

Synchronous Demodulator and Configurable Analog Filter

FEATURES

Demodulates signal input bandwidths to 30 kHz Programmable filter enables variable bandwidths Filter tracks input carrier frequency Programmable reference clock frequency Flexible system interface Single-ended/differential signal inputs and outputs Rail-to-rail outputs directly drive analog-to-digital converters (ADCs) Phase detection sensitivity of 9.3m[°]θ_{REL} rms Configurable with 3-wire and 4-wire serial port interface (SPI) or seamless boot from I 2C EEPROMs Very low power operation 395 μA at fCLKIN = 500 kHz Single supply: 2.7 V to 3.6 V Specified temperature range: −40°C to +85°C 16-lead TSSOP package

APPLICATIONS

Synchronous demodulation Sensor signal conditioning Lock-in amplifiers Phase detectors Precision tunable filters Signal recovery Control systems

GENERAL DESCRIPTION

The $ADA2200$ is a sampled analog technology¹ synchronous demodulator for signal conditioning in industrial, medical, and communications applications. The [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) is an analog input, sampled analog output device. The signal processing is performed entirely in the analog domain by charge sharing among capacitors, which eliminates the effects of quantization noise and rounding errors. The [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) includes an analog domain, low-pass decimation filter, a programmable infinite impulse response (IIR) filter, and a mixer. This combination of features reduces ADC sample rates and lowers the downstream digital signal processing requirements.

Th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) acts as a precision filter when the demodulation function is disabled. The filter has a programmable bandwidth and tunable center frequency. The filter characteristics are highly stable over temperature, supply, and process variation.

Single-ended and differential signal interfaces are possible on both input and output terminals, simplifying the connection to other

¹ Patent pending.

Rev. 0 [Document Feedback](https://form.analog.com/Form_Pages/feedback/documentfeedback.aspx?doc=ADA2200.pdf&product=ADA2200&rev=0)

Information furnished by Analog Devices is believed to be accurate and reliable. However, no responsibility is assumed by Analog Devices for its use, nor for any infringements of patents or other rights of third parties that may result from its use. Specifications subject to change without notice. No license is granted by implication or otherwise under any patent or patent rights of Analog Devices. Trademarks and registered trademarks are the property of their respective owners.

Data Sheet **[ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf)**

components of the signal chain. The low power consumption and rail-to-rail operation is ideal for battery-powered and low voltage systems.

The [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) can be programmed over its SPI-compatible serial port or can automatically boot from the EEPROM through its I2 C interface. On-chip clock generation produces a mixing signal with a programmable frequency and phase. In addition, th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) synchronization output signal eases interfacing to other sampled systems, such as data converters and multiplexers.

The [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) is available in a 16-lead TSSOP package. Its performance is specified over the industrial temperature range of −40°C to +85°C. Note that throughout this data sheet, multifunction pins, such as SCLK/SCL, are referred to either by the entire pin name or by a single function of the pin, for example, SCLK, when only that function is relevant.

One Technology Way, P.O. Box 9106, Norwood, MA 02062-9106, U.S.A. Tel: 781.329.4700 ©2014 Analog Devices, Inc. All rights reserved. [Technical Support](http://www.analog.com/en/content/technical_support_page/fca.html) www.analog.com

TABLE OF CONTENTS

REVISION HISTORY

8/14-Revision 0: Initial Version

SPECIFICATIONS

 $V_{DD} = 3.3$ V, $V_{OCM} = V_{DD}/2$, $f_{CLKIN} = f_{SI} = 500$ kHz, default register configuration, differential input/output, R_L = 1 M Ω to GND, T_A = 25°C, unless otherwise noted.

Table 1.

¹ See th[e Terminology](#page-9-0) section.

² Common-mode signal swept from f_{MOD} − 1 kHz to f_{MOD} + 1 kHz. Output measured at frequency offset from f_{MOD}. For example, a common-mode signal at f_{MOD} − 500 Hz is measured at 500 Hz.

 3 The input impedance is equal to a 4 pF capacitor switched at f_{CLKIN}. Therefore, the input impedance = $10^{12}/(2\pi f_{CLKIN} \times 4)$.

SPI TIMING CHARACTERISTICS

V_{DD} = 2.7 V to 3.6 V, default register configuration, $T_A = -40$ to +85°C, unless otherwise noted.

Table 2. SPI Timing

Figure 3. SPI Write Timing Diagram (SPI Master Write to the [ADA2200\)](http://www.analog.com/ADA2200?doc=ADA2200.pdf)

Table 3. EEPROM Master FC Boot Timing					
Parameter ¹	Symbol	Min	Typical	Max	Unit
BOOT					
Load from BOOT Complete				9600	CLKIN cycles
RST to BOOT Setup Time	t ₂			2	CLKIN cycles
BOOT Pulse Width	t_3				CLKIN cycles
RESET					
Minimum RST Pulse Width	t1	25			ns
START CONDITION					
BOOT Low Transition to Start Condition	t ₄				CLKIN cycles

 T . **I.1.** α . **EEPROM Master** I^2CD **C Boot Timing**

¹ CLKIN cycles with CLKIN DIV[2:0] set to 000.

Table 4. Output, SYNCO, and RCLK Timing, Default Register Settings

Figure 6. Input, Output, SYNCO, and RCLK Timing Relative to CLKIN

ABSOLUTE MAXIMUM RATINGS

Table 5.

Stresses at or above those listed under Absolute Maximum Ratings may cause permanent damage to the product. This is a stress rating only; functional operation of the product at these or any other conditions above those indicated in the operational section of this specification is not implied. Operation beyond the maximum operating conditions for extended periods may affect product reliability.

THERMAL RESISTANCE

 θ_{IA} is specified for a device in a natural convection environment, soldered on a 4-layer JEDEC printed circuit board (PCB).

Table 6.

ESD CAUTION

ESD (electrostatic discharge) sensitive device. Charged devices and circuit boards can discharge without detection. Although this product features patented or proprietary protection circuitry, damage may occur on devices subjected to high energy ESD. Therefore, proper ESD precautions should be taken to avoid performance degradation or loss of functionality.

PIN CONFIGURATION AND FUNCTION DESCRIPTIONS

Table 7. Pin Function Descriptions

TYPICAL PERFORMANCE CHARACTERISTICS

Figure 8. Histogram of Device-to-Device Phase Delay Variation

Figure 9. Output Settling from SYNCO Falling Edge, 3.7 V Steps

Figure 10. Magnitude Measurement Error vs. Relative Phase

Figure 13. Phase Measurement Error vs. Relative Phase

TERMINOLOGY

Cycle Mean

The cycle mean is the average of all the output samples (OUTP/OUTN) over one RCLK period. In the default configuration, there are eight output samples per RCLK cycle; thus, the cycle mean is the average of eight consecutive output samples. If the device is reconfigured such that the frequency of RCLK is fso/4, then the cycle mean is the average of four consecutive output samples.

Conversion Gain

Conversion gain is calculated as follows:

$$
Conversion Gain = \frac{\sqrt{I^2 + Q^2}}{V_{IN}}
$$

where:

I is the offset corrected cycle mean, PHASE90 bit = 0. *Q* is the offset corrected cycle mean, PHASE90 bit = 1. *VIN* is the rms value of the input voltage.

The offset corrected cycle mean = cycle mean − output offset.

Relative Phase (θ_{REL})

Relative phase is the phase difference between the rising positive zero crossing of a sine wave at the INN/INP inputs relative to the next rising edge of RCLK.

Figure 14. Example Showing Relative Phase, θ*REL, of 37°*

Phase Delay ($^{\circ}$ θ_{DELAY})

The phase delay is the relative phase (θ_{REL}) that produces a zero cycle mean output value for a sine wave input with a frequency equal to fRCLK. The phase delay is the relative phase value that corresponds to the positive zero crossing of the phase measurement transfer function.

Phase Measurement Transfer Function

[Figure 15](#page-9-1) shows the cycle mean value of the output for a 1 V rms input sine wave as θ_{REL} is swept from 0° to 360°.

Figure 15. Phase Transfer Function with Phase Delay of 83°, 1 V rmsInput

THEORY OF OPERATION

The [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) is a synchronous demodulator and tunable filter implemented with sampled analog technology (SAT). Synchronous demodulators, also known as lock-in amplifiers, enable accurate measurement of small ac signals in the presence of noise interference orders of magnitude greater than the signal amplitude. Synchronous demodulators use phase sensitive detection to isolate the component of the signal at a specific reference frequency and phase. Noise at frequencies that are offset from the reference frequency are easily rejected and do not significantly impair the measurement.

SAT works on the principle of charge sharing. A sampled analog signal is a stepwise continuous signal without amplitude quantization. This contrasts with a signal sampled by an ADC, which becomes a discrete time signal with quantized amplitude.

With SAT, the input signal is sampled by holding the voltage on a capacitor at the sampling instant. Basic signal processing can then be performed in the analog domain by charge sharing among capacitors. Th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) includes an analog domain low-pass decimation filter, a programmable IIR filter, and a mixer. This combination of features enables reduced ADC sample rates and lowers the downstream digital signal processing requirements if the signal is digitized.

The output of th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) can also be used in an all analog signal path. In these applications, add a reconstruction filter following the [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) in the signal path.

SYNCHRONOUS DEMODULATION BASICS

Employing synchronous demodulation as a sensor signaling conditioning technique can result in improved sensitivity when compared to other methods. Synchronous demodulation adds two key benefits for recovering small sensor output signals in the presence of noise. The first benefit being the addition of an excitation signal, which enables the sensor output signal to be moved to a lower noise frequency band. The second benefit is that synchronous demodulation enables a simple low-pass filter to remove most of the remaining undesired noise components.

[Figure 16](#page-10-2) shows a basic synchronous demodulation system used for measuring the output of a sensor.

Figure 16. Basic Synchronous Demodulator Block Diagram

A carrier signal (f_{MOD}) excites the sensor. This shifts the signal generated by the physical parameter being measured by the sensor to the carrier frequency. This shift allows the desired signal to be placed in a frequency band with lower noise, improving the accuracy of the measurement. A band-pass filter (BPF) removes some of the out of band noise. A synchronous demodulator (or mixer) shifts the signal frequency back to dc. The last stage low-pass filter removes much of the remaining noise[. Figure 17](#page-10-3) an[d Figure 18](#page-10-4) show the frequency spectrum of the signal at different points in the synchronous demodulator.

Figure 17. Output Spectrum of Synchronous Demodulator Before Demodulation

Figure 18. Output Spectrum of Synchronous Demodulator After Demodulation

Phase Sensitive Detection

Synchronous demodulation uses the principle of phase sensitive detection to separate the signal of interest from unwanted signals. I[n Figure 16,](#page-10-2) the mixer performs the phase sensitive detection. The signal at the mixer output (C) is the product of the reference signal and a filtered version of the sensor output (B). If the reference signal is a sine wave, the physical parameter is a constant and there is no noise in the system. The signal at the output of the BPF is a sine wave that can be expressed as

 V_B sin(ω_{REF} *t* + φ_B)

The output of the mixer (if implemented as a multiplier) is then

 $\frac{1}{2}V_BV_{REF}cos(\phi_B - \phi_{REF}) - \frac{1}{2}V_BV_{REF}cos(2\omega_{REF} + \phi_B + \phi_{REF})$

This signal is a dc signal and an ac signal at twice the reference frequency. If the LPF is sufficient to remove the ac signal, the signal at the LPF output (D) is

 $\frac{1}{2}V_BV_{REF}cos(\varphi_B-\varphi_{REF})$

The LPF output is a dc signal that is proportional to both the magnitude and phase of the signal at the BPF output (B). When the input amplitude is held constant, the LPF output enables can be used to measure the phase. When the input phase is held constant, the LPF can be used to measure amplitude.

Note that the reference signal is not required to be a pure sine wave. The excitation signal and demodulation signal must only share a common frequency and phase to employ phase sensitive detection. In some applications, it may be possible to use the square wave output from th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) RCLK output directly.

Internal to th[e ADA2200,](http://www.analog.com/ADA2200?doc=ADA2200.pdf) the demodulation is performed not by multiplying the REFCLK signal with the input signal, but by holding the output constant for ½ the sample output periods. This operation is similar to a half wave demodulation of the input signal. For more information on signal detection using this function, see the [Applications](#page-15-0) Information section.

[ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) ARCHITECTURE

The signal path for the [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) consists of a high impedance input buffer followed by a fixed low-pass filter (FIR decimation filter), a programmable IIR filter, a mixer function, and a differential pin driver[. Figure 19](#page-11-2) shows a detailed block diagram of th[e ADA2200.](http://www.analog.com/ADA2200?doc=ADA2200.pdf) The signal processing blocks are all implemented using a charge sharing technique.

DECIMATION FILTER

The clock signal divider (after CLKIN) determines the input sampling frequency, f_{SI} , of the decimation filter. The decimation filter produces one filtered sample for every eight input samples. [Figure 20](#page-11-3) shows the wideband frequency response of the decimation filter. Because the filter operates on sampled data, images of the filter appear at multiples of the input sample rate, f_{SI}. The stop band of the decimation filter begins around $\frac{1}{2}$ of the output data rate, fso. Because an image pass band exists around f_{SI} , any undesired signals in the pass band around f_{SI} alias to dc and are indistinguishable from the low frequency input signal.

To preserve the full dynamic range of th[e ADA2200,](http://www.analog.com/ADA2200?doc=ADA2200.pdf) use an input antialiasing filter if noise at frequencies above 7.5 fs is not lower than the noise floor of the frequencies of interest. A firstorder low-pass filter is usually sufficient for the antialiasing filter.

Figure 20. Decimation Filter Frequency Response

[Figure 21](#page-11-4) shows a more narrow bandwidth view of the decimation transfer function. The stop band of the decimation filter starts at ½ of the output sample rate. The stop band rejection of the decimator low-pass filter is approximately 55 dB. The pass band of the decimation filter extends to $1/4th$ of the output sample rate or $1/32nd$ of the decimator input sample rate.

Figure 21. Decimation Filter Transfer Function, fsI = 800 kHz

IIR FILTER

The IIR block operates at the output sample rate, fso, which is at $1/8th$ of the input sample rate (f_{SI}). By default, the IIR filter is configured as a band-pass filter with a center frequency at fso/8 ($f_{SI}/64$). This frequency corresponds to the default mixing frequency and assures that input signals in the center of the pass band mix down to dc.

[Figure 22](#page-12-2) shows the default frequency response of the IIR filter.

Figure 22. Default IIR Filter Frequency Response (fso/8 BPF)

If a different frequency response is required, the IIR can be programmed for a different response. Register 0x0011 through Register 0x0027 contain coefficient values that program the filter response. To program the filter, first load the configuration registers (Register 0x0011 through Register 0x0027) with the desired coefficients. The coefficients can then be loaded into the filter by writing 0x03 to Register 0x0010.

The IIR filter can be configured for all pass operation by loading the coefficients listed i[n Table 8.](#page-12-3)

MIXER

The [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) performs the mixing function by holding the output samples constant for ½ of the RCLK period. This is similar to a half-wave rectification function except that the output does not return to zero for ½ the output period, but retains the value of the previous sample.

In the default configuration, there are eight output sample periods during each RCLK cycle. There are four updated output samples while the RCLK signal is high. While RCLK is low, the fourth updated sample is held constant for four additional output sample periods. The timing of the output samples in the default configuration is shown in [Table 4.](#page-5-0)

The RCLK divider, RCLK DIV[1:0], can be set to divide fso by 4. When this mode is selected, four output sample periods occur during each RCLK cycle. Two output samples occur while the RCLK signal is high. While RCLK is low, the second updated sample is held constant for two additional output sample periods.

The mixer can be bypassed. When the mixer is bypassed, the output produces an updated sample value every output sample period.

Phase Shifter

It is possible to change the timing of the output samples with respect to RCLK by writing to the PHASE90 bit in Register 0x002A. When the alternative timing option is selected, two output samples are updated while RCLK is low, and two are updated while RCLK is high. The second sample, which is taken while RCLK is high, is held four additional output sample periods. The timing is shown i[n Figure 5.](#page-5-1)

Applying a 90° phase shift can be useful in a number of instances. It enables a pair o[f ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) devices to perform in phase and quadrature demodulation. A 90° phase shift can also be useful in control systems for selecting an appropriate error signal output.

DS0-X 3034A, MY51340519; Tue Mar 18 23:36:39 2014 Agilent normai
250MSa/s To turn on cursors, press the [Cursors] key on the front panel. **(B)**

Figure 23. Output Sample Timing Relative to RCLK, (A) PHASE90 = 0, (B) PHASE90 = 1

CLOCKING OPTIONS

The [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) has several clocking options to make system integration easier.

Clock Dividers

The [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) has a pair of on-chip clock dividers to generate the system clocks. The input clock divider, CLKIN DIV[2:0], sets the input sample rate of the decimator (f_{SI}) by dividing the CLKIN signal. The value of CLKIN DIV[2:0] can be set to 1, 16, 64, or 256.

The output sample rate (f_{so}) is always $1/8th$ of the decimator input sample rate.

The RCLK divider, RCLK DIV[1:0], sets the frequency of the mixer frequency, f_M (which is also the frequency of RCLK) by dividing f_{SO} by either 4 or 8.

Synchronization Pulse Output

The [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) generates an output pulse (SYNCO), which can be used by a microprocessor or directly by an ADC to initiate an analog to digital conversion of th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) output. The SYNCO signal ensures that the ADC sampling occurs at an optimal time during the [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) output sample window.

One output sample of th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) is 8 fsI clock cycles long. The SYNCO pulse is 1 fsI clock cycle in duration. As shown in [Figure 24,](#page-13-1) the SYNCO pulse can be programmed to occur at 1 of 16 different timing offsets. The timing offsets are spaced at ½ f_{SI} clock cycle intervals and span the full output sample window.

The SYNCO pulse can be inverted, or the SYNCO output can be disabled. The operation of the SYNCO timing generation configuration settings are contained in Register 0x0029.

12295-024

205-2024

INPUT AND OUTPUT AMPLIFIERS

Single-Ended Configurations

If a single-ended input configuration is desired, the input signal must have a common-mode voltage near midsupply. Decouple the other inputs to the common-mode voltage of the input signal.

Note that differences between the common-mode levels between the INP and INN inputs result in an offset voltage inside the device. Even though the BPF removes the offset, minimize the offset to avoid reducing the available signal swing internal to the device.

For single-ended outputs, either OUTP or OUTN can be used. Leave the unused output floating.

Differential Configurations

Using th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) in differential mode utilizes the full dynamic range of the device and provides the best noise performance and common-mode rejection.

APPLICATIONS INFORMATION

The signal present at the output of the [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) depends on the amplitude and relative phase of the signal applied at it inputs. When the amplitude or phase is known and constant, any output variations can be attributed to the modulated parameter. Therefore, when the relative phase of the input is constant, the [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) performs amplitude demodulation. When the amplitude is constant, the [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) performs phase demodulation.

The sampling and demodulation processes introduce additional frequency components onto the output signal. If the output signal of th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) is used in the analog domain or if it is sampled asynchronously to th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) sample clock, these high frequency components can be removed by following the [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) with a reconstruction filter.

If th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) output is sampled synchronously to the [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) output sample rate, an analog reconstruction filter is not required because the ADC inherently rejects sampling artifacts. The frequency artifacts introduced by the demodulation process can be removed by digital filtering.

AMPLITUDE MEASUREMENTS

If the relative phase of the input signal to th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) remains constant, the output amplitude is directly proportional to the amplitude of the input signal. Note that the signal gain is a function of the relative phase of the input signal[. Figure 15](#page-9-1) shows the relationship between the cycle mean output and the relative phase. The cycle mean output voltage is

 $V_{\text{CYCLEMEAN}} = \text{Conversion Gain} \times V_{\text{IN(RMS)}} \times \sin(\theta_{\text{REL}} - \theta_{\text{DEL}}) =$

 $1.05 \times V_{IN(RMS)} \times \sin(\theta_{REL} - \theta_{DEL})$

Therefore, the highest gain, and thus the largest signal-to-noise ratio measurement, is obtained when operating th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) with $\theta_{\text{REL}} = \theta_{\text{DEL}} + 90^{\circ} = 173^{\circ}$. This value of θ_{REL} is also the operating point with the lowest sensitivity to changes in the relative phase. Operating with $\theta_{REL} = \theta_{DEL} - 90^{\circ} = -7^{\circ}$ offers the same gain and measurement accuracy, but with a sign inversion.

PHASE MEASUREMENTS

If the amplitude of the input signal to th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) remains constant, the output amplitude is a function of the relative phase of the input signal. The relative phase can be measured as

 $\theta_{\text{REL}} = \sin^{-1}(V_{\text{CYCLEMEAN}}/(\text{Conversion Gain} \times V_{\text{IN(RMS)}})) + \theta_{\text{DEL}} =$

$$
\sin^{-1}(V_{CYCLEMEAN}/(1.05\times V_{IN(RMS)})) + \theta_{DEL}
$$

Note that the output voltage scales directly with the input signal amplitude. A full-scale input signal provides the greatest phase sensitivity (V / Θ _{REL}) and thus the largest signal-to-noise ratio measurement.

The phase sensitivity also varies with relative phase. The sensitivity is at a maximum when $\theta_{\text{REL}} = 83^{\circ}$. For this reason, the optimal measurement range is for input signals with a relative phase equal to the phase delay of ±45°. This range provides the highest gain and thus the largest signal-to-noise ratio measurement. This range is also the operating point with the lowest sensitivity to changes in the relative phase. Operating at a relative phase equal to the phase delay of −135° to −225° offers the same gain and measurement accuracy, but with a sign inversion.

The phase sensitivity with a 4 V p-p differential input operating with a relative phase that is equal to the phase delay results in a phase sensitivity of 36.6 mV/ Θ REL.

AMPLITUDE AND PHASE MEASUREMENTS

When both the amplitude and relative phase of the input signals are unknown, it is necessary to obtain two orthogonal components of the signal to determine its amplitude, relative phase, or both. These two signal components are referred to as the in-phase (I) and quadrature (Q) components of the signal.

A signal with two known rectangular components is represented as a vector or phasor with an associated amplitude and phase (see [Figure 25\)](#page-15-4).

Figure 25. Rectangular and Polar Representation of a Signal

If the signal amplitude remains nearly constant for the duration of the measurement, it is possible to measure both the I and the Q components of the signal by toggling the PHASE90 bit between two consecutive measurements. To measure the I component, set the PHASE90 bit to 0. To measure the Q component, set the PHASE90 bit to 1.

After both the I and Q components have been obtained, it is possible to separate the effects of the amplitude and phase variations. Then, calculate the magnitude and relative phase using the following formulas:

$$
A = \sqrt{I^2 + Q^2}
$$

$$
\theta_{REL} = \cos^{-1}\left[\frac{Q}{A}\right] + \theta_{DEL}
$$

Or alternatively

$$
\theta_{REL} = \sin^{-1}\left[\frac{I}{A}\right] + \theta_{DEL}
$$

The inverse sine or inverse cosine functions linearize the relationship between the relative phase of the signal and the measured angle. Because the inverse sine and inverse cosine are only defined in two quadrants, the sign of I and Q must be considered to map the result over the entire 360° range of possible relative phase values. The use of the inverse tangent function is not recommended because the phase measurements become extremely sensitive to noise as the calculated phase approaches ±90°.

ANALOG OUTPUT SYSTEMS

When the output signal of th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) is used in the analog domain or if it is sampled asynchronously to th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) sample clock, it is likely that a reconstruction filter is required.

Reconstruction Filters

The bandwidth of the analog reconstruction filter sets the demodulation bandwidth of the analog output. There is a direct trade-off between the noise and demodulation bandwidth. Therefore, it is recommended to ensure that the reconstruction filter cutoff frequency is as low as possible while minimizing the attenuation of the demodulated signal of interest.

Similar to a digital-to-analog converter (DAC), the output of the [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) is a stepwise continuous output. This waveform contains positive and negative images of the desired signal at multiples of fso. In most cases, the images are undesired noise components that must be attenuated.

The lowest frequency image to appear in the output spectrum appears at a frequency of $f_{SO} - f_{IN}$. The image amplitude is reduced by the $sin(x)/x$ roll-off. System accuracy requirements may dictate that additional low-pass filtering is required to remove the output sample images.

INTERFACING TO ADCS *Settling Time Considerations*

If the ADC is coherently sampling th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) outputs, design the output filter to ensure that the output samples settle prior to ADC sampling. The output filter does not need to remove the sampling images generated by th[e ADA2200.](http://www.analog.com/ADA2200?doc=ADA2200.pdf) The images are inherently rejected by the ADC sampling process.

Clock Synchronization

The SYNCO output can trigger the ADC sampling process directly, or a microcontroller can use SYNCO to adjust the ADC sampling time. Adjusting the SYNCO pulse timing can maximize the available time for th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) outputs to settle prior to ADC sampling.

Multichannel ADCs

In multichannel systems that require simultaneous sampling, the [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) can provide per channel programmable filtering and simultaneous sampling.

[Figure 26](#page-16-3) shows an 8-channel system with a 1 MHz aggregate throughput rate. Th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) samples each channel at 1 MSPS and produces filtered samples at an output sample rate of 125 kHz each. Th[e AD7091R-8](http://www.analog.com/AD7091R-8?doc=ADA2200.pdf) is an 8-channel, 1 MHz ADC with multiplexed inputs, which cycle through the eight channels at 125 kHz, producing an aggregate output sample rate of 1 MHz.

Figure 26[. ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) in an 8-Channel Simultaneous Sampling Application

LOCK-IN AMPLIFIER APPLICATION

[Figure 27](#page-16-4) shows th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) in a lock-in amplifier application. The 80 kHz master clock signal sets the input sample rate of the decimation filter, f_{SI}. The output sample rate is 10 kHz. In the default configuration, the excitation signal generated by RCLK is 1.25 kHz. This is also the center frequency of the on-chip IIR filter.

In many cases, the RCLK signal is buffered to provide a square wave excitation signal to the sensor. It may also be desirable to provide further signal conditioning to provide a sine wave excitation signal to the sensor.

A low noise instrumentation amplifier provides sufficient gain to amplify the signal so that the noise floor of the signal into the [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) is above the combined noise floor of th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) and the ADC referred to the [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) inputs.

In default mode, the [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) produces eight output samples for every cycle of the excitation (RCLK) signal. There are four unique output sample values. The fourth value appears on the output for five consecutive output sample periods.

There are several ways of digitally processing the output samples to optimize measurement accuracy, bandwidth, and throughput rate. One method is to take the sum of eight samples to return a value. A moving average filter lowers the noise floor of the returned values. The length of the moving average filter is determined by the noise floor and settling time requirements.

INTERFACING TO MICROCONTROLLERS

The diagram i[n Figure 28](#page-17-3) shows basic circuit configuration driven by a low power microcontroller (th[e ADuCM361\)](http://www.analog.com/ADuCM361?doc=ADA2270.pdf). In this case, th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) reduces the ADC sampling rate by a factor of 8, and reduces the subsequent signal processing required by the microcontroller.

EEPROM BOOT CONFIGURATION

The diagram i[n Figure 29](#page-17-4) shows a standalone configuration with an EEPROM boot for the [ADA2200.](http://www.analog.com/ADA2200?doc=ADA2200.pdf) The standard oscillator circuit between CLKIN and XOUT generates the clock signal. Holding BOOT low during a power-on reset (POR) forces th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) to load its configuration from a preprogrammed EEPROM. An EEPROM boot is also initiated by bringing the BOOT pin low while the device in not in reset.

Figure 29. Standalone Configuration

POWER DISSIPATION

The [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) current draw is composed of two main components, the amplifier bias currents and the switched capacitor currents. The amplifier currents are independent of clock frequency; the switched capacitor currents scale in direct proportion to fsi.

[Figure 30](#page-17-5) shows th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) measured typical current draw at supply voltages of 2.7 V and 3.3 V, as the input clock varies from 1 kHz to 1 MHz, with CLKIN DIV $[2:0] = 1$. With a 3.3 V supply voltage, the current draw can be estimated with the following equation:

 $I_{DD} = 290 \times 0.2 \times f_{CLKIN} \mu A$

where f_{CLKIN} is specified in kHz.

Figure 30. Typical Current Draw vs. CLKIN Frequency at V_{DD} = 2.7 V and 3.3 V

DEVICE CONFIGURATION

The [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) has several registers that can be programmed to customize the device operation. There are two methods for programming the registers: the device can be programmed over the serial port interface, or the I^2C master can be used to read the configuration from a serial EEPROM.

SERIAL PORT OPERATION

The serial port is a flexible, synchronous serial communications port that allows easy interfacing to many industry-standard microcontrollers and microprocessors. The serial I/O is compatible with most synchronous transfer formats, including both the Motorola SPI and Intel® SSR protocols. The interface allows read/write access to all registers that configure th[e ADA2200.](http://www.analog.com/ADA2200?doc=ADA2200.pdf) Single-byte or multiple-byte transfers are supported, as well as MSB first or LSB first transfer formats. The serial port interface can be configured as a single-pin I/O (SDIO) or as two unidirectional pins for input and output (SDIO and SDO).

A communication cycle with th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) has two phases. Phase 1 is the instruction cycle (the writing of an instruction byte into the device), coincident with the first 16 SCLK rising edges. The instruction byte provides the serial port controller with information regarding the data transfer cycle—Phase 2 of the communication cycle. The Phase 1 instruction byte defines whether the upcoming data transfer is a read or write, along with the starting register address for the first byte of the data transfer. The first 16 SCLK rising edges of each communication cycle are used to write the instruction byte into the device.

A logic high on the CS/A0 pin followed by a logic low resets the serial port timing to the initial state of the instruction cycle. From this state, the next 16 rising SCLK edges represent the instruction bits of the current I/O operation.

The remaining SCLK edges are for Phase 2 of the communication cycle. Phase 2 is the actual data transfer between the device and the system controller. Phase 2 of the communication cycle is a transfer of one or more data bytes. Registers change immediately upon writing to the last bit of each transfer byte.

DATA FORMAT

The instruction byte contains the information shown i[n Table 9.](#page-18-5)

Table 9. Serial Port Instruction Byte

R/W, Bit 15 of the instruction byte, determines whether a read or a write data transfer occurs after the instruction byte write. Logic 1 indicates a read operation, and Logic 0 indicates a write operation.

A14 to A0, Bit 14 to Bit 0 of the instruction byte, determine the register that is accessed during the data transfer portion of the communication cycle. For multibyte transfers, A14 is the starting byte address. The remaining register addresses are generated by the device based on the LSB first bit (Register 0x0000, Bit 6).

SERIAL PORT PIN DESCRIPTIONS

Serial Clock (SCLK/SCL)

The serial clock pin synchronizes data to and from the device and runs the internal state machines. The maximum frequency of SCLK is 20 MHz. All data input is registered on the rising edge of the SCLK signal. All data is driven out on the falling edge of the SCLK signal

Chip Select (CS/A0)

An active low input starts and gates a communication cycle. It allows more than one device to be used on the same serial communications lines. When the $\overline{\text{CS}}$ /A0 pin is high, the SDO and SDIO signals go to a high impedance state. Keep the CS/A0 pin low throughout the entire communication cycle.

Serial Data I/O (SDIO/SDA)

Data is always written into the device on this pin. However, this pin can be used as a bidirectional data line. The configuration of this pin is controlled by Register 0x0000, Bit 3 and Bit 4. The default is Logic 0, configuring the SDIO/SDA pin as unidirectional.

Serial Data Output (RCLK/SDO)

If the [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) is configured for 4-wire SPI operation, this pin can be used as the serial data output pin. If the device is configured for 3-wire SPI operation, this pin can be used as an output for the reference clock (RCLK) signal. Setting the RCLK select bit (Register 0x002A, Bit 3) high activates the RCLK signal.

SERIAL PORT OPTIONS

The serial port can support both MSB first and LSB first data formats. This functionality is controlled by the LSB first bit (Register 0x0000, Bit 6). The default is MSB first (LSB first $= 0$).

When the LSB first bit $= 0$ (MSB first), the instruction and data bits must be written from MSB to LSB. Multibyte data transfers in MSB first format start with an instruction byte that includes the register address of the most significant data byte. Subsequent data bytes follow from high address to low address. In MSB first mode, the serial port internal byte address generator decrements for each data byte of the multibyte communication cycle.

When the LSB first $bit = 1$, the instruction and data bits must be written from LSB to MSB. Multibyte data transfers in LSB first format start with an instruction byte that includes the register address of the least significant data byte. Subsequent data bytes follow from the low address to the high address. In LSB first mode, the serial port internal byte address generator increments for each data byte of the multibyte communication cycle.

If the MSB first mode is active, the data address is decremented for each successive read or write operation performed in a multibyte register access. If the LSB first mode is active, the data address increments for each successive read or write operation performed in a multibyte register access.

BOOTING FROM EEPROM

The device can load the internal registers from the EEPROM using the internal I²C master to customize the operation of th[e ADA2200.](http://www.analog.com/ADA2200?doc=ADA2200.pdf) To enable this feature, the user must control either the RST pin or the BOOT pin. In either case, the device boots from the EEPROM only when it is out of reset and the master clock is active.

Enabling Load from Memory

A boot from the EEPROM is initiated by two methods.

To initiate loading via the \overline{BOOT} pin, the device must be out of reset, and the BOOT pin is brought low for a minimum of two clock cycles of the master clock. After it is initiated, the boot completes irrespective of the state of the BOOT pin. To initiate subsequent boots, the \overline{BOOT} pin must be brought high and then low for a minimum of two clock cycles of the master clock.

To initiate loading via the RST pin, the BOOT pin must be low. The RST pin can be tied high and th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) loads from the EEPROM when the device is powered up and the internal POR cycle completes. To initiate subsequent boots, th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) can be power cycled or the $\overline{\text{RST}}$ pin can be brought low and then high.

The SPI interface is disabled while the [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) is loading the EEPROM.

Load from Memory Cycle

The [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) reads the first 28 bytes of the EEPROM. The first 27 bytes represent the contents to be loaded into Register 0x0011 to Register 0x0027. Byte 28 contains the checksum stored in the EEPROM.

The [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) calculates the checksum for the first 27 bytes that it reads back and compares it to the checksum in the EEPROM. The [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) calculated checksum is accessible by reading the EEPROM checksum register (Register 0x002E). If th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) checksum matches the checksum stored in the EEPROM, the load from the EEPROM was successful. The load from the EEPROM pass or fail status is recorded in the EEPROM status register (Register 0x002F).

In addition, the LSB of the EEPROM status register indicates whether the load cycle is complete. Logic 1 represents successful completion of the load cycle. Logic 0 represents the occurrence of a timeout violation during the loading cycle. In the event of a timeout or the successful completion of the load from a memory cycle, the [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) I 2 C master interface disables, and th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) SPI interface reenables, allowing the user communication access to the device.

The load cycle completes within 10,000 clock cycles of CLKIN (or CLKIN divided by the current value of CLKIN DIV[2:0] if the load cycle is being initiated by the BOOT pin).

Dual Configuration/Dual Device Memory Load

The CS/A0 pin allows a single EEPROM device to support a dual configuration for a singl[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) device or different configurations for two differen[t ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) devices. To ensure reliable operation, set the $\overline{\text{CS}}$ /A0 pin to the desired state before initiating a boot, and then hold the state for the entire duration of the boot.

To configure a singl[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) device, the EEPROM must have a word page size that supports a minimum of 32 words, each of 8 bits per word. To support two devices, or a dual configuration for a single device, the EEPROM must have at least two word pages. The [ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) configuration data for each device must be allocated to the EEPROM memory within a single word page.

Using SPI Master with EEPROM Loading

The load from a memory cycle requires an $I²C$ communication bus between th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) and the EEPROM device; however, th[e ADA2200](http://www.analog.com/ADA2200?doc=ADA2200.pdf) can still be controlled by the SPI interface after the load from the memory cycle is complete. It is recommended that the CS/A0 pin return to logic high after the load from the memory cycle and before the first SPI read or write command. This allows the user to ensure that the proper setup time elapses before the initiation of a SPI read/write command (se[e Table 2\)](#page-3-2).

DEVICE CONFIGURATION REGISTER MAP AND DESCRIPTIONS

Table 10. Device Configuration Register Map1

¹ X means don't care.

² N/A means not applicable.

Table 11. Device Configuration Register Descriptions

Data Sheet **ADA2200**

¹ NA/ means not applicable.

² The filter coefficients listed are the default values programmed into the filter on reset. The value read back from the registers is 0x00.

Figure 33. Detailed Block Diagram

OUTLINE DIMENSIONS

COMPLIANT TO JEDEC STANDARDS MO-153-AB

Figure 34. 16-Lead Thin Shrink Small Outline Package [TSSOP] (RU-16) Dimensions shown in millimeters

ORDERING GUIDE

 $1 Z =$ RoHS-Compliant Part.

I 2 C refers to a communications protocol originally developed by Philips Semiconductors (now NXP Semiconductors).

www.analog.com

ООО "ЛайфЭлектроникс" "LifeElectronics" LLC

ИНН 7805602321 КПП 780501001 Р/С 40702810122510004610 ФАКБ "АБСОЛЮТ БАНК" (ЗАО) в г.Санкт-Петербурге К/С 30101810900000000703 БИК 044030703

 Компания «Life Electronics» занимается поставками электронных компонентов импортного и отечественного производства от производителей и со складов крупных дистрибьюторов Европы, Америки и Азии.

С конца 2013 года компания активно расширяет линейку поставок компонентов по направлению коаксиальный кабель, кварцевые генераторы и конденсаторы (керамические, пленочные, электролитические), за счёт заключения дистрибьюторских договоров

 Мы предлагаем:

- *Конкурентоспособные цены и скидки постоянным клиентам.*
- *Специальные условия для постоянных клиентов.*
- *Подбор аналогов.*
- *Поставку компонентов в любых объемах, удовлетворяющих вашим потребностям.*
- *Приемлемые сроки поставки, возможна ускоренная поставка.*
- *Доставку товара в любую точку России и стран СНГ.*
- *Комплексную поставку.*
- *Работу по проектам и поставку образцов.*
- *Формирование склада под заказчика.*
- *Сертификаты соответствия на поставляемую продукцию (по желанию клиента).*
- *Тестирование поставляемой продукции.*
- *Поставку компонентов, требующих военную и космическую приемку.*
- *Входной контроль качества.*
- *Наличие сертификата ISO.*

 В составе нашей компании организован Конструкторский отдел, призванный помогать разработчикам, и инженерам.

Конструкторский отдел помогает осуществить:

- *Регистрацию проекта у производителя компонентов.*
- *Техническую поддержку проекта.*
- *Защиту от снятия компонента с производства.*
- *Оценку стоимости проекта по компонентам.*
- *Изготовление тестовой платы монтаж и пусконаладочные работы.*

 Tел: +7 (812) 336 43 04 (многоканальный) Email: org@lifeelectronics.ru

www[.lifeelectronics.ru](http://lifeelectronics.ru/)