

WM8782A

24-Bit, 192kHz Stereo ADC

DESCRIPTION

The WM8782A is a high performance, low cost stereo audio ADC designed for recordable media applications.

The device offers stereo line level inputs along with two control input pins (FORMAT, IWL) to allow operation of the audio interface in three industry standard modes. An internal op-amp is integrated on the front end of the chip to accommodate analogue input signals greater than $1V_{rms}$. The device also has a high pass filter to remove residual DC offsets.

WM8782A offers a Slave mode clocking scheme. A stereo 24-bit multi-bit sigma-delta ADC is used with 128x, 64x or 32x over-sampling, according to sample rate. Digital audio output word lengths from 16-24 bits and sampling rates from 8kHz to 192kHz are supported.

The device is a hardware controlled device and is supplied in a 20-lead SSOP.

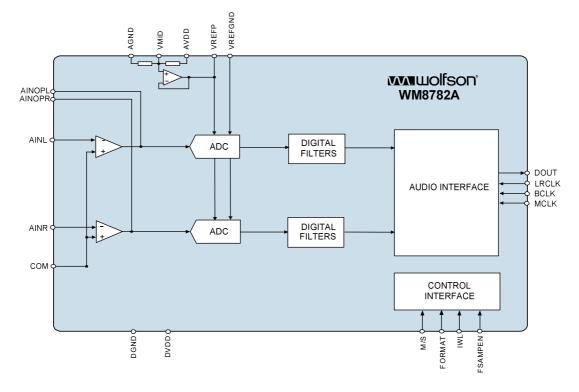
The device is available over a functional temperature range of -40°C to +85°C

FEATURES

- SNR 100dB ('A' weighted @ 48kHz)
- THD -93dB (at –1dB)
- Sampling Frequency: 8 192kHz
- Slave Clocking Mode
- System Clock (MCLK): 128fs, 192fs, 256fs, 384fs, 512fs, 768fs
- Audio Data Interface Modes
- 16-24 bit I²S, 16-24 bit Left, 16-24 bit Right Justified
- Supply Voltages
 - Analogue: 2.7 to 5.5V
 - Digital core: 2.7V to 3.6V
- 20-lead SSOP package
- Accelerated Lifetime Screened Devices available.

APPLICATIONS

- Recordable DVD Players
- Personal Video Recorders
- STB
- Studio Audio Processing Equipment
- Automotive



WOLFSON MICROELECTRONICS plc

Production Data, April 2010, Rev 4.8

To receive regular email updates, sign up at http://www.wolfsonmicro.com/enews

BLOCK DIAGRAM

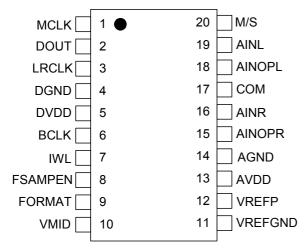
TABLE OF CONTENTS

DESCRIPTION	1
FEATURES	1
APPLICATIONS	1
BLOCK DIAGRAM	1
TABLE OF CONTENTS	2
PIN CONFIGURATION	3
ORDERING INFORMATION	3
PIN DESCRIPTION	4
ABSOLUTE MAXIMUM RATINGS	5
THERMAL PERFORMANCE	
RECOMMENDED OPERATING CONDITIONS	6
ELECTRICAL CHARACTERISTICS	6
TERMINOLOGY	
SIGNAL TIMING REQUIREMENTS	8
SYSTEM CLOCK TIMING	
AUDIO INTERFACE TIMING – SLAVE MODE	
DEVICE DESCRIPTION	-
INTRODUCTION	
ADC DIGITAL FILTER DIGITAL AUDIO INTERFACE	. 10
POWER ON RESET	
DIGITAL FILTER CHARACTERISTICS	
ADC FILTER RESPONSES	-
ADC HIGH PASS FILTER	
APPLICATIONS INFORMATION	
RECOMMENDED EXTERNAL COMPONENTS	
RECOMMENDED EXTERNAL COMPONENTS VALUES	
PACKAGE DIMENSIONS	
IMPORTANT NOTICE	
ADDRESS:	. 20



Production Data

PIN CONFIGURATION



ORDERING INFORMATION

DEVICE	TEMPERATURE RANGE	PACKAGE	MOISTURE SENSITIVITY LEVEL	PEAK SOLDERING TEMPERATURE
WM8782ASEDS/V	-40°C to +85°C	20-lead SSOP (Pb-free)	MSL2	260°C
WM8782ASEDS/RV	-40°C to +85°C	20-lead SSOP (Pb-free, tape and reel)	MSL2	260°C

Note:

Reel quantity = 2,000



WM8782A

PIN DESCRIPTION

	NAME	TYPE	DESCRIPTION			
PIN NO.						
1	MCLK	Digital Input	Master Clock			
2	DOUT	Digital Output	ADC Digital Audio Data			
3	LRCLK	Digital Input	Audio Interface Left / Right Clock			
4	DGND	Supply	Digital Negative Supply			
5	DVDD	Supply	Digital Positive Supply			
6	BCLK	Digital Input	Audio Interface Bit Clock			
7	IWL	Digital Tristate Input	Word Length			
			0 = 16 bit			
			1 = 20 bit			
			Z = 24 bit			
8	FSAMPEN	Digital Tristate Input	Fast Sampling Rate Enable			
			0 = 48kHz			
			1= 96kkHz			
			Z= 192kHz			
9	FORMAT	Digital Tristate Input	Audio Mode Select			
			0 = RJ			
			1 = LJ			
			Z = 12S			
10	VMID	Analogue Output	Mid rail Voltage Decoupling Capacitor			
11	VREFGND	Supply	Negative Supply and Substrate Connection			
12	VREFP	Analogue Output	Positive Reference Voltage Decoupling Pin; 10uF external decoupling			
13	AVDD	Supply	Analogue Positive Supply			
14	AGND	Supply	Analogue Negative Supply and Substrate Connection			
15	AINOPR	Analogue Output	Right Channel Internal Op-Amp Output			
16	AINR	Analogue Input	Right Channel Input			
17	COM	Analogue Input	Common mode high impedance input should be set to midrail.			
18	AINOPL	Analogue Output	Left Channel Internal Op-Amp Output			
19	AINL	Analogue Input	Left Channel Input			
20	M/S	Digital Input	Interface Mode Select			
			0 = Slave mode (128fs, 192fs, 256fs, 384fs, 512fs, 768fs)			
			1 = Reserved			
			(fs=word clock frequency)			



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:

$$\label{eq:MSL1} \begin{split} \mathsf{MSL1} &= \mathsf{unlimited} \ \text{floor} \ \text{life} \ at <\!\!30^\circ\text{C} \ / \ 85\% \ \text{Relative} \ \text{Humidity}. \ \text{Not normally stored in moisture barrier bag}. \\ \mathsf{MSL2} &= \mathsf{out} \ \text{of} \ \text{bag storage for 1 year} \ at <\!\!30^\circ\text{C} \ / \ 60\% \ \text{Relative} \ \text{Humidity}. \ \text{Supplied in moisture barrier bag}. \\ \mathsf{MSL3} &= \mathsf{out} \ \text{of} \ \text{bag storage for 168 hours} \ at <\!\!30^\circ\text{C} \ / \ 60\% \ \text{Relative} \ \text{Humidity}. \ \text{Supplied in moisture barrier bag}. \\ \end{split}$$

The Moisture Sensitivity Level is specified in Ordering Information.

CONDITION	MIN	MAX
Digital supply voltage	-0.3V	+4.5V
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
Ambient temperature (supplies applied)	-55°C	+125°C
Storage temperature	-65°C	+150°C
Pb free package body temperature (reflow 10 seconds)		+260°C
Package body temperature (soldering 2 minutes)		+183°C

Notes:

1. Analogue and digital grounds must always be within 0.3V of each other.

THERMAL PERFORMANCE

PARAMETER	SYMBOL	TEST CONDITIONS	MIN	TYP	MAX	UNIT
Thermal resistance – junction to	$R_{ extsf{ heta}JA}$			81		°C/W
ambient				See note 1		

Notes:

1. Figure given for package mounted on 4-layer FR4 according to JESD51-7. (No forced air flow is assumed).

2. Thermal performance figures are estimated.



RECOMMENDED OPERATING CONDITIONS

PARAMETER	SYMBOL	TEST CONDITIONS	MIN	TYP	MAX	UNIT
Digital supply range	DVDD	WM8782SEDS,	2.7		3.6	V
		WM8782SEDS/R				
Analogue supply range	AVDD	WM8782SEDS,	2.7		5.5	V
		WM8782SEDS/R				
Ground	DGND,AGND			0		V
Operating temperature range	T _A	WM8782SEDS,	-40		+85	°C
		WM8782SEDS/R				

Notes:

1. Digital supply DVDD must never be more than 0.3V greater than AVDD.

ELECTRICAL CHARACTERISTICS

Test Conditions

DVDD = 3.3V, AVDD = 5.0V, T_A = $+25^{\circ}$ C, 1kHz signal, A-weighted, fs = 48kHz, MCLK = 256fs, 24-bit audio data, Slave Mode unless otherwise stated.

PARAMETER	SYMBOL	TEST CONDITIONS	MIN	ТҮР	MAX	UNIT
ADC Performance – WM8782SE	DS, WM8782SE	EDS/R (+25°C)				
Full Scale Input Signal Level (for ADC 0dB Input)				1.0		V _{rms}
Input resistance, using recommended external resistor network on p22.				10		kΩ
Input capacitance				20		pF
Signal to Noise Ratio (see Terminology note 1,2,4)	SNR	weighted, @ fs = 48kHz	93	100		dB
		Unweighted, @ fs = 48kHz		98		dB
		weighted, @ fs = 48kHz, AVDD = 3.3V		98		dB
Signal to Noise Ratio (see Terminology note 1,2,4)	SNR	weighted, @ fs = 96kHz		98		dB
		Unweighted, @ fs = 96kHz		98		dB
		weighted, @ fs = 96kHz AVDD = 3.3V		98		dB
Total Harmonic Distortion	THD	1kHz, -1dB Full Scale @ fs = 48kHz		-93		dB
		1kHz, -1dB Full Scale @ fs = 96kHz		-93		dB
		1kHz, -1dB Full Scale @ fs = 192kHz		-92		dB
Dynamic Range	DNR	-60dBFS	93	100		dB
Channel Separation (see Terminology note 4)		1kHz Input		90		dB
Channel Level Matching		1kHz signal		0.1		dB
Channel Phase Deviation		1kHz signal		0.0001		Degree
Power Supply Rejection Ratio	PSRR	1kHz 100mVpp, applied to AVDD, DVDD		50		dB



Production Data

WM8782A

Test Conditions

DVDD = 3.3V, AVDD = 5.0V, $T_A = +25^{\circ}C$, 1kHz signal, A-weighted, fs = 48kHz, MCLK = 256fs, 24-bit audio data, Slave Mode unless otherwise stated.

PARAMETER	SYMBOL	TEST CONDITIONS	MIN	TYP	MAX	UNIT
Digital Logic Levels (TTL Levels)						
Input LOW level	V _{IL}				0.8	V
Input HIGH level	V _{IH}		2.0			V
Input leakage current – digital pad			-1	±0.2	+1	μA
Input leakage current – digital tristate input (Note 3)				85		μA
Input capacitance				5		pF
Output LOW	Vol	I _{OL} =1mA			0.1 x DVDD	V
Output HIGH	V _{OH}	I _{OH} = -1mA	0.9 x DVDD			V
Analogue Reference Levels						
Midrail Reference Voltage	VMID	AVDD to VMID and VMID to VREFN	-4%	AVDD/2	+4%	V
Potential Divider Resistance	RVMID			70		kΩ
Buffered Reference Voltage	VREFP		-4%	AVDD/2	+4%	V
VREF source current	IVREF				5	mA
VREF sink current	IVREF				5	mA
Supply Current						
Analogue supply current		AVDD = 5V		26		mA
Digital supply current		DVDD = 3.3V		5		mA
Power Down				0.5		mA

Notes:

- All performance measurements are done with a 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 this is not audible, it may affect
 dynamic specification values.
- 2. VMID is decoupled with 10uF and 0.1uF capacitors close to the device package. Smaller capacitors may reduce performance.
- 3. This high leakage current is due to the topology of the instate pads. The pad input is connected to the midpoint of an internal resistor string to pull input to vmid if undriven.

TERMINOLOGY

- Signal-to-noise ratio (dB) Ratio of output level with 1kHz full scale input, to the output level with all zeros into the digital input, over a 20Hz to 20kHz bandwidth. (No Auto-zero or Automute function is employed in achieving these results).
- Dynamic range (dB) DR 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. 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.



SIGNAL TIMING REQUIREMENTS

SYSTEM CLOCK TIMING

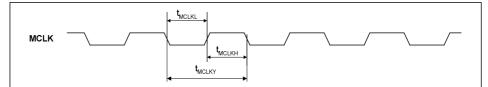


Figure 1 System Clock Timing Requirements

Test Conditions

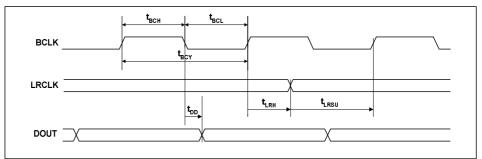
DVDD = 3.3V, DGND = 0V, T_A = +25°C, fs = 48kHz, Slave Mode, MCLK = 256fs, 24-bit data, unless otherwise stated.

PARAMETER	SYMBOL	MIN	TYP	MAX	UNIT
System Clock Timing Information					
MCLK System clock pulse width high	T _{MCLKL}	11			ns
MCLK System clock pulse width low	T _{MCLKH}	11			ns
MCLK System clock cycle time	T _{MCLKY}	28			ns
MCLK duty cycle	T _{MCLKDS}	40:60		60:40	

Table 1 Master Clock Timing Requirements



AUDIO INTERFACE TIMING - SLAVE MODE





Test Conditions

DVDD = 3.3V, DGND = 0V, T_A = +25°C, Slave Mode, fs = 48kHz, MCLK = 256fs, 24-bit data, unless otherwise stated.

PARAMETER	SYMBOL	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
DOUT propagation delay from BCLK falling edge	t _{DD}	0		10	ns

Table 2 Digital Audio Data Timing – Slave Mode

Note:

LRCLK should be synchronous with MCLK, although the WM8782A interface is tolerant of phase variations or jitter on these signals.



DEVICE DESCRIPTION

INTRODUCTION

The WM8782A is a stereo 24-bit ADC designed for demanding recording applications such as DVD recorders, studio mixers, PVRs, and AV amplifiers. The WM8782A consists of stereo line level inputs, followed by a sigma-delta modulator and digital filtering.

The device offers stereo line level inputs along with two control input pins (FORMAT, IWL) to allow operation of the audio interface in three industry standard modes (left justified, right justified or I^2S). An internal op-amp is integrated on the front end of the chip to accommodate analogue input signals greater than $1V_{rms}$. The device also has a high pass filter to remove residual DC offsets.

The WM8782A offers Slave mode clocking schemes. The WM8782A supports master clock rates from 128fs to 768fs and digital audio output word lengths from 16-24 bits. Sampling rates from 8kHz to 192kHz are supported, delivering high SNR operating with 128x, 64x or 32x over-sampling, according to the sample rate.

The line inputs are biased internally through the operational amplifier to V_{MID}.

ADC

The WM8782A uses a multi-bit over sampled sigma-delta ADC. A single channel of the ADC is illustrated in Figure 3 Multi-Bit Oversampling Sigma Delta ADC Schematic.

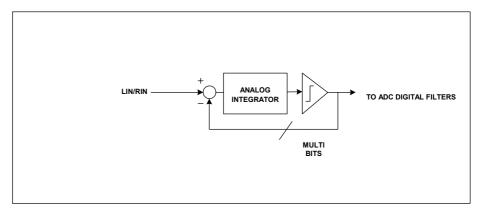


Figure 3 Multi-Bit Oversampling Sigma Delta ADC Schematic

The use of multi-bit feedback and high oversampling rates reduces the effects of jitter and high frequency noise.

The ADC Full Scale input is 1.0V rms at AVDD = 5.0 volts. Any input voltage greater than full scale will possibly overload the ADC and cause distortion. Note that the full scale input has a linear relationship with AVDD. The internal op-amp and appropriate resistors can be used to reduce signals greater than 1Vrms before they reach the ADC.

The ADC filters perform true 24 bit signal processing to convert the raw multi-bit oversampled data from the ADC to the correct sampling frequency to be output on the digital audio interface.

ADC OUTPUT PHASE

In the input to output data-path, the digital output data DOUT, is a phase inverted representation of the analogue input signal.

ADC DIGITAL FILTER

The ADC digital filters contain a digital high pass filter. The high-pass filter response detailed in Digital Filter Characteristics. The operation of the high pass filter removes residual DC offsets that are present on the audio signal.



The digital audio interface uses three pins:

- DOUT: ADC data output
- LRCLK: ADC data alignment clock
- BCLK: Bit clock, for synchronisation

The digital audio interface takes the data from the internal ADC digital filters and places it on DOUT and LRCLK. DOUT is the formatted digital audio data stream output from the ADC digital filters with left and right channels multiplexed together. LRCLK is an alignment clock that controls whether Left or Right channel data is present on the DOUT line. DOUT and LRCLK are synchronous with the BCLK signal with each data bit transition signified by a BCLK high to low transition. DOUT is always an output. BCLK and LRCLK are always inputs.

Three different audio data formats are supported:

- Left justified
- Right justified
- l²S

SLAVE MODE OPERATION

The WM8782A can be configured as a slave mode device. In slave mode, the WM8782A responds with data to clocks it receives over the digital audio interface. The mode is selected by setting the MS input pin (see Table 3 Slave selection below). Slave mode is illustrated below.

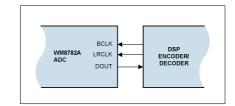


Figure 4 Slave Mode

	PIN	DESCRIPTION
M/S		Slave Selection
		0 = Slave Mode
		1= Reserved

Table 3 Slave selection

AUDIO INTERFACE CONTROL

The Input Word Length and Audio Format mode can be selected by using IWL and FORMAT pins.

PIN	DESCRIPTION
IWL	Word Length
	0 = 16 bit
	1 = 20 bit
	Z = 24 bit
FORMAT	Audio Mode Select
	0 = RJ
	1 = LJ
	Z = 12S

Table 4 Audio Data Format Control



AUDIO DATA FORMATS

In Left Justified mode, the MSB is available on the first rising edge of BCLK following an LRCLK transition. The other bits up to the LSB are then transmitted in order. Depending on word length, BCLK frequency and sample rate, there may be unused BCLK cycles before each LRCLK transition.

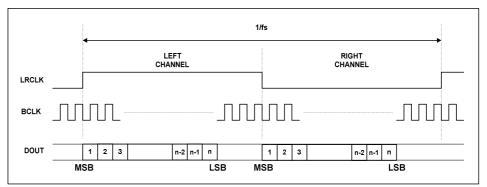


Figure 5 Left Justified Audio Interface (assuming n-bit word length)

In Right Justified mode, the LSB is available on the last rising edge of BCLK before an LRCLK transition. All other bits are transmitted before (MSB first). Depending on word length, BCLK frequency and sample rate, there may be unused BCLK cycles after each LRCLK transition.

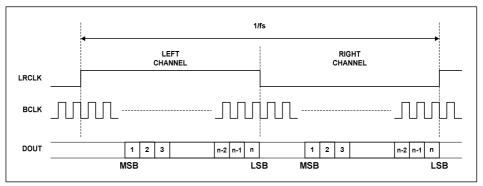


Figure 6 Right Justified Audio Interface (assuming n-bit word length)

In I²S mode, the MSB is available on the second rising edge of BCLK following an LRCLK transition. The other bits up to the LSB are then transmitted in order. Depending on word length, BCLK frequency and sample rate, there may be unused BCLK cycles between the LSB of one sample and the MSB of the next.

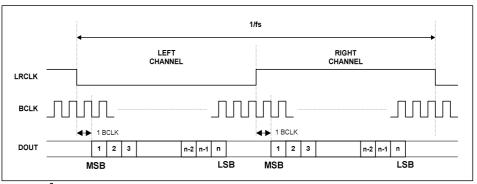


Figure 7 I²S Audio Interface (assuming n-bit word length)



MASTER CLOCK AND AUDIO SAMPLE 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 (MCLK). The external master system clock can be applied directly through the MCLK input pin. In a system where there are a number of possible sources for the reference clock it is recommended that the clock source with the lowest jitter be used to optimise the performance of the ADC.

The master clock is used to operate the digital filters and the noise shaping circuits. The WM8782A supports master clocks of 128fs, 192fs, 256fs, 384fs, 512fs and 768fs, where fs is the audio sampling frequency (LRCLK). In Slave Mode, the WM8782A automatically detects the audio sample rate.

SAMPLING RATE	Master Clock Frequency (MHz)					
(LRCLK)	128fs	192fs	256fs	384fs	512fs	768fs
8kHz	1.024	1.536	2.048	3.072	4.096	6.144
16kHz	2.048	3.072	4.096	6.144	8.192	12.288
32kHz	4.096	6.144	8.192	12.288	16.384	24.576
44.1kHz	5.6448	8.467	11.2896	16.9340	22.5792	33.8688
48kHz	6.144	9.216	12.288	18.432	24.576	36.864
96kHz	12.288	18.432	24.576	36.864	-	-
192kHz	24.576	36.864	-	-	-	-

Table 5 shows the common MCLK frequencies for different sample rates.

Table 5 Master Clock Frequency Selection

In Slave mode, the WM8782A has a master detection circuit that automatically determines the relationship between the master clock frequency and the sampling rate (to within +/- 32 system clocks). If there is a greater than 32 clocks error the interface sets itself to the highest rate available (768fs). There must be a fixed number of MCLKS per LRCLK, although the WM8782A is tolerant of phase variations or jitter on these clocks.

FSAMPEN

The FSAMPEN pin controls the over sampling rate of the ADC. The WM8782A can operate at sample rates from 8kHz to 192kHz. The WM8782A uses a sigma-delta modulator that operates at an optimal frequency of 6.144MHz.

By default the WM8782A generates the ADC frequency at 128xOSR. At fs=48kHz, the ADC frequency is 128xOSR = 128x48kHz = 6.144MHz.

If fs=96KHz, the FSAMPEN pin must be set to 1. In this case, the ADC frequency is 64xOSR = 64x96kHz = 6.144MHz.

If fs=192KHz, the FSAMPEN pin must be set to z. In this case, the ADC frequency is 32xOSR = 32x192kHz = 6.144MHz.

PIN	DESCRIPTION
FSAMPEN	Fast sampling rate enable
	0 = 48ken (128x OSR)
	1= 96ken (64x OSR)
	z= 192ken (32x OSR)

Table 6 Sampling Rate Enable Selection



POWER DOWN CONTROL

The WM8782A can be powered down by stopping MCLK. Power down mode using MCLK is entered after 65536/fs clocks. On power-up, the WM8782A applies the power-on reset sequence described below.

When MCLK is stopped DOUT is forced to zero.

POWER ON RESET

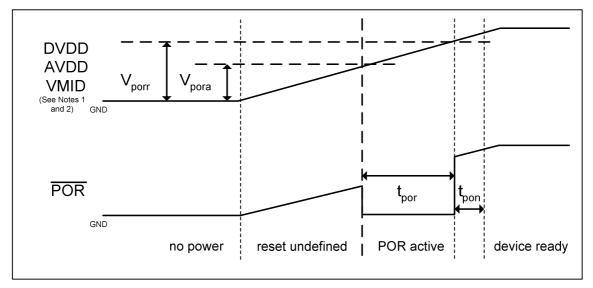


Figure 8 Power Supply Timing Requirements – Power-on

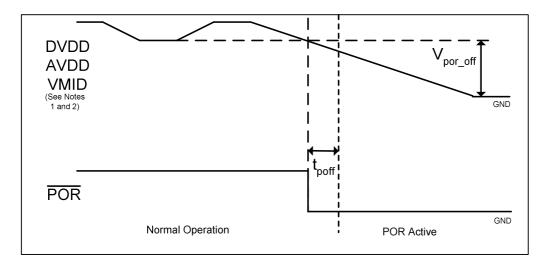


Figure 9 Power Supply Timing Requirements – Power-down



Production Data

Test Conditions

AVDD = 5V, DVDD = 3.3V, AGND = DGND = 0V, T_A = +25°C

PARAMETER	SYMBOL	TEST CONDITIONS	MIN	ТҮР	MAX	UNIT
Power Supply Input Timing Information						
DVDD level to activate POR – power on	V _{pora}	Measured from DGND		0.7		V
AVDD level to activate POR – power on	V _{pora}	Measured from AGND		0.7		V
VMID level to activate POR – power on	V _{pora}	Measured from AGND		0.7		V
DVDD level to release POR – power on (see notes 1 and 2)	V _{porr}	Measured from DGND		DVDD Min		V
AVDD level to release POR – power on (see notes 1 and 2)	V _{porr}	Measured from AGND		AVDD Min		V
VMID level to release POR – power on (see notes 1 and 2)	V _{porr}	Measured from AGND		1		V
POR active period (see notes 1 and 2)	t _{por}	Measured from POR active to POR release	30 (note 6)	Defined by DVDD VMID Rise T		μs
DVDD level to activate POR – power off (see note 5)	V_{por_off}	Measured from DGND		0.8		V
AVDD level to activate POR – power off (see note 5)	V_{por_off}	Measured from AGND		0.8		V
VMID level to activate POR – power off (see note 5)	V _{por_off}	Measured from AGND		0.7		V
Power on – POR propagation delay through device	t _{pon}	Measured from rising EDGE of POR		30		μs
Power down – POR propagation delay through device	t _{poff}	Measured from falling EDGE of POR		30		μs

Notes:

- 1
- POR is activated when DVDD or AVDD or VMID reach their stated V_{pora} level (Figure 8) POR is only released when DVDD and AVDD and VMID have all reached their stated V_{porr} levels (Figure 8). 2. 3. The rate of rise of VMID depends on the rate of rise of AVDD, the internal 50k Ω resistance and the external decoupling capacitor. Typical tolerance of 50K resistor can be taken as +/-20%.
- If AVDD, DVDD or VMID suffer a brown-out (i.e. drop below the minimum recommended operating level but do not go below V_{por_off}), then the chip will not reset and will resume normal operation when the voltage is back to the 4. recommended level again.
- The chip will enter reset at power down when AVDD or DVDD or VMID falls below Vpor_off. This may be important if the 5.
- supply is turned on and off frequently by a power management system. The minimum t_{por} period is maintained even if DVDD, AVDD and VMID have zero rise time. This specification is guaranteed by design rather than test. 6.

DIGITAL FILTER CHARACTERISTICS

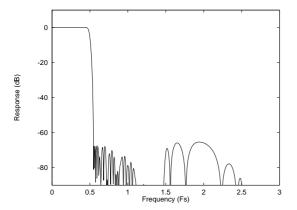
The WM8782A digital filter characteristics scale with sample rate.

PARAMETER	TEST CONDITIONS	MIN	TYP	MAX	UNIT
ADC Sample Rate (Single Rate)	ate – 48Hz typically)				
Passband	+/- 0.01dB	0		0.4535fs	
	-6dB		0.4892fs		
Passband Ripple				+/- 0.01	dB
Stopband		0.5465fs			
Stopband Attenuation	f > 0.5465fs	-65			dB
Group Delay			22		fs
ADC Sample Rate (Dual Rat	e – 96kHz typically)				
Passband	+/- 0.01dB	0		0.4535fs	
	-6dB		0.4892fs		
Passband Ripple				+/- 0.01	dB
Stopband		0.5465fs			
Stopband Attenuation	f > 0.5465fs	-65			dB
Group Delay			22		fs

0.02

Table 7 Digital Filter Characteristics

ADC FILTER RESPONSES



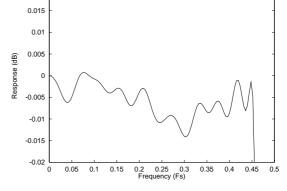


Figure 10 Digital Filter Frequency Response

Figure 11 ADC Digital Filter Ripple



Production Data

ADC HIGH PASS FILTER

The WM8782A has a digital highpass filter to remove DC offsets. The filter response is characterised by the following polynomial.

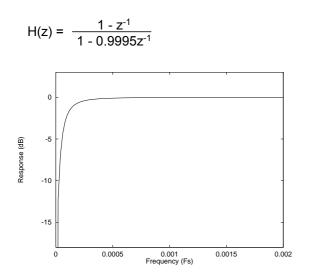


Figure 12 ADC Highpass Filter Response



APPLICATIONS INFORMATION

RECOMMENDED EXTERNAL COMPONENTS

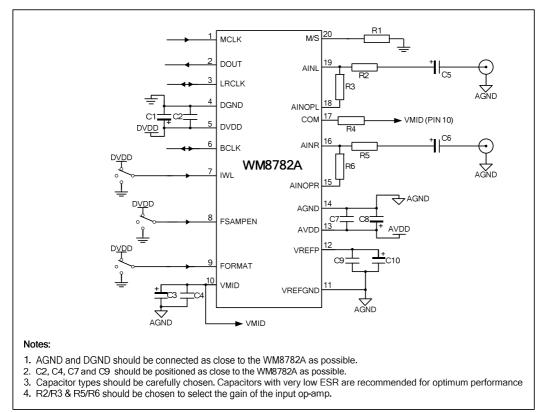


Figure 13 External Components Diagram

RECOMMENDED EXTERNAL COMPONENTS VALUES

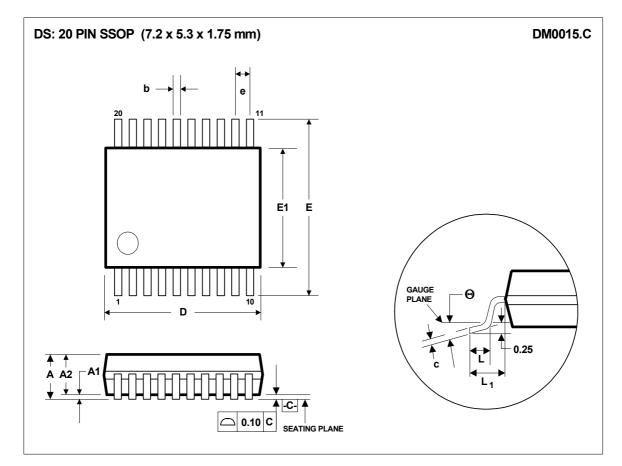
COMPONENT REFERENCE	SUGGESTED VALUE	DESCRIPTION	
C1 and C8	10µF	De-coupling for DVDD and AVDD	
C2 and C7	0.1µF	De-coupling for DVDD and AVDD	
C5 and C6	10μF	Analogue input AC coupling caps	
R1	10kΩ	Current limiting resistors	
R2 and R5	10kΩ	Internal op-amp input resistor	
R3 and R6	5kΩ	Internal op-amp feedback resistor	
R4	3.3kΩ	Common mode resistor	
C4	0.1µF	Reference de-coupling capacitors for VMID pin	
C3	10μF		
C9	0.1µF	Reference de-coupling capacitors for VREFP pin	
C10	10μF		

Table 8 External Components Description

The above Table 8 shows resistor values which will give a gain of 0.5. This assumes an input signal of 2Vrms to C4 and C5.



PACKAGE DIMENSIONS



Symbols	Dimensions (mm)				
	MIN NOM		MAX		
Α			2.0		
A ₁	0.05				
A ₂	1.65	1.75	1.85		
b	0.22	0.30	0.38		
С	0.09		0.25		
D	6.90	7.20	7.50		
е	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	1.25 REF				
θ	0°	4 ^o	8°		
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 = AE. REFER TO THIS SPECIFICATION FOR FURTHER DETAILS.



IMPORTANT NOTICE

Wolfson Microelectronics plc ("Wolfson") products and services are sold subject to Wolfson's terms and conditions of sale, delivery and payment supplied at the time of order acknowledgement.

Wolfson warrants performance of its products to the specifications in effect at the date of shipment. Wolfson reserves the right to make changes to its products and specifications or to discontinue any product or service without notice. Customers should therefore obtain the latest version of relevant information from Wolfson to verify that the information is current.

Testing and other quality control techniques are utilised to the extent Wolfson deems necessary to support its warranty. Specific testing of all parameters of each device is not necessarily performed unless required by law or regulation.

In order to minimise risks associated with customer applications, the customer must use adequate design and operating safeguards to minimise inherent or procedural hazards. Wolfson is not liable for applications assistance or customer product design. The customer is solely responsible for its selection and use of Wolfson products. Wolfson is not liable for such selection or use nor for use of any circuitry other than circuitry entirely embodied in a Wolfson product.

Wolfson's products are not intended for use in life support systems, appliances, nuclear systems or systems where malfunction can reasonably be expected to result in personal injury, death or severe property or environmental damage. Any use of products by the customer for such purposes is at the customer's own risk.

Wolfson does not grant any licence (express or implied) under any patent right, copyright, mask work right or other intellectual property right of Wolfson covering or relating to any combination, machine, or process in which its products or services might be or are used. Any provision or publication of any third party's products or services does not constitute Wolfson's approval, licence, warranty or endorsement thereof. Any third party trade marks contained in this document belong to the respective third party owner.

Reproduction of information from Wolfson datasheets is permissible only if reproduction is without alteration and is accompanied by all associated copyright, proprietary and other notices (including this notice) and conditions. Wolfson is not liable for any unauthorised alteration of such information or for any reliance placed thereon.

Any representations made, warranties given, and/or liabilities accepted by any person which differ from those contained in this datasheet or in Wolfson's standard terms and conditions of sale, delivery and payment are made, given and/or accepted at that person's own risk. Wolfson is not liable for any such representations, warranties or liabilities or for any reliance placed thereon by any person.

ADDRESS:

Wolfson Microelectronics plc Westfield House 26 Westfield Road Edinburgh EH11 2QB United Kingdom

Tel :: +44 (0)131 272 2000 Fax :: +44 (0)131 272 7001 Email :: <u>sales@wolfsonmicro.com</u>



X-ON Electronics

Largest Supplier of Electrical and Electronic Components

Click to view similar products for Audio A/D Converter ICs category:

Click to view products by Cirrus manufacturer:

Other Similar products are found below :

WM8781GEDSV WM8788GEDT WM8782SEDSV CS5346-CQZR CS4461-CZZ WM8782ASEDS/RV AD1877JRZ AD1974WBSTZ ADAU1977WBCPZ ADAU7112ACBZR7 ADAU7118ACPZRL ADAU1979WBCPZ-RL CS5345-CQZ WM8950CGEFL/V CS5340-CZZ CS5340-DZZ CS5341-CZZ CS5342-CZZ CS5344-CZZ CS5346-CQZ CS5351-DZZ CS5361-KSZ CS5361-KZZ CS5364-DQZ CS5381-KSZ CS5381-KZZ CS53L21-CNZ CS53L30-CNZ WM8782ASEDS/V WM8786GEDS/RV 6PADC3101TRGERQ1 PCM1800E PCM1801U PCM1803ADB PCM1803ADBG4 PCM1804DB PCM1804DBG4 PCM1804DBR PCM1804S1IDBRQ1 PCM1807PW PCM1807PWR PCM1808PW PCM1808PWG4 PCM1808PWRG4 PCM1851APJT PCM1851APJTR PCM1861DBT PCM1862DBTR PCM1864DBT PCM1865DBT