WM8768

24-bit, 192kHz 8-Channel DAC

DESCRIPTION

The WM8768 is a multi-channel audio DAC ideal for DVD and surround sound processing applications for home hi-fi, automotive and other audio visual equipment.

Four stereo 24-bit multi-bit sigma delta DACs are used with oversampling digital interpolation filters. Digital audio input word lengths from 16-32 bits and sampling rates from 8kHz to 192kHz are supported. Each DAC channel has independent digital volume and mute control.

The audio data interface supports I^2S, left justified, right justified and DSP digital audio formats.

The device is controlled via either via a 3 wire serial interface or directly using the hardware interface. These interfaces provide access to features including channel selection, volume controls, mutes, de-emphasis and power management facilities. The device is available in a 28-pin SSOP.

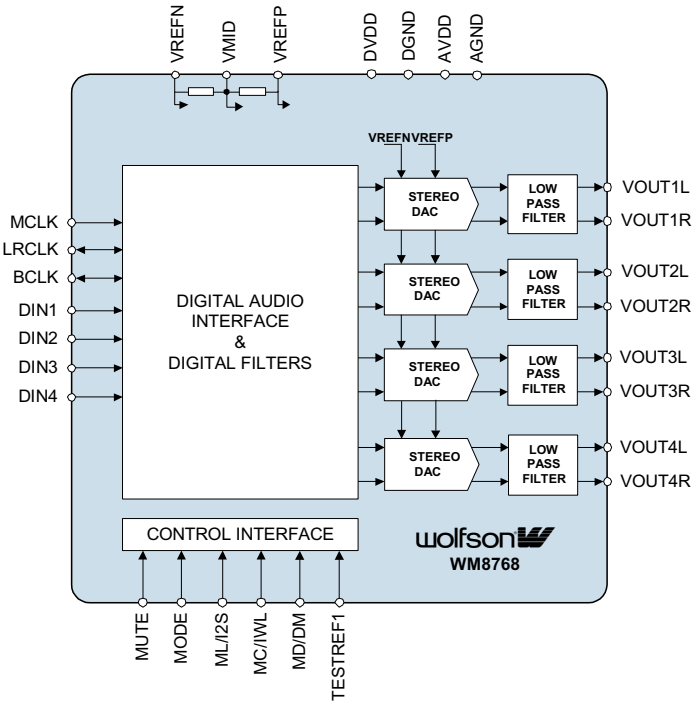
FEATURES

- 8-Channel DAC with PCM.
- Audio Performance
  - 103dB SNR ('A' weighted @ 48kHz) DAC
- DAC Sampling Frequency: 8kHz – 192kHz
- 3-Wire SPI Serial or Hardware Control Interface
- Programmable Audio Data Interface Modes
  - I^2S, Left, Right Justified or DSP
  - 16/20/24/32 bit Word Lengths
- Four Independent stereo DAC outputs with independent digital volume controls
- Master or Slave Audio Data Interface
- 2.7V to 5.5V Analogue, 2.7V to 3.6V Digital supply Operation
- 28 pin SSOP Package

APPLICATIONS

- DVD Players
- Surround Sound AV Processors and Hi-Fi systems
- Automotive Audio

BLOCK DIAGRAM



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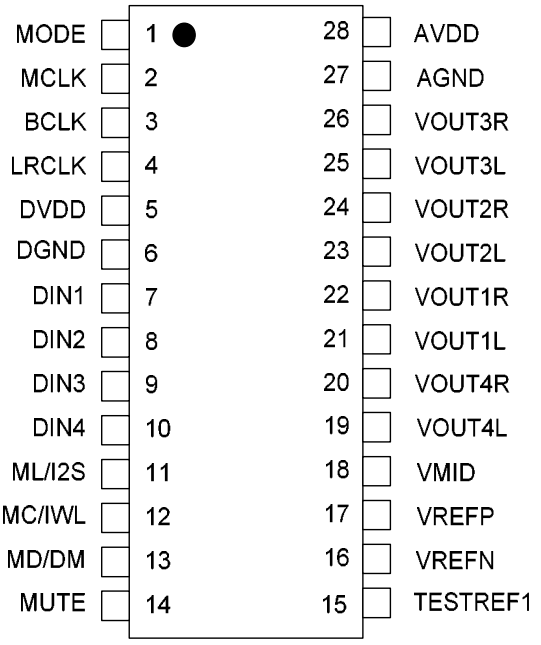
PIN CONFIGURATION 28 LEAD SSOP



ORDERING INFORMATION

DEVICE	TEMP. RANGE	PACKAGE	MOISTURE SENSITIVITY LEVEL
WM8768EDS	-25 to +85°C	28-pin SSOP	MSL1
WM8768GEDS	-25 to +85°C	28-pin SSOP (lead free)	MSL1
WM8768EDS/R	-25 to +85°C	28-pin SSOP (tape and reel)	MSL1
WM8768GEDS/R	-25 to +85°C	28-pin SSOP (lead free, tape and reel)	MSL1

Note:

Reel quantity = 2,000

PIN DESCRIPTION – 28 PIN SSOP

PIN	NAME	TYPE	DESCRIPTION
1	MODE	Digital input	Control format selection 0 = Software control 1 = Hardware control
2	MCLK	Digital input	Master clock; 128, 192, 256, 384, 512, 768fs or 1152fs (fs = word clock frequency)
3	BCLK	Digital input/output	Audio interface bit clock
4	LRCLK	Digital input/output	Audio left/right word clock
5	DVDD	Supply	Digital positive supply
6	DGND	Supply	Digital negative supply
7	DIN1	Digital input	DAC channel 1 data input
8	DIN2	Digital input	DAC channel 2 data input
9	DIN3	Digital input	DAC channel 3 data input
10	DIN4	Digital input	DAC channel 4 data input
11	ML/I2S	Digital input	Software Mode: Serial interface Latch signal Hardware Mode: Input Audio Data Format
12	MC/IWL	Digital input	Software Mode: Serial control interface clock Hardware Mode: Audio data input word length
13	MD/DM	Digital input	Software Mode: Serial interface data Hardware Mode: De-emphasis selection
14	MUTE	Digital input/output	DAC Zero Flag output or DAC mute input
15	TESTREF1	Digital input	Test Pin
16	VREFN	Analogue input	DAC negative reference supply
17	VREFP	Analogue input	DAC positive reference supply
18	VMID	Analogue output	Midrail divider decoupling pin; 10uF external decoupling
19	VOUT4L	Analogue output	DAC channel 1 left output
20	VOUT4R	Analogue output	DAC channel 1 right output
21	VOUT3L	Analogue output	DAC channel 2 left output
22	VOUT3R	Analogue output	DAC channel 2 right output
23	VOUT2L	Analogue output	DAC channel 3 left output
24	VOUT2R	Analogue output	DAC channel 3 right output
25	VOUT1L	Analogue output	DAC channel 4 left output
26	VOUT1R	Analogue output	DAC channel 4 right output
27	AGND	Supply	Analogue negative supply and substrate connection
28	AVDD	Supply	Analogue positive supply

Note : Digital input pins have Schmitt trigger input buffers.

ABSOLUTE MAXIMUM RATINGS

Absolute Maximum Ratings are stress ratings only. Permanent damage to the device may be caused by continuously operating at or beyond these limits. Device functional operating limits and guaranteed performance specifications are given under Electrical Characteristics at the test conditions specified.

ESD Sensitive Device. This device is manufactured on a CMOS process. It is therefore generically susceptible to damage from excessive static voltages. Proper ESD precautions must be taken during handling and storage of this device.

Wolfson tests its package types according to IPC/JEDEC J-STD-020B for Moisture Sensitivity to determine acceptable storage conditions prior to surface mount assembly. These levels are:

MSL1 = unlimited floor life at <30°C / 85% Relative Humidity. Not normally stored in moisture barrier bag.
MSL2 = out of bag storage for 1 year at <30°C / 60% Relative Humidity. Supplied in moisture barrier bag.
MSL3 = out of bag storage for 168 hours at <30°C / 60% Relative Humidity. Supplied in moisture barrier bag.

The Moisture Sensitivity Level for each package type is specified in Ordering Information.

CONDITION	MIN	MAX
Digital supply voltage	-0.3V	+5V
Analogue supply voltage	-0.3V	+7V
Voltage range digital inputs	DGND -0.3V	DVDD +0.3V
Voltage range analogue inputs	AGND -0.3V	AVDD +0.3V
Master Clock Frequency		37MHz
Operating temperature range, T_A	-25°C	+85°C
Storage temperature after soldering	-65°C	+150°C
Package body temperature (soldering 10 seconds)		+240°C
Package body temperature (soldering 2 minutes)		+183°C

Notes:

1. Analogue and digital grounds must always be within 0.3V of each other for normal operation of the device.

RECOMMENDED OPERATING CONDITIONS

PARAMETER	SYMBOL	TEST CONDITIONS	MIN	TYP	MAX	UNIT
Digital supply range	DVDD		2.7		3.6	V
Analogue reference supply	VREFP		2.7		5.5	V
Analogue supply range	AVDD		2.7		5.5	V
Ground	AGND, VREFN, DGND			0		V
Difference DGND to AGND			-0.3	0	+0.3	V

Note: Digital supply DVDD must never be more than 0.3V greater than AVDD for normal operation of the device.

ELECTRICAL CHARACTERISTICS

Test Conditions

AVDD, VREFP = 5V, DVDD = 3.3V, AGND, VREFN = 0V, DGND = 0V, T_A = +25°C, fs = 48kHz, MCLK = 256fs, unless otherwise stated.

PARAMETER	SYMBOL	TEST CONDITIONS	MIN	TYP	MAX	UNIT
Digital Logic Levels (CMOS Levels)						
Input LOW level	V_{IL}				0.3 x DVDD	V
Input HIGH level	V_{IH}		0.7 x DVDD			V
Output LOW	V_{OL}	I_{OL}=1mA			0.1 x DVDD	V
Output HIGH	V_{OH}	I_{OH}= -1mA	0.9 x DVDD			V
Analogue Reference Levels						
Reference voltage	V_{VMID}			VREFP/2		V
Potential divider resistance	R_{VMID}	VREFP to VMID and VMID to VREFN		100k		Ω
DAC Performance (Load = 10kΩ, 50pF)						
0dBFs Full scale output voltage				1.0 x VREFP/5		Vrms
SNR (Note 1,2,4)		A-weighted, @ fs = 48kHz	95	103		dB
SNR (Note 1,2,4)		A-weighted @ fs = 96kHz		101		dB
SNR (Note 1,2,4)		A-weighted @ fs = 192kHz		101		dB
SNR (Note 1,2,4)		A-weighted @ fs = 48kHz, AVDD = 3.3V		101		dB
SNR (Note 1,2,4)		A-weighted @ fs = 96kHz, AVDD = 3.3V		96		dB
Dynamic Range (Note 2,4)	DNR	A-weighted, -60dB full scale input	95	103		dB
Total Harmonic Distortion (THD) (Note 4)		1kHz, 0dBFs		-90	-85	dB
Mute Attenuation		1kHz Input, 0dB gain		100		dB
DAC channel separation				100		dB
Power Supply Rejection Ratio	PSRR	1kHz 100mVpp		50		dB
		20Hz to 20kHz 100mVp-p		45		dB
Supply Current						
Analogue supply current		AVDD = 5V		18.4		mA
Digital Supply Current		DVDD = 3.3V		14.6		mA

Notes:

1. Ratio of output level with 1kHz full scale input, to the output level with all zeros into the digital input, measured 'A' weighted.
2. All performance measurements done with 20kHz low pass filter, and where noted an A-weight filter. Failure to use such a filter will result in higher THD+N and lower SNR and Dynamic Range readings than are found in the Electrical Characteristics. The low pass filter removes out of band noise; although it is not audible it may affect dynamic specification values.
3. VMID decoupled with 10uF and 0.1uF capacitors (smaller values may result in reduced performance).
4. The performance of each DAC is measured separately

TERMINOLOGY

1. Signal-to-noise ratio (dB) - SNR is a measure of the difference in level between the full scale output and the output with no signal applied. (No Auto-zero or Automute function is employed in achieving these results).
2. Dynamic range (dB) - DNR is a measure of the difference between the highest and lowest portions of a signal. Normally a THD+N measurement at 60dB below full scale. The measured signal is then corrected by adding the 60dB to it. (e.g. THD+N @ -60dB= -32dB, DR= 92dB).
3. THD+N (dB) - THD+N is a ratio, of the rms values, of (Noise + Distortion)/Signal.
4. Stop band attenuation (dB) - Is the degree to which the frequency spectrum is attenuated (outside audio band).
5. Channel Separation (dB) - Also known as Cross-Talk. This is a measure of the amount one channel is isolated from the other. Normally measured by sending a full scale signal down one channel and measuring the other.
6. Pass-Band Ripple - Any variation of the frequency response in the pass-band region.

MASTER CLOCK TIMING

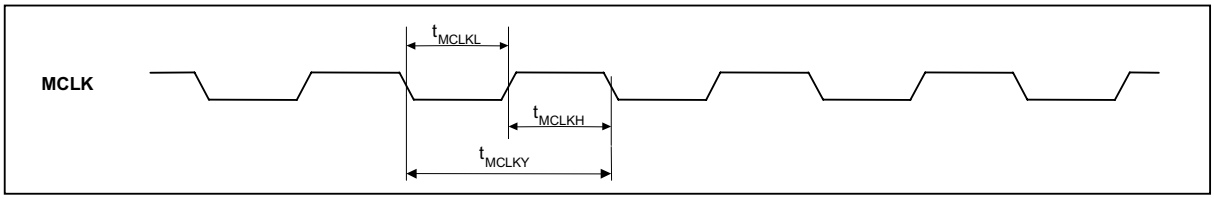


Figure 1 DAC Master Clock Timing Requirements

Test Conditions

AVDD, VREFP = 5V, DVDD = 3.3V, AGND, VREFN = 0V, DGND = 0V, T_A = +25°C, fs = 48kHz, MCLK = 256fs unless otherwise stated.

PARAMETER	SYMBOL	TEST CONDITIONS	MIN	TYP	MAX	UNIT
System Clock Timing Information						
MCLK System clock pulse width high	t_{MCLKH}		11			ns
MCLK System clock pulse width low	t_{MCLKL}		11			ns
MCLK System clock cycle time	t_{MCLKY}		28			ns
MCLK Duty cycle			40:60		60:40	

Table 1 Master Clock Timing Requirements

DIGITAL AUDIO INTERFACE – MASTER MODE

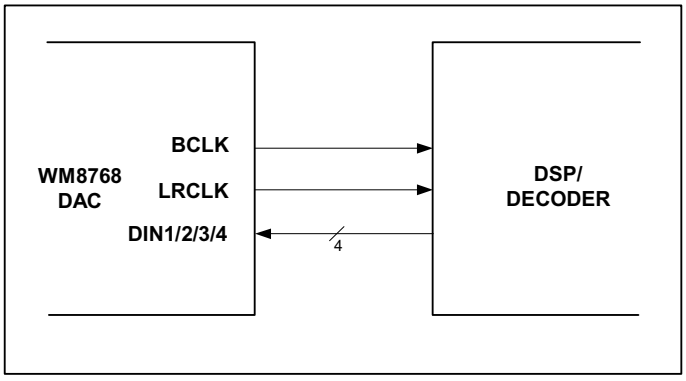


Figure 2 Audio Interface - Master Mode

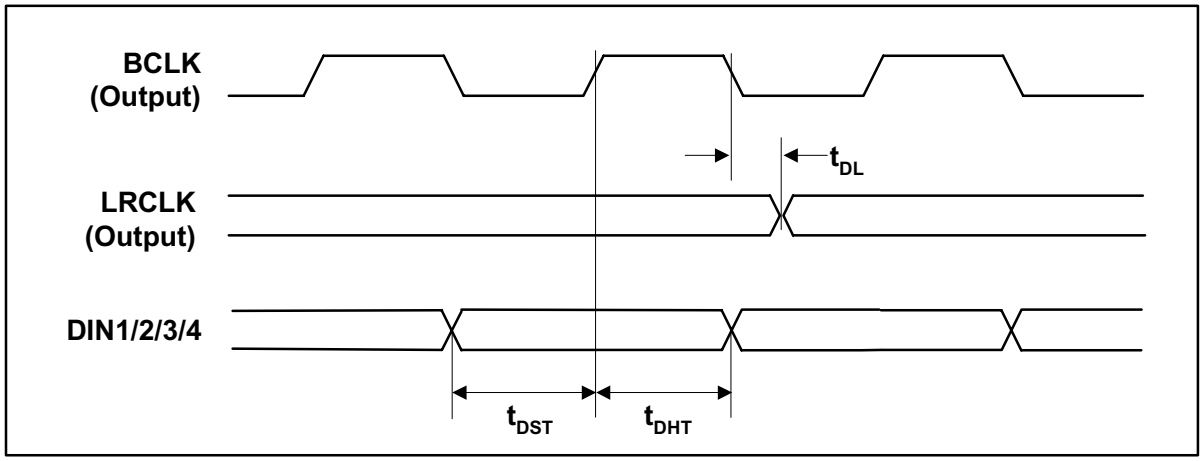


Figure 3 Digital Audio Data Timing – Master Mode

Test Conditions

AVDD, VREFP = 5V, DVDD = 3.3V, AGND, VREFN, DGND = 0V, T_A = +25°C, Master Mode, fs = 48kHz, MCLK = 256fs unless otherwise stated.

PARAMETER	SYMBOL	TEST CONDITIONS	MIN	TYP	MAX	UNIT
Audio Data Input Timing Information						
LRCLK propagation delay from BCLK falling edge	t_{DL}		0		10	ns
DIN1/2/3/4 setup time to BCLK rising edge	t_{DST}		10			ns
DIN1/2/3/4 hold time from BCLK rising edge	t_{DHT}		10			ns

Table 2 Digital Audio Data Timing – Master Mode

DIGITAL AUDIO INTERFACE – SLAVE MODE

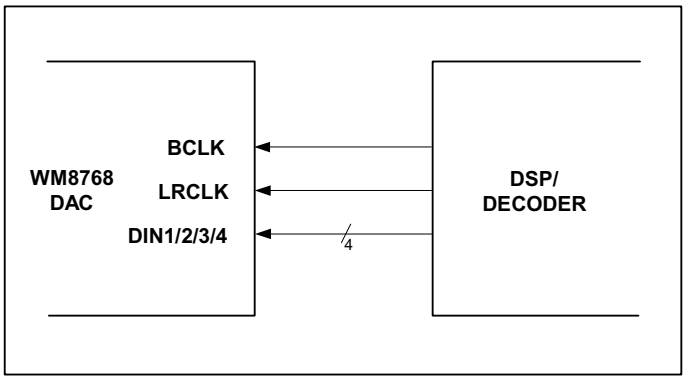


Figure 4 Audio Interface – Slave Mode

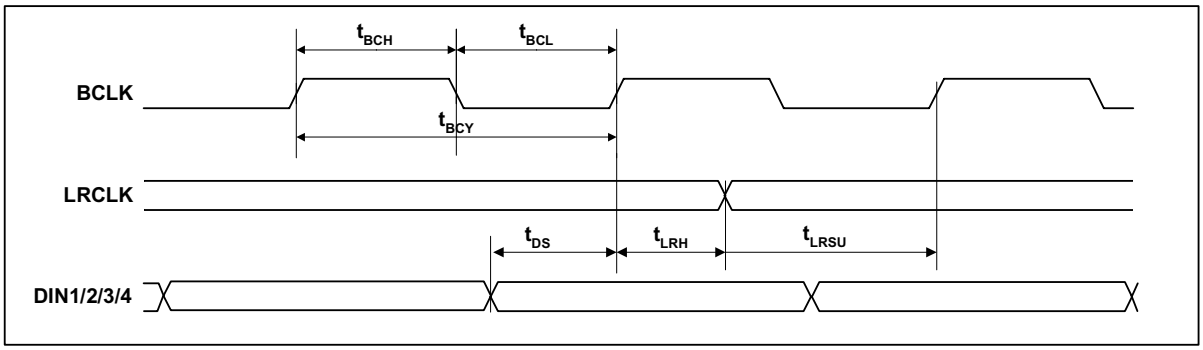


Figure 5 Digital Audio Data Timing – Slave Mode

Test Conditions

AVDD = 5V, DVDD = 3.3V, AGND = 0V, DGND = 0V, T_A = +25°C, Slave Mode, fs = 48kHz, MCLK = 256fs unless otherwise stated.

PARAMETER	SYMBOL	TEST CONDITIONS	MIN	TYP	MAX	UNIT
Audio Data Input Timing Information						
BCLK cycle time	t_{BCY}		50			ns
BCLK pulse width high	t_{BCH}		20			ns
BCLK pulse width low	t_{BCL}		20			ns
LRCLK set-up time to BCLK rising edge	t_{LRSU}		10			ns
LRCLK hold time from BCLK rising edge	t_{LRH}		10			ns
DIN1/2/3/4 set-up time to BCLK rising edge	t_{DS}		10			ns
DIN1/2/3/4 hold time from BCLK rising edge	t_{DH}		10			ns

Table 3 Digital Audio Data Timing – Slave Mode

MPU INTERFACE TIMING

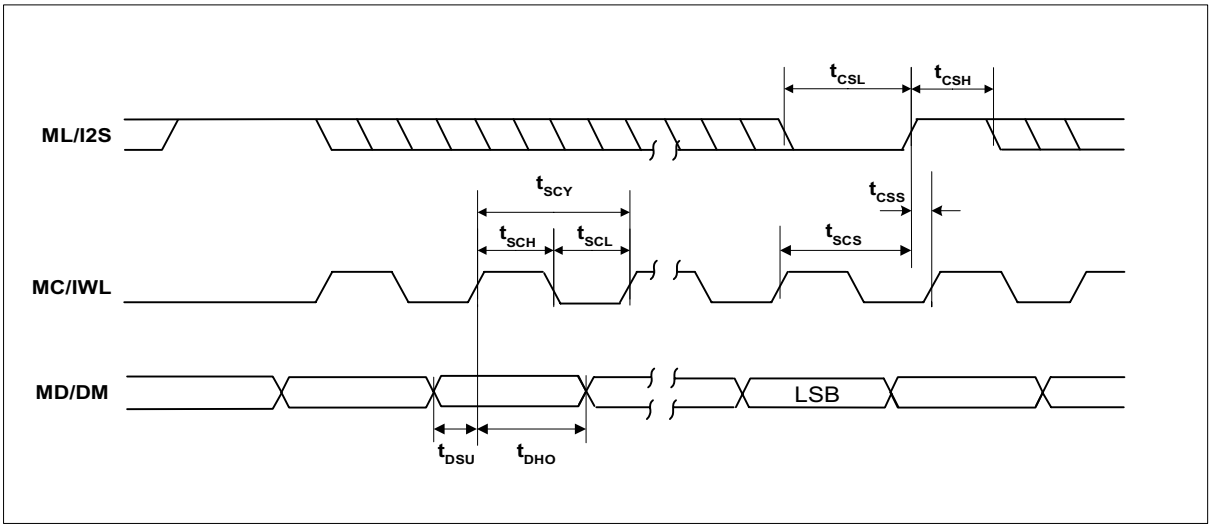


Figure 6 SPI Compatible Control Interface Input Timing

Test Conditions AVDD = 5V, DVDD = 3.3V, AGND, DGND = 0V, T_A = +25°C, fs = 48kHz, MCLK = 256fs unless otherwise stated					
PARAMETER	**SYMBOL**	**MIN**	**TYP**	**MAX**	**UNIT**
MC/IWL rising edge to ML/I2S rising edge	t_{SCS}	60			ns
MC/IWL pulse cycle time	t_{SCY}	80			ns
MC/IWL pulse width low	t_{SCL}	30			ns
MC/IWL pulse width high	t_{SCH}	30			ns
MD/DM to MC/IWL set-up time	t_{DSU}	20			ns
MC/IWL to MD/DM hold time	t_{DHO}	20			ns
ML/I2S pulse width low	t_{CSL}	20			ns
ML/I2S pulse width high	t_{CSH}	20			ns
ML/I2S rising to MC/IWL rising	t_{CSS}	20			ns

Table 4 3-wire SPI Compatible Control Interface Input Timing Information

DEVICE DESCRIPTION

INTRODUCTION

WM8768 is a complete 8-channel DAC including digital interpolation and decimation filters and switched capacitor multi-bit sigma delta DACs with digital volume controls on each channel and output smoothing filters.

The device is implemented as four separate stereo DACs in a single package and controlled by a single interface.

Each stereo DAC has its own data input DIN1/2/3/4. DAC word clock LRCLK, DAC bit clock BCLK and DAC master clock MCLK are shared between them.

The Audio Interface may be configured to operate in either master or slave mode. In Slave mode, LRCLK and BCLK are all inputs. In Master mode, LRCLK and BCLK are all outputs.

Each DAC has its own digital volume control that is adjustable in 0.5dB steps. The digital volume controls may be operated independently. In addition, a zero cross detect circuit is provided for each DAC for the digital volume controls. The digital volume control detects a transition through the zero point before updating the volume. This minimises audible clicks and 'zipper' noise as the gain values change.

Control of internal functionality of the device is by 3-wire serial or pin programmable control interface. The software control interface may be asynchronous to the audio data interface as control data will be re-synchronised to the audio processing internally.

Operation using master clocks of 128fs, 192fs, 256fs, 384fs, 512fs, 768fs or 1152fs is provided for the DAC In Slave mode selection between clock rates is automatically controlled. Audio sample rates (fs) from less than 8ks/s up to 192ks/s are allowed for the DAC, provided the appropriate master clock is input.

In PCM mode, the audio data interface supports right justified, left justified and I^2S (Philips left justified, one bit delayed) interface formats along with a highly flexible DSP serial port interface.

PCM AUDIO DATA SAMPLING RATES

In a typical digital audio system there is only one central clock source producing a reference clock to which all audio data processing is synchronised. This clock is often referred to as the audio system's Master Clock. The external master system clock can be applied directly through the DAC MCLK input pin(s) with no software configuration necessary

The DAC master clock for WM8768 supports audio sampling rates from 128fs to 1152fs, where fs is the audio sampling frequency (LRCLK) typically 32kHz, 44.1kHz, 48kHz, 96kHz or 192kHz (for DAC operation only). The master clock is used to operate the digital filters and the noise shaping circuits.

In Slave mode the WM8768 has a master clock detection circuit that automatically determines the relationship between the system clock frequency and the sampling rate (to within +/- 32 master clocks). If there is a greater than 32 clocks error the interface defaults to 1152fs mode. The WM8768 is tolerant of phase variations or jitter on the master clock. Table 5 shows the typical master clock frequency inputs for the WM8768.

The signal processing for the WM8768 typically operates at an oversampling rate of 128fs. The exception to this is for operation with a 128/192fs system clock, e.g. for 192kHz operation, when the oversampling rate is 64fs.

SAMPLING RATE (LRCLK)	SYSTEM CLOCK FREQUENCY (MHZ)						
	128fs	192fs	256fs	384fs	512fs	768fs	1152fs
32kHz	4.096	6.144	8.192	12.288	16.384	24.576	36.864
44.1kHz	5.6448	8.467	11.2896	16.9340	22.5792	33.8688	Unavailable
48kHz	6.144	9.216	12.288	18.432	24.576	36.864	Unavailable
96kHz	12.288	18.432	24.576	36.864	Unavailable	Unavailable	Unavailable
192kHz	24.576	36.864	Unavailable	Unavailable	Unavailable	Unavailable	Unavailable

Table 5 System Clock Frequencies Versus Sampling Rate

HARDWARE CONTROL MODES

When the MODE pin is held high, the following hardware modes of operation are available.

MUTE AND AUTOMUTE OPERATION

In both hardware and software modes, MUTE controls the selection of MUTE directly, and can be used to enable and disable the automute function. This pin becomes an output when left floating and indicates infinite ZERO detect (IZD) has been detected.

	DESCRIPTION
0	Normal Operation
1	Mute DAC channels
Floating	Enable IZD, MUTE becomes an output to indicate when IZD occurs. L=IZD detected, H=IZD not detected.

Table 6 Mute and Automute Control

Figure 7 shows the application and release of MUTE whilst a full amplitude sinusoid is being played at 48kHz sampling rate. When MUTE (lower trace) is asserted, the output (upper trace) begins to decay exponentially from the DC level of the last input sample. The output will decay towards V_{MID} with a time constant of approximately 64 input samples. When MUTE is de-asserted, the output will restart almost immediately from the current input sample.

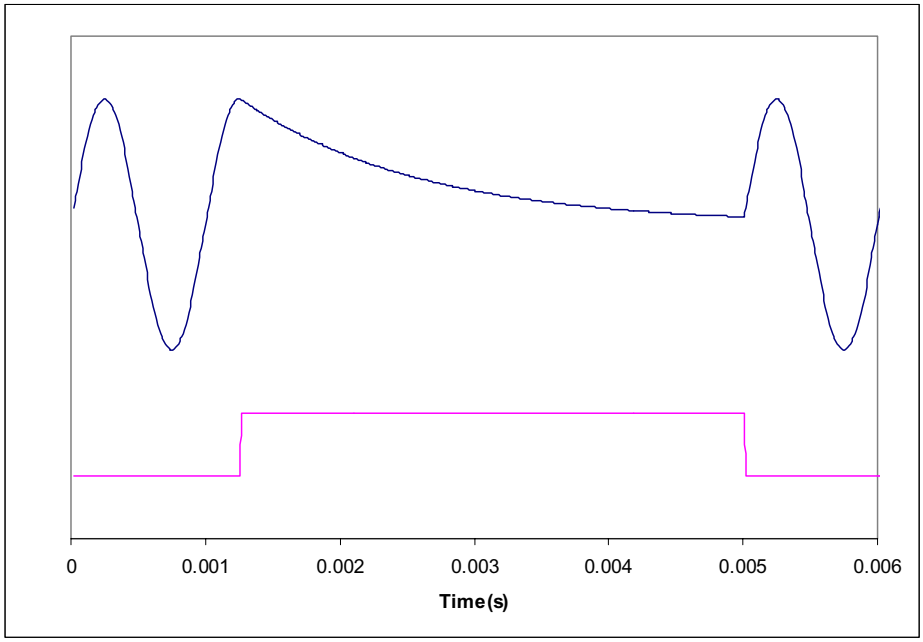


Figure 7 Application and Release of Soft Mute

The MUTE pin is an input to select mute or not mute. MUTE is active high; taking the pin high causes the filters to soft mute, ramping down the audio signal over a few milliseconds. Taking MUTE low again allows data into the filter.

The automute function detects a series of ZERO value audio samples of 1024 samples long being applied to both channels. After such an event, a latch is set whose output (AUTOMUTED) is wire OR'ed through a 10kΩ resistor to the MUTE pin. Thus if the MUTE pin is not being driven, the automute function will assert mute.

If MUTE is tied low, AUTOMUTED is overridden and will not mute unless the IZD register bit is set. If MUTE is driven from a bi-directional source, then both MUTE and automute functions are available. If MUTE is not driven, AUTOMUTED appears as a weak output (10kΩ source impedance) and can be used to drive external mute circuits. AUTOMUTED will be removed as soon as any channel receives a non-ZERO input.

A diagram showing how the various Mute modes interact is shown below Figure 8.

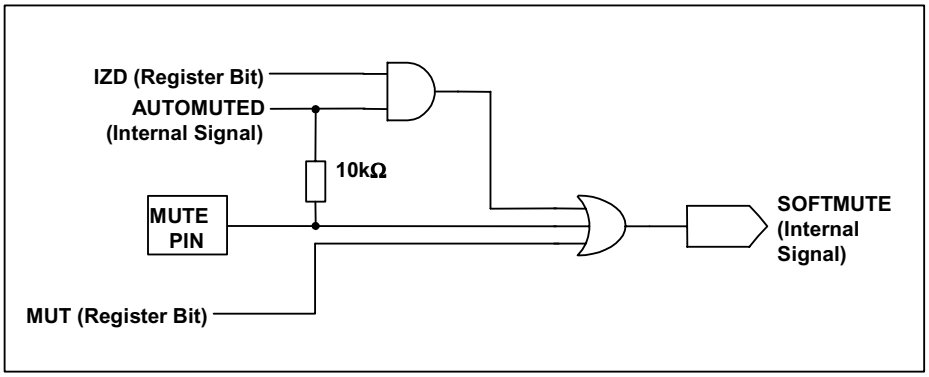


Figure 8 Selection Logic for MUTE Modes

INPUT FORMAT SELECTION

In hardware mode, ML/I2S and MC/IWL become input controls for selection of input data format type and input data word length for the DAC.

ML/I2S	MC/IWL	INPUT DATA MODE
0	0	24-bit right justified
0	1	20-bit right justified
1	0	16-bit I^2S
1	1	24-bit I^2S

Table 7 Input Format Selection

Note:

In 24 bit I^2S mode, any width of 24 bits or less is supported provided that the left/right clocks (LRCLK) are high for a minimum of 24 bit clocks (BCLK) and low for a minimum of 24 bit clocks. If exactly 32 bit clocks occur in one left/right clock (16 high, 16 low) the chip will auto detect and run a 16 bit data mode.

DE-EMPHASIS CONTROL

In hardware mode, the MD/DM pin becomes an input control for selection of de-emphasis filtering to be applied.

MD/DM	DE-EMPHASIS
0	Off
1	On

Table 8 De-emphasis Control

DIGITAL AUDIO INTERFACE

MASTER AND SLAVE MODES

The audio interface operates in either Slave or Master mode, selectable using the MS control bit. In both Master and Slave modes DIN1/2/3/4 are always inputs to the WM8768. The default is Slave mode.

In Slave mode, LRCLK and BCLK are inputs to the WM8768 (Figure 9). DIN1/2/3/4 and LRCLK are sampled by the WM8768 on the rising edge of BCLK.

By setting the control bit BCP the polarity of BCLK may be reversed so that DIN1/2/3/4 and LRCLK are sampled on the falling edge of BCLK .

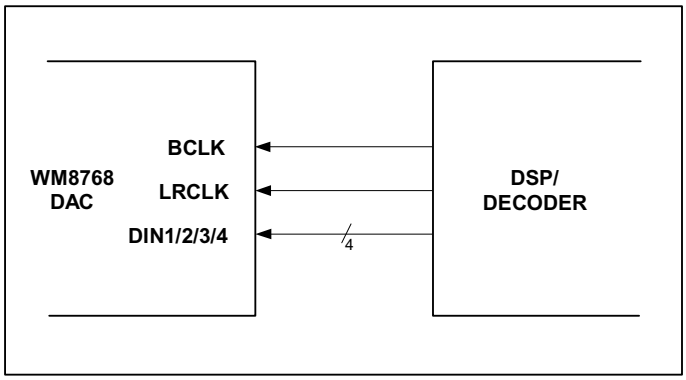


Figure 9 Slave Mode

In Master mode, LRCLK and BCLK are outputs from the WM8768 (Figure 10). LRCLK and BCLK are generated by the WM8768. DIN1/2/3/4 are sampled by the WM8768 on the rising edge of BCLK so the controller must output DAC data that changes on the falling edge of BCLK.

By setting control bit BCP the polarity of BCLK may be reversed so that DIN1/2/3/4 are sampled on the falling edge of BCLK..

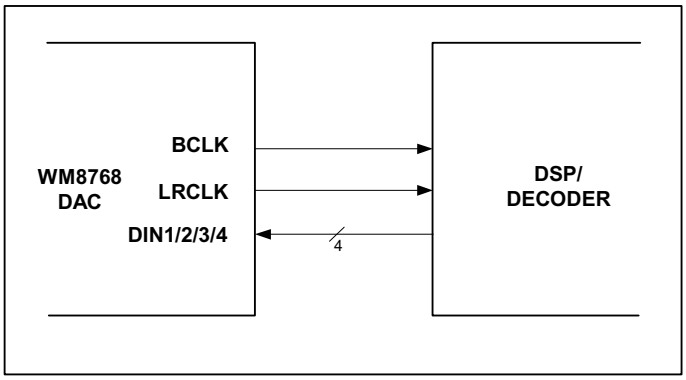


Figure 10 Master Mode

AUDIO INTERFACE FORMATS

Audio data is applied to the internal DAC filters via the Digital Audio Interface. 5 popular interface formats are supported:

- Left Justified mode
- Right Justified mode
- I^2S mode
- DSP Early mode
- DSP Late mode

All 5 formats send the MSB first and support word lengths of 16, 20, 24 and 32 bits, with the exception of 32 bit right justified mode, which is not supported.

In left justified, right justified and I^2S modes, the digital audio interface receives DAC data on the DIN1/2/3/4 inputs. Audio Data for each stereo channel is time multiplexed with LRCLK indicating whether the left or right channel is present. LRCLK is also used as a timing reference to indicate the beginning or end of the data words.

In left justified, right justified and I^2S modes, the minimum number of BCLKs per LRCLK period is 2 times the selected word length. LRCLK must be high for a minimum of word length BCLKs and low for a minimum of word length BCLKs. Any mark to space ratio on LRCLK is acceptable provided the above requirements are met.

In DSP early or DSP late mode, all 8 DAC channels are time multiplexed onto DIN1. LRCLK is used as a frame sync signal to identify the MSB of the first word. The minimum number of BCLKs per LRCLK period is 8 times the selected word length. Any mark to space ratio is acceptable on LRCLK provided the rising edge is correctly positioned.

LEFT JUSTIFIED MODE

In left justified mode, the MSB of DIN1/2/3/4 is sampled by the WM8768 on the first rising edge of BCLK following a LRCLK transition. LRCLK is high during the left samples and low during the right samples (Figure 11).

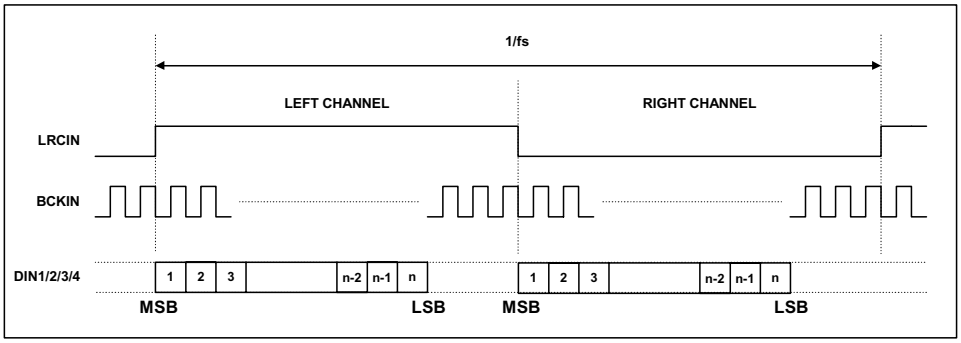


Figure 11 Left Justified Mode Timing Diagram

RIGHT JUSTIFIED MODE

In right justified mode, the LSB of DIN1/2/3/4 is sampled by the WM8768 on the rising edge of BCLK preceding a LRCLK transition. LRCLK are high during the left samples and low during the right samples (Figure 12).

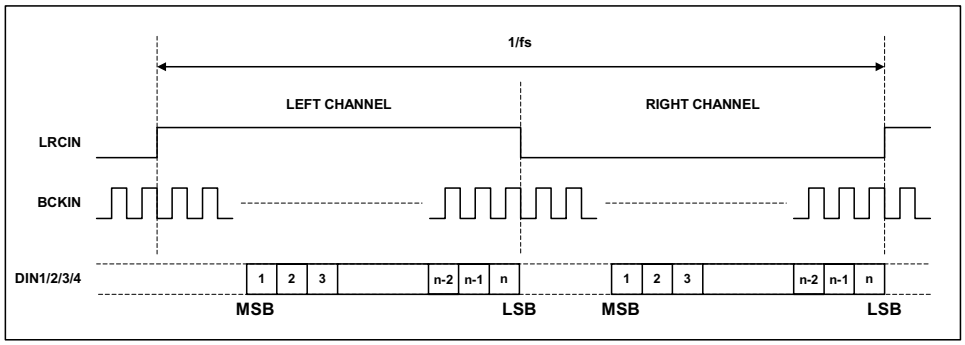


Figure 12 Right Justified Mode Timing Diagram

I2S MODE

In I^2S mode, the MSB of DIN1/2/3/4 is sampled by the WM8768 on the second rising edge of BCLK following a LRCLK transition. LRCLK are low during the left samples and high during the right samples.

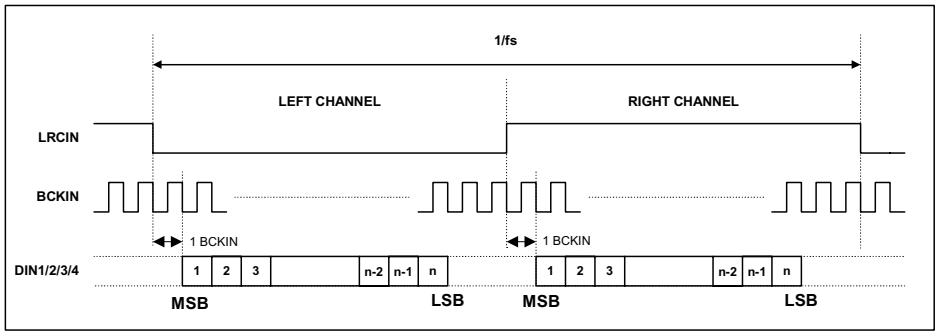


Figure 13 I^2S Mode Timing Diagram

DSP EARLY MODE

In DSP early mode, the MSB of DAC channel 1 left data is sampled by the WM8768 on the second rising edge on BCLK following a LRCLK rising edge. DAC channel 1 right and DAC channels 2/3/4 data follow DAC channel 1 left data (Figure 14).

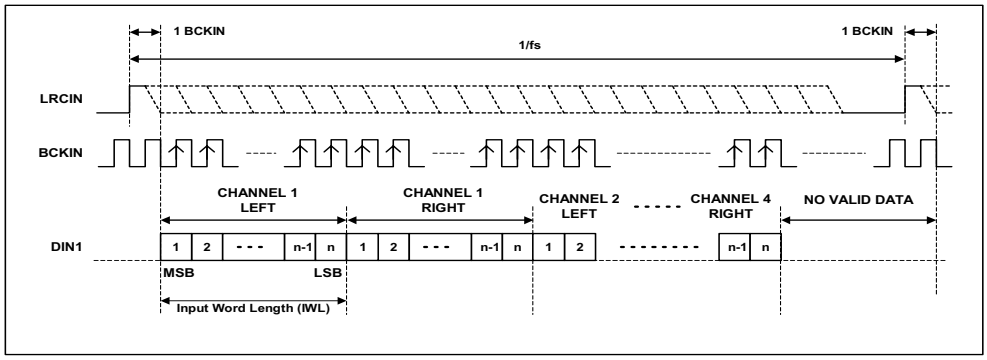


Figure 14 DSP Early Mode Timing Diagram – DAC data input

DSP LATE MODE

In DSP late mode, the MSB of DAC channel 1 left data is sampled by the WM8768 on the first BCLK rising edge following a LRCLK rising edge. DAC channel 1 right and DAC channels 2/3/4 data follow DAC channel 1 left data (Figure 15).

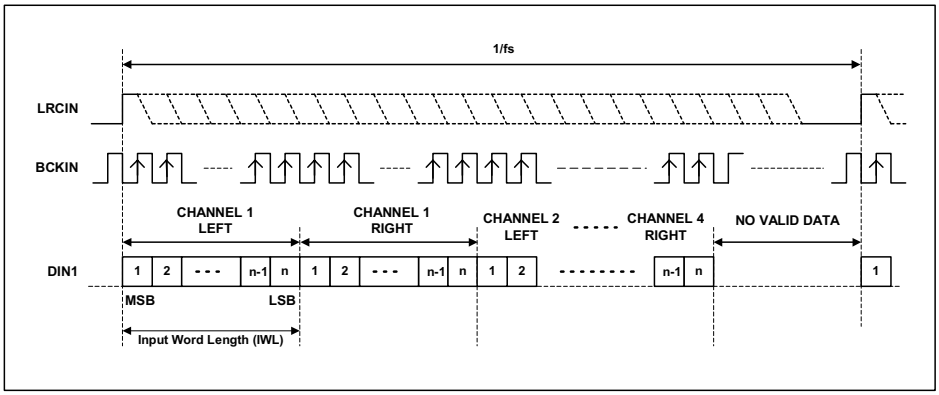


Figure 15 DSP Late Mode Timing Diagram – DAC data input

In both early and late DSP modes, DACL1 is always sent first, followed immediately by DACR1 and the data words for the other 8 channels. No BCLK edges are allowed between the data words. The word order is DAC1 left, DAC1 right, DAC2 left, DAC2 right, DAC3 left, DAC3 right, DAC4 left, DAC4 right .

POWERDOWN MODES

The WM8768 has powerdown control bits allowing specific parts of the WM8768 to be powered off when not being used. The four stereo DACs each have a separate powerdown control bit, DACD[2:0] & DACD4, allowing individual stereo DACs to be powered off when not in use. Setting DACD will power down everything except the references VMID and REFADC. These may be powered down by setting PDWN. Setting PDWN will override all other powerdown control bits. It is recommended that the DACs are powered down before setting PDWN.

ZERO DETECT

The WM8768 has a zero detect circuit for each DAC channel that detects when 1024 consecutive zero samples have been input. The MUTE pin output may be programmed to output the zero detect signal (see Table 9) which may then be used to control external muting circuits. A '1' on MUTE indicates a zero detect. The zero detect may also be used to automatically enable DAC mute by setting IZD.

{DZFM4,DZFM[1:0]}	MUTE
000	All channels zero
001	Channel 1 zero
010	Channel 2 zero
011	Channel 3 zero
100	Channel 4 zero
101	Channels 1 and 2 zero
110	Channels 1, 2 and 3 zero
111	Channels 3 and 4 zero

Table 9 Zero Flag Output Select

SOFTWARE CONTROL INTERFACE OPERATION

The WM8768 is controlled using a 3-wire serial interface in software mode or pin programmable in hardware mode.

The control mode is selected by the state of the MODE pin.

3-WIRE (SPI COMPATIBLE) SERIAL CONTROL MODE

MD/DM is used for the program data, MC/IWL is used to clock in the program data and ML/I2S is used to latch the program data. MD/DM is sampled on the rising edge of MC/IWL. The 3-wire interface protocol is shown in Figure 16.

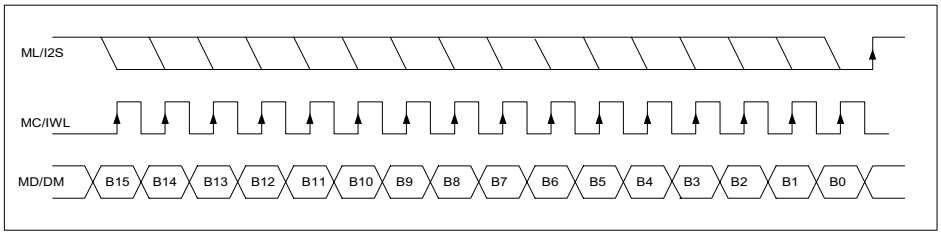


Figure 16 3-wire SPI Compatible Interface

1. B[15:9] are Control Address Bits
2. B[8:0] are Control Data Bits
3. ML/I2S is edge sensitive – the data is latched on the rising edge of ML/I2S.

CONTROL INTERFACE REGISTERS

ATTENUATOR CONTROL MODE

Setting the ATC register bit causes the left channel attenuation settings to be applied to both left and right channel DACs from the next audio input sample. No update to the attenuation registers is required for ATC to take effect.

REGISTER ADDRESS	BIT	LABEL	DEFAULT	DESCRIPTION
0000010 DAC Channel Control	3	ATC	0	Attenuator Control Mode: 0 : Right channels use Right attenuations 1: Right Channels use Left Attenuations

INFINITE ZERO DETECT ENABLE (PCM)

Setting the IZD register bit will enable the internal infinite zero detect function:

REGISTER ADDRESS	BIT	LABEL	DEFAULT	DESCRIPTION
0000010 DAC Channel Control	4	IZD	0	Infinite zero Mute Enable 0 : disable inifinite zero mute 1: enable infinite zero Mute

With IZD enabled, applying 1024 consecutive zero input samples each stereo channel will cause that stereo channel outputs to be muted to V_{MID}. Mute will be removed as soon as that stereo channel receives a non-zero input.

DAC OUTPUT CONTROL

The DAC output control word determines how the left and right inputs to the audio Interface are applied to the left and right DACs:

REGISTER ADDRESS	BIT	LABEL	DEFAULT	DESCRIPTION		
0000010 DAC Control	8:5	PL[3:0]	1001	PL[3:0]	Left Output	Right Output
				0000	Mute	Mute
				0001	Left	Mute
				0010	Right	Mute
				0011	(L+R)/2	Mute
				0100	Mute	Left
				0101	Left	Left
				0110	Right	Left
				0111	(L+R)/2	Left
				1000	Mute	Right
				1001	Left	Right
				1010	Right	Right
				1011	(L+R)/2	Right
				1100	Mute	(L+R)/2
				1101	Left	(L+R)/2
				1110	Right	(L+R)/2
				1111	(L+R)/2	(L+R)/2

DAC DIGITAL AUDIO INTERFACE CONTROL REGISTER

Interface format is selected via the FMT[1:0] register bits:

REGISTER ADDRESS	BIT	LABEL	DEFAULT	DESCRIPTION
0000011 Interface Control	1:0	FMT [1:0]	00	Interface format Select 00 : right justified mode 01: left justified mode 10: I^2S mode 11: DSP (early or late) mode

In left justified, right justified or I^2S modes, the LRP register bit controls the polarity of LRCLK. If this bit is set high, the expected polarity of LRCLK will be the opposite of that shown Figure 11, Figure 12 and Figure 13. Note that if this feature is used as a means of swapping the left and right channels, a 1 sample phase difference will be introduced. In DSP modes, the LRP register bit is used to select between early and late modes.

REGISTER ADDRESS	BIT	LABEL	DEFAULT	DESCRIPTION
0000011 Interface Control	2	LRP	0	In left/right/I^2S modes: LRCLK Polarity (normal) 0 : normal LRCLK polarity 1: inverted LRCLK polarity
				In DSP mode: 0 : Early DSP mode 1: Late DSP mode

By default, LRCLK and DIN1/2/3/4 are sampled on the rising edge of BCLK and should ideally change on the falling edge. By default, LRCLK and DOUT are sampled on the rising edge of BCLK and should ideally change on the falling edge. Data sources that change LRCLK and DOUT on the rising edge of BCLK can be supported by setting the BCP register bit. Data sources that change LRCLK and DIN1/2/3/4 on the rising edge of BCLK can be supported by setting the BCP register bit. Setting BCP to 1 inverts the polarity of BCLK to the inverse of that shown in Figure 11, Figure 12, Figure 13, Figure 14, Figure 15 and Figure 16.

REGISTER ADDRESS	BIT	LABEL	DEFAULT	DESCRIPTION
0000011 Interface Control	3	BCP	0	BCLK Polarity 0 : normal BCLK polarity 1: inverted BCLK polarity

The IWL[1:0] bits are used to control the input word length.

REGISTER ADDRESS	BIT	LABEL	DEFAULT	DESCRIPTION
0000011 Interface Control	5:4	IWL [1:0]	00	Input Word Length 00 : 16 bit data 01: 20 bit data 10: 24 bit data 11: 32 bit data

Note: If 32-bit mode is selected in right justified mode, the WM8768 defaults to 24 bits.

In all modes, the data is signed 2's complement. The digital filters always input 24-bit data. If the DAC is programmed to receive 16 or 20 bit data, the WM8768 pads the unused LSBs with zeros. If the DAC is programmed into 32 bit mode, the 8 LSBs are ignored.

Note: In 24 bit I^2S mode, any width of 24 bits or less is supported provided that LRCLK is high for a minimum of 24 BCLKs and low for a minimum of 24 BCLKs. If exactly 32 bit clocks occur in one left/right clock (16 high, 16 low) the chip will auto detect and run a 16 bit data mode.

A number of options are available to control how data from the Digital Audio Interface is applied to the DAC channels.

DAC OUTPUT PHASE

The DAC Phase control word determines whether the output of each DAC is non-inverted or inverted

REGISTER ADDRESS	BIT	LABEL	DEFAULT	DESCRIPTION		
0000011 DAC Phase	8:6	PHASE [2:0]	000	Bit	DAC	Phase
				0	DAC1L/R	1 = invert
				1	DAC2L/R	1 = invert
				2	DAC3L/R	1 = invert
0001111 DAC4 Control	3	PHASE4	0	N/A	DAC4L/R	1 = invert

DIGITAL ZERO CROSS-DETECT

The Digital volume control also incorporates a zero cross detect circuit which detects a transition through the zero point before updating the digital volume control with the new volume. This is enabled by control bit DZCEN.

REGISTER ADDRESS	BIT	LABEL	DEFAULT	DESCRIPTION
0001001 DAC Control	0	DZCEN	0	DAC Digital Volume Zero Cross Enable: 0: Zero cross detect enabled 1: Zero cross detect disabled

MUTE FLAG OUTPUT

The {DZFM4,DZFM[1:0]} control bits allow the selection of the 8 DAC channel zero flag bits for output on the MUTEB pin. A '1' on MUTE indicates 1024 consecutive zero input samples to the DAC channels selected.

REGISTER ADDRESS	BIT	LABEL	DEFAULT	DESCRIPTION
0001001 Zero Flag	2:1	DZFM[1:0]	00	Selects the output MUTE pin (A '1' indicates 1024 consecutive zero input samples on the DAC channels selected.
0001111 DAC4 Control	1	DZFM4	0	Selects the output MUTE pin (A '1' indicates 1024 consecutive zero input samples on the DAC channels selected.

{DZFM4,DZFM[1:0]}	MUTE
000	All channels zero
001	Channel 1 zero
010	Channel 2 zero
011	Channel 3 zero
100	Channel 4 zero
101	Channels 1 and 2 zero
110	Channels 1, 2 and 3 zero
111	Channels 3 and 4 zero

Table 10 Zero Flag Output Select

DAC MUTE MODES

The WM8768 has individual mutes for each of the four DAC channels. Setting MUTE for a channel will apply a 'soft' mute to the input of the digital filters of the channel muted.

REGISTER ADDRESS	BIT	LABEL	DEFAULT	DESCRIPTION
0001001 DAC Mute	5:3	DMUTE [2:0]	000	DAC 3,2 and 1 Soft Mute select
0001111 DAC4 Control	2	DMUTE4	0	DAC 4 Soft Mute select

{DMUTE4,DMUTE [2:0]}	DAC CHANNEL 4	DAC CHANNEL 3	DAC CHANNEL 2	DAC CHANNEL 1
0000	Not MUTE	Not MUTE	Not MUTE	Not MUTE
0001	Not MUTE	Not MUTE	Not MUTE	MUTE
0010	Not MUTE	Not MUTE	MUTE	Not MUTE
0011	Not MUTE	Not MUTE	MUTE	MUTE
0100	Not MUTE	MUTE	Not MUTE	Not MUTE
0101	Not MUTE	MUTE	Not MUTE	MUTE
0110	Not MUTE	MUTE	MUTE	Not MUTE
0111	Not MUTE	MUTE	MUTE	MUTE
1000	MUTE	Not MUTE	Not MUTE	Not MUTE
1001	MUTE	Not MUTE	Not MUTE	MUTE
1010	MUTE	Not MUTE	MUTE	Not MUTE
1011	MUTE	Not MUTE	MUTE	MUTE
1100	MUTE	MUTE	Not MUTE	Not MUTE
1101	MUTE	MUTE	Not MUTE	MUTE
1110	MUTE	MUTE	MUTE	Not MUTE
1111	MUTE	MUTE	MUTE	MUTE

Setting the MUTEALL register bit will apply a 'soft' mute to the input of all the DAC digital filters:

REGISTER ADDRESS	BIT	LABEL	DEFAULT	DESCRIPTION
0000010 DAC Mute	0	MUTEALL	0	Soft Mute select 0 : Normal Operation 1: Soft mute all channels

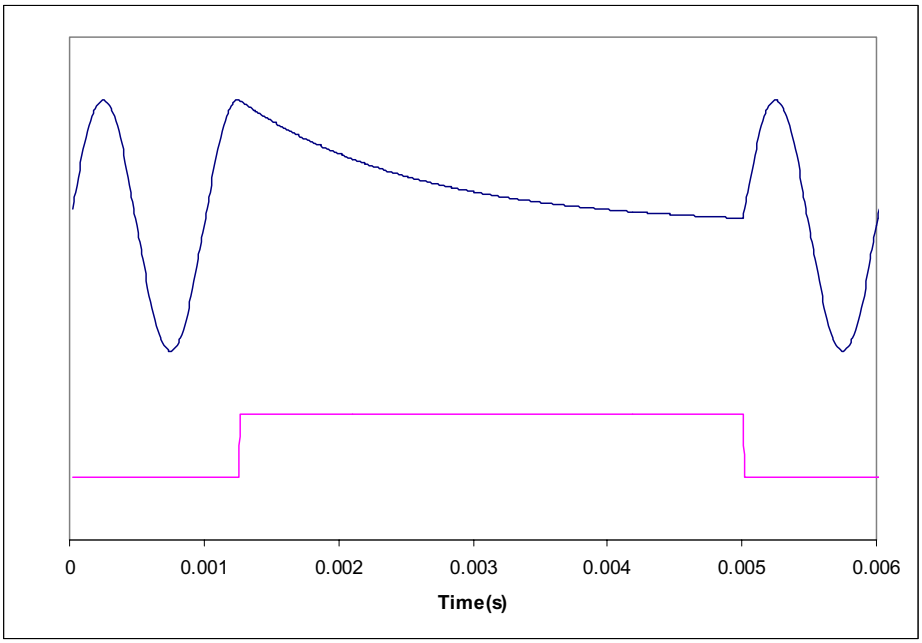


Figure 17 Application and Release of Soft Mute

Figure 17 shows the application and release of MUTE whilst a full amplitude sinusoid is being played at 48kHz sampling rate. When MUTE (lower trace) is asserted, the output (upper trace) begins to decay exponentially from the DC level of the last input sample. The output will decay towards V_{MID} with a time constant of approximately 64 input samples. If MUTE is applied to all channels for 1024 or more input samples the outputs will be connected directly to V_{MID} if IZD is set. When MUTE is de-asserted, the output will restart immediately from the current input sample.

Note that all other means of muting the DAC channels: setting the PL[3:0] bits to 0, setting the PDWN bit or setting attenuation to 0 will cause much more abrupt muting of the output.

DE-EMPHASIS MODE

Each stereo DAC channel has an individual de-emphasis control bit:

REGISTER ADDRESS	BIT	LABEL	DEFAULT	DESCRIPTION
0001001 DAC De-emphahsis Control	[8:6]	DEEMPH [2:0]	000	DAC 3, 2 and 1 De-emphasis channel selection select:
0001111 DAC4 Control	4	DEEMPH4	0	DAC 4 De-emphasis channel selection select:

{DEEMPH4,DEEMPH [2:0]}	DAC CHANNEL 4	DAC CHANNEL 3	DAC CHANNEL 2	DAC CHANNEL 1
0000	Not-DEEMPH	Not-DEEMPH	Not-DEEMPH	Not-DEEMPH
0001	Not-DEEMPH	Not-DEEMPH	Not-DEEMPH	DEEMPH
0010	Not-DEEMPH	Not-DEEMPH	DEEMPH	Not-DEEMPH
0011	Not-DEEMPH	Not-DEEMPH	DEEMPH	DEEMPH
0100	Not-DEEMPH	DEEMPH	Not-DEEMPH	Not-DEEMPH
0101	Not-DEEMPH	DEEMPH	Not-DEEMPH	DEEMPH
0110	Not-DEEMPH	DEEMPH	DEEMPH	Not-DEEMPH
0111	Not-DEEMPH	DEEMPH	DEEMPH	DEEMPH
1000	DEEMPH	Not-DEEMPH	Not-DEEMPH	Not-DEEMPH
1001	DEEMPH	Not-DEEMPH	Not-DEEMPH	DEEMPH
1010	DEEMPH	Not-DEEMPH	DEEMPH	Not-DEEMPH
1011	DEEMPH	Not-DEEMPH	DEEMPH	DEEMPH
1100	DEEMPH	DEEMPH	Not-DEEMPH	Not-DEEMPH
1101	DEEMPH	DEEMPH	Not-DEEMPH	DEEMPH
1110	DEEMPH	DEEMPH	DEEMPH	DEEMPH
1111	DEEMPH	DEEMPH	DEEMPH	DEEMPH

Refer to Figure 22, Figure 23, Figure 24, Figure 25, Figure 26 and Figure 27 for details of the De-Emphasis performance at different sample rates.

REGISTER ADDRESS	BIT	LABEL	DEFAULT	DESCRIPTION
0000010 DAC DEMP	1	DEEMP ALL	0	DEMMP select 0 : Normal Operation 1: De-emphasis all channels

POWERDOWN MODE AND DAC DISABLE

Setting the PDWN register bit immediately powers down the DAC's on the WM8768, overriding the DACD powerdwen bits control bits. All trace of the previous input samples are removed, but all control register settings are preserved. When PDWN is cleared the digital filters will be reinitialised

REGISTER ADDRESS	BIT	LABEL	DEFAULT	DESCRIPTION
0000010 Powerdown Control	2	PDWN	0	Power Down all DAC's Select: 0: All DAC's enabled 1: All DAC's disabled

The DACs may also be powered down individually by setting the DACPD disable bits. Each Stereo DAC channel has a separate disable DACPD[2:0]. Setting DACPD for a channel will disable the DACs and select a low power mode.

REGISTER ADDRESS	BIT	LABEL	DEFAULT	DESCRIPTION
0001010 Powerdown Control	3:1	DACD[2:0]	000	DAC Disable
0001111 DAC4 Control	1	DACD4	0	DAC 4 Powerdown

{DACD4,DACD [2:0]}	DAC CHANNEL 4	DAC CHANNEL 3	DAC CHANNEL 2	DAC CHANNEL 1
0000	Active	Active	Active	Active
0001	Active	Active	Active	DISABLE
0010	Active	Active	DISABLE	Active
0011	Active	Active	DISABLE	DISABLE
0100	Active	DISABLE	Active	Active
0101	Active	DISABLE	Active	DISABLE
0110	Active	DISABLE	DISABLE	Active
0111	Active	DISABLE	DISABLE	DISABLE
1000	DISABLE	Active	Active	Active
1001	DISABLE	Active	Active	DISABLE
1010	DISABLE	Active	DISABLE	Active
1011	DISABLE	Active	DISABLE	DISABLE
1100	DISABLE	DISABLE	Active	Active
1101	DISABLE	DISABLE	Active	DISABLE
1110	DISABLE	DISABLE	DISABLE	Active
1111	DISABLE	DISABLE	DISABLE	DISABLE

MASTER POWERDOWN

Control bit PWRDNALL overrides the {DACD4,DACD[2:0]} bits and powers everything down including the reference VMID. It is recommended that the DAC's are powered down first before setting this bit.

REGISTER ADDRESS	BIT	LABEL	DEFAULT	DESCRIPTION
0001010 Interface Control	4	PWRDNALL	0	Master Power down bit 0: Not powered down 1: Powered down

MASTER MODE SELECT

Control bit MS selects between audio interface Master and Slave Modes. In Master mode LRCLK and BCLK are outputs and are generated by the WM8768. In Slave mode LRCLK and BCLK are inputs to WM8768.

REGISTER ADDRESS	BIT	LABEL	DEFAULT	DESCRIPTION
0001010 Interface Control	5	MS	0	Audio Interface Master/Slave Mode select: 0: Slave Mode 1: Master Mode

MASTER MODE LRCLK FREQUENCY SELECT

In Master mode the WM8768 generates LRC and BCLK. These clocks are derived from the master clock and the ratio of MCLK to LRC is set by RATE.

REGISTER ADDRESS	BIT	LABEL	DEFAULT	DESCRIPTION
0001010 Interface Control	8:6	RATE[2:0]	010	Master Mode MCLK: LRCLK ratio select: 000: 128fs 001: 192fs 010: 256fs 011: 384fs 100: 512fs 101: 768fs

MUTE PIN DECODE

The MUTE pin can either be used an output or an input. When used as an input the MUTE pins action can controlled by setting the DZFM and DZFM4 bit to select the corresponding DAC for applying the MUTE to. As an output its meaning is selected by the DZFM and DZFM4 control bits. By default selecting the MUTE to represent if DAC1 has received more than 1024 midrail samples will cause the MUTE to be asserted a softmute on DAC1. Disabling the decode block will cause any logical high on the MUTE pin to apply a softmute to all DAC's. For compatibility with the WM8772 register the MUTE pin decode bit is also found in the ADC control register, which is redundant on this chip. The OR of these two register bit is taken internally.

REGISTER ADDRESS	BIT	LABEL	DEFAULT	DESCRIPTION
0001100 ADC Control	6	MPD	0	MUTE pin decode disable: 0: MUTE pin decode enable 1: MUTE pin decode disable
0001111 DAC4 control	5	MPD	0	MUTE pin decode disable: 0: MUTE pin decode enable 1: MUTE pin decode disable

DAC DIGITAL VOLUME CONTROL

The DAC volume may also be adjusted in the digital domain using independent digital attenuation control registers

REGISTER ADDRESS	BIT	LABEL	DEFAULT	DESCRIPTION
0000000 Digital Attenuation DACL1	7:0	LDA1[7:0]	11111111 (0dB)	Digital Attenuation data for Left channel DACL1 in 0.5dB steps. See Table 11
	8	UPDATE	Not latched	Controls simultaneous update of all Attenuation Latches 0: Store LDA1 in intermediate latch (no change to output) 1: Store LDA1 and update attenuation on all channels
0000001 Digital Attenuation DACR1	7:0	RDA1[6:0]	11111111 (0dB)	Digital Attenuation data for Right channel DACR1 in 0.5dB steps. See Table 11
	8	UPDATE	Not latched	Controls simultaneous update of all Attenuation Latches 0: Store RDA1 in intermediate latch (no change to output) 1: Store RDA1 and update attenuation on all channels.
0000100 Digital Attenuation DACL2	7:0	LDA2[7:0]	11111111 (0dB)	Digital Attenuation data for Left channel DACL2 in 0.5dB steps. See Table 11
	8	UPDATE	Not latched	Controls simultaneous update of all Attenuation Latches 0: Store LDA2 in intermediate latch (no change to output) 1: Store LDA2 and update attenuation on all channels.
0000101 Digital Attenuation DACR2	7:0	RDA2[7:0]	11111111 (0dB)	Digital Attenuation data for Right channel DACR2 in 0.5dB steps. See Table 11
	8	UPDATE	Not latched	Controls simultaneous update of all Attenuation Latches 0: Store RDA2 in intermediate latch (no change to output) 1: Store RDA2 and update attenuation on all channels.
0000110 Digital Attenuation DACL3	7:0	LDA3[7:0]	11111111 (0dB)	Digital Attenuation data for Left channel DACL3 in 0.5dB steps. See Table 11
	8	UPDATE	Not latched	Controls simultaneous update of all Attenuation Latches 0: Store LDA3 in intermediate latch (no change to output) 1: Store LDA3 and update attenuation on all channels.
0000111 Digital Attenuation DACR3	7:0	RDA3[7:0]	11111111 (0dB)	Digital Attenuation data for Right channel DACR3 in 0.5dB steps. See Table 11
	8	UPDATE	Not latched	Controls simultaneous update of all Attenuation Latches 0: Store RDA3 in intermediate latch (no change to output) 1: Store RDA3 and update attenuation on all channels.
0001101 Digital Attenuation DACL4	7:0	LDA3[7:0]	11111111 (0dB)	Digital Attenuation data for Left channel DACL4 in 0.5dB steps. See Table 11
	8	UPDATE	Not latched	Controls simultaneous update of all Attenuation Latches 0: Store LDA4 in intermediate latch (no change to output) 1: Store LDA4 and update attenuation on all channels.
0001110 Digital Attenuation DACR4	7:0	RDA3[7:0]	11111111 (0dB)	Digital Attenuation data for Right channel DACR4 in 0.5dB steps. See Table 11
	8	UPDATE	Not latched	Controls simultaneous update of all Attenuation Latches 0: Store RDA3 in intermediate latch (no change to output) 1: Store RDA3 and update attenuation on all channels.
0001000 Master Digital Attenuation (all channels)	7:0	MASTDA [7:0]	11111111 (0dB)	Digital Attenuation data for all DAC channels in 0.5dB steps. See Table 11
	8	UPDATE	Not latched	Controls simultaneous update of all Attenuation Latches 0: Store gain in intermediate latch (no change to output) 1: Store gain and update attenuation on all channels.

L/RDAX[7:0]	ATTENUATION LEVEL
00(hex)	-∞ dB (mute)
01(hex)	-127.5dB
:	:
:	:
:	:
FE(hex)	-0.5dB
FF(hex)	0dB

Table 11 Digital Volume Control Attenuation Levels

SOFTWARE REGISTER RESET

Writing to register 11111 will cause a register reset, resetting all register bits to their default values. This reset will last either 2*MCLK periods or until another write is made to the serial interface.

REGISTER MAP

The complete register map is shown below. The detailed description can be found in the relevant text of the device description. The WM8768 can be configured using the Control Interface. All unused bits should be set to '0'.

REGISTER	B15	B14	B13	B12	B11	B10	B9	B8	B7	B6	B5	B4	B3	B2	B1	B0	DEFAULT
R0(00h)	0	0	0	0	0	0	0	UPDATE	LDA1[7:0]								X11111111
R1(01h)	0	0	0	0	0	0	1	UPDATE	RDA1[7:0]								X11111111
R2(02h)	0	0	0	0	0	1	0	PL[8:5]				IZD	ATC	PDWN All DAC	DEEMP ALL DAC	MUTE All DAC	100100000
R3(03h)	0	0	0	0	0	1	1	PHASE[8:6]			DACIWL[5:4]		DACBCP	DACLRP	DACFMT[1:0]		000000000
R4(04h)	0	0	0	0	1	0	0	UPDATE	LDA2[7:0]								X11111111
R5(05h)	0	0	0	0	1	0	1	UPDATE	RDA2[7:0]								X11111111
R6(06h)	0	0	0	0	1	1	0	UPDATE	LDA3[7:0]								X11111111
R7(07h)	0	0	0	0	1	1	1	UPDATE	RDA3[7:0]								X11111111
R8(08h)	0	0	0	1	0	0	0	UPDATE	MASTDA[7:0]								X11111111
R9(09h)	0	0	0	1	0	0	1	DEEMP[8:6]			DMUTE[5:3]			DZFM[2:1]		ZCD	000000000
R10(0Ah)	0	0	0	1	0	1	0	DACRATE[8:6]			DACMS	PWRDN ALL	DACD[3:1]			0	010000000
R12(0Ch)	0	0	0	1	1	0	0	0	0	MPD	0	0	0	0	0	0	000000000
R13(0Dh)	0	0	0	1	1	0	1	UPDATE	LDA4[7:0]								X11111111
R14(0Eh)	0	0	0	1	1	1	0	UPDATE	RDA4[7:0]								X11111111
R15(0Fh)	0	0	0	1	1	1	1	0	0	0	MPD	DEEMP 4	PHASE4	DMUTE 4	DZFM4	DACD4	000000000
R31(1Fh)	0	0	1	1	1	1	1	RESET									000000000

DIGITAL FILTER CHARACTERISTICS

PARAMETER	TEST CONDITIONS	MIN	TYP	MAX	UNIT
DAC Filter					
Passband	±0.05 dB			0.444fs	
	-3dB		0.487fs		
Passband ripple				±0.05	dB
Stopband		0.555fs			
Stopband Attenuation	f > 0.555fs	-60			dB
Group Delay			21		fs

Table 12 Digital Filter Characteristics

DAC FILTER RESPONSES

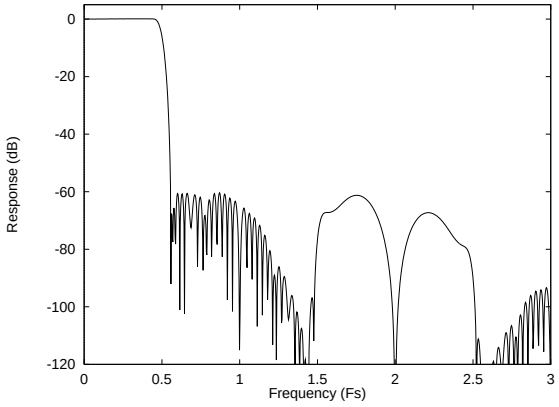


Figure 18 DAC Digital Filter Frequency Response – 44.1, 48 and 96kHz

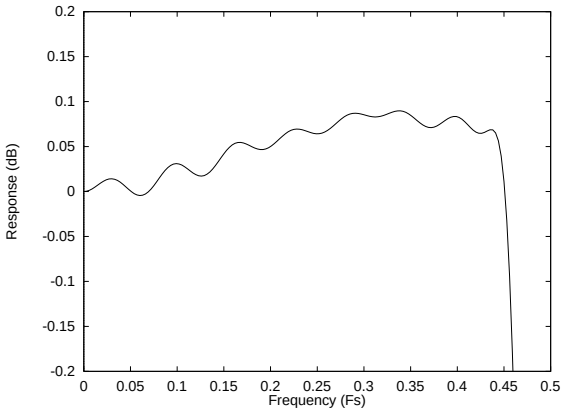


Figure 19 DAC Digital Filter Ripple –44.1, 48 and 96kHz

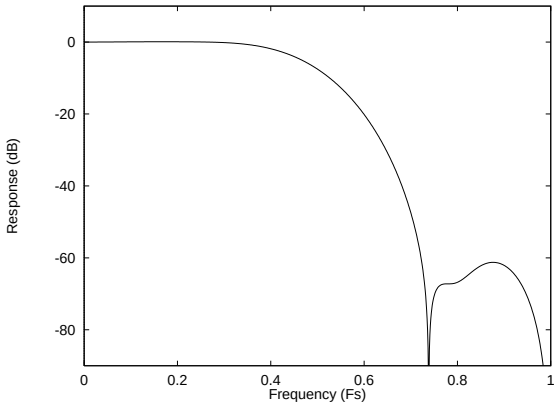


Figure 20 DAC Digital Filter Frequency Response – 192kHz

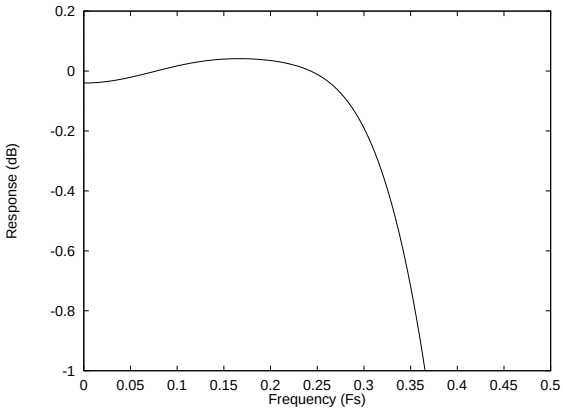


Figure 21 DAC Digital filter Ripple – 192kHz

DIGITAL DE-EMPHASIS CHARACTERISTICS

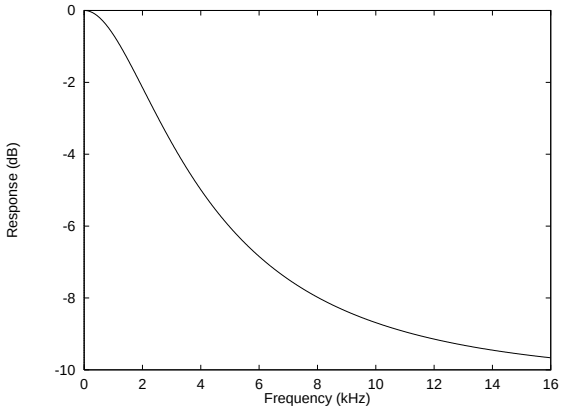


Figure 22 De-Emphasis Frequency Response (32kHz)

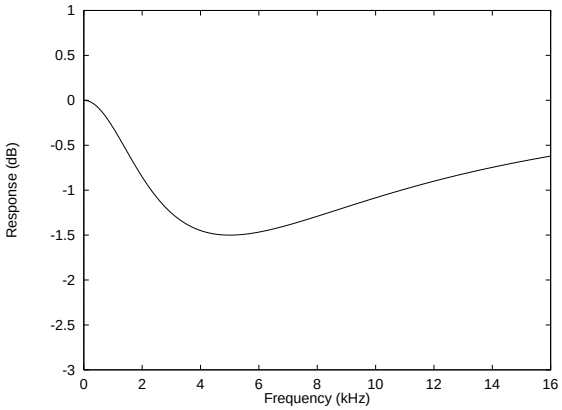


Figure 23 De-Emphasis Error (32kHz)

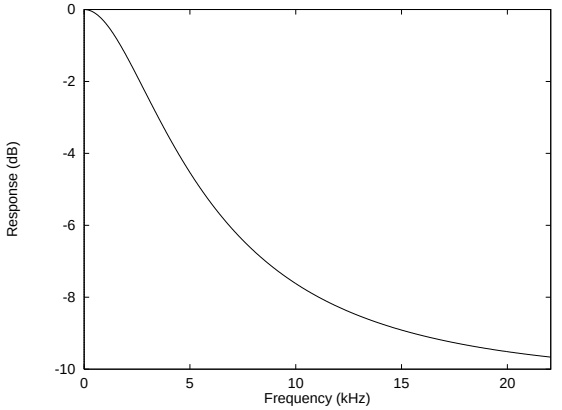


Figure 24 De-Emphasis Frequency Response (44.1kHz)

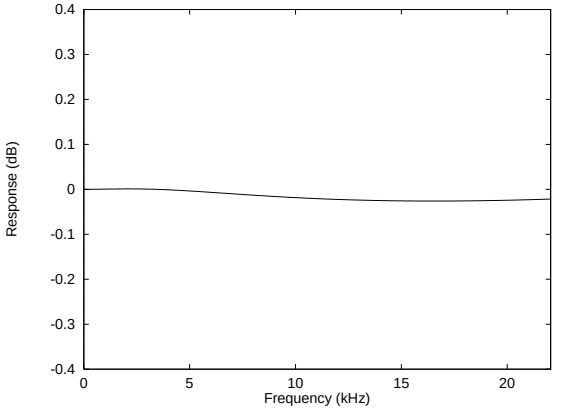


Figure 25 De-Emphasis Error (44.1kHz)

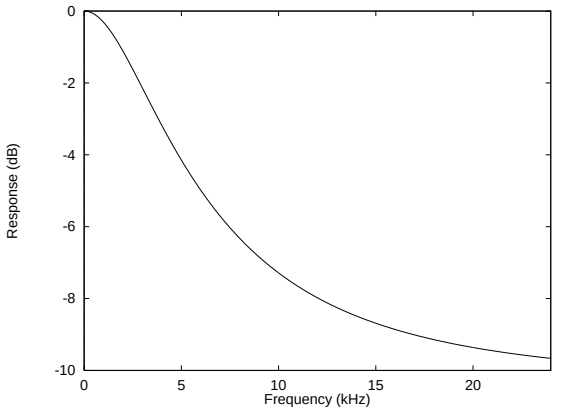


Figure 26 De-Emphasis Frequency Response (48kHz)

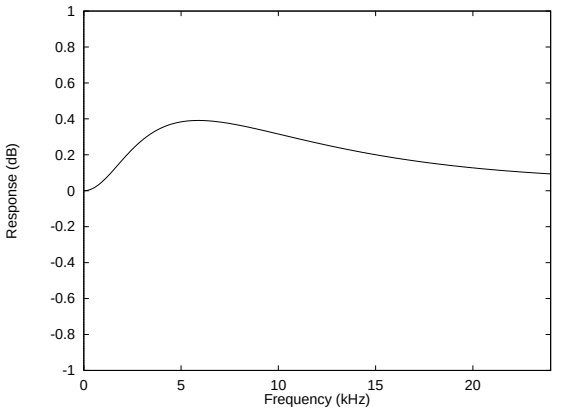


Figure 27 De-Emphasis Error (48kHz)

APPLICATIONS INFORMATION

RECOMMENDED EXTERNAL COMPONENTS

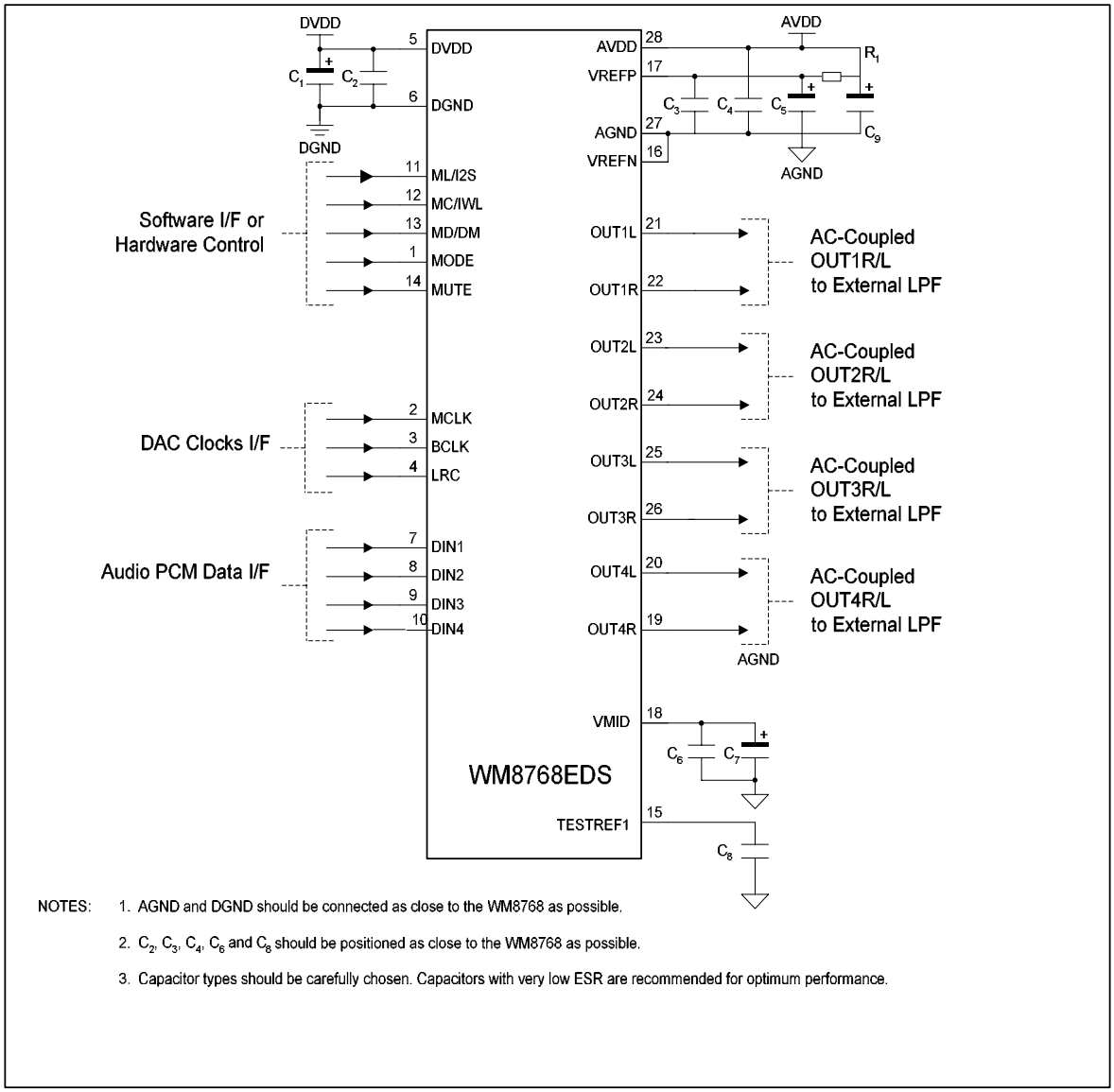


Figure 28 Recommended External Component Diagram

RECOMMENDED EXTERNAL COMPONENTS VALUES

COMPONENT REFERENCE	SUGGESTED VALUE	DESCRIPTION
C1 and C5	10µF	De-coupling for DVDD and AVDD.
C2 to C4	0.1µF	De-coupling for DVDD and AVDD.
C6	0.1µF	Reference de-coupling capacitors for VMID.
C7	10µF	
C8	0.1µF	De-coupling for TESTREF1
C9	10µF	Filtering for VREFP. Omit if AVDD low noise.
R1	33VΩ	Filtering for VREP. Use 0Ω if AVDD low noise.

Table 13 External Components Description

SUGGESTED ANALOGUE LOW PASS POST DAC FILTERS

It is recommended that a lowpass filter be applied to the output from each DAC channel for Hi Fi applications. Typically a second order filter is suitable and provides sufficient attenuation of high frequency components (the unique low order, high bit count multi-bit sigma delta DAC structure used in WM8768 produces much less high frequency output noise than normal sigma delta DACs. This filter is typically also used to provide the 2x gain needed to provide the standard 2Vrms output level from most consumer equipment. Figure 29 shows a suitable post DAC filter circuit, with 2x gain. Alternative inverting filter architectures might also be used with as good results.

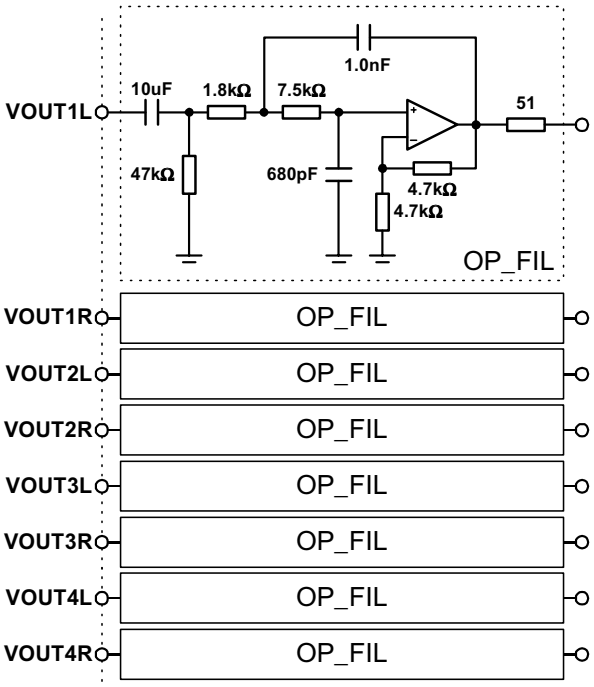
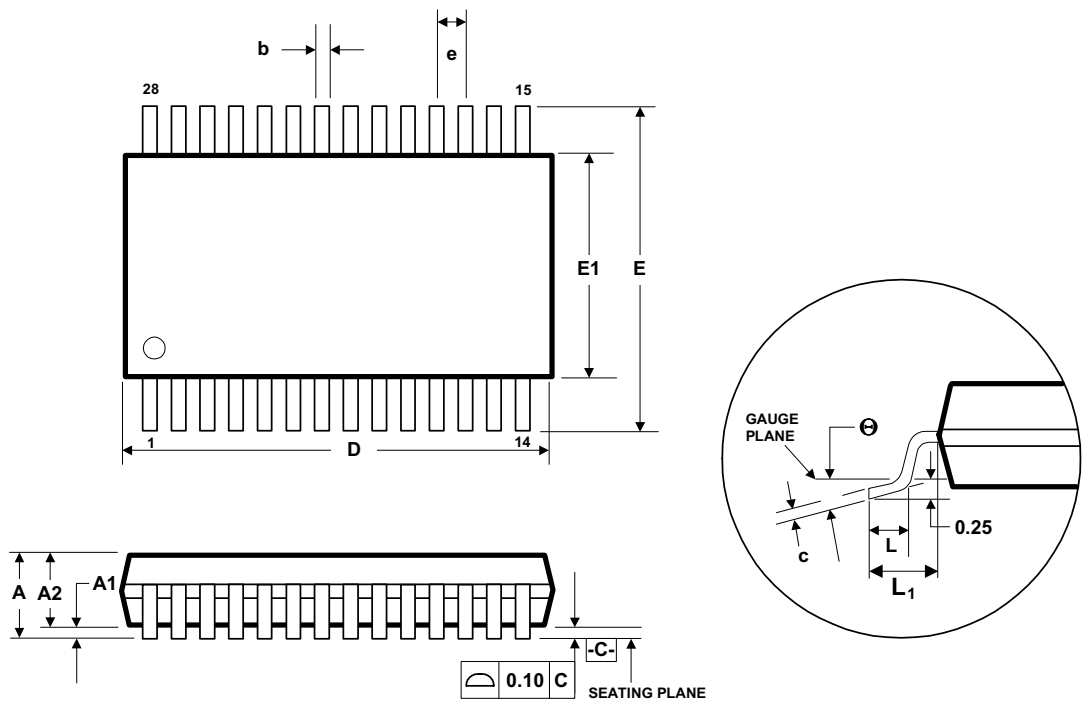


Figure 29 Recommended Post DAC Filter Circuit

PACKAGE DRAWING

DS: 28 PIN SSOP (10.2 x 5.3 x 1.75 mm) **DM007.D**



Symbols	Dimensions (mm)		
	MIN	NOM	MAX
A	-----	-----	2.0
A_1	0.05	-----	0.25
A_2	1.65	1.75	1.85
b	0.22	0.30	0.38
c	0.09	-----	0.25
D	9.90	10.20	10.50
e		0.65 BSC	
E	7.40	7.80	8.20
E_1	5.00	5.30	5.60
L	0.55	0.75	0.95
L_1		0.125 REF	
θ	0^o	4^o	8^o
REF:	JEDEC.95, MO-150		

NOTES:
A. ALL LINEAR DIMENSIONS ARE IN MILLIMETERS.
B. THIS DRAWING IS SUBJECT TO CHANGE WITHOUT NOTICE.
C. BODY DIMENSIONS DO NOT INCLUDE MOLD FLASH OR PROTRUSION, NOT TO EXCEED 0.20MM.
D. MEETS JEDEC.95 MO-150, VARIATION = AH. REFER TO THIS SPECIFICATION FOR FURTHER DETAILS.

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