

# IF Diversity Receiver

# Data Sheet **[AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf)**

# <span id="page-0-0"></span>**FEATURES**

**SNR = 73.0 dBFS in a 95 MHz bandwidth at 185 MHz AIN and 245.76 MSPS SFDR = 85 dBc at 185 MHz AIN and 250 MSPS Noise density = −151.2 dBFS/Hz input at 185 MHz, −1 dBFS AIN and 250 MSPS Total power consumption: 1 W with fixed-frequency NCO, 95 MHz FIR filter 1.8 V supply voltages LVDS (ANSI-644 levels) outputs Integer 1-to-8 input clock divider (625 MHz maximum input) Integrated dual-channel ADC Sample rates of up to 250 MSPS IF sampling frequencies to 400 MHz Internal ADC voltage reference Flexible input range 1.4 V p-p to 2.1 V p-p (1.75 V p-p nominal) ADC clock duty cycle stabilizer 95 dB channel isolation/crosstalk Integrated wideband digital processor 32-bit complex numerically controlled oscillator (NCO) FIR filter with 2 modes Real output from an fs/4 output NCO Amplitude detect bits for efficient AGC implementation**

- **Energy saving power-down modes**
- <span id="page-0-3"></span>**Decimated, interleaved real LVDS data outputs**

# <span id="page-0-1"></span>**APPLICATIONS**

**Communications Diversity radio systems Multimode digital receivers (3G) TD-SCDMA, WiMax, WCDMA, CDMA2000, GSM, EDGE, LTE General-purpose software radios Broadband data applications**

# <span id="page-0-2"></span>**GENERAL DESCRIPTION**

Th[e AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) is a mixed-signal intermediate frequency (IF) receiver consisting of dual 14-bit, 250 MSPS ADCs and a wideband digital downconverter (DDC). The [AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) is designed to support communications applications, where low cost, small size, wide bandwidth, and versatility are desired.

The dual ADC cores feature a multistage, differential pipelined architecture with integrated output error correction logic. Each ADC features wide bandwidth inputs supporting a variety of user-selectable input ranges. An integrated voltage reference eases design considerations. A duty cycle stabilizer is provided to compensate for variations in the ADC clock duty cycle, allowing the converters to maintain excellent performance.



#### <span id="page-0-4"></span>**Rev. C [Document Feedback](https://form.analog.com/Form_Pages/feedback/documentfeedback.aspx?doc=AD6649.pdf&product=AD6649&rev=C)**

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# <span id="page-1-0"></span>**REVISION HISTORY**

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## $1/13$ -Rev. A to Rev. B







4/11-Revision 0: Initial Version

ADC data outputs are internally connected directly to the digital downconverter (DDC) of the receiver. The digital receiver has two channels and provides processing flexibility. Each receive channel has four cascaded signal processing stages: a 32-bit frequency translator (numerically controlled oscillator (NCO)), an optional sample rate converter, a fixed FIR filter, and an  $f_s/4$ fixed-frequency NCO.

In addition to the receiver DDC, th[e AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) has several functions that simplify the automatic gain control (AGC) function in the system receiver. The programmable threshold detector allows monitoring of the incoming signal power using the fast detect output bits of the ADC. If the input signal level exceeds the programmable threshold, the fast detect indicator goes high. Because this threshold indicator has low latency, the user can quickly turn down the system gain to avoid an overrange condition at the ADC input.

After digital processing, data is routed directly to the 14-bit output port. These outputs operate at ANSI or reduced swing LVDS signal levels.

Th[e AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) receiver digitizes a wide spectrum of IF frequencies. Each receiver is designed for simultaneous reception of the main channel and the diversity channel. This IF sampling architecture greatly reduces component cost and complexity compared with traditional analog techniques or less integrated digital methods. In diversity applications, the output data format is real due to the final NCO, which shifts the output center frequency to  $f_s/4$ .

Flexible power-down options allow significant power savings, when desired.

Programming for setup and control is accomplished using a 3-pin SPI-compatible serial interface.

Th[e AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) is available in a 64-lead LFCSP and is specified over the industrial temperature range of −40°C to +85°C. This product is protected by a U.S. patent.

# <span id="page-2-0"></span>**PRODUCT HIGHLIGHTS**

- 1. Integrated dual, 14-bit, 250 MSPS ADCs.
- 2. Integrated wideband filter and 32-bit complex NCO.
- 3. Fast overrange and threshold detect.
- 4. Proprietary differential input maintains excellent SNR performance for input frequencies of up to 400 MHz.
- 5. SYNC input allows synchronization of multiple devices.
- 6. 3-pin, 1.8 V SPI port for register programming and register readback.

# <span id="page-3-0"></span>**SPECIFICATIONS ADC DC SPECIFICATIONS**

<span id="page-3-1"></span> $AVDD = 1.8$  V, DRVDD = 1.8 V, maximum sample rate, VIN = -1.0 dBFS differential input,<sup>1</sup> 1.75 V p-p full-scale input range, duty cycle stabilizer (DCS) enabled, NCO enabled, FIR filter enabled, unless otherwise noted.



<sup>1</sup> A −1.0 dBFS input level at the analog inputs corresponds to an output level of −2.5 dBFS when using the fixed-frequency NCO and 95 MHz FIR filter. When using the tunable-frequency NCO and 100 MHz FIR filter, the output level is −1.3 dBFS. These respective output level reductions are due to FIR filter losses. See th[e FIR Filters](#page-23-0) section for more details.

<sup>2</sup> Input capacitance refers to the effective capacitance between one differential input pin and AGND.

<sup>3</sup> Input resistance refers to the effective resistance between one differential input pin and its complement.

<sup>4</sup> Measured with a 185 MHz, full-scale sine wave input on both channels and an NCO frequency of 62.5 MHz (fs/4).

<sup>5</sup> Standby power is measured with a dc input and the CLK pin inactive (set to AVDD or AGND).

# <span id="page-4-0"></span>**ADC AC SPECIFICATIONS**

 $AVDD = 1.8$  V, DRVDD = 1.8 V, maximum sample rate,  $VIN = -1.0$  dBFS differential input,<sup>1</sup> 1.75 V p-p full-scale input range, DCS enabled, NCO enabled, FIR filter enabled, unless otherwise noted.



<sup>1</sup> A −1.0 dBFS input level at the analog inputs corresponds to an output level of −2.5 dBFS when using the fixed-frequency NCO and 95 MHz FIR filter. When using the tunable-frequency NCO and 100 MHz FIR filter, the output level is −1.3 dBFS. These respective output level reductions are due to FIR filter losses. See th[e FIR Filters](#page-23-0) section for more details.

<sup>2</sup> See the AN-835 Application Note, *[Understanding High Speed ADC Testing and Evaluation](http://www.analog.com/AN-835?doc=AD6649.pdf)*, for a complete set of definitions.

<sup>3</sup> SNR specifications are for filtered 95 MHz bandwidth.

<sup>4</sup> Crosstalk is measured at 100 MHz with −1 dBFS on one channel and with no input on the alternate channel.

# <span id="page-5-0"></span>**DIGITAL SPECIFICATIONS**

AVDD = 1.8 V, DRVDD = 1.8 V, maximum sample rate, VIN = −1.0 dBFS differential input,<sup>1</sup> 1.0 V internal reference, DCS enabled, unless otherwise noted.



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<sup>1</sup> A −1.0 dBFS input level at the analog inputs corresponds to an output level of −2.5 dBFS when using the fixed-frequency NCO and 95 MHz FIR filter. When using the tunable-frequency NCO and 100 MHz FIR filter, the output level is −1.3 dBFS. These respective output level reductions are due to FIR filter losses. See th[e FIR Filters](#page-23-0) section for more details.

<sup>2</sup> Pull-up.

<sup>3</sup> Pull-down.

# <span id="page-7-0"></span>**SWITCHING SPECIFICATIONS**

# **Table 4.**



<sup>1</sup> Conversion rate is the clock rate after the divider.

# <span id="page-8-0"></span>**TIMING SPECIFICATIONS**

## <span id="page-8-2"></span>**Table 5.**



## *Timing Diagrams*

<span id="page-8-1"></span>

*Figure 2. Interleaved LVDS Mode Data Output Timing*



# <span id="page-9-0"></span>ABSOLUTE MAXIMUM RATINGS

#### **Table 6.**



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

# <span id="page-9-1"></span>**THERMAL CHARACTERISTICS**

The exposed paddle must be soldered to the ground plane for the LFCSP package. Soldering the exposed paddle to the customer board increases the reliability of the solder joints, maximizing the thermal capability of the package.

#### <span id="page-9-3"></span>**Table 7. Thermal Resistance**



<sup>1</sup> Per JEDEC 51-7, plus JEDEC 25-5 2S2P test board.

2 Per JEDEC JESD51-2 (still air) or JEDEC JESD51-6 (moving air). 3 Per MIL-Std 883, Method 1012.1. 4 Per JEDEC JESD51-8 (still air).

Typical  $\theta_{JA}$  is specified for a 4-layer PCB with solid ground plane. As shown in [Table 7,](#page-9-3) airflow increases heat dissipation, which reduces  $\theta_{JA}$ . In addition, metal in direct contact with the package leads from metal traces, through holes, ground, and

power planes, reduces the  $\theta_{JA}$ .

## <span id="page-9-2"></span>**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.

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<span id="page-10-0"></span>



Figure 4. LFCSP Interleaved Parallel LVDS Pin Configuration (Top View)

**Table 8. Pin Function Descriptions (Interleaved Parallel LVDS Mode)** 

**NOTES**





# <span id="page-12-0"></span>TYPICAL PERFORMANCE CHARACTERISTICS

AVDD = 1.8 V, DRVDD = 1.8 V, sample rate = 250 MSPS, DCS enabled, 1.75 V p-p differential input, VIN = −1.0 dBFS, 32k sample,  $T_A = 25^{\circ}$ C, fixed-frequency NCO, 95 MHz BW FIR filter, unless otherwise noted. In the FFT plots that follow, the location of the second and third harmonics is noted when they fall in the pass band of the filter. A −1.0 dBFS input level at the analog inputs corresponds to an output level of −2.5 dBFS when using the fixed-frequency NCO and 95 MHz FIR filter. When using the tunable-frequency NCO and 100 MHz FIR filter, the output level is −1.3 dBFS. These respective output level reductions are due to FIR filter losses. See the FIR Filters section for more details.









*Figure 10[. AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) Single-Tone FFT with f<sub>IN</sub> = 305.1 MHz* 

**0 10 20 30 40 50 60 70 80 90 100 110 120 FREQUENCY (MHz)**

**–140**



*Figure 11[. AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) Single-Tone SNR/SFDR vs. Input Amplitude (AIN) with fIN = 90.1 MHz*



*Figure 12[. AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) Single-Tone SNR/SFDR vs. Input Frequency (f<sub>IN</sub>)* 







*Figure 14[. AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) Two-Tone SFDR/IMD3 vs. Input Amplitude (AIN) with f<sub>IN1</sub>* = 184.12 MHz, f<sub>IN2</sub> = 187.12 MHz, f<sub>S</sub> = 250 MSPS



*Figure 15[. AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) Two-Tone FFT with fIN1 = 89.12 MHz, fIN2 = 92.12 MHz, fS = 250 MSPS*



*Figure 16[. AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) Two-Tone FFT with f<sub>IN1</sub> = 184.12 MHz*,  $f_{IN2} = 187.12 \text{ MHz}, f_S = 250 \text{ MSPS}$ 

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Figure 17[. AD6649 S](http://www.analog.com/AD6649?doc=AD6649.pdf)ingle-Tone SNR/SFDR vs. Sample Rate (f<sub>s</sub>) with  $f_{IN}$  = 90.1 MHz



Figure 18[. AD6649 G](http://www.analog.com/AD6649?doc=AD6649.pdf)rounded Input Histogram

# <span id="page-15-0"></span>EQUIVALENT CIRCUITS



*Figure 19. Equivalent Analog Input Circuit*



*Figure 20. Equivalent Clock Input Circuit* 



*Figure 21. Equivalent LVDS Output Circuit*



*Figure 22. Equivalent SDIO Circuit*



*Figure 23. Equivalent SCLK, PDWN, or OEB Input Circuit*



*Figure 24. Equivalent CSB Input Circuit* 



*Figure 25. Equivalent SYNC Input Circuit*

# <span id="page-16-0"></span>THEORY OF OPERATION

The [AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) has two analog input channels, two filter channels, and two digital output channels. The intermediate frequency (IF) input signal passes through several stages before appearing at the output port(s) as a filtered and optionally decimated digital signal.

The dual ADC design can be used for diversity reception of signals, where the ADCs operate identically on the same carrier but from two separate antennae. The ADCs can also be operated with independent analog inputs. The user can sample frequencies from dc to 300 MHz using appropriate low-pass or band-pass filtering at the ADC inputs with little loss in ADC performance. Operation to 400 MHz analog input is permitted but occurs at the expense of increased ADC noise and distortion.

Synchronization capability is provided to allow synchronized timing between multiple devices.

Programming and control of the [AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) are accomplished using a 3-pin SPI-compatible serial interface.

# <span id="page-16-1"></span>**ADC ARCHITECTURE**

The [AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) architecture consists of a dual front-end sampleand-hold circuit, followed by a pipelined switched-capacitor ADC. The quantized outputs from each stage are combined into a final 14-bit result in the digital correction logic. The pipelined architecture permits the first stage to operate on a new input sample and the remaining stages to operate on the preceding samples. Sampling occurs on the rising edge of the clock.

Each stage of the pipeline, excluding the last, consists of a low resolution flash ADC connected to a switched-capacitor digitalto-analog converter (DAC) and an interstage residue amplifier (MDAC). The MDAC magnifies the difference between the reconstructed DAC output and the flash input for the next stage in the pipeline. One bit of redundancy is used in each stage to facilitate digital correction of flash errors. The last stage simply consists of a flash ADC.

The input stage of each channel contains a differential sampling circuit that can be ac- or dc-coupled in differential or singleended modes. The output staging block aligns the data, corrects errors, and passes the data to the output buffers. The output buffers are powered from a separate supply, allowing digital output noise to be separated from the analog core. During power-down, the output buffers go into a high impedance state.

# <span id="page-16-2"></span>**ANALOG INPUT CONSIDERATIONS**

The analog input to the [AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) is a differential switchedcapacitor circuit that has been designed for optimum performance while processing a differential input signal.

The clock signal alternatively switches the input between sample mode and hold mode (see the configuration shown i[n Figure 26\)](#page-16-3). When the input is switched into sample mode, the signal source

must be capable of charging the sampling capacitors and settling within 1/2 clock cycle.

A small resistor in series with each input can help reduce the peak transient current required from the output stage of the driving source. A shunt capacitor can be placed across the inputs to provide dynamic charging currents. This passive network creates a low-pass filter at the ADC input; therefore, the precise values are dependent on the application.

In intermediate frequency (IF) undersampling applications, the shunt capacitors should be reduced. In combination with the driving source impedance, the shunt capacitors limit the input bandwidth. Refer to the [AN-742 Application Note,](http://www.analog.com/AN-742?doc=AD6649.pdf) *Frequency [Domain Response of Switched-Capacitor ADCs](http://www.analog.com/AN-742?doc=AD6649.pdf)*; the [AN-827](http://www.analog.com/AN-827?doc=AD4469.pdf)  Application Note, *[A Resonant Approach to Interfacing Amplifiers](http://www.analog.com/AN-827?doc=AD4469.pdf)  [to Switched-Capacitor ADCs](http://www.analog.com/AN-827?doc=AD4469.pdf)*; and the *Analog Dialogue* article, ["Transformer-Coupled Front-End for Wideband A/D Converters,"](http://www.analog.com/dialogue/transformer-coupled_front-end) for more information on this subject.





<span id="page-16-3"></span>For best dynamic performance, the source impedances driving VIN+ and VIN− should be matched, and the inputs should be differentially balanced.

### *Input Common Mode*

The analog inputs of the [AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) are not internally dc biased. In ac-coupled applications, the user must provide this bias externally. Setting the device so that  $V_{CM} = 0.5 \times AVDD$  (or 0.9 V) is recommended for optimum performance. An on-board common-mode voltage reference is included in the design and is available from the VCM pin. Using the VCM output to set the input common mode is recommended. Optimum performance is achieved when the common-mode voltage of the analog input is set by the VCM pin voltage (typically  $0.5 \times$  AVDD). The VCM pin must be decoupled to ground by a 0.1 µF capacitor, as described in th[e Applications Information](#page-38-0) section. This decoupling capacitor should be placed close to the pin to minimize the series resistance and inductance between the part and this capacitor.

### **Differential Input Configurations**

Optimum performance is achieved while driving th[e AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) in a differential input configuration. For baseband applications, the [AD8138,](http://www.analog.com/AD8138?doc=AD6649.pdf) [ADA4937-2,](http://www.analog.com/ADA4937-2?doc=AD6649.pdf) [ADA4938-2,](http://www.analog.com/ADA4938-2?doc=AD6649.pdf) an[d ADA4930-2](http://www.analog.com/ADA4930-2?doc=AD6649.pdf) differential drivers provide excellent performance and a flexible interface to the ADC.

The output common-mode voltage of the [ADA4930-2 i](http://www.analog.com/ADA4930-2?doc=AD6649.pdf)s easily set with the VCM pin of the [AD6649 \(](http://www.analog.com/AD6649?doc=AD6649.pdf)see [Figure 27\)](#page-17-0), and the driver can be configured in a Sallen-Key filter topology to provide band-limiting of the input signal.



Figure 27. Differential Input Configuration Using th[e ADA4930-2](http://www.analog.com/ADA4930-2?doc=AD6649.pdf)

<span id="page-17-0"></span>For baseband applications where SNR is a key parameter, differential transformer coupling is the recommended input configuration. An example is shown in [Figure 28.](#page-17-1) To bias the analog input, the VCM voltage can be connected to the center tap of the secondary winding of the transformer.



Figure 28. Differential Transformer-Coupled Configuration

<span id="page-17-1"></span>The signal characteristics must be considered when selecting a transformer. Most RF transformers saturate at frequencies below a few megahertz. Excessive signal power can also cause core saturation, which leads to distortion.

At input frequencies in the second Nyquist zone and above, the noise performance of most amplifiers is not adequate to achieve the true SNR performance of th[e AD6649.](http://www.analog.com/AD6649?doc=AD6649.pdf) For applications where SNR is a key parameter, differential double balun coupling is the recommended input configuration (see [Figure 30\)](#page-17-2). In this configuration, the input is ac-coupled and the CML is provided to each input through a 33  $\Omega$  resistor. These resistors compensate for losses in the input baluns to provide a 50  $\Omega$  impedance to the driver.

In the double balun and transformer configurations, the value of the input capacitors and resistors is dependent on the input frequency and source impedance. Based on these parameters the value of the input resistors and capacitors may need to be adjusted or some components may need to be removed. [Table 9 d](#page-17-3)isplays recommended values to set the RC network for different input frequency ranges. However, these values are dependent on the input signal and bandwidth and should be used only as a starting guide. Note that the values given i[n Table 9 a](#page-17-3)re for each R1, R2, C2, and R3 component shown i[n Figure 28](#page-17-1) and [Figure 30.](#page-17-2) 

#### <span id="page-17-3"></span>**Table 9. Example RC Network**



An alternative to using a transformer-coupled input at frequencies in the second Nyquist zone is to use an amplifier with variable gain. Th[e AD8375 o](http://www.analog.com/AD8375?doc=AD6649.pdf)r [AD8376 d](http://www.analog.com/AD8376?doc=AD6649.pdf)igital variable gain amplifier (DVGAs) provides good performance for driving th[e AD6649.](http://www.analog.com/AD6649?doc=AD6649.pdf)  [Figure 29 s](#page-17-4)hows an example of the [AD8376 d](http://www.analog.com/AD8376?doc=AD6649.pdf)riving the [AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) through a band-pass antialiasing filter.



**NOTES**

**1. ALL INDUCTORS ARE COILCRAFT® 0603CS COMPONENTS WITH THE EXCEPTION OF THE 1µH CHOKE INDUCTORS (COILCRAFT 0603LS). 2. FILTER VALUES SHOWN ARE FOR A 20MHz BANDWIDTH FILTER CENTERED AT 140MHz.**

<span id="page-17-4"></span>Figure 29. Differential Input Configuration Using th[e AD8376](http://www.analog.com/AD8376?doc=AD6649.pdf)

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<span id="page-17-2"></span>

### <span id="page-18-0"></span>**VOLTAGE REFERENCE**

A stable and accurate voltage reference is built into th[e AD6649.](http://www.analog.com/AD6649?doc=AD6649.pdf) The full-scale input range can be adjusted by varying the reference voltage via SPI. The input span of the ADC tracks reference voltage changes linearly.

## <span id="page-18-1"></span>**CLOCK INPUT CONSIDERATIONS**

For optimum performance, the [AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) sample clock inputs, CLK+ and CLK−, should be clocked with a differential signal. The signal is typically ac-coupled into the CLK+ and CLK− pins via a transformer or via capacitors. These pins are biased internally (see [Figure 31\)](#page-18-2) and require no external bias. If the inputs are floated, the CLK− pin is pulled low to prevent spurious clocking.





#### <span id="page-18-2"></span>**Clock Input Options**

The [AD6649 h](http://www.analog.com/AD6649?doc=AD6649.pdf)as a very flexible clock input structure. Clock input can be a CMOS, LVDS, LVPECL, or sine wave signal. Regardless of the type of signal being used, clock source jitter is of the most concern, as described in th[e Jitter Considerations](#page-19-1)  section.

[Figure 32 a](#page-18-3)n[d Figure 33 s](#page-18-4)how two preferable methods for clocking the [AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) (at clock rates of up to 625 MHz). A low jitter clock source is converted from a single-ended signal to a differential signal using an RF balun or RF transformer.

The RF balun configuration is recommended for clock frequencies between 125 MHz and 625 MHz, and the RF transformer is recommended for clock frequencies from 10 MHz to 200 MHz. The back-to-back Schottky diodes across the transformer secondary limit clock excursions into th[e AD6649 t](http://www.analog.com/AD6649?doc=AD6649.pdf)o approximately 0.8 V p-p differential. This limit helps prevent the large voltage swings of the clock from feeding through to other portions of th[e AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) while preserving the fast rise and fall times of the signal, which are critical to low jitter performance.



<span id="page-18-3"></span>Figure 32. Transformer-Coupled Differential Clock (Up to 200 MHz)



Figure 33. Balun-Coupled Differential Clock (Up to 625 MHz)

<span id="page-18-4"></span>If a low jitter clock source is not available, another option is to ac couple a differential PECL signal to the sample clock input pins as shown i[n Figure 34.](#page-18-5) Th[e AD9510,](http://www.analog.com/AD9510?doc=AD6649.pdf) [AD9511,](http://www.analog.com/AD9511?doc=AD6649.pdf) [AD9512,](http://www.analog.com/AD9512?doc=AD6649.pdf) [AD9513,](http://www.analog.com/AD9513?doc=AD6649.pdf)  [AD9514,](http://www.analog.com/AD9514?doc=AD6649.pdf) [AD9515,](http://www.analog.com/AD9515?doc=AD6649.pdf) [AD9516,](http://www.analog.com/AD9516?doc=AD6649.pdf) [AD9517,](http://www.analog.com/AD9517?doc=AD6649.pdf) [AD9518,](http://www.analog.com/AD9518?doc=AD6649.pdf) [AD9520,](http://www.analog.com/AD9520?doc=AD6649.pdf) [AD9522,](http://www.analog.com/AD9522?doc=AD6649.pdf) [AD9523,](http://www.analog.com/AD9523?doc=AD6649.pdf) [AD9524,](http://www.analog.com/AD9524?doc=AD6649.pdf) an[d ADCLK905/](http://www.analog.com/ADCLK905?doc=AD6649.pdf)[ADCLK907/](http://www.analog.com/ADCLK907?doc=AD6649.pdf)[ADCLK925 c](http://www.analog.com/ADCLK925?doc=AD6649.pdf)lock drivers offer excellent jitter performance.



Figure 34. Differential PECL Sample Clock (Up to 625 MHz)

<span id="page-18-5"></span>A third option is to ac-couple a differential LVDS signal to the sample clock input pins, as shown i[n Figure 35.](#page-18-6) Th[e AD9510,](http://www.analog.com/AD9510?doc=AD6649.pdf) [AD9511,](http://www.analog.com/AD9511?doc=AD6649.pdf) [AD9512,](http://www.analog.com/AD9512?doc=AD6649.pdf) [AD9513,](http://www.analog.com/AD9513?doc=AD6649.pdf) [AD9514,](http://www.analog.com/AD9514?doc=AD6649.pdf) [AD9515,](http://www.analog.com/AD9515?doc=AD6649.pdf) [AD9516,](http://www.analog.com/AD9516?doc=AD6649.pdf) [AD9517,](http://www.analog.com/AD9517?doc=AD6649.pdf)  [AD9518,](http://www.analog.com/AD9518?doc=AD6649.pdf) [AD9520,](http://www.analog.com/AD9520?doc=AD6649.pdf) [AD9522,](http://www.analog.com/AD9522?doc=AD6649.pdf) [AD9523,](http://www.analog.com/AD9523?doc=AD6649.pdf) an[d AD9524 c](http://www.analog.com/AD9524?doc=AD6649.pdf)lock drivers offer excellent jitter performance.



#### <span id="page-18-6"></span>**Input Clock Divider**

The [AD6649 c](http://www.analog.com/AD6649?doc=AD6649.pdf)ontains an input clock divider with the ability to divide the input clock by integer values between 1 and 8. The duty cycle stabilizer (DCS) is enabled by default on power-up.

Th[e AD6649 c](http://www.analog.com/AD6649?doc=AD6649.pdf)lock divider can be synchronized using the external SYNC input. Bit 1 and Bit 2 of Register 0x3A allow the clock divider to be resynchronized on every SYNC signal or only on the first SYNC signal after the register is written. A valid SYNC causes the clock divider to reset to its initial state. This synchronization feature allows multiple parts to have their clock dividers aligned to guarantee simultaneous input sampling.

# *Clock Duty Cycle*

Typical high speed ADCs use both clock edges to generate a variety of internal timing signals and, as a result, may be sensitive to clock duty cycle. Commonly, a ±5% tolerance is required on the clock duty cycle to maintain dynamic performance characteristics.

The [AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) contains a duty cycle stabilizer (DCS) that retimes the nonsampling (falling) edge, providing an internal clock signal with a nominal 50% duty cycle. This allows the user to provide a wide range of clock input duty cycles without affecting the performance of the [AD6649.](http://www.analog.com/AD6649?doc=AD6649.pdf) 

Jitter on the rising edge of the input clock is still of paramount concern and is not reduced by the duty cycle stabilizer. The duty cycle control loop does not function for clock rates less than 40 MHz nominally. The loop has a time constant associated with it that must be considered when the clock rate can change dynamically. A wait time of 1.5 µs to 5 µs is required after a dynamic clock frequency increase or decrease before the DCS loop is relocked to the input signal. During the time period that the loop is not locked, the DCS loop is bypassed, and internal device timing is dependent on the duty cycle of the input clock signal. In such applications, it may be appropriate to disable the duty cycle stabilizer. In all other applications, enabling the DCS circuit is recommended to maximize ac performance.

## <span id="page-19-1"></span>*Jitter Considerations*

High speed, high resolution ADCs are sensitive to the quality of the clock input. The degradation in SNR at a given input frequency ( $f_{IN}$ ) due to jitter ( $t_J$ ) can be calculated by

 $SNR_{HF} = -10 \log[(2\pi \times f_{IN} \times t_{JRMS})^2 + 10^{(-SNR_{LF}/10)}]$ 

In the equation, the rms aperture jitter represents the rootmean-square of all jitter sources, which include the clock input, the analog input signal, and the ADC aperture jitter specification. IF undersampling applications are particularly sensitive to jitter, as shown i[n Figure 36.](#page-19-2)



*Figure 36. SNR (95 MHz BW) vs. Input Frequency and Jitter*

<span id="page-19-2"></span>The clock input should be treated as an analog signal in cases where aperture jitter may affect the dynamic range of the AD6649. Power supplies for clock drivers should be separated from the

ADC output driver supplies to avoid modulating the clock signal with digital noise. Low jitter, crystal controlled oscillators make the best clock sources. If the clock is generated from another type of source (by gating, dividing, or another method), it should be retimed by the original clock at the last step.

Refer to th[e AN-501 Application Note,](http://www.analog.com/AN-501?doc=AD6649.pdf) *Aperture Uncertainty [and ADC System Performance](http://www.analog.com/AN-501?doc=AD6649.pdf)*, and the [AN-756 Application](http://www.analog.com/AN-756?doc=AD6649.pdf)  Note, *[Sampled Systems and the Effects of Clock Phase Noise and](http://www.analog.com/AN-756?doc=AD6649.pdf)  [Jitter](http://www.analog.com/AN-756?doc=AD6649.pdf)*, for more information about jitter performance as it relates to ADCs.

# <span id="page-19-0"></span>**POWER DISSIPATION AND STANDBY MODE**

As shown in [Figure 37,](#page-19-3) the power dissipated by th[e AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) is proportional to its sample rate. The data in [Figure 37](#page-19-3) was taken using the same operating conditions as those used for the [Typical](#page-12-0)  [Performance Characteristics.](#page-12-0)





<span id="page-19-3"></span>By asserting PDWN (either through the SPI port or by asserting the PDWN pin high), th[e AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) is placed in power-down mode. In this state, the ADC typically dissipates 10 mW. During power-down, the output drivers are placed in a high impedance state. Asserting the PDWN pin low returns the [AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) to its normal operating mode. Note that PDWN is referenced to the digital output driver supply (DRVDD) and should not exceed that supply voltage.

Low power dissipation in power-down mode is achieved by shutting down the reference, reference buffer, biasing networks, and clock. Internal capacitors are discharged when entering power-down mode and then must be recharged when returning to normal operation. As a result, wake-up time is related to the time spent in power-down mode, and shorter power-down cycles result in proportionally shorter wake-up times.

When using the SPI port interface, the user can place the ADC in power-down mode or standby mode. Standby mode allows the user to keep the internal reference circuitry powered when faster wake-up times are required. See the [Memory Map Register](#page-35-0)  [Description](#page-35-0) section and th[e AN-877 Application Note,](http://www.analog.com/AN-877?doc=AD6649.pdf) *Interfacing [to High Speed ADCs via SPI](http://www.analog.com/AN-877?doc=AD6649.pdf)*, for additional details.

# <span id="page-20-0"></span>**DIGITAL OUTPUTS**

The [AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) output drivers can be configured for either ANSI LVDS or reduced drive LVDS using a 1.8 V DRVDD supply.

As detailed in the [AN-877 Application Note,](http://www.analog.com/AN-877?doc=AD6649.pdf) *Interfacing to High [Speed ADCs via SPI](http://www.analog.com/AN-877?doc=AD6649.pdf)*, the data format can be selected for offset binary, twos complement, or gray code when using the SPI control.

## *Digital Output Enable Function (OEB)*

The [AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) has a flexible three-state ability for the digital output pins. The three-state mode is enabled using the OEB pin or through the SPI interface. If the OEB pin is low, the output data drivers are enabled. If the OEB pin is high, the output data drivers are placed in a high impedance state. This OEB function is not intended for rapid access to the data bus. Note that OEB is referenced to the digital output driver supply (DRVDD) and should not exceed that supply voltage.

When using the SPI interface, the data and fast detect outputs of each channel can be independently three-stated by using the output enable bar bit (Bit 4) in Register 0x14. Because the output data is interleaved, if only one of the two channels is disabled, the data of the remaining channel is repeated in both the rising and falling output clock cycles.

## *Timing*

Th[e AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) provides latched data with a pipeline delay of 23 or 43 input sample clock cycles, depending on the mode of operation. Data outputs are available one propagation delay  $(t_{PD})$  after the rising edge of the clock signal.

The length of the output data lines and loads placed on them should be minimized to reduce transients within th[e AD6649.](http://www.analog.com/AD6649?doc=AD6649.pdf) These transients can degrade converter dynamic performance.

The lowest typical conversion rate of th[e AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) is 40 MSPS. At clock rates below 40 MSPS, dynamic performance may degrade.

## *Data Clock Output (DCO)*

The [AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) also provides data clock output (DCO) intended for capturing the data in an external register[. Figure 2](#page-8-1) shows a graphical timing diagram of th[e AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) output modes.

# <span id="page-20-1"></span>**OVERRANGE (OR)**

The overrange indicator is asserted when an overrange is detected on the input of th[e AD6649.](http://www.analog.com/AD6649?doc=AD6649.pdf) The overrange condition is determined at the output of the pipeline ADC and, therefore, is subject to a latency of 10 ADC clocks. An overrange at the input is indicated by this bit, 10 clock cycles after it occurs.

#### **Table 10. Output Data Format**



# <span id="page-21-0"></span>DIGITAL PROCESSING

The [AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) includes a digital processing section that provides filtering. This digital processing section includes a numerically controlled oscillator (NCO), a selectable FIR filter (high performance or low latency), and a second coarse NCO (fs/4 fixed value) for output frequency translation (complex to real). These blocks can be configured in several modes to implement a signal processing function. Refer to [Figure 1 f](#page-0-4)or the functional block diagram of th[e AD6649.](http://www.analog.com/AD6649?doc=AD6649.pdf) 

# <span id="page-21-1"></span>**NUMERICALLY CONTROLLED OSCILLATOR (NCO)**

Frequency translation is accomplished with an NCO shared between the two channels. Amplitude and phase dither can be enabled on chip to improve the noise and spurious performance of the NCO.

Because the filtering prevents usage of part of the Nyquist spectrum, a means is needed to translate the sampled input spectrum into the usable range of the decimation filter. To achieve this, a 32-bit, tuning, complex NCO is provided. This NCO/mixer allows the input spectrum to be tuned to dc, where it can be effectively filtered by the subsequent filter blocks to prevent aliasing.

When using the low latency FIR, the NCO must be tuned to fs/4 (0x40000000). This prevents unwanted aliases from falling back into the band of interest.

# <span id="page-21-2"></span>**NCO AND FIR FILTER MODES**

The NCO and FIR blocks can be used in two modes depending on the bandwidth and latency requirement of the application. The two modes of operation of these blocks are summarized i[n Table 11.](#page-21-4) 

### <span id="page-21-4"></span>**Table 11. Signal Path Modes**



Two fixed-coefficient FIR filters provide filtering capability. A low latency FIR or a high performance FIR can be selected. It removes the negative frequency images to avoid aliasing negative frequencies for real outputs. [Figure 38,](#page-21-5) [Figure 39,](#page-21-6) an[d Figure 40](#page-21-7)  show the progression of a 95 MHz bandwidth signal through the filter stages when using the fixed-frequency NCO and 95 MHz FIR filter with a sample rate of 245.76 MSPS. The tunable-frequency NCO can be used instead and operates in a similar fashion. In these modes, the output is centered at 61.44 MHz, assuming a 245.76 MSPS sample rate.

# <span id="page-21-3"></span>**f<sub>s</sub>/4 FIXED-FREQUENCY NCO**

A fixed-frequency fs/4 NCO is provided to translate the filtered, decimated signal from dc to fs/4 to allow a real output. The fs/4 NCO is required in all operation modes because complex output from the part is not supported.

<span id="page-21-5"></span>

<span id="page-21-7"></span><span id="page-21-6"></span>**13.940 61.44 108.94 122.88** 09635-247 Figure 40. Exampl[e AD6649 9](http://www.analog.com/AD6649?doc=AD6649.pdf)5 MHz Bandwidth Output Signal Tuned to fs/4 (NCO Frequency = 61.44 MHz)

# <span id="page-22-0"></span>NUMERICALLY CONTROLLED OSCILLATOR (NCO)

# <span id="page-22-1"></span>**FREQUENCY TRANSLATION**

This processing stage comprises a digital tuner consisting of a 32-bit complex numerically controlled oscillator (NCO). The NCO is always enabled. This NCO block accepts a real input from the ADC stage and outputs a frequency translated complex (I and Q) output.

The NCO frequency is programmed in Register 0x52 through Register 0x55. These four 8-bit registers make up a 32-bit unsigned frequency programming word. Frequencies between −CLK/2 and +CLK/2 are represented using the following frequency words:

- 0x80000000 represents a frequency given by −CLK/2.
- $0x00000000$  represents dc (frequency = 0 Hz).
- 0x7FFFFFFF represents CLK/2 − CLK/232.

Use the following equation to calculate the NCO frequency:

$$
NCO\_FREQ = 2^{32} \times \frac{Mod(f, f_{CLK})}{f_{CLK}}
$$

where:

*NCO\_FREQ* is a 32-bit twos complement number representing the NCO frequency register.

*f* is the desired carrier frequency in hertz.

*f<sub>CLK</sub>* is the [AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) ADC clock rate in hertz.

# <span id="page-22-2"></span>**NCO SYNCHRONIZATION**

The [AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) NCOs within a single part or across multiple parts can be synchronized using the external SYNC input. Bit 0 and Bit 1 of Register 0x58 allow the NCO to be resynchronized on every SYNC signal or only on the first SYNC signal after the register is written. A valid SYNC causes the NCO to restart at the programmed phase offset value.

# <span id="page-22-3"></span>**NCO AMPLITUDE AND PHASE DITHER**

The NCO block contains amplitude and phase dither to improve the spurious performance. Amplitude dither improves performance by randomizing the amplitude quantization errors within the angular-to-Cartesian conversion of the NCO. This option reduces spurs at the expense of a slightly raised noise floor. With amplitude dither enabled, the NCO has an SNR of greater than 93 dB and an SFDR of greater than 115 dB. With amplitude dither disabled, the SNR is increased to greater than 96 dB at the cost of SFDR performance, which is reduced to 100 dB. The NCO amplitude and phase dither are recommended and can be enabled by setting Bit 1 and Bit 2 in Register 0x51.

# <span id="page-23-0"></span>FIR FILTERS

The two FIR filters that can be used are either a 47-tap, high performance, fixed-coefficient FIR filter or a 21-tap, low latency, fixed-coefficient FIR filter. These filters are useful in providing alias protection at the device output. The high performance FIR is a simple sum-of-products FIR filter with 47 filter taps and 21-bit fixed coefficients. Note that this filter does not decimate. The normalized coefficients used in the implementation and the decimal equivalent value of the coefficients are listed for the low latency FIR i[n Table 12 a](#page-23-3)nd the high performance FIR i[n Table 13.](#page-23-4)



<span id="page-23-3"></span>

<span id="page-23-4"></span>

# <span id="page-23-1"></span>**FIR SYNCHRONIZATION**

Th[e AD6649 f](http://www.analog.com/AD6649?doc=AD6649.pdf)ilters within a single part or across multiple parts can be synchronized using the external SYNC input. The filters can be configured to be resynchronized on every SYNC signal or only on the first SYNC signal after the SPI control register is written. A valid SYNC causes the FIR filter to restart at the programmed decimation phase value. Bit 4 and Bit 5 of Register 0x58 allow the FIR to be resynchronized on every SYNC signal or only on the first SYNC signal after the register is written.

# <span id="page-23-2"></span>**FILTER PERFORMANCE**

When using the fixed-frequency NCO and a 95 MHz FIR filter, the output rate is equal to the sample clock rate. The composite response of this mode is shown i[n Figure 41.](#page-23-5) The detailed passband response for this mode is shown in [Figure 42.](#page-24-1) To place the part in this mode, set SPI Register 0x50 to 0xB0. When operating in this mode, the NCO must be placed at f<sub>s</sub>/4, and the low latency NCO select bit (Bit 0) in Register 0x5A must be set. It is important to note that a −1.0 dBFS input level at the analog inputs corresponds to an output level of −2.5 dBFS when using the low latency FIR filter. This output level reduction is a result of the −1.5 dB passband attenuation in the FIR filter in this mode and does not result in loss in the dynamic range of the converter.



<span id="page-23-5"></span>Figure 41. Low Latency FIR Filter Composite Response at 245.76 MSPS (Fixed-Frequency NCO, 95 MHz FIR Filter Mode)

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<span id="page-24-1"></span>*Figure 42. Low Latency FIR Filter Pass-Band Response at 245.76 MSPS (Fixed-Frequency NCO, 95 MHz FIR Filter Mode)*

When using the tunable-frequency NCO and 100 MHz FIR filter, the output rate is equal to the sample clock rate. The response of the high performance FIR filter is shown in [Figure 43.](#page-24-2) The detailed pass-band response for this mode is shown in [Figure 44.](#page-24-3) To place the part into this mode, set SPI Register 0x50 to 0xA0. When using the high performance FIR filter, the output level is −1.3 dBFS for a corresponding input level of −1.0 dBFS at the analog inputs. This is a result of the −0.3 dB pass-band attenuation of the FIR filter in this mode and does not result in loss in the dynamic range of the converter.



<span id="page-24-2"></span>*Figure 43. High Performance FIR Filter Pass-Band Response at 245.76 MSPS (Tunable-Frequency NCO, 100 MHz FIR Filter)*



<span id="page-24-3"></span>*Figure 44. High Performance FIR Filter Pass-Band Response at 245.76 MSPS (Tunable-Frequency NCO, 100 MHz FIR Filter)*

# <span id="page-24-0"></span>**OUTPUT NCO**

The output of the 32-bit fine-tuning NCO is complex and typically centered in frequency around dc. This complex output is carried through the stages of either the 95 MHz or 100 MHz FIR filter to provide proper antialiasing filtering. The final NCO provides a means to move this complex output signal away from dc so that a real output can be provided from the [AD6649.](http://www.analog.com/AD6649?doc=AD6649.pdf) The output NCO translates the output from dc to a frequency equal to the output frequency divided by  $4(f_s/4)$ . This provides the user with an output signal centered at fs/4 in frequency.

Th[e AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) output NCOs within a single part or across multiple parts can be synchronized using the external SYNC input. Bit 7 and Bit 6 of Register 0x58 allow the output NCO to be resynchronized on every SYNC signal or only on the first SYNC signal after the register is written.

# <span id="page-25-0"></span>ADC OVERRANGE AND GAIN CONTROL

In receiver applications, it is desirable to have a mechanism to reliably determine when the converter is about to be clipped. The standard overflow indicator provides delayed information on the state of the analog input that is of limited value in preventing clipping. Therefore, it is helpful to have a programmable threshold below full scale that allowstime to reduce the gain before the clip occurs. In addition, because input signals can have significant slew rates, latency of this function is of concern.

Using the SPI port, the user can provide a threshold above which the FD output is active. Bit 0 of SPI Register 0x45 allows the user to select the threshold level. As long as the signal is below the selected threshold, the FD output remains low. In this mode, the magnitude of the data is considered in the calculation of the condition, but the sign of the data is not considered. The threshold detection responds identically to positive and negative signals outside the desired range (magnitude).

# <span id="page-25-1"></span>**ADC OVERRANGE (OR)**

The ADC overrange indicator is asserted when an overrange is detected on the input of the ADC. The overrange condition is determined at the output of the ADC pipeline and, therefore, is subject to a latency of 7 ADC clock cycles. An overrange at the input is indicated by this bit 7 clock cycles after it occurs.

# <span id="page-25-2"></span>**GAIN SWITCHING**

The [AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) includes circuitry that is useful in applications either where large dynamic ranges exist or where gain ranging amplifiers are employed. This circuitry allows digital thresholds to be set such that an upper threshold and a lower threshold can be programmed.

One such use is to detect when an ADC is about to reach full scale with a particular input condition. The result is to provide an indicator that can be used to quickly insert an attenuator that prevents ADC overdrive.

# *Fast Threshold Detection (FDA and FDB)*

The FD indicator is asserted if the input magnitude exceeds the value programmed in the fast detect upper threshold register, located in Register 0x47 and Register 0x48. The selected threshold register is compared with the signal magnitude at the output of the ADC. The fast upper threshold detection has a latency of 4 clock cycles. The upper threshold magnitude is defined by the following equation:

*Upper Threshold Magnitude* (dBFS) = 20 log(*Threshold Magnitude*/213 )

The FD indicators are not cleared until the signal drops below the lower threshold for the programmed dwell time. The lower threshold is programmed in the fast detect lower threshold register, located at Register 0x49 and Register 0x4A. The fast detect lower threshold register is a 15-bit register that is compared with the signal magnitude at the output of the ADC. This comparison is subject to the ADC pipeline latency but is accurate in terms of converter resolution. The lower threshold magnitude is defined by the following equation:

*Lower Threshold Magnitude* (dBFS) = 20 log(*Threshold Magnitude*/213 )

The dwell time can be programmed from 1 to 65,535 sample clock cycles by placing the desired value in the fast detect dwell time register, located in Register 0x4B and Register 0x4C.

The operation of the upper threshold and lower threshold registers, along with the dwell time, is shown i[n Figure 45.](#page-25-3)

<span id="page-25-3"></span>

*Figure 45. Threshold Settings for FDA and FDB Signals*

# <span id="page-26-0"></span>DC CORRECTION

Because the dc offset of the ADC may be significantly larger than the signal being measured, a dc correction circuit is included to null the dc offset before measuring the power. The dc correction circuit can also be switched into the main signal path, but this may not be appropriate if the ADC is digitizing a time-varying signal with significant dc content, such as GSM.

# *DC Correction Bandwidth*

The dc correction circuit is a high-pass filter with a programmable bandwidth (ranging between 0.29 Hz and 2.387 kHz at 245.76 MSPS). The bandwidth is controlled by writing the 4-bit dc correction bandwidth select register, located at Register 0x40, Bits[5:2]. The following equation can be used to compute the bandwidth value for the dc correction circuit:

$$
DC\_Corr\_BW = 2^{-k-14} \times \frac{f_{CLK}}{2 \times \pi}
$$

where:

*k* is the 4-bit value programmed in Bits[5:2] of Register 0x40 (values between 0 and 13 are valid for k; programming 14 or 15 provides the same result as programming 13). *fCLK* is the [AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) ADC sample rate in hertz.

# *DC Correction Readback*

The current dc correction value can be read back in Register 0x41 and Register 0x42 for each channel. The dc correction value is a 16-bit value that can span the entire input range of the ADC.

# *DC Correction Freeze*

Setting Bit 6 of Register 0x40 freezes the DC correction at its current state and continues to use the last updated value as the dc correction value. Clearing this bit restarts dc correction and adds the currently calculated value to the data.

# *DC Correction Enable Bits*

Setting Bit 1 of Register 0x40 enables dc correction for use in the output data signal path.

# <span id="page-27-0"></span>CHANNEL/CHIP SYNCHRONIZATION

The [AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) has a SYNC input that allows the user flexible synchronization options for synchronizing the internal blocks. The SYNC feature is useful for guaranteeing synchronized operation across multiple ADCs. The input clock divider, NCO, FIR filters, and the output fs/4 NCO can be synchronized using the SYNC input. Each of these blocks can be enabled to synchronize on a single occurrence of the SYNC signal or on every occurrence by setting the appropriate bits in Register 0x58.

The SYNC input is internally synchronized to the sample clock. However, to ensure that there is no timing uncertainty between multiple parts, the SYNC input signal should be synchronized to the input clock signal. The SYNC input should be driven using a single-ended CMOS type signal.

If Bit 1 in Register 0x59 is used, the SYNC input can be set to either level or edge sensitive mode. If the SYNC input is set to edge sensitive mode, Bit 0 of Register 0x59 can be used to determine whether the rising or falling edge is used. The settings written to Register 0x59 apply only to the FIR filters and the NCOs.

# <span id="page-28-0"></span>SERIAL PORT INTERFACE (SPI)

Th[e AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) serial port interface (SPI) allows the user to configure the converter for specific functions or operations through a structured register space provided inside the ADC. The SPI gives the user added flexibility and customization, depending on the application. Addresses are accessed via the serial port and can be written to or read from via the port. Memory is organized into bytes that can be further divided into fields. These fields are documented in th[e Memory Map](#page-30-0) section. For detailed operational information, see th[e AN-877 Application Note,](http://www.analog.com/AN-877?doc=AD6649.pdf) *Interfacing to [High Speed ADCs via SPI](http://www.analog.com/AN-877?doc=AD6649.pdf)*.

# <span id="page-28-1"></span>**CONFIGURATION USING THE SPI**

Three pins define the SPI of this ADC: the SCLK pin, the SDIO pin, and the CSB pin (se[e Table 14\)](#page-28-3). The SCLK (serial clock) pin is used to synchronize the read and write data presented from/to the ADC. The SDIO (serial data input/output) pin is a dual-purpose pin that allows data to be sent and read from the internal ADC memory map registers. The CSB (chip select bar) pin is an active low control that enables or disables the read and write cycles.

#### <span id="page-28-3"></span>**Table 14. Serial Port Interface Pins**



The falling edge of the CSB, in conjunction with the rising edge of the SCLK, determines the start of the framing. An example of the serial timing and its definitions can be found in [Figure 46](#page-29-2) an[d Table 5.](#page-8-2) 

Other modes involving the CSB are available. The CSB can be held low indefinitely, which permanently enables the device; this is called streaming. The CSB can stall high between bytes to allow for additional external timing. When CSB is tied high, SPI functions are placed in a high impedance mode. This mode turns on any SPI pin secondary functions.

During an instruction phase, a 16-bit instruction is transmitted. Data follows the instruction phase, and its length is determined by the W0 and W1 bits.

All data is composed of 8-bit words. The first bit of each individual byte of serial data indicates whether a read or write command is issued. This allows the serial data input/output (SDIO) pin to change direction from an input to an output.

In addition to word length, the instruction phase determines whether the serial frame is a read or write operation, allowing the serial port to be used both to program the chip and to read the contents of the on-chip memory. If the instruction is a readback operation, performing a readback causes the serial data input/ output (SDIO) pin to change direction from an input to an output at the appropriate point in the serial frame.

Data can be sent in MSB first mode or in LSB first mode. MSB first is the default on power-up and can be changed via the SPI port configuration register. For more information about this and other features, see the [AN-877 Application Note,](http://www.analog.com/AN-877?doc=AD6649.pdf) *Interfacing [to High Speed ADCs via SPI](http://www.analog.com/AN-877?doc=AD6649.pdf)*.

# <span id="page-28-2"></span>**HARDWARE INTERFACE**

The pins described in [Table 14](#page-28-3) comprise the physical interface between the user programming device and the serial port of the [AD6649.](http://www.analog.com/AD6649?doc=AD6649.pdf) The SCLK pin and the CSB pin function as inputs when using the SPI interface. The SDIO pin is bidirectional, functioning as an input during write phases and as an output during readback.

The SPI interface is flexible enough to be controlled by either FPGAs or microcontrollers. One method for SPI configuration is described in detail in the [AN-812 Application Note,](http://www.analog.com/AN-812?doc=AD6649.pdf) *Micro[controller-Based Serial Port Interface \(SPI\) Boot Circuit](http://www.analog.com/AN-812?doc=AD6649.pdf)*.

The SPI port should not be active during periods when the full dynamic performance of the converter is required. Because the SCLK signal, the CSB signal, and the SDIO signal are typically asynchronous to the ADC clock, noise from these signals can degrade converter performance. If the on-board SPI bus is used for other devices, it may be necessary to provide buffers between this bus and th[e AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) to prevent these signals from transitioning at the converter inputs during critical sampling periods.

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# <span id="page-29-0"></span>**SPI ACCESSIBLE FEATURES**

[Table 15](#page-29-1) provides a brief description of the general features that are accessible via the SPI. These features are described in detail in the AN-877 Application Note, *[Interfacing to High Speed ADCs](http://www.analog.com/AN-877?doc=AD6649.pdf)  [via SPI](http://www.analog.com/AN-877?doc=AD6649.pdf)*. The [AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) part-specific features are described in the [Memory Map Register Description](#page-35-0) section.

#### <span id="page-29-1"></span>**Table 15. Features Accessible Using the SPI**



<span id="page-29-2"></span>

*Figure 46. Serial Port Interface Timing Diagram*

# <span id="page-30-0"></span>MEMORY MAP **READING THE MEMORY MAP REGISTER TABLE**

<span id="page-30-1"></span>Each row in the memory map register table has eight bit locations. The memory map is roughly divided into four sections: the chip configuration registers (Address 0x00 to Address 0x02); the channel index and transfer registers (Address 0x05 and Address 0xFF); the ADC functions registers, including setup, control, and test (Address 0x08 to Address 0x3A); and the digital feature control registers (Address 0x40 to Address 0x5A).

The memory map register table (se[e Table](#page-31-1) 16) documents the default hexadecimal value for each hexadecimal address shown. The column with the heading Bit 7 (MSB) is the start of the default hexadecimal value given. For example, Address 0x14, the output mode register, has a hexadecimal default value of 0x05. This means that Bit  $0 = 1$  and the remaining bits are 0s. This setting is the default output format value, which is twos complement. For more information on this function and others, see the [AN-877 Application Note,](http://www.analog.com/AN-877?doc=AD6649.pdf) *Interfacing to High Speed [ADCs via SPI](http://www.analog.com/AN-877?doc=AD6649.pdf)*. This document details the functions controlled by Register 0x00 to Register 0x20. The remaining registers, from Register 0x3A to Register 0x5A, are documented in the [Memory](#page-35-0)  [Map Register Description](#page-35-0) section.

## *Open and Reserved Locations*

All address and bit locations that are not included in [Table](#page-31-1) 16 are not currently supported for this device. Unused bits of a valid address location should be written with 0s. Writing to these locations is required only when part of an address location is open (for example, Address 0x18). If the entire address location is open (for example, Address 0x13), this address location should not be written.

# *Default Values*

After the [AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) is reset, critical registers are loaded with default values. The default values for the registers are given in the memory map register table[, Table](#page-31-1) 16.

# *Logic Levels*

An explanation of logic level terminology follows:

- "Bit is set" is synonymous with "bit is set to Logic 1" or "writing Logic 1 for the bit."
- "Clear a bit" is synonymous with "bit is set to Logic 0" or "writing Logic 0 for the bit."

# *Transfer Register Map*

Address 0x08 to Address 0x20, Address 0x3A, Address 0x40 to Address 0x42, Address 0x45 to 0x4C, and Address 0x50 to Address 0x5A are shadowed. Writes to these addresses do not affect part operation until a transfer command is issued by writing 0x01 to Address 0xFF, setting the transfer bit. This allows these registers to be updated internally and simultaneously when the transfer bit is set. The internal update takes place when the transfer bit is set, and then the bit autoclears.

# *Channel-Specific Registers*

Some channel setup functions, such as the signal monitor thresholds, can be programmed to a different value for each channel. In these cases, channel address locations are internally duplicated for each channel. These registers and bits are designated in [Table](#page-31-1) 16 as local. These local registers and bits can be accessed by setting the appropriate Channel A or Channel B bits in Register 0x05. If both bits are set, the subsequent write affects the registers of both channels. In a read cycle, only Channel A or Channel B should be set to read one of the two registers. If both bits are set during an SPI read cycle, the part returns the value for Channel A. Registers and bits designated as global in [Table](#page-31-1) 16 affect the entire part and the channel features for which independent settings are not allowed between channels. The settings in Register 0x05 do not affect the global registers and bits.

# <span id="page-31-0"></span>**MEMORY MAP REGISTER TABLE**

All address and bit locations that are not included in [Table](#page-31-1) 16 are not currently supported for this device.

# <span id="page-31-1"></span>**Table 16. Memory Map Registers**



# Data Sheet **AD6649**



# AD6649 Data Sheet



# <span id="page-34-0"></span>Data Sheet **AD6649**



<sup>1</sup> The channel index register at Address 0x05 should be set to 0x03 (default) when writing to Address 0x00.

# <span id="page-35-0"></span>**MEMORY MAP REGISTER DESCRIPTION**

For more information on functions controlled in Register 0x00 to Register 0x20, see th[e AN-877 Application Note,](http://www.analog.com/AN-877?doc=AD6649.pdf) *Interfacing [to High Speed ADCs via SPI](http://www.analog.com/AN-877?doc=AD6649.pdf)*.

## *Sync Control (Register 0x3A)*

**Bits[7:3]—Reserved**

### **Bit 2—Clock Divider Next Sync Only**

If the master sync buffer enable bit (Address 0x3A, Bit 0) and the clock divider sync enable bit (Address 0x3A, Bit 1) are high, Bit 2 allows the clock divider to sync to the first sync pulse that it receives and to ignore the rest. The clock divider sync enable bit (Address 0x3A, Bit 1) resets after it syncs.

### **Bit 1—Clock Divider Sync Enable**

Bit 1 gates the sync pulse to the clock divider. The sync signal is enabled when Bit 1 is high and Bit 0 is high. This is continuous sync mode.

## **Bit 0—Master Sync Buffer Enable**

Bit 0 must be set high to enable any of the sync functions. If the sync capability is not used, this bit should remain low to conserve power.

## *DC Correction Control (Register 0x40)*

**Bit 7—Reserved**

### **Bit 6—DC Correction Freeze**

When Bit 6 is set high, the dc correction is no longer updated to the signal monitor block, which holds the last dc value calculated.

# **Bits[5:2]—DC Correction Bandwidth Select**

Bits[5:2] set the averaging time of the signal monitor dc correction function. This 4-bit word sets the bandwidth of the correction block, according to the following equation:

$$
DC\_Corr\_BW = 2^{-k-14} \times \frac{f_{CLK}}{2 \times \pi}
$$

where:

*k* is the 4-bit value programmed in Bits[5:2] of Register 0x40 (values between 0 and 13 are valid for k; programming 14 or 15 provides the same result as programming 13). *f<sub>CLK</sub>* is the [AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) ADC sample rate in hertz.

### **Bit 1—DC Correction Enable**

Setting this bit high causes the output of the dc measurement block to be summed with the data in the signal path to remove the dc offset from the signal path.

### **Bit 0—Reserved**

# *Fast Detect Control (Register 0x45)*

**Bits[7:4]—Reserved**

## **Bit 3—Force FD Output Enable**

Setting this bit high forces the FD output pin to the value written to Bit 2 of this register (Register 0x45). This enables the user to force a known value on the FD pin for debugging.

### **Bit 2—Force FD Output Value**

The value written to Bit 2 is forced on the FD output pin when Bit 3 is written high.

#### **Bit 1—Reserved**

### **Bit 0—Enable Fast Detect Output**

Setting this bit high enables the output of the upper threshold FD comparator to drive the FD output pin.

## *Fast Detect Upper Threshold (Register 0x47 and Register 0x48)*

**Register 0x48, Bits[7:5]—Reserved**

### **Register 0x48, Bits[4:0]—Fast Detect Upper Threshold[12:8]**

### **Register 0x47, Bits[7:0]—Fast Detect Upper Threshold[7:0]**

These registers provide an upper limit threshold. The 13-bit value is compared with the output magnitude from the ADC block. If the ADC magnitude exceeds this threshold value, the FD output pin is set if Bit 0 in Register 0x45 is set.

# *Fast Detect Lower Threshold (Register 0x49and Register 0x4A)*

**Register 0x4A, Bits[7:5]—Reserved**

**Register 0x4A, Bits[4:0]—Fast Detect Lower Threshold[12:8]**

### **Register 0x49, Bits[7:0]—Fast Detect Lower Threshold[7:0]**

These registers provide a lower limit threshold. The 13-bit value is compared with the output magnitude from the ADC block. If the ADC magnitude is less than this threshold value for the number of cycles programmed in the dwell time register, the FD output bit is cleared.

# *Fast Detect Dwell Time (Register 0x4B and Register 0x4C)*

### **Register 0x4C, Bits[7:0]—Fast Detect Dwell Time[15:8]**

### **Register 0x4B, Bits[7:0]—Fast Detect Dwell Time[7:0]**

These register values set the minimum time in ADC sample clock cycles (after clock divider) that a signal needs to stay below the lower threshold limit before the FD output bits are cleared.

### *Filter Control (Register 0x50)*

**Bit 7—Reserved (Reads Back as 1)**

**Bit 6—Reserved**

**Bit 5—Reserved (Reads Back as 1)**

### **Bit 4—FIR Mode**

Setting this bit low enables the high performance FIR filter. Setting this bit high enables the low latency FIR.

### **Bit 3—Output Gain**

Setting this bit high sets the output gain to −6 dB. A 0 value on this bit sets the gain at 0 dB.

## **Bit 2—9-bit Output Mode Enable**

If this bit is set, the NCOs and filters are bypassed and the part outputs nine bits of data. These nine bits are presented on the nine MSBs of the output bus (that is, Bit D13 through Bit D5).

## **Bits[1:0]—Datapath Gain**

These bits set the datapath gain as follows:

 $00 = 0$  dB gain

01 = −6 dB gain

 $10 = -12$  dB gain

11 = −18 dB gain

## *NCO Control (Register 0x51)*

#### **Bit 7—Reserved**

## Bit 6-NCO32 to f<sub>S</sub>/4 NCO Sync Enable

This bit should be set high when NCO32 is set to fs/4 using the fixed-frequency NCO and the 95 MHz FIR filter. It should be disabled when using the tunable-frequency NCO and 100 MHz FIR filter.

## **Bit 5—Spectral Reversal**

This bit should be set high to reverse the output frequency spectrum.

### **Bit 4—Reserved (Reads Back as 1)**

**Bit 3—Reserved**

### **Bit 2—NCO32 Amplitude Dither Enable**

When Bit 2 is set, amplitude dither in the NCO is enabled. When Bit 2 is cleared, amplitude dither is disabled.

### **Bit 1—NCO32 Phase Dither Enable**

When Bit 2 is set, phase dither in the NCO is enabled. When Bit 2 is cleared, phase dither is disabled.

### **Bit 0—Reserved (Reads Back as 1)**

### *NCO Frequency (Register 0x52 to Register 0x55)*

```
Register 0x52, Bits[7:0]—NCO Frequency Value[31:24]
```

```
Register 0x53, Bits[7:0]—NCO Frequency Value[23:16]
```

```
Register 0x54, Bits[7:0]—NCO Frequency Value[15:8]
```
### **Register 0x55, Bits[7:0]—NCO Frequency Value[7:0]**

This 32-bit value is used to program the NCO tuning frequency. The frequency value to be programmed is given by the following equation:

$$
NCO\_FREQ = 2^{32} \times \frac{Mod(f, f_{CLK})}{f_{CLK}}
$$

### where:

*NCO\_FREQ* is a 32-bit twos complement number representing the NCO frequency register.

*f* is the desired carrier frequency in hertz. *f<sub>CLK</sub>* is the [AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) ADC clock rate in hertz.

# *NCO Phase Offset (Register 0x56 and Register 0x57)*

**Register 0x56, Bits[7:0]—NCO Phase Value[15:8]**

# **Register 0x57, Bits[7:0]—NCO Phase Value[7:0]**

The 16-bit value programmed into the NCO phase value register is loaded into the NCO block each time the NCO is started or when an NCO SYNC signal is received. This process allows the NCO to be started with a known nonzero phase.

Use the following equation to calculate the NCO phase offset value:

### $NCO$  *PHASE* =  $2^{16}$   $\times$  *PHASE*/360

where *NCO* PHASE is a decimal number equal to the 16-bit binary number to be programmed at Register 0x56 and Register 0x57, and *PHASE* is the desired NCO phase in degrees.

### *SYNC Control (Register 0x58)*

### Bit 7-f<sub>S</sub>/4 NCO Next Sync Only

If the master sync buffer enable bit (Register 0x3A, Bit 0) and the f<sub>S</sub>/4 NCO sync enable bit (Register 0x58, Bit 6) are high, Bit 7 allows the fs/4 NCO to synchronize following the first sync pulse that it receives and ignore the rest. If Bit 7 is set, Bit 6 of Register 0x58 resets after this sync occurs.

## Bit 6-f<sub>S</sub>/4 NCO Sync Enable

Bit 6 gates the sync pulse to the fs/4 NCO. When Bit 6 is set high, the sync signal causes the  $f_s/4$  NCO to synchronize. This sync is active only when the master sync buffer enable bit (Register 0x3A, Bit 0) is high. This is continuous sync mode.

### **Bit 5—FIR Next Sync Only**

If the master sync buffer enable bit (Register 0x3A, Bit 0) and the FIR sync enable bit (Register 0x58, Bit 4) are high, Bit 5 allows the FIR to synchronize following the first sync pulse that it receives and to ignore the rest. If Bit 5 is set, Bit 4 of Register 0x3A resets after this sync occurs.

### **Bit 4—FIR Sync Enable**

Bit 4 gates the sync pulse to the FIR filter. When Bit 4 is set high, the sync signal causes the half-band to resynchronize. This sync is active only when the master sync buffer enable bit (Register 0x3A, Bit 0) is high. This is continuous sync mode.

### **Bits[3:2]—Reserved**

### **Bit 1—NCO32 Next Sync Only**

If the master sync buffer enable bit (Register 0x3A, Bit 0) and the NCO32 sync enable bit (Register 0x58, Bit 0) are high, Bit 1 allows the NCO32 to synchronize following the first sync pulse that it receives and to ignore the rest. Bit 0 of Register 0x58 resets after a sync occurs if Bit 1 is set.

### **Bit 0—NCO32 Sync Enable**

Bit 0 gates the sync pulse to the 32-bit NCO. When this bit is set high, the sync signal causes the NCO to resynchronize, starting at the NCO phase offset value. This sync is active only when the master sync buffer enable bit (Register 0x3A, Bit 0) is high. This is continuous sync mode.

## *NCO/FIR SYNC Pin Control (Register 0x59)*

**Bits[7:2]—Reserved**

### **Bit 1—SYNC Pin Sensitivity**

If Bit 1 is set to a 0, the SYNC input responds to a level. If this bit is set low, the SYNC input responds to the edge (rising or falling) set in Bit 0 of Address 0x59.

## **Bit 0—SYNC Pin Edge Sensitivity**

If Bit 1 is set high, setting Bit 0 to a 0 causes the SYNC input to respond to a falling edge. If this bit is set, the SYNC input respond to a rising edge.

## *NCO Control 2 (Register 0x5A)*

## **Bits[7:1]—Reserved**

## **Bit 0—Low Latency NCO Select**

If Bit 0 is set to a 1, the low latency NCO is selected. This bit should be selected for the fixed-frequency NCO, 95 MHz FIR filter mode of operation. When this bit is set, the NCO value must be set to either 0x40000000 or 0xC0000000.

# <span id="page-38-0"></span>APPLICATIONS INFORMATION

# <span id="page-38-1"></span>**DESIGN GUIDELINES**

Before starting system level design and layout of the [AD6649,](http://www.analog.com/AD6649?doc=AD6649.pdf) it is recommended that the designer become familiar with these guidelines, which discuss the special circuit connections and layout requirements needed for certain pins.

### *Power and Ground Recommendations*

When connecting power to the [AD6649,](http://www.analog.com/AD6649?doc=AD6649.pdf) it is recommended that two separate 1.8 V supplies be used: one supply should be used for analog (AVDD), and a separate supply should be used for the digital outputs (DRVDD). The designer can employ several different decoupling capacitors to cover both high and low frequencies. These capacitors should be located close to the point of entry at the PC board level and close to the pins of the part with minimal trace length.

A single PCB ground plane should be sufficient when using the [AD6649.](http://www.analog.com/AD6649?doc=AD6649.pdf) With proper decoupling and smart partitioning of the PCB analog, digital, and clock sections, optimum performance is easily achieved.

## *Exposed Paddle Thermal Heat Slug Recommendations*

It is mandatory that the exposed paddle on the underside of the ADC be connected to analog ground (AGND) to achieve the best electrical and thermal performance. A continuous, exposed (no solder mask) copper plane on the PCB should mate to the [AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) exposed paddle, Pin 0.

The copper plane should have several vias to achieve the lowest possible resistive thermal path for heat dissipation to flow through the bottom of the PCB. These vias should be filled or plugged with nonconductive epoxy.

To maximize the coverage and adhesion between the ADC and the PCB, a silkscreen should be overlaid to partition the continuous plane on the PCB into several uniform sections. This provides several tie points between the ADC and the PCB during the reflow process. Using one continuous plane with no partitions guarantees only one tie point between the ADC and the PCB. See the evaluation board for a PCB layout example. For detailed information about packaging and PCB layout of chip scale packages, refer to the [AN-772 Application Note,](http://www.analog.com/AN-772?doc=AD6649.pdf) *A [Design and Manufacturing Guide for the Lead Frame Chip Scale](http://www.analog.com/AN-772?doc=AD6649.pdf)  [Package \(LFCSP\)](http://www.analog.com/AN-772?doc=AD6649.pdf)*.

# *VCM*

The VCM pin should be decoupled to ground with a 0.1 μF capacitor, as shown i[n Figure 28.](#page-17-1) For optimal channel-to-channel isolation, a 33  $\Omega$  resistor should be included between th[e AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) VCM pin and the Channel A analog input network connection and between th[e AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) VCM pin and the Channel B analog input network connection.

# *SPI Port*

The SPI port should not be active during periods when the full dynamic performance of the converter is required. Because the SCLK, CSB, and SDIO signals are typically asynchronous to the ADC clock, noise from these signals can degrade converter performance. If the on-board SPI bus is used for other devices, it may be necessary to provide buffers between this bus and the [AD6649](http://www.analog.com/AD6649?doc=AD6649.pdf) to keep these signals from transitioning at the converter inputs during critical sampling periods.

<span id="page-39-0"></span>

### <span id="page-39-1"></span>**ORDERING GUIDE**



<sup>1</sup> Z = RoHS Compliant Part.

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