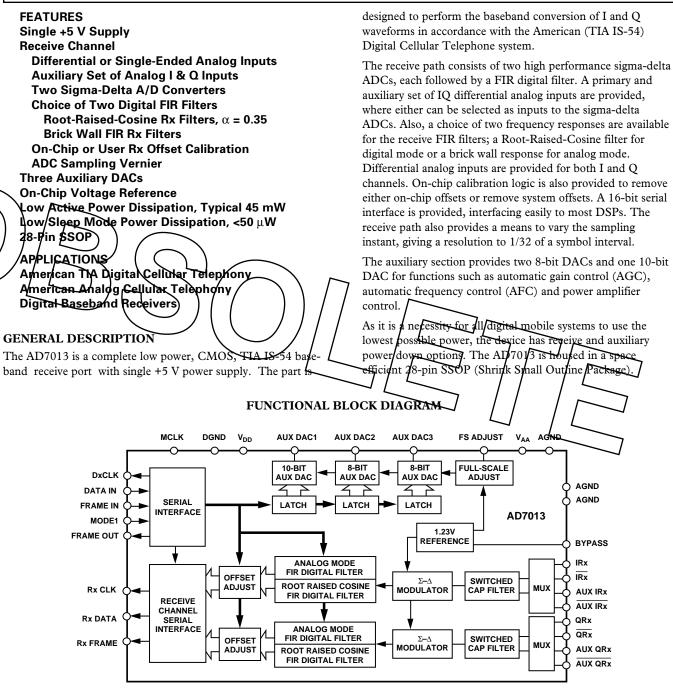


CMOS TIA IS-54 Baseband Receive Port

AD7013



REV. A

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Parameter	AD7013A	Units	Test Conditions/Comments
RECEIVE SECTION			
ADC SPECIFICATION			
Number of Input Channels	4		$(IRx-\overline{IRx})$ and
-			QRx-QRx; $CR12 = 0$
			(AUX IRx-AUX IRx) and
			$(AUX QRx - \overline{AUX QRx}); CR12 = 1$
Number of ADC Channels	2		
Resolution	15	Bits	
ADC Signal Range	2.6	Volts p-p	Measured Using an Input Sine Wave of 3 kH
Differential Signal Range	$V_{BIAS} \pm 0.65$	Volts	For Both Noninverting and
	BIAS		Inverting Analog Inputs
Single-Ended Signal Range	$V_{BIAS} \pm 1.3$	Volts	For Noninverting Analog Inputs;
omgre znava orgini range	BIAS = 113	, 0115	Inverting Analog Inputs = V_{BIAS}
V_{BIAS}	0.65 to (V _{AA} -0.65)	Volts min/max	Differential
BIAS	1.3 to $(V_{AA} - 1.3)$	Volts min/max	Single-Ended
Input Range Accuracy	± 7.5	%	Single-Ended
Accuracy	1.5	70	
Bias Offset Error	±7.5	mV	Autocalibration; $V_{BIAS} = min/max$
Dias Offset Enor	±55	mV	User Calibration; I & Q Offset
		111 V	Adjust Registers Equal to Zero
Dynamic Spacifications			Adjust Registers Equal to Zero
CMKR		dB typ	Measured Using an Input Sine Wave of
	$\sqrt{7^{40}}$		3 kHz with Both Noninverting and
			Inverting Inputs Tied Together
Dynamic Range			Digital Mode Filter; CR11 = 0
Dynamic Range		dlø typ	Analog Mode Eilter; $CRH = 1$
SNR ²		dB typ	Digital Mode Filter; $CR11 = 0$
SINK		AB min	Digital Hode Filler; CHI = 0
	68 L	dB-typ dB-min	Analog Mode Filter; <i>Q</i> R11 = 1
			Analog Wode Filler, CKI I – I
Input Sampling Rate	63 1.5552/1.28	dB typ MHz	MCEK = 6.2208 MHz/5.12 MHz; MCEK
Output Word Rate	97.2/80	kHz	MCLK = 6.2208 MHz/5.12 MHz;
Output word Nate	91.2/00	KI IZ	$4 \times \text{Sampling of the Symbol Rate, MCLK/6}$
	48.6/40	kHz	MCLK = 6.2208 MHz/5.12 MHz;
	10.0/10	KIIZ	$2 \times \text{Sampling of the Symbol Rate, MCLK/12}$
RECEIVE DIGITAL FILTERS			
Digital Mode			MCLK = 6.2208 MHz
Root-Raised-Cosine	$\alpha = 0.35$		
Settling Time	329.2	μs	
Absolute Group Delay	164.6	μs	
Frequency Response			
0–7.8975 kHz	±0.05	dB max	
11.9 kHz	-3.0	dB	
16.4025 kHz	-19	dB	
> 30 kHz	-66	dB max	
Analog Mode			MCLK = 5.12 MHz
Brick Wall Filter			
Settling Time	400	μs	
Absolute Group Delay	200	μs	
Frequency Response	200	pus	
0–8 kHz	0 to -0.5	dB max	
11.4 kHz	-3.0	dB	
15 kHz	-24	dB	
>17 kHz	-24 -68	dB max	
TIA IS-54 RECEIVE SPECIFICATIONS			
Error Vector Magnitude ³ Error Offset Magnitude ³	2	% rms typ	Measured Using a Full-Scale Input
	1	% rms typ	1

Parameter		AD7013A		Units	Test Conditions/Comments
AUXILIARY SECTION					
	AUX DAC1	AUX DAC2	AUX DAC3		
Resolution	10	8	8	Bits	
DC Accuracy	10	0	0	210	
Integral	±3	±1	±1	LSBs max	
Differential	-1.5/+4	± 1	±1 ±1	LSBs max	AUX DAC2 & AUX DAC3 Guarantee
Differential	-1.5/+4	± 1	± 1	LSDS max	Monotonic
Zero Code Leakage	±500	±500	±500	nA max	Wonotome
Gain Error	±7.5	±7.5	±500 ±7.5	% max	
Output Full-Scale Current	566				P = 1010
-	000	280	280	μA	$R_{SET} = 18 \text{ k}\Omega$
Output Impedance ⁴		2		MΩ typ	
Output Voltage Compliance		2.6		Volts max	
Coding		Binary			
Power Down Option		Yes			
REFERENCE SPECIFICATIONS					
V _{REF}		1.23		Volts typ	
Reference Accuracy		±5		% max	
Reference Impedance		20		kΩ typ	
		20		itte typ	
LOGICINIUTS)	\checkmark				
N _{INH} , Input High Voltage		V_{DD} -0.9		V min	
VINI, Input Low Valtage	4/~	0,2		V max	
Inthe, Input Current		\10\ /	7	μA max	
C _{IN} , Input Capacitance) / /)o) / /	' r	pF max	
LOGIC OUTPUTS			/		
V _{OH} , Output High Voltage		V _{DD} =04	/	Lymin	$ \mathbf{I}_{00T} \leq 40 \mu$
V_{OL} , Output Low Voltage		0.4		Vmax	$ I_{OUT} \le 40 \ \mu \text{m}$
V _{OL} , Output Low Voltage	\vdash		\sim		
POWER SUPPLIES					
V _{DD}		4.5/5.5	<u> </u>	V _{MIN} /V _{MAX}	
I_{DD}^{5}					
All Sections Active		10.5		mA max	CR14 = CR15 = CR16 = CR17 = 1
		9		mA typ	MCLK = 6.2208 MHz; 80 pF
					Load on DxCLK
ADCs Active Only		8.6		mA max	CR14 = 1; CR15 = CR16 = CR17 = 0
		-			MCLK = 6.2208 MHz; 80 pF
					Load on DxCLK
AUX DACs Active Only		2.2		mA max	CR14 = 0; CR15 = CR16 = CR17 = 1
NON DIAS ACtive Only		2.2			MCLK Inactive, MCLK = $0 V$
10-Bit AUX DAC Active		1.6		mA max	CR14 = CR15 = CR16 = 0; CR17 = 1
To bit non bho neive		1.0			MCLK Inactive, MCLK = 0 V
All Sections Powered Down ⁶		2		mA max	CR14 = CR15 = CR16 = CR17 = 0
All Sections Fowered Dowll ^o		4			
					MCLK = 6.2208 MHz; 80 pF
				.	Load on DxCLK
		30		μA typ	MCLK =100 kHz; 80 pF
		10		μA max	Load on DxCLK MCLK Inactive, MCLK = 0 V

NOTES

¹Operating temperature ranges as follows: A version: -40°C to +85°C. ²SNR calculation includes noise and distortion components.

⁻SNK calculation includes noise and distortion components.
³See Terminology.
⁴Sampled tested only.
⁵Measured while the digital inputs are static and equal to 0 V or V_{DD}.
⁶With all sections powered down, I_{DD} is proportional to the capacitive load on DxCLK. For example, I_{DD} is typically 1.7 mA with 80 pF load and 600 μA with 10 pF load.

Specifications subject to change without notice.

TERMINOLOGY

Sampling Rate

This is the rate at which the modulators on the receive channels sample the analog input.

Output Rate

This is the rate at which data words are made available at the RxDATA pin.

Integral Nonlinearity

This is the maximum deviation from a straight line passing through the endpoints of the DAC or ADC transfer function.

Differential Nonlinearity

This is the difference between the measured and the ideal 1 LSB change between any two adjacent codes in the DAC or ADC.

Dynamic Range

Dynamic Range is the ratio of the maximum rms input signal to the rms noise of the converter, expressed logarithmically, in decibels $(dR = 20 \log_{10} [ratio])$.

Signal to (Notse + Distortion) Ratio

This is the measured ratio of signal to (noise + distortion) at the output of the receive channel. The signal is the rms amplitude of the fundamental. Noise is the rms sum of all nonfundamental signals up to half the sampling frequency $(f_{s}/2)$, excluding de. The ratio is dependent upon the number of quantization levels in the digitization process; the more levels, the smaller the quantization noise. The theoretical signal to (noise + distortion) ratio for a sine wave is given by:

Signal to (Noise + Distortion) = (6.02N)

Settling Time

This is the digital filter settling time in the AD7013 receive section.

Bias Offset Error

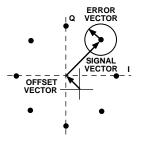
This is the amount of offset in the receive channel ADC when the differential inputs are tied together.

Receive Error Vector Magnitude

This is a measure of the rms signal error vector introduced by the receive Root-Raised Cosine digital filter. This is measured by applying an ideal transmit signal (i.e., an ideal $\pi/4$ DQPSK modulator and an ideal transmit Root-Raised Cosine filter) to the receive channel and measuring the resulting rms error vector.

Offset Vector Magnitude

This is a measure of the offset vector introduced by the AD7013 as illustrated in the figure below. The offset vector is calculated so as to minimize the rms error vector for each of the constellation points.



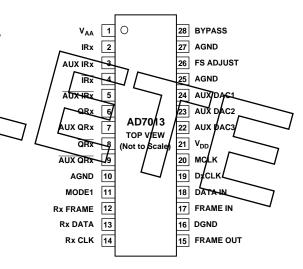
ABSOLUTE MAXIMUM RATINGS¹

$(T_A = +25^{\circ}C \text{ unless otherwise noted})$
V_{AA} , V_{DD} to GND
AGND to DGND
Digital I/O Voltage to DGND $\dots -0.3$ V to V _{DD} +0.3 V
Analog I/O Voltage to AGND $\dots \dots \dots$
Operating Temperature Range
Industrial (A Version)
Storage Temperature Range
Maximum Junction Temperature
SSOP θ_{JA} Thermal Impedance+122°C/W
Lead Temperature Soldering
Vapor Phase (60 sec)+215°C
Infrared (15 sec)+220°C

NOTES

¹Stresses above 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 any other conditions above those listed in the operational sections of this specification is not implied. Exposure to absolute maximum rating conditions extended periods may affect device reliability.

PIN CONFIGURATION



ORDERING GUIDE

Model	Temperature Range	Package Option*
AD7013ARS	–40°C to +85°C	RS-28

*RS = SSOP.

CAUTION_

ESD (electrostatic discharge) sensitive device. Electrostatic charges as high as 4000 V readily accumulate on the human body and test equipment and can discharge without detection. Although this device features proprietary ESD protection circuitry, permanent damage may occur on devices subjected to high energy electrostatic discharges. Therefore, proper ESD precautions are recommended to avoid performance degradation or loss of functionality.

1.76 1



		PIN FUNCTION DESCRIPTIONS
SSOP Pin		
Number	Mnemonic	Function
POWER SU	PPLY	
1	V_{AA}	Positive Power Supply for Analog section. A 0.1 μ F decoupling capacitor should be connected between this pin and AGND.
21	V _{DD}	Positive Power Supply for Digital section. A 0.1 μ F decoupling capacitor should be connected between this pin and DGND. Both V _{AA} and V _{DD} should be externally tied together.
10, 25, 27	AGND	Analog Ground.
16	DGND	Digital Ground. Both AGND and DGND should be externally tied together.
ANALOG S	IGNAL AND REFERENCE	
28	BYPASS	Reference Decoupling Output. A 10 nF decoupling capacitor should be connected between this pin and AGND.
2, 4	IRx, IRx	Differential Analog Inputs for the I receive channel. These are the primary receive
6,8	QRx, \overline{QRx}	analog inputs and are selected by setting CR12 to a zero in the command register. Differential Analog Inputs for the Q receive channel. These are the primary receive
}, 5) / [AUXIRX, AUX IRX	analog inputs and are selected by setting CR12 to a zero in the command register. Auxiliary Differential Analog Inputs for the I receive channel. The Auxiliary inputs
	ALX QRY, AUX QRY	are selected by setting CR12 to a one in the command register.
7,9 [Add OKK, AND OKX	are selected by setting CR12 to a one in the command register.
24	AUX IDAGI	Analog output from the 10-bit auxiliary DAC.
3, 22	AUX DACZAUX DAC3	Analog outputs from the 8-bit/auxiliary DACs.
26	FS ADJUST	An external resistor is connected from this pin/to ground to determine the full-
20		scale current for AUX DAC1, AUX DAC2, and AUX DAC3
SEDIAL IN	FERFACE AND CONTROL	
20	MCLK	Master Clock, Digital Input. When operating in S-54 Digital mote this pin should be driven by a 6.2208 MHz CMOS compatible clock source and 5 12 MHz clock
19	DxCLK	source for Analog Mode. Transmit Clock, Digital Output. This is a continuous clock equal to MCLK/2 which can be used to clock the serial port of a DSP.
17	FRAME IN	Digital Input. This is used to frame the clocking in of 16-bit words for the control registers serial interface.
18	DATA IN	Digital Input. Transmit Serial Data, digital input. This pin is used to clock in data for the serial interface on the rising edge of DxCLK.
15	FRAME OUT	Digital Output. This output represents a buffered version of FRAME IN and is controlled by the MODE1 pin. This pin can be used to daisy chain the FRAME IN signal.
11	MODE1	Digital Input. This pin determines the state of FRAME OUT. When MODE1 is his FRAME IN is buffered and made available on FRAME OUT. When MODE1 is low, FRAME OUT is in 3-STATE.
DECEIVE	NTEDEACE AND CONTROL	
RECEIVE I	NTERFACE AND CONTROI RxCLK	Output Clock for the receive section interface.
14	RxFRAME	Synchronization output for framing I and Q data at the receive interface.
12	RXDATA	Receive Data, digital output. I and Q data are available at this pin via a 16-bit serial
1.7		interface. Data is valid on the falling edge of RxCLK. I and Q data are clocked out as two 16-bits words, with the I word being clocked first. The last bit in each 16-bit word is a I/\overline{Q} flag bit, indicating whether that word is an I word or a Q word.

$\begin{array}{l} \textbf{CONTROL SERIAL INTERFACE TIMING}^{1} \quad \begin{array}{l} (V_{AA}=+5~V\pm10\%;~V_{DD}=+5~V\pm10\%;~AGND=DGND=0~V,\\ f_{MCLK}=6.2208~MHz;~T_{A}=T_{MIN}~to~T_{MAX},~unless~otherwise~noted) \end{array}$

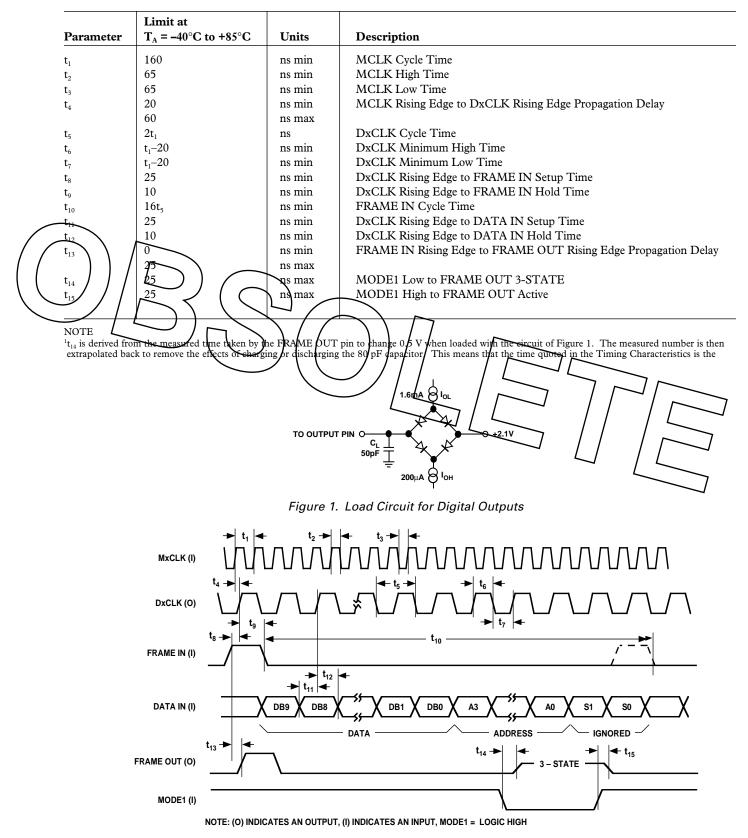


Figure 2. 16-Bit Serial Interface for Writing to the AD7013 Internal Registers

$\label{eq:relation} \begin{array}{l} \textbf{RECEIVE SECTION TIMING} & (V_{AA}=+5~V\pm10\%;~V_{DD}=+5~V\pm10\%;~AGND=DGND=0~V, f_{MCLK}=6.2208~MHz; \\ \textbf{T}_{A}=\textbf{T}_{MIN}~to~\textbf{T}_{MAX},~unless~otherwise~noted) \end{array}$

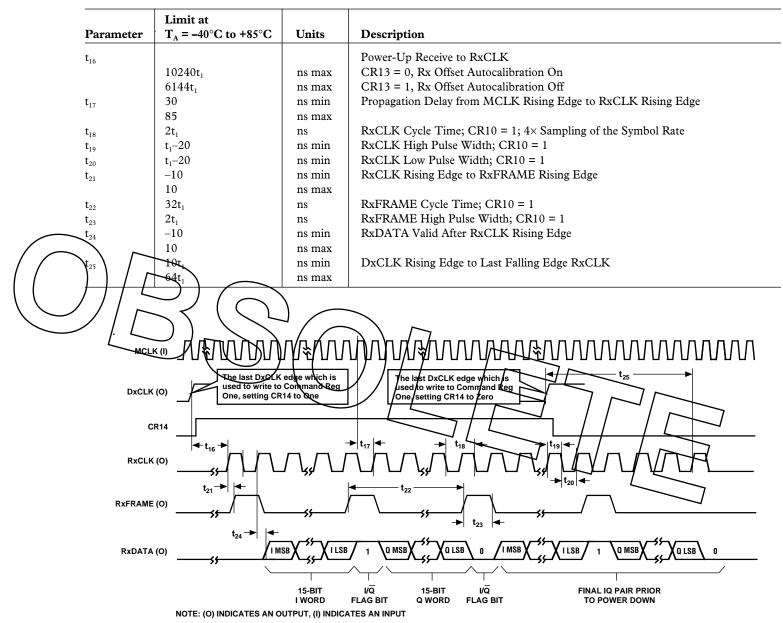


Figure 3. Receive Serial Interface Timing with 4× Sampling of the Symbol Rate (CR10 = 1)

$(V_{AA} = +5 V \pm 10\%; V_{DD} = +5 V \pm 10\%; AGND = DGND = 0 V, f_{MCLK} = 6.2208 \text{ MHz};$ **RECEIVE SECTION TIMING** T_A = T_{MIN} to T_{MAX}, unless otherwise noted)

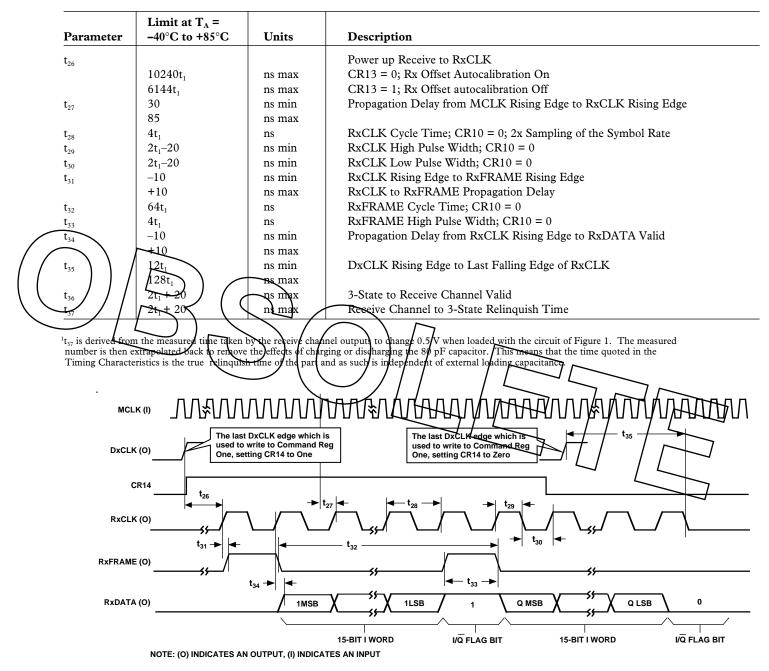


Figure 4. Receive Serial Interface Timing with $2 \times$ Sampling of the Symbol Rate (CR10 = 0)

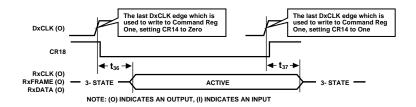


Figure 5. Receive Serial Interface 3-State Timing

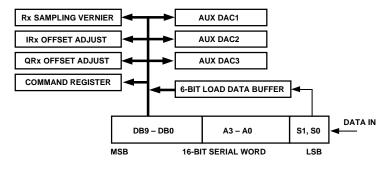
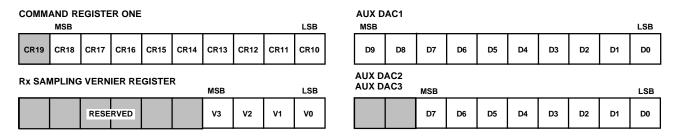


Figure 6. AD7013 Registers



Register Name	A3		ldress A1	A0	Register Size	Reset State	Description
				-			
COMMAND VERNER				_م 	9 Bits 4 Bits	All Zeros All Zeros	The COMMAND register is used to select various operating modes of the AD7013. A detailed description of the COMMAND register is given in Table II. The VERNIER register allows additional group delay to be introduced into the I and Q ADCs. This provides a means to vary
\sim / \sim	\square		_ \		/	\frown	the ADC sampling instant.
IRx OFFSET	0	1)	10 Bits	All Zeros	The contents of the IRx OFFSET register are substracted from the i channel ADC word. When autocalibration is selected, this register is automatically loaded by the AD7013 at the beginning of a normal operation. When user calibration is selected, this register can be externally loaded with a two somplement offset 10-bit
							word to be subtracted from subsequent ADC samples.
QRx OFFSET	. 0	1	1	0	10 Bits	All Zeros	The contents of the QRX OFFSET register are substracted from the Q channel ADC word. When auto calibration is selected, this register is automatically loaded by the AD7013 at the beginning of a normal operation. When user calibration is selected this register can be externally loaded with a twos complement offset 10-bit word to be subtracted from subsequent ADC samples.
AUX DAC1	0	1	1	1	10 Bits	All Zeros	The 10-bit auxiliary DAC current output is determined by this register. The output current is equal to {AUX DAC1 _{FULL SCALE} * N/2 ¹⁰ } where N is the 10-bit word contained in the AUX DAC1 register and AUX DAC1 _{FULL SCALE} is determined by the value of R_{SET} connected between FSADJUST and AGND.
AUX DAC2	1	0	0	0	8 Bits	All Zeros	The 8-bit auxiliary DAC current output is determined by this register. The output current is equal to $\{AUX DAC1_{FULL SCALE} * N/2^8\}$ where N is the 8-bit word contained in the AUX DAC2 register and AUX DAC2 _{FULL SCALE} is determined by the value of RSET connected between FS ADJUST and AGND.
AUX DAC3	1	0	0	1	8 Bits	All Zeros	The 8-bit auxiliary DAC current output is determined by this register. The output current is equal to $\{AUX DAC3_{FULL SCALE} * N/2^8\}$ where N is the 8-bit word contained in the AUX DAC3 register and AUX DAC3_{FULL SCALE} is determined by the value of R _{SET} connected between FS ADJUST and AGND.
RESET	0	0	0	1	N/A	N/A	When this address in selected, all of the internal registers are initialized to their reset state.
6-Bit LOAD	0	0	1	1	N/A	N/A	When this address is used, a special loading sequence, as shown in Table IV, is used to write to any of the internal registers.
	0	0	0	0	N/A	N/A	No Action.
N/A	1 0	0	0	0	1 1 1/11	1 1 1/11	



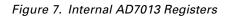


Table II. Command Register One

CR10	= 0 = 1	Low ADC sample rate. The sample rate of the receive ADCs are equal to $2\times$ the symbol rate or equal to MCLK/128. High ADC sample rate. The sample rate of the ADCs are equal to $4\times$ the symbol rate or equal to MCLK/64.
CRN		RRC Receive FIR filter. This selects the root-raised consine filter response for the receive sigma-delta ADCs. This is used to match the transmit RRC filter as required by the IS-54 standard. The frequency response is shown in Figure 16. Analog Mode FIR filter. This selects a filter response which has a sharper roll-off than the RRC FIR filter and the frequency response has also been scaled to operate at a master clock frequency of 5.12 MHz. This allows the sampling rate of the receive ADCs to be a multiple of 10 kHz as required for analog cellular. The frequency response is shown in Figure 17.
CR12	= 0 = 1	Primary ADC inputs. This selects IRx and IRx as the I channel inputs and QRx and QRx as the Q channel inputs. Auxiliary ADC inputs. This selects AUX IRx and AUX IRx as the I channel inputs and AUX QRx and AUX QRx and AUX QRx as the Q channel inputs.
CR13	= 0 = 1	Auto ADC offset calibration. If auto calibration is selected, then an offset word for both ADCs is calculated each time the receive ADCs are brought out of sleep mode. This allows ADC offsets within the AD7013 to be automatically calibrated out. User ADC offset calibration. When user calibration is selected, then contents of the offset registers are not updated by the AD7013 when brought out of sleep mode. This allows the user to load the offset register externally thereby allowing the AD7013 to also calibrate out external offsets.
CR14	= 0 = 1	Receive ADC sleep mode. This enters the I and Q ADCs into a low power sleep mode after outputting the current IQ sample. Receive ADC active mode. This activates the receive ADCs for normal operation.
CR15	= 0 = 1	8-Bit AUX DAC3 sleep mode. This enters the 8-bit auxiliary DAC into a low power sleep mode. 8-Bit AUX DAC3 active mode. This activates the 8-bit auxiliary DAC for normal operation.
CR16	= 0 = 1	8-Bit AUX DAC2 sleep mode. This enters the 8-bit auxiliary DAC into a low power sleep mode 8-Bit AUX DAC2 active mode. This activates the 8-bit auxiliary DAC for normal operation.
CR17	= 0 = 1	10-Bit AUX DAC1 sleep mode. This enters the 10-bit auxiliary DAC into a low power sleep mode. 10-Bit AUX DAC1 active mode. This activates the 10-bit auxiliary DAC for normal operation.
CR18	= 0 = 1	3-State Enable. This enables the 3-state buffers on the receive serial interface.3-State Disable. This disables the 3-state buffers on the receive serial interface, entering the serial interface into 3-state.
CR19	= X	No Action.

RECEIVE SECTION

The receive section consists of I and Q receive channels, each comprising of a simple switched-capacitor filter followed by a 15-bit sigma-delta ADC. The data is available on a 16-bit serial interface, interfacing easily to most DSPs. On-board digital filters, which form part of the sigma-delta ADCs, also perform system level filtering. A choice of two digital filter responses are available, optimized for either $\pi/4$ DQPSK digital mode or the existing analog cellular system. For digital mode, Root-Raised Cosine digital filters can be selected; whereas for analog mode, digital filters with a -3 dB point of 11.4 kHz can be selected. Their amplitude and phase response characteristics provide excellent adjacent channel rejection. A means is also provided to calibrate either on-chip or receive path offsets in both the I and Q channels. The receive section is also provided with a low power sleep mode, drawing only minimal current between receive bursts.

Switched Capacitor Input

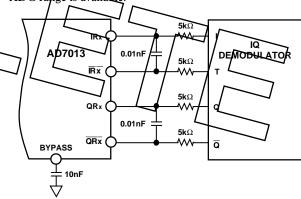
The receive section analog front-end is sampled at MCLK/4 by a switched-capacitor filter. The filter has a zero at MCLK/8 as shown in Figure 8a. The receive channel also contains a digital low-pase filter (further details are contained in the following section which operates at a clock frequency of MCLK/8. Due to sampling nature of the digital filter, the pass band is repeated th bout the operating dock frequency (MCLK/8) and at multiples of Because the first null of the the clock frequency (Figure 8b). switched-capacitor filter coincides with the first image of the digital filter, this image is attenuated by an additional 30 dBs (Figure 8c) further simplifying the external antialiasing requirements. A simple R-C Network can be used to attenuate the digital filter image MCLK/8 as shown in Figure 9.



The receive channel uses differential inputs to interface more easily to IQ demodulators and also to provide common-mode noise rejection. However, if required the receive channel inputs can also be configured for single ended operation. The primary and auxiliary channels have similar performance and either can be used for differential operation or single-ended operation. The CR12 control bit determines whether the primary or auxiliary inputs are connected to the differential inputs of the sigma-delta modulator.

Figure 9 illustrates an antialiasing filter comprised of a single pole RC network with a -3 dB frequency of 159 kHz. The low-pass filter provides sufficient rejection at images of the FIR digital filter illustrated in Figure 10c.

For single ended operation, the inverting input should be connected to a bias voltage and the noninverting input should swing ± 1.3 V around this bias voltage in order to exercise the entire ADC range. In applications where the full ± 1.3 V range is not required, the on-chip 1.23 V reference can be used to provide the bias voltage. For instance as in Figure 10, an OP295 rail-to-rail low power op amp is used to buffer the BYPASS pin in order to generate a 1.23 V_{BIAS}. The V_{BIAS} is connected to the inverting input thereby setting the single-ended input range equal to 0 V to 2.46 V. Also with the addition of an attenuator circuit the input range can be expanded to 0 V to 4.92 V as shown on the second ADC channel. If the inverting input is tied to AGND, then only half the ADC range is available.



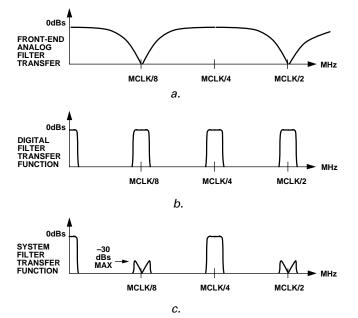


Figure 8. Switched Capacitor and Digital Filter Transfer Functions

Figure 9. External RC Network for Differential Signals

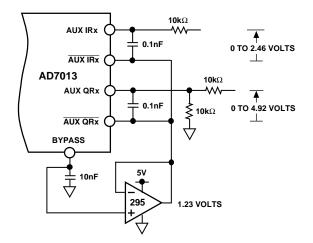


Figure 10. External RC Network for Single-Ended Signals

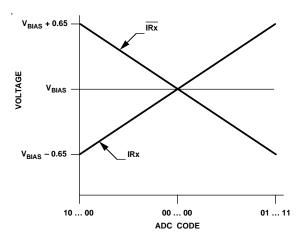


Figure 11. ADC Transfer Function for Differential Operation

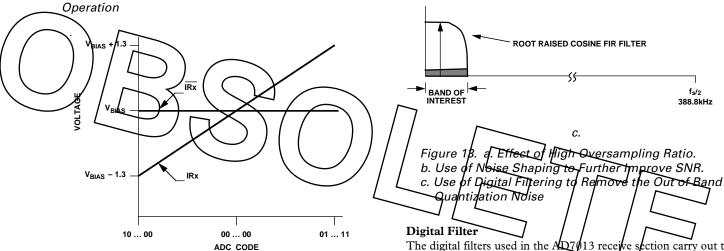


Figure 12. ADC Transfer Function for Single-Ended Operation

SIGMA-DELTA ADC

The AD7013 receive channels employ a sigma-delta conversion technique, which provides a high resolution 15-bit output for both I and Q channels with system filtering being implemented on-chip.

The output of the switched-capacitor filter is continuously sampled at MCLK/8, by a charge-balanced modulator, and is converted into a digital pulse train whose duty cycle contains the digital information. Due to the high oversampling rate which spreads the quantization noise from 0 to $f_s/2$, the noise energy which is contained in the band of interest is reduced (Figure 13a). To reduce the quantization noise still further, a high order modulator is employed to shape the noise spectrum, so that most of the noise energy is shifted out of the band of interest (Figure 13b).

The digital filter that follows the modulator removes the large out of band quantization noise (Figure 13c), while converting the digital pulse train into parallel 15-bit wide binary data. The 15-bit I and Q data plus an I/Q flag bit is made available, via a serial interface, as a 16-bit word, MSB first.

The digital filters used in the AD7013 receive section carry out two important functions. First, they remove the out of band guantization noise which is shaped by the analog modulator. Second, they are also designed to perform system level filtering, providing the Root-Raised Cosine filter as required for TIA IS-54.

Since digital filtering occurs after the A/D conversion process, it can remove noise injected during the conversion process. Analog filtering cannot do this. Also, the digital filter combines low passband ripple with a steep roll off, while also maintaining a linear phase response. This is very difficult to achieve with analog filters.

Filter Characteristics

QUANTIZATION NOISE

NOISE SHAPING

a.

b.

BAND OF

BAND OF

The digital filter is a 256-tap FIR filter, clocked at 1/8 the master clock frequency. A choice of two frequency responses are available: a Root-Raised Cosine response (CR11 = 0) and a brick wall response at 11.4 kHz (CR11 = 1) for analog mode. Figure 16 and Figure 17 illustrate the respective frequency responses for both digital mode and analog mode while Figure 18 compares the low frequency response of the digital filters.

Due to the low-pass nature of the receive filters there is a settling time associated with step input functions. Output data will not be meaningful until all the digital filter taps have been loaded with data samples taken after the step change. Hence, the AD7013 digital filters have a settling time of $256 \times 8t_1$ (i.e., 329.2μ s when MCLK = 6.2208 MHz and 400μ s when MCLK = 5.12 MHz).

f_s/2

388.8kHz

fs/2

388.8kHzMHz

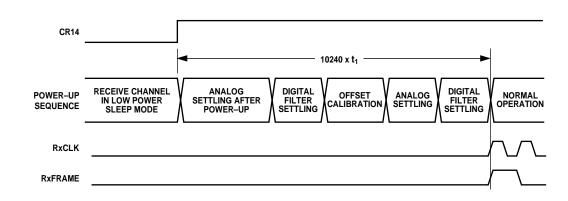
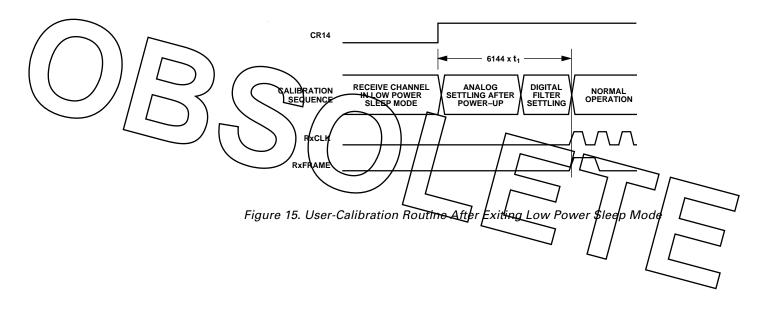
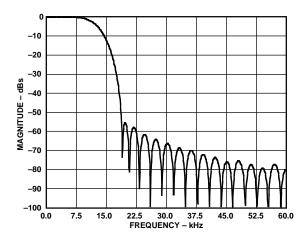
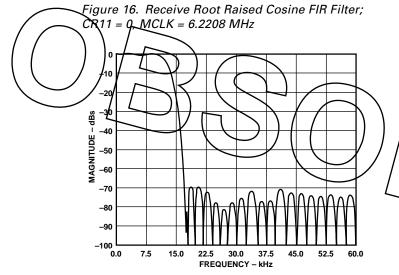


Figure 14. Autocalibration Routine After Exiting Low Power Sleep Mode









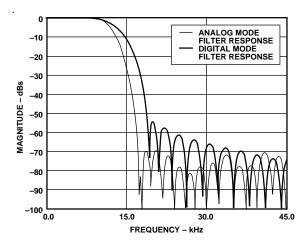
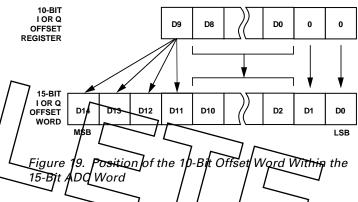


Figure 18. Comparision of the Two Frequency Responses Where Digital Mode was Clocked at 6.2208 MHz and Analog Mode was Clocked at 5.12 MHz

Receive Offset Calibration

Included in the digital filter is a means by which receive signal offsets may be calibrated out. Each channel of the digital low-pass filter section has an offset register. The offset register can be made to contain a value representing the dc offset of the preceding analog circuitry. In normal operation, the value stored in the offset register is subtracted from the filter output data before the data appears on the serial output pin. By so doing, dc offsets in the I and Q channels get calibrated out. Autocalibration or user calibration can be selected. Autocalibration will remove internal offsets only while user calibration allows the user to write to the offset register in order to also remove external offsets.

The offset registers have enough resolution to hold the value of any dc offset between ± 153 mV (1/8th of the input range). The 10-bit offset register represents a twos-complement value which is mapped to a 15-bit twos-complement word as shown in Figure 19. The contents of the offset registers are subtracted from their respective ADC samples.



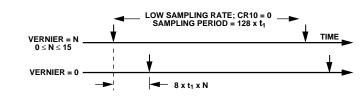
Receive Offset Adjust: Alto-Calibration (CR1 $\beta = p$) If receive autocalibration has been selected (CR13 = 0), then the AD7013 will initiate an autocalibration routine each time the receive path is brought out of the low power sleep mode (CR14 = 0). The AD7013 internally disconnects the differential inputs from the input pins and shorts the differential inputs to measure the resulting ADC offset. This is then averaged 16 times to reduce ADC noise, and the averaged result is then placed in the offset register. The input to the ADC is then switched back for normal operation, and after allowing for both analog settling and digital filter settling, the first IQ sample pair is output (Figure 14). Autocalibration will only remove on-chip offsets.

Receive Offset Adjust: User Calibration (CR13 = 1)

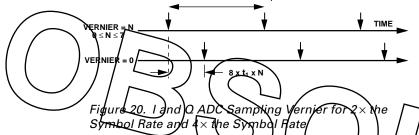
When user calibration has been selected, the receive offset register can be written to, allowing offsets in the IF/RF demodulation circuitry to be also calibrated out. However, the user is now responsible for calibrating out receive offsets belonging to the AD7013. When the receive path enters the low power mode (CR14 = 0), the offset registers remain valid. After powering up, the first IQ sample pair is output once time has elapsed for both the analog circuitry to settle and also for the output of the digital filter to settle as shown in Figure 15.

ADC Sampling Vernier

Also included in the digital filter is the means to vary the sampling instant, as Figure 20 illustrates. The absolute group delay can be varied from a minimum of four symbols to a maximum of four and a half symbols allowing the user to define the sampling instant to a resolution 1/32 of the symbol rate. The vernier can be used to seek the optimum sampling instant for minimum Inter-Symbol-Interference (ISI).







A 4-bit vernier register is used to set the sampling instant for both the I and Q receive ADCs. When the vernier register is programmed with zero the ADCs will have a minimum group delay of approximately 165 μ s. Nonzero values in the vernier register will add additional group delay thereby moving the sampling instant for both ADCs. After programming the sampling vernier it takes eight symbols (\approx 330 μ s) for the digital filter to settle. When the ADC is operating at the high rate, vernier values from 8 to 15 yield similar sampling instants as vernier values from 0 to 7, but delayed by an additional 1/4 of a symbol period.

Table III. Loading Sequence for the 16-Bit Interface

DB9-DB0	A3-A0	S1, S 0	Action
D9-D0	Destination Address	Ignored	Destination Reg←D9–D0
-			

Table IV. Loading Sequence for the 6-Bit Interface

DB9-DB0	A3-A0	S1, S0	Action
Ignored	0011	D9, D8	D9←S1 and D8←S0
Ignored	Destination Address	D7, D6	$D7 \leftarrow S1$ and $D6 \leftarrow S0$
Ignored	Destination Address	D5, D4	D5←S1 and D4←S0
Ignored	Destination Address	D3, D2	D3←S1 and D2←S0
Ignored	Destination Address	D1, D0	D1←S1 and D0←S0
			Destination Reg←D9–D0

Receive Section Digital Interface

The receive interface can be connected to DSP processors requiring the use of only one serial port. The 15-bit I and Q samples are made available as 16-bit words, where the last bit in each word is an I/Q flag bit.

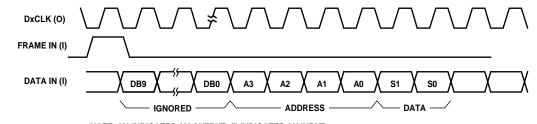
The serial data is made available on the RxDATA pin, with the I/Q flag indicating whether the 16-bit word being clocked out is an I sample or a Q sample. Although the I data is clocked out before the Q data, internally both samples are processed together. The receive interface (RxCLK, RxFRAME & RxDATA) can be 3-Stated by setting CR18 to zero, CR18 should be set high for normal operation.

When the receive section is put into sleep mode, by setting CR14 to zero, the receive interface will complete the current IQ cycle before entering into a low power sleep mode.

High Sampling Rate (CR 10=1)

The timing diagram for the receive interface is shown in Figure 3. The output word rate per channel is equal to 97.2 kHz (MCLK/64) which corresponds to 4 times the symbol rate.

When the receive section is brought out of sleep mode (CR14 = 1), the receive section will initiate an offset autocalibration routine if CR13 = 0. Once the receive offset calibration routine is complete then RxCLK will continuously shift out I and Q data, always beginning with I data. RxFRAME provides a framing signal that is used to indicate the beginning of an I or Q, 16-bit data word that is valid on the next falling edge of RxCLK. On coming out of sleep, RxFRAME goes high one clock cycle before the beginning of I data, and subsequently goes high in the same clock cycle as the last bit of each 16-bit word (both I and Q). RxDATA is valid on the falling edge of RxCLK and is clocked out MSB first, with the I/Q flag bit indicating whether the 16-bit word is an I sample or a Q sample.



NOTE: (O) INDICATES AN OUTPUT, (I) INDICATES AN INPUT

Figure 21. 6-Bit Serial Interface for Internal AD7013 Registers

Low Sampling Rate (CR10 = 0)

The timing diagram for the receive interface is shown in Figure 4. The output word rate per channel is equal to 48.6 kHz (MCLK/ 128) which corresponds to two times the symbol rate. The low sampling rate operates in a similar manner to that described for the high sampling rate.

AUXILIARY DACs

One 10-bit auxiliary DAC and two 8-bit auxiliary DACs are provided for extra control functions such as automatic gain control, automatic frequency control and power control. Figure 22 illustrates a simplified block diagram of the auxiliary DACs. The AUX DACs consist of high impedance current sources, designed to operate at very low currents while maintaining their DC accuracy. The DACs are designed using a current segmented architecture. The bit currents corresponding to each digital input are either routed to the analog output (bit = 1) or to AGND (bit = 0).

Each of the auxiliary DACs has independent low power sleep modes. The command register has three control bits CR17, CR16 and CR 5 which control AUX DAC1, AUX DAC2 and AUX DAC3 respectively. A logit 0 represents low power sleep mode and a logic 1 represents normal operation.

The full-scale currents of the auxiliary DACs are controlled by a single external resistor, R_{str} , connected between the FS ADJUST pin and AGND. The relationship between full-scale current and R_{str} is given as follows:

10-Bit AUX DAC $AUX DAC_{FULL SCALE} (mA) = 7992 \times V_{REF} (\Omega) / R_{SET} (\Omega)$ 8-Bit AUX DACs

 $AUX DAC_{FULL SCALE}(mA) = 3984 \times V_{REF}(V) / R_{SET}(\Omega)$

By using smaller values of R_{SET} , thereby increasing AUX DAC fullscale current, improved INL and DNL performance is possible as shown in Table V.

Table V. AUX DAC1 INL and DNL as a Function of $R_{\mbox{\scriptsize SET}}$

R _{set}	Worst Case INL (LSBs)	Worst Case DNL (LSBs)		
18 kΩ	-1.45	+1.83		
9 kΩ	+1.22	+1.59		
4.5 kΩ	+1.18	+1.38		

Digital Interface

Communication with the Command register, auxiliary DACs, ADC offset registers and ADC vernier is accomplished via the 3-pin serial interface. Either one of two loading formats may be used to write to any of the AD7013's internal registers. The first format consists of a single 16-bit serial word to write to any internal register (Table III). The second format consists of five 16-bit serial words, where only the last 6 bits in each 16-bit word are used to load five 2-bit data nibbles. The load sequence for this format is given is Table IV. The second format is only enable when the Register Address 3 is used as the destination register as shown in Table I.

PCB Layout Considerations

The use of an analog ground plane is recommended, where the ground plane extends around the analog circuitry. Both AGND and DGND should be externally tied together and connected to the analog ground plane.

Good power supply decoupling is very important for best ADC performance. A 0.1 μ F ceramic decoupling capacitor should be connected between V_{AA} and the ground plane. The physical placement of the capacitor (surface mount if possible) is important and should be placed as close to the pin of the device as is physically possible. This is also applied to the V_{DD} pin. Poor power supply decoupling can lead to a degradation in ADC offsets and SNR.

The Bypass pin should be decoupled to the ground plane using a 10 nF capacitor. Large capacitor values are not recommended as this can cause the reference not to reach its final value, on power up, before ADC autocalibration has commenced.

Capacitive loading of digital outputs should be minimized as much as possible if power dissipation is a critical factor. The charging and discharging of external load capacitances can be a significant contribution to power dissipation, especially when the AD7013 is in a low power sleep mode as the DxCLK remains active.

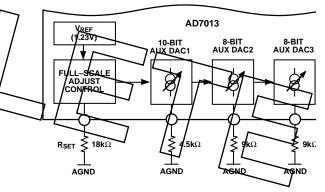


Figure 22. AUX DACs

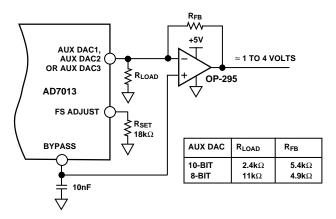
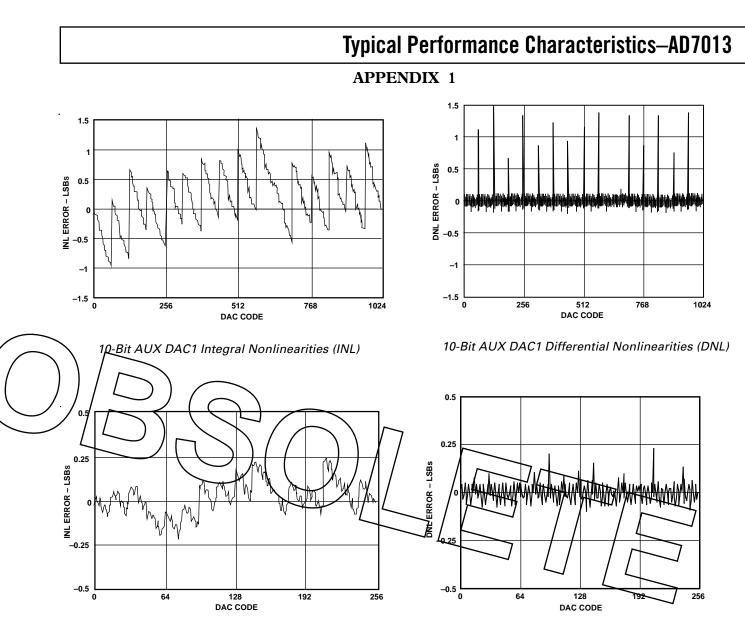
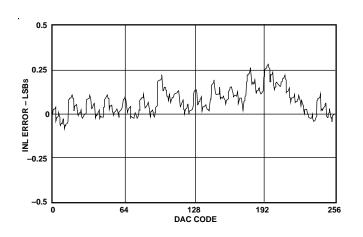


Figure 23. External Op Amp Circuitry to Extend Output Voltage Range

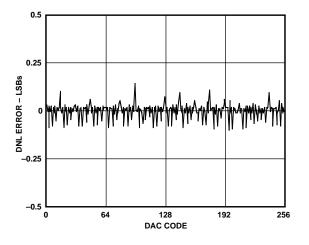


8-Bit AUX DAC2 Integral Nonlinearities (INL)

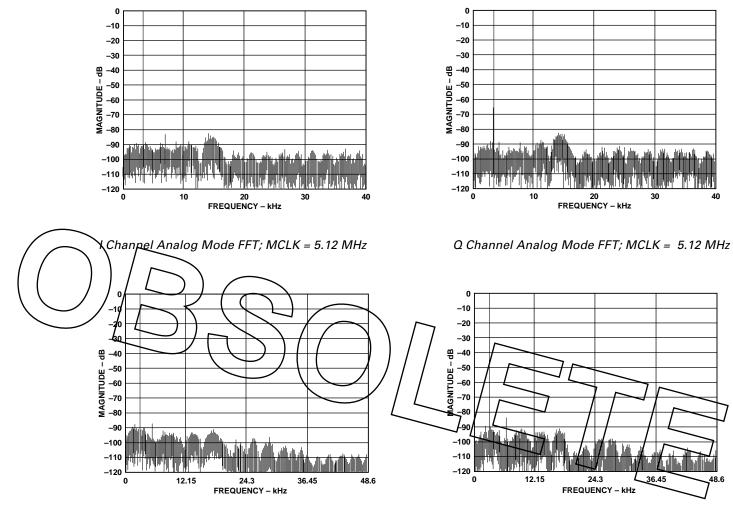


8-Bit AUX DAC3 Integral Nonlinearities (INL)

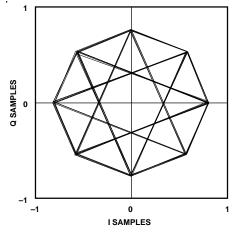
8-Bit AUX DAC2 Differential Nonlinearities (DNL)



8-Bit AUX DAC3 Differential Nonlinearities (DNL)

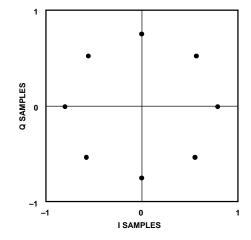


I Channel Digital Mode FFT; MCLK = 6.2208 MHz

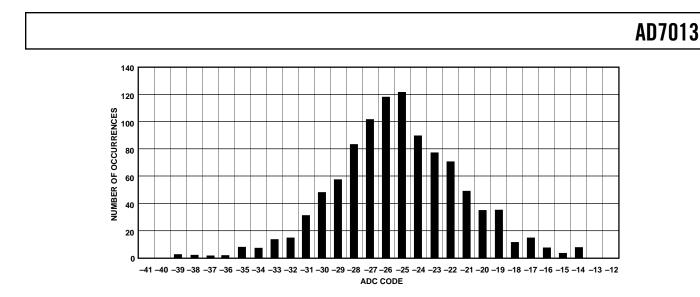


 $\pi/4$ DQPSK I and Q Receive Samples

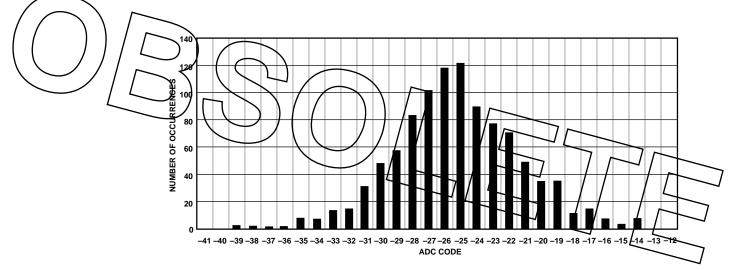
Q Channel Digital Mode FFT; MCLK = 6.2208 MHz



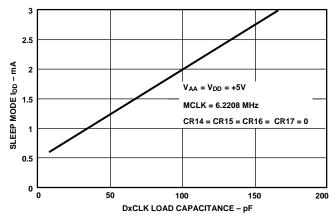
 $\pi/4$ DQPSK Constellation Diagram; Typical Error Vector 2% RMS



I Channel ADC Noise Histogram with IRx and \overline{IRx} Tied Together and Offset Register = 0; Number Codes = 1000, Standard Deviation = 4.44 Codes



Q Channel ADC Noise Histogram with QRx and \overline{QRx} Tied Together and Offset Register = 0; Number Codes = 1000, Standard Deviation = 3.82 Codes



AD7013 Sleep Current as a Function of DxCLK Load Capacitance

OUTLINE DIMENSIONS

Dimensions shown in inches and (mm).

