multibit sigma-delta modulator already provides extremely low noise tone energy.

Master clock is from 256 or 384 times the input data rate, allowing choice of input data rate up to 50 kHz, including standard audio rates of 48, 44.1, 32, 16 and 8 kHz.

The DAC section is followed by a volume and mute control and can be simultaneously played back directly through a stereo 32 Ohm headset pair of drivers.



## Power Management for Mobiles (PM)

## AT73C213 Audio Interface for Portable Handsets

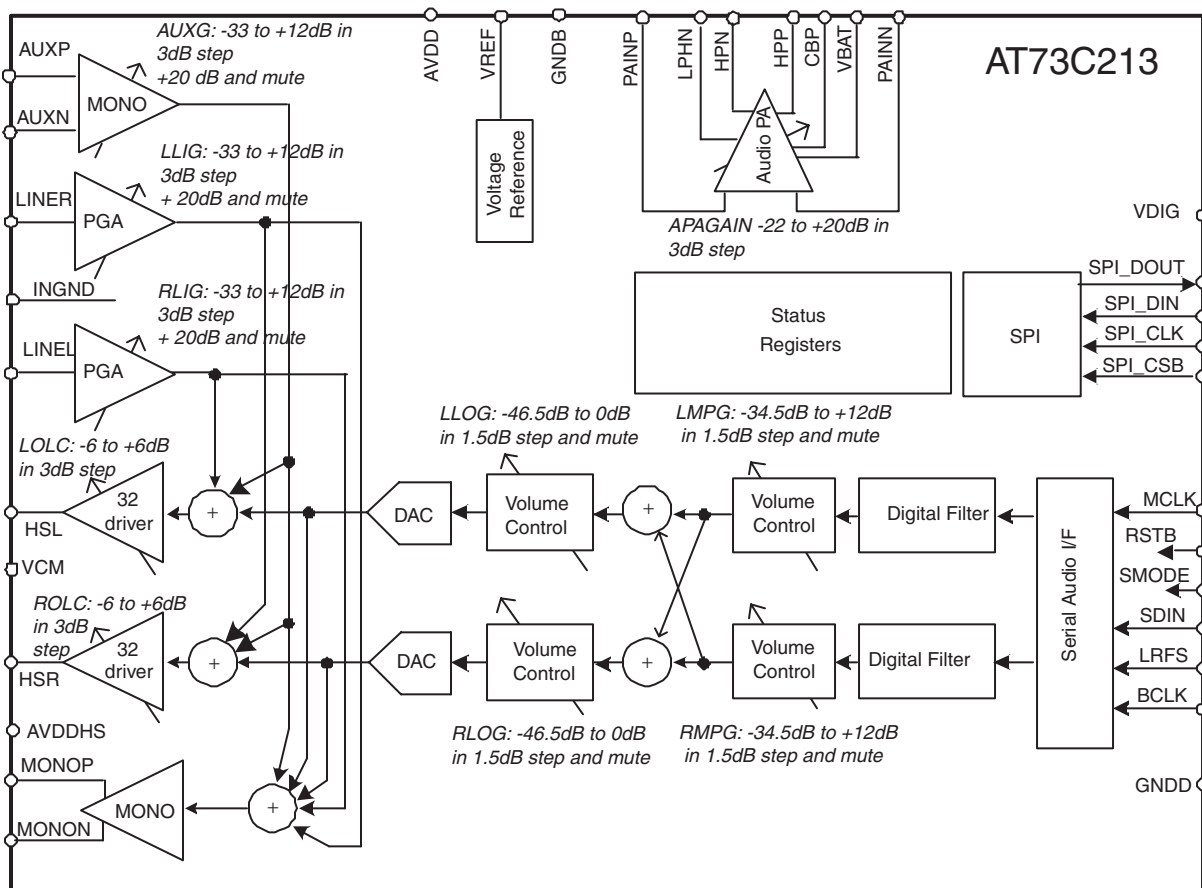
The stereo 32 Ohm headset pair of drivers also includes a mixer of a LINEL and LINER pair of stereo inputs, as well as a differential monaural auxiliary input (line level).

The DAC output can be connected through a buffer stage to the input of the audio power amplifier, using 2x coupling capacitors. The mono buffer stage also includes a mixer of the LINEL and LINER inputs, as well as a differential monaural auxiliary input (line level) which can be, for example, the output of a voice Codec output driver in mobile phones.

The audio power amplifier is a dual-mode AB class amplifier with differential output and programmable volume control. In full power mode, it is capable of driving an 8-ohm loudspeaker at maximum power of 1W at 5V supply and 440 mW at 3.6V supply. In low power mode, it can drive the same loudspeaker as an earpiece, making it suitable as a handsfree speaker driver in wireless handset application. The volume, mute, power down, de-emphasis controls and 16-bit, 18-bit and 20-bit audio formats are digitally programmable via a 4-wire SPI bus and the digital audio data are provided through a multi-format I2S interface.

## 2. Block Diagram

Figure 2-1. AT73C213 Functional Block Diagram



### 3. Pin Description

**Table 3-1.** Pin Description

Pin Name	I/O	Pin	Type	Function
LPHN	O	10	Analog	Low power audio stage output
HPN	O	11	Analog	Negative speaker output
VBAT	I	12	Supply	Audio amplifier supply
HPP	O	13	Analog	Positive speaker output
CBP	O	14	Analog	Audio amplifier common mode voltage decoupling
PAINN	I	15	Analog	Audio amplifier negative input
PAINP	I	16	Analog	Audio amplifier positive input
SDIN	I	17	Digital	Audio interface serial data input
BCLK	I	18	Digital	Audio interface bit clock
LRFS	I	19	Digital	Audio interface left/right channel synchronization frame pulse
MCLK	I	20	Digital	Audio interface master clock input
RSTB	I	21	Digital	Master reset (active low)
SMODE	I	22	Digital	Serial interface selection (to connect to ground)
GNDD	GND	23	Ground	Digital ground
VDIG	I	24	Supply	Digital supply
SPI_DOUT	O	25	Digital	SPI data output
SPI_DIN	I/O	26	Digital	SPI data input
SPI_CLK	I	27	Digital	SPI clock
SPI_CSB	I	28	Digital	SPI chip select
MONON	O	29	Analog	Negative monaural driver output
MONOP	O	30	Analog	Positive monaural driver output
AUXP	I	31	Analog	Audio mono auxiliary positive input
AUXN	I	32	Analog	Audio mono auxiliary negative input
VREF	I	1	Analog	Voltage reference pin for decoupling
AVDD	I	2	Supply	Analog supply (DAC + Line in + Aux + Mono out)
HSL	O	3	Analog	Left channel headset driver output
HSR	O	4	Analog	Right channel headset driver output
AVDDHS	I	5	Supply	Headset driver analog supply
LINEL	I	6	Analog	Left channel line in
LINER	I	7	Analog	Right channel line in
INGND	I	8	Analog	Line signal ground pin for decoupling
VCM	I	9	Analog	Common mode reference for decoupling
GNDB	GND	33 (Bottom)	Ground	Analog ground

## 4. Electrical Characteristics

**Table 4-1.** Absolute Maximum Ratings\*

Operating Temperature (Industrial)-40°C to +85°C
Storage Temperature-55°C to +150°C
Power Supply Input
on VBAT-0.3V to +5.5V
on VDIG, AVDD-0.3V to +3.6V

\*NOTICE: Stresses beyond those listed under “Absolute Maximum Ratings” may cause permanent damage to the device. This is a stress rating only and functional operation of the device at these or other conditions beyond those indicated in the operational sections of this specification is not implied. Exposure to absolute maximum rating conditions for extended periods may affect device reliability.

## 5. Digital IOs

All the digital IOs: SDIN, BCLK, LRFS, MCLK, RSTB, SMODE, SPI\_DOUT, SPI\_DIN, SPI\_CLK, SPI\_CSB are referred to as VDIG.

**Table 5-1.** Digital IOs

Symbol	Parameter	Conditions	VDIG	Min	Max	Unit
VIL	Low level input voltage	Guaranteed input low Voltage	from 2.4Vto 3.3 V	-0.3	0.2 x VDIG	V
VIH	High level input voltage	Guaranteed input high Voltage	from 2.4Vto 3.3 V	0.8 x VDIG	VDIG + 0.3	V
VOL	Low level output voltage	IOL = 2 mA	from 2.4Vto 3.3 V		0.4	V
VOH	High level output voltage	IOH = 2 mA	from 2.4Vto 3.3 V	VDIG - 0.5V		V

## 6. Audio Power Amplifier

### 6.1 Electrical Specifications

V<sub>BAT</sub> = 3.6V, T<sub>A</sub> = 25°C unless otherwise noted. High power mode, 100 nF capacitor connected between CBP and GND  
 Audio, load = 8 Ohms.

**Table 6-1.** Audio Power Amplifier Electrical Specifications (General Conditions: V<sub>DD</sub> = 3.6V, T<sub>A</sub> = 25°C)

Symbol	Parameter	Conditions	Min	Typ	Max	Unit
V <sub>DD</sub>	Supply voltage	Unloaded, 100 nF decoupling capacitor to GND	3	3.6	5.5	V
I <sub>DD</sub>	Quiescent current	Inputs shorted, no load		6	8	mA
I <sub>DDstby</sub>	Standby current				2	μA
V <sub>Cbp</sub>	DC reference			V <sub>DD</sub> /2		V
VOS	Output differential offset	Full gain	-20	0	20	mV
Z <sub>IN</sub>	Input impedance	Active state	12K	20k	30k	Ohm
Z <sub>LFP</sub>	Output load	Full Power mode	6	8	32	Ohm
Z <sub>LLP</sub>	Output load	Low-Power mode, including R1	100	150	300	Ohm
C <sub>L</sub>	Capacitive load				100	pF
PSRR	Power supply rejection ratio	200 to 2 kHz differential output		60		dB
BW <sub>min</sub>	Low Frequency Cutoff	1 kHz reference frequency 3 dB attenuation 470 nF input coupling capacitors		50		Hz
BW <sub>max</sub>	High Frequency Cutoff	1 KHz reference frequency 3 dB attenuation	20			kHz
t <sub>UP</sub>	Output setup time	Off to on mode Voltage already settled Input capacitors precharged			10	ms
V <sub>N</sub>	Output noise	Max gain, A weighted		120	500	μV <sub>RMS</sub>
THD <sub>HP</sub>	Output distortion	High power mode, 1 kHz, P <sub>out</sub> = 100mW, gain = 0dB		0.3		%
THD <sub>LP</sub>	Output distortion	Low power mode, 1KHz, V <sub>out</sub> = 100m Vpp, Max gain, load 8 ohms in series with 200 ohms		1		%
P <sub>max</sub>	Maximum power	High power mode, 1 KHz, V <sub>out</sub> = 100 mVpp, Max gain, load 8 ohms in series with 200 ohms		440		mW
G <sub>ACC</sub>	Overall Gain accuracy		-2	0	2	dB
G <sub>STEP</sub>	Gain Step Accuracy		-0.7	0	0.7	dB

## 7. Audio DAC

### 7.1 Electrical Specifications

AVDD, AVDDHS = 2.8 V,  $T_A = 25^\circ\text{C}$ , typical case, unless otherwise noted.

All noise and distortion specifications are measured in the 20 Hz to 0.425xfs and A-weighted filtered.

Full-scale levels scale proportionally with the analog supply voltage.

**Table 7-1.** Electrical Specifications

	Min	Typ	Max	Units
<b>Overall</b>				
Operating Temperature (ambient)	-40	+25	+85	$^\circ\text{C}$
Analog Supply Voltage (AVDD, AVDDHS)	2.7	2.8	3.3	V
Digital Supply Voltage (VDIG)	2.4	2.8	3.3	V
<b>DIGITAL INPUTS/OUTPUTS</b>				
Resolution	20			bits
Logic Family	CMOS			
Logic Coding	2's Complement			
<b>Analog Performance - DAC to Line-out/Headphone Output</b>				
Output level for full scale input (for AVDD, AVDDHS = 2.8 V)		1.65		Vpp
Output common mode voltage		0.5 x AVDDHS		V
Output load resistance (on HSL, HSR)				
Headphone load	16	32		Ohm
Line load	7	10		kOhm
Output load capacitance (on HSL, HSR)				
Headphone load		30	1000	pF
Line load		30	150	pF
Signal to Noise Ratio (-1dBFS @ 1kHz input and 0dB Gain) Line and Headphone loads	87	92		dB
Total Harmonic Distortion (-1dBFS @ 1kHz input and 0dB Gain)				
Line Load		0.01	0.016	%
Headphone Load		0.06	0.1	%
Headphone Load (16 Ohm)		0.5	1	%
Dynamic Range (measured with -60 dBFS @ 1kHz input, extrapolated to full-scale)				
Line Load	88	93		dB
Headphone Load	70	74		dB
Interchannel mismatch		0.1	1	dB
Left-channel to right-channel crosstalk (@ 1kHz)		-90	-80	dB
Output Headset Driver Level Control Range	-6		6	dB
Output Headset Driver Level Control Step		3		dB

**Table 7-1. Electrical Specifications (Continued)**

	Min	Typ	Max	Units
PSRR				
1 kHz		55		dB
20 kHz		50		dB
Maximum output slope at power up (100 to 220 $\mu$ F coupling capacitor)			3	V/s
<b>Analog Performance - Line-in/Microphone Input to Line-out/Headphone Output</b>				
Input level for full scale output - 0dBFS Level @ AVDD, AVDDHS = 2.8 V and 0 dB gain		1.65 583		Vppm Vrms
@ AVDD, AVDDHS = 2.8 V and 20 dB gain		0.165 58.3		Vppm Vrms
Input common mode voltage		0.5 x AVDD		V
Input impedance	7	10		kOhm
Signal to Noise Ratio				
-1 dBFS @ 1kHz input and 0 dB gain	81	85		dB
-21 dBFS @ 1kHz input and 20 dB gain		71		
Dynamic Range (extrapolated to full scale level)				
-60 dBFS @ 1kHz input and 0 dB gain	82	86		dB
-60 dBFS @ 1kHz input and 20 dB gain		72		
Total Harmonic Distortion				
-1dBFS @ 1kHz input and 0 dB gain		0.01	0.016	%
-1dBFS @ 1kHz input and 20 dB gain		0.018	0.04	%
Interchannel mismatch		0.1	1	dB
Left-channel to right-channel crosstalk (@ 1kHz)		-90	-80	dB
<b>Analog Performance - Differential mono input amplifier</b>				
Differential input level for full scale output - 0dBFS Level @ AVDD, AVDDHS = 2.8 V and 0 dB gain		1.65 583		Vppdif mVrms
Input common mode voltage		0.5xAVDD		V
Input impedance	7	10		kOhm
Signal to Noise Ratio (-1 dBFS @ 1kHz input and 0 dB gain)	76	80		dB
Total Harmonic Distortion (-1dBFS @ 1kHz input and 0 dB gain)		-85	-81	dB
<b>Analog Performance - PA Driver</b>				
Differential output level for full scale input (for AVDD, AVDDHS = 2.8 V)		3.3		Vppdif
Output common mode voltage		0.5 x AVDDHS		V
Output load		10		kOhm
			30	pF
Signal to Noise Ratio (-1dBFS @ 1kHz input and 0dB Gain)	76	80		dB



**Table 7-1. Electrical Specifications (Continued)**

	Min	Typ	Max	Units
Total Harmonic Distortion (-1dBFS @ 1kHz input and 0dB Gain)		-75	-71	dB
<b>Master Clock</b>				
Master Clock Maximum Long Term Jitter			1.5	ns <sub>pp</sub>
<b>Digital Filter Performance</b>				
Frequency response (10 Hz to 20 kHz)		± 0.1		dB
Deviation from linear phase (10 Hz to 20 kHz)		± 0.1		deg
Passband 0.1 dB corner		0.4535		Fs
Stopband	0.5465			Fs
Stopband Attenuation	65			dB
<b>De-emphasis Filter Performance (for 44.1kHz Fs)</b>				
MAX deviation from ideal response	-1		1	dB
<b>POWER PERFORMANCE</b>				
Current consumption from Analog supply in power on		9.5		mA
Current consumption from Analog supply in power down			10	mA
Power on Settling Time				
From full power down to full power up (Vref and VCM decoupling capacitors charge)		500		ms
Line in amplifier (line in coupling capacitors charge)		50		ms
Driver amplifier (out driver DC blocking capacitors charge)		500		ms

## 7.2 Digital Filters Transfer Function

Figure 7-1. Channel Filter

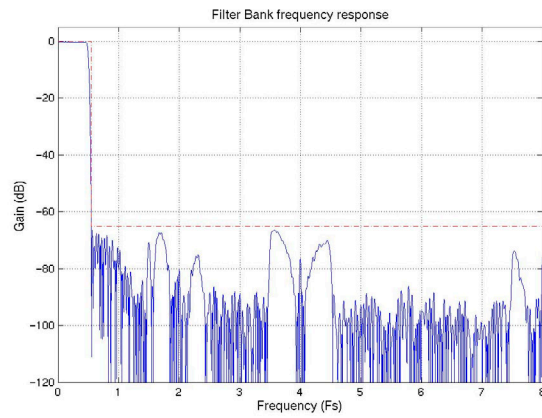


Figure 7-2. Channel Filter

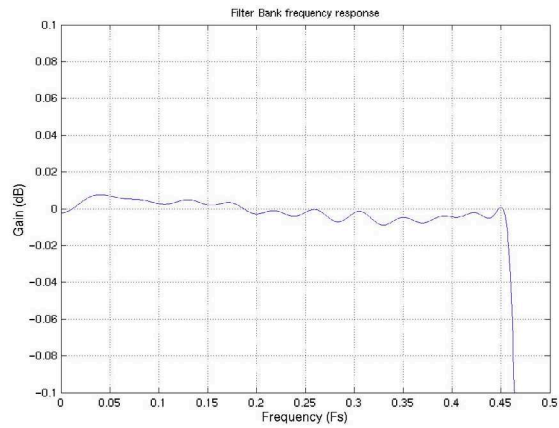
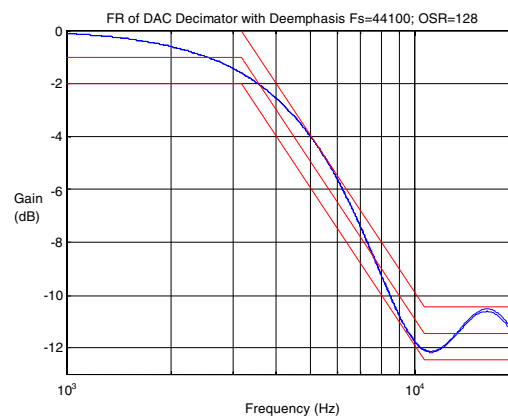


Figure 7-3. De-emphasis Filter



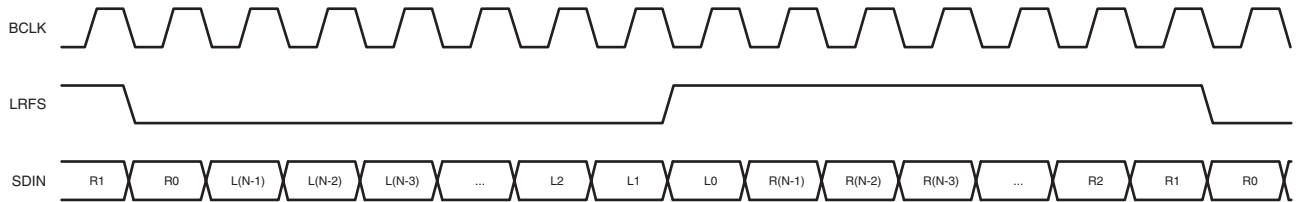
### 7.3 Data Interface

Normal operation is entered by applying correct LRFS, BCLK and SDIN waveforms to the serial interface, as illustrated in [Figure 7-4](#), [Figure 7-5](#) and [Figure 7-6](#).

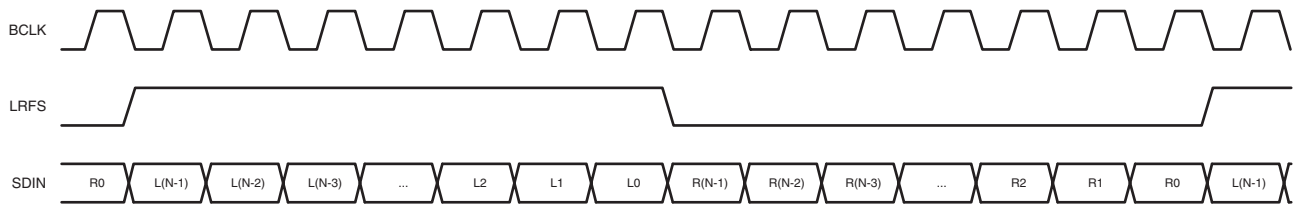
To avoid noise at the output, the reset state is maintained until proper synchronization is achieved in the serial interface.

The data interface allows three different data transfer modes. See [Figure 7-4](#), [Figure 7-5](#) and [Figure 7-6](#).

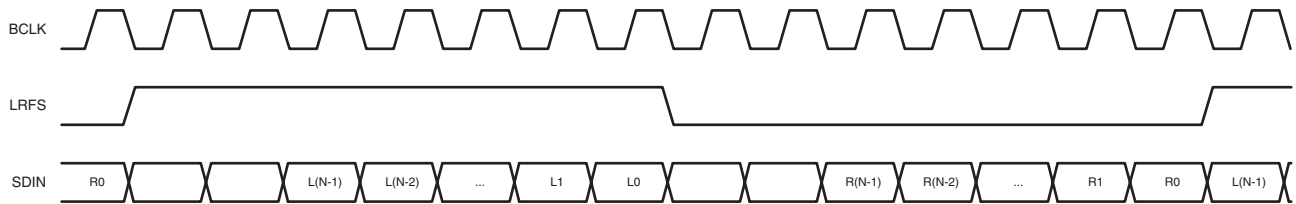
**Figure 7-4.** 20-bit I2S Justified Mode



**Figure 7-5.** 20-bit MSB Justified Mode



**Figure 7-6.** 20-bit LSB Justified Mode



The selection between modes is done using the DINTSEL<1:0> signal.

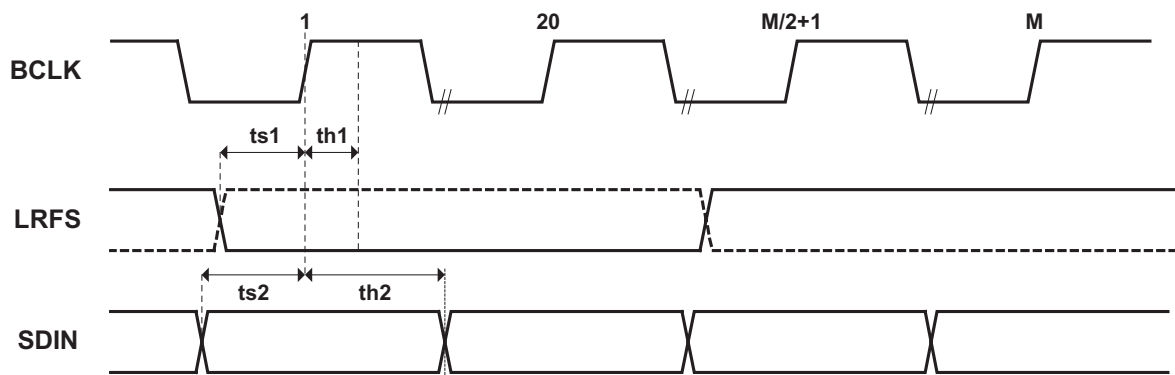
DINTSEL <1:0>	Format
00	I2S Justified
01	MSB Justified
1x	LSB Justified

The data interface always works in slave mode. This means that the LRFS and the BCLK signals are provided by the host controller. In order to achieve proper operation, the LRFS and the BCLK signals must be synchronous with the MCLK master clock signal and their frequency relationship must reflect the selected data mode. For example, if the data mode selected is the 20-bit MSB Justified, then the BCLK frequency must be 40 times higher than the LRFS frequency.

### 7.4 Timing Specifications

The timing constraints of the data interface are described in [Figure 7-7](#) and [Table 7-2](#) below.

**Figure 7-7.** Data Interface Timing Diagram



**Table 7-2.** Data Interface Timing Parameters

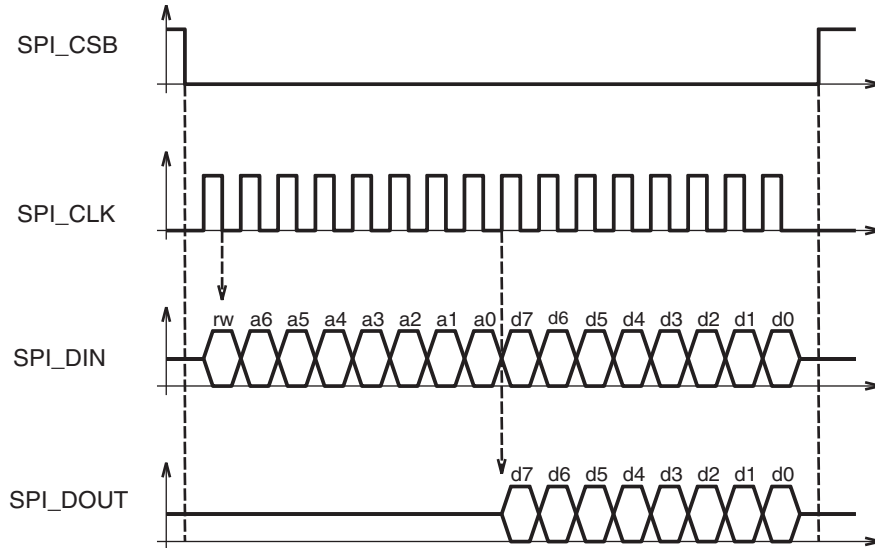
	Parameter	Min	Typ	Max	Unit
ts1	LRFS set-up time before BCLK rising edge	10			ns
th1	LRFS hold time after BCLK rising edge	10			ns
ts2	DIN set-up time before BCLK rising edge	10			ns
th2	DIN hold time after BCLK rising edge	10			ns

## 8. SPI Interface

### 8.1 Architecture

The SPI is a three-wire bi-directional asynchronous serial link. It works only in slave mode. The protocol is the following:

**Figure 8-1.** SPI Architecture



### 8.2 SPI Protocol

On SPI\_DIN, the first bit is a read/write bit. 0 indicates a write operation, while 1 is for a read operation. The seven following bits are used for the register address and the eight last ones are the write data. For both address and data, the most significant bit is the first one.

In case of a read operation, SPI\_DOUT provides the contents of the read register, MSB first.

The transfer is enabled by the CSB signal active low. When no operation is being carried out, SPI\_DOUT is set high impedance to allow sharing of MCU serial interface with other devices. The interface is reset at every rising edge of SPI\_CS# in order to come back to an idle state, even if the transfer does not succeed. The SPI is synchronized with the serial clock SPI\_CLK. Falling edge latches SPI\_DIN input and rising edge shifts SPI\_DOUT output bits.

Note that MCLK must run during any SPI write access from address 0x00 to 0x0D.

### 8.3 SPI Interface Timing

Figure 8-2. SPI Timing

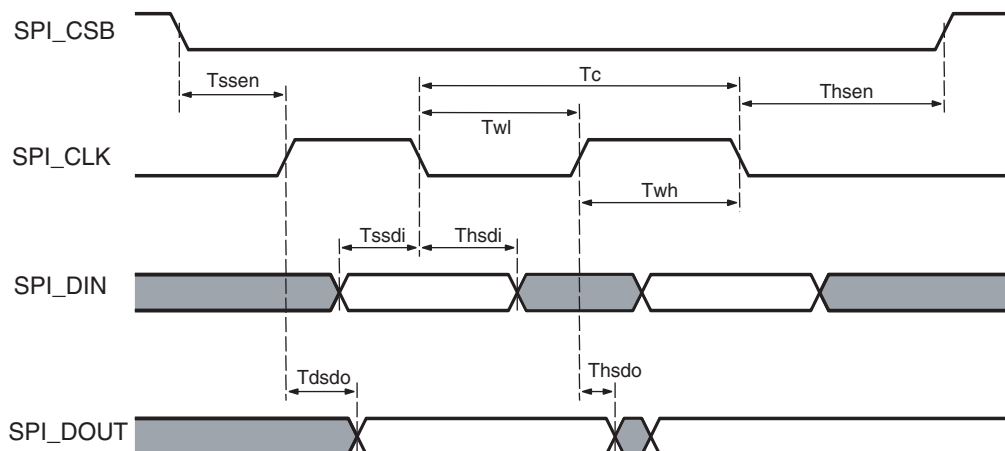


Table 8-1. SPI Timing Parameters

Parameter	Description	Min	Max
$T_c$	SPI_CLK min period	100 ns	-
$T_{wl}$	SPI_CLK min pulse width low	50 ns	-
$T_{wh}$	SPI_CLK min pulse width high	50 ns	-
$T_{ssen}$	Setup time SPI_CS# falling to SPI_CLK rising	50 ns	-
$T_{hsen}$	Hold time SPI_CLK falling to SPI_CS# rising	50 ns	-
$T_{ssdi}$	Setup time SPI_DIN valid to SPI_CLK falling	20 ns	-
$T_{hsdi}$	Hold time SPI_CLK falling to SPI_DIN not valid	20 ns	-
$T_{dsdo}$	Delay time SPI_CLK rising to SPI_DOUT valid	-	20 ns
$T_{hsdo}$	Hold time SPI_CLK rising to SPI_DOUT not valid	0 ns	-

## 8.4 SPI User Interface

**Table 8-2.** SPI Register Mapping

Address	Register	Name	Access	Reset State
0x00	DAC_CTRL	DAC Control	Read/Write	0x00
0x01	DAC_LLIG	DAC Left Line In Gain	Read/Write	0x05
0x02	DAC_RLIG	DAC Right Line In Gain	Read/Write	0x05
0x03	DAC_LPMG	DAC Left Master Playback Gain	Read/Write	0x08
0x04	DAC_RPMG	DAC Right Master Playback Gain	Read/Write	0x08
0x05	DAC_LLOG	DAC Left Line Out Gain	Read/Write	0x00
0x06	DAC_RLOG	DAC Right Line Out Gain	Read/Write	0x00
0x07	DAC_OLC	DAC Output Level Control	Read/Write	0x22
0x08	DAC_MC	DAC Mixer Control	Read/Write	0x09
0x09	DAC_CSFC	DAC Clock and Sampling Frequency Control	Read/Write	0x00
0x0A	DAC_MISC	DAC Miscellaneous	Read/Write	0x00
0x0C	DAC_PRECH	DAC Precharge Control	Read/Write	0x00
0x0D	DAC_AUXG	DAC Auxiliary Input Gain Control	Read/Write	0x05
0x10	DAC_RST	DAC Reset	Read/Write	0x00
0x11	PA_CRTL	Power Amplifier Control	Read/Write	0x00

Note: MSB = Bit 7, LSB = Bit 0

### 8.5 DAC Control Register

Register Name: DAC\_CTRL

Reset State: 0x00

Access: Read/Write

7	6	5	4	3	2	1	0
ONPADRV	ON AUXIN	ONDACR	ONDA CL	ONLNOR	ONLNOL	ONLNIR	ONLNIL

- **ONLNIL**  
Left channel line in amplifier (L to power down, H to power up)
- **ONLNIR**  
Right channel line in amplifier (L to power down, H to power up)
- **ONLNOL**  
Left channel line out driver (L to power down, H to power up)
- **ONLNOR**  
Right channel line out driver (L to power down, H to power up)
- **ONDA CL**  
Left channel DAC (L to power down, H to power up)
- **ONDACR**  
Right channel DAC (L to power down, H to power up)
- **ON AUXIN**  
Differential mono auxiliary input amplifier (L to power down, H to power up)
- **ONPADRV**  
Differential mono PA driver (L to power down, H to power up)

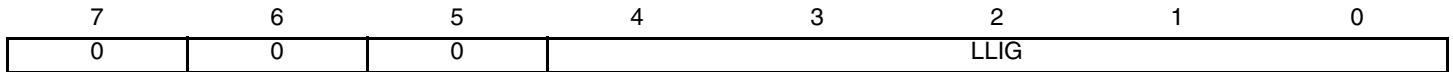


## 8.6 DAC Left Line In Gain Register

Register Name: DAC\_LLIG

Reset State: 0x05

Access: Read/Write



- **LLIG: Left channel line in analog gain selector**

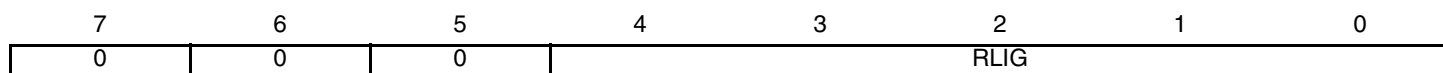
LLIG<4:0>	Gain	Unit
00000	20	dB
00001	12	dB
00010	9	dB
00011	6	dB
00100	3	dB
00101	0	dB
00110	-3	dB
00111	-6	dB
01000	-9	dB
01001	-12	dB
01010	-15	dB
01011	-18	dB
01100	-21	dB
01101	-24	dB
01110	-27	dB
01111	-30	dB
10000	-33	dB
≥10001	< -60	dB

## 8.7 DAC Right Line In Gain Register

Register Name: DAC\_RLIG

Reset State: 0x05

Access: Read/Write



- **RLIG:** Right channel line in analog gain selector

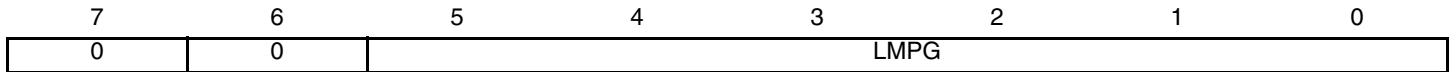
RLIG<4:0>	Gain	Unit
00000	20	dB
00001	12	dB
00010	9	dB
00011	6	dB
00100	3	dB
00101	0	dB
00110	-3	dB
00111	-6	dB
01000	-9	dB
01001	-12	dB
01010	-15	dB
01011	-18	dB
01100	-21	dB
01101	-24	dB
01110	-27	dB
01111	-30	dB
10000	-33	dB
≥10001	< -60	dB

## 8.8 DAC Left Master Playback Gain Register

Register Name: DAC\_LMPG

Reset State: 0x08

Access: Read/Write



- **LMPG: Left channel master playback digital gain selector**

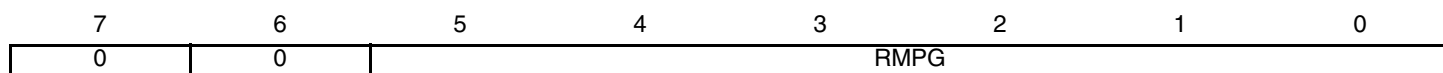
LMPG<5:0>	Gain	Unit	LMPG<5:0>	Gain	Unit
000000	12.0	dB	010001	-13.5	dB
000001	10.5	dB	010010	-15.0	dB
000010	9.0	dB	010011	-16.5	dB
000011	7.5	dB	010100	-18.0	dB
000100	6.0	dB	010101	-19.5	dB
000101	4.5	dB	010110	-21.0	dB
000110	3.0	dB	010111	-22.5	dB
000111	1.5	dB	011000	-24.0	dB
001000	0.0	dB	011001	-25.5	dB
001001	-1.5	dB	011010	-27.0	dB
001010	-3.0	dB	011011	-28.5	dB
001011	-4.5	dB	011100	-30.0	dB
001100	-6.0	dB	011101	-31.5	dB
001101	-7.5	dB	011110	-33.0	dB
001110	-9.0	dB	011111	-34.5	dB
001111	-10.5	dB	≥ 100000	mute	dB
010000	-12.0	dB			

### 8.9 DAC Right Master Playback Gain Register

Register Name: DAC\_RMPG

Reset State: 0x08

Access: Read/Write



- **RMPG: Right channel master playback digital gain selector**

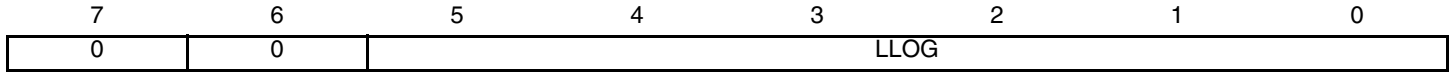
RMPG<5:0>	Gain	Unit	RMPG<5:0>	Gain	Unit
000000	12.0	dB	010001	-13.5	dB
000001	10.5	dB	010010	-15.0	dB
000010	9.0	dB	010011	-16.5	dB
000011	7.5	dB	010100	-18.0	dB
000100	6.0	dB	010101	-19.5	dB
000101	4.5	dB	010110	-21.0	dB
000110	3.0	dB	010111	-22.5	dB
000111	1.5	dB	011000	-24.0	dB
001000	0.0	dB	011001	-25.5	dB
001001	-1.5	dB	011010	-27.0	dB
001010	-3.0	dB	011011	-28.5	dB
001011	-4.5	dB	011100	-30.0	dB
001100	-6.0	dB	011101	-31.5	dB
001101	-7.5	dB	011110	-33.0	dB
001110	-9.0	dB	011111	-34.5	dB
001111	-10.5	dB	≥ 100000	mute	dB
010000	-12.0	dB			

## 8.10 DAC Left Line Out Gain Register

Register Name: DAC\_LLOG

Reset State: 0x00

Access: Read/Write



- **LLOG: Left channel line out digital gain selector**

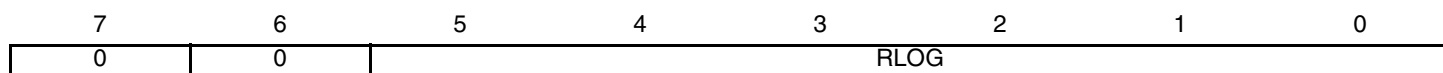
LLOG<5:0>	Gain	Unit	LLOG<5:0>	Gain	Unit
000000	0.0	dB	010000	-24.0	dB
000001	-1.5	dB	010001	-25.5	dB
000010	-3.0	dB	010010	-27.0	dB
000011	-4.5	dB	010011	-28.5	dB
000100	-6.0	dB	010100	-30.0	dB
000101	-7.5	dB	010101	-31.5	dB
000110	-9.0	dB	010110	-33.0	dB
000111	-10.5	dB	010111	-34.5	dB
001000	-12.0	dB	011000	-36.0	dB
001001	-13.5	dB	011001	-37.5	dB
001010	-15.0	dB	011010	-39.0	dB
001011	-16.5	dB	011011	-40.5	dB
001100	-18.0	dB	011100	-42.0	dB
001101	-19.5	dB	011101	-43.5	dB
001110	-21.0	dB	011110	-45.0	dB
001111	-22.5	dB	011111	-46.5	dB
			≥ 100000	mute	dB

## 8.11 DAC Right Line Out Gain Register

Register Name: DAC\_RLOG

Reset State: 0x00

Access: Read/Write



• **RLOG: Right channel line out digital gain selector**

RLOG<5:0>	Gain	Unit	RLOG<5:0>	Gain	Unit
000000	0.0	dB	010000	-24.0	dB
000001	-1.5	dB	010001	-25.5	dB
000010	-3.0	dB	010010	-27.0	dB
000011	-4.5	dB	010011	-28.5	dB
000100	-6.0	dB	010100	-30.0	dB
000101	-7.5	dB	010101	-31.5	dB
000110	-9.0	dB	010110	-33.0	dB
000111	-10.5	dB	010111	-34.5	dB
001000	-12.0	dB	011000	-36.0	dB
001001	-13.5	dB	011001	-37.5	dB
001010	-15.0	dB	011010	-39.0	dB
001011	-16.5	dB	011011	-40.5	dB
001100	-18.0	dB	011100	-42.0	dB
001101	-19.5	dB	011101	-43.5	dB
001110	-21.0	dB	011110	-45.0	dB
001111	-22.5	dB	011111	-46.5	dB
			≥ 100000	mute	dB

## 8.12 DAC Output Level Control Register

Register Name: DAC\_OLC

Reset State: 0x22

Access: Read/Write

7	6	5	4	3	2	1	0
RSHORT	ROLC			LSHORT	LOLC		

- **LOLC: Left channel output level control selector**

LOLC	Gain	Unit
000	6	dB
001	3	dB
010	0	dB
011	-3	dB
≥ 100	-6	dB

- **LSHORT: Left channel short circuit indicator**

Persistent; after being set, bit is not cleared automatically even after the short circuit is eliminated; must be cleared by reset cycle or direct register write operation.

- **ROLC: Right channel output level control selector**

ROLC	Gain	Unit
000	6	dB
001	3	dB
010	0	dB
011	-3	dB
≥ 100	-6	dB

- **RSHORT: Right channel short circuit indicator**

Persistent; after being set, bit is not cleared automatically even after the short circuit is eliminated; must be cleared by reset cycle or direct register write operation.

### 8.13 DAC Mixer Control Register

Register Name: DAC\_MC

Reset State: 0x09

Access: Read/Write

7	6	5	4	3	2	1	0
0	0	INVR	INVL	RMSMIN2	RMSMIN1	LMSMIN2	LMSMIN1

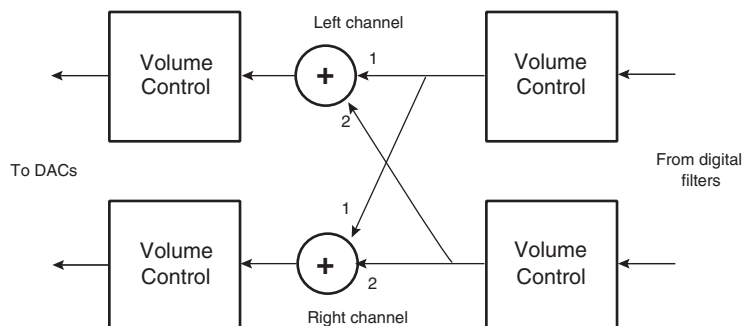
- **LMSMIN1: Left Channel Mono/Stereo Mixer Left Mixed input enable (H to enable, L to disable)**
- **LMSMIN2: Left Channel Mono/Stereo Mixer Right Mixed input enable (H to enable, L to disable)**
- **RMSMIN1: Right Channel Mono/Stereo Mixer Left Mixed input enable (H to enable, L to disable)**
- **RMSMIN2: Right Channel Mono/Stereo Mixer Right Mixed input enable (H to enable, L to disable)**
- **INVL: Left channel mixer output invert (H to enable, L to disable)**
- **INVR: Right channel mixer output invert (H to enable, L to disable)**

#### 8.13.1 Digital Mixer Control

The Audio DAC features a digital mixer that allows the mixing and selection of multiple input sources.

The mixing/multiplexing functions are described in [Figure 8-3](#) and [Table 8-3](#).

**Figure 8-3.** Digital Mixer Functions



Note: When the two mixer inputs are selected, a -6 dB gain is applied to the output signal. When only one input is selected, no gain is applied.

**Table 8-3.** Digital Mixer Signal Description

Signal	Description
LMSMIN1	Left Channel Mono/Stereo Mixer Left Mixed input enable - High to enable, Low to disable
LMSMIN2	Left Channel Mono/Stereo Mixer Right Mixed input enable - High to enable, Low to disable
RMSMIN1	Right Channel Mono/Stereo Mixer Left Mixed input enable - High to enable, Low to disable
RMSMIN2	Right Channel Mono/Stereo Mixer Right Mixed input enable - High to enable, Low to disable



## 8.14 DAC Clock and Sampling Frequency Control Register

Register Name: DAC\_CSFC

Reset State: 0x00

Access: Read/Write

7	6	5	4	3	2	1	0
0	0	0	OVRSEL	0	0	0	0

- **OVRSEL: Master clock selector**

L to 256 x Fs, H to 384 x Fs

Master clock and sampling frequency selection

[Table 8-4](#) describes the modes available for master clock and sampling frequency selection.

**Table 8-4.** Master Clock Modes

OVRSEL	Master Clock
0	256 x Fs
1	384 x Fs

### 8.15 DAC Miscellaneous Register

Register Name: DAC\_MISC

Reset State: 0x00

Access: Read/Write



- **NBITS<1:0>: Data interface word length**

The selection of input sample size is done using the NBITS field.

NBITS <1:0>	Format
00	16 bits
01	18 bits
10	20 bits

- **DEEMPEN: De-emphasis enable (L to disable, H to enable)**

To enable the de-emphasis filtering the DEEMPEN signal must be set to high.

- **DITHEN: Dither enable (L to disable, H to enable)**

The dither option (added in the playback channel) is enabled by setting the DITHEN signal to high.

- **DINTSEL<1:0>: I2S data format selector**

The selection between modes is done using the DINTSEL<1:0> signal.

DINTSEL<1:0>	Format
00	I2S Justified
01	MSB Justified
1x	LSB Justified

- **VCMCAPSEL: VCM decoupling capacitor selector**

L for 10 µF, H for 100 µF