DC to 50 MHz, Dual I/Q Demodulator and Phase Shifter

Data Sheet **[AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf)**

NEXALOG
DEVICES

FEATURES

Dual integrated I/Q demodulator 16 phase select options on each output (22.5° per step) Quadrature demodulation accuracy Phase accuracy: ±0.1° Amplitude balance: ±0.05 dB Bandwidth 4 × LO: 10 kHz to 200 MHz RF: dc to 50 MHz Baseband: determined by external filtering Output dynamic range: 159 dB/Hz LO drive > 0 dBm (50 Ω); 4 × LO > 1 MHz Supply: ±5 V Power consumption: 190 mW/channel (380 mW total) Power-down

APPLICATIONS

Medical imaging (CW ultrasound beamforming) Phased array systems (radar and adaptive antennas) Communication receivers

GENERAL DESCRIPTION

The [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf)¹ is a dual phase-shifter and I/Q demodulator that enables coherent summing and phase alignment of multiple analog data channels. It is the first solid-state device suitable for beamformer circuits, such as those used in high performance medical ultrasound equipment featuring CW Doppler. The RF inputs interface directly with the outputs of the dual-channel, low noise preamplifiers included in th[e AD8332.](http://www.analog.com/AD8332?doc=AD8333.pdf)

A divide-by-4 circuit generates the internal 0° and 90° phases of the local oscillator (LO) that drive the mixers of a pair of matched I/Q demodulators.

The [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) can be applied as a major element in analog beamformer circuits in medical ultrasound equipment.

The [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) features an asynchronous reset pin. When used in arrays, the reset pin sets all the LO dividers in the same state. Sixteen discrete phase rotations in 22.5° increments can be selected independently for each channel. For example, if Channel 1 is used as a reference and the RF signal applied to Channel 2 has an I/Q phase lead of 45°, Channel 2 can be phase aligned with Channel 1 by choosing the correct code.

¹ Protected by US Patent 7,760,833.

FUNCTIONAL BLOCK DIAGRAM

Phase shift is defined by the output of one channel relative to another. For example, if the code of Channel 1 is adjusted to 0000 and that of Channel 2 is adjusted to 0001 and the same signal is applied to both RF inputs, the output of Channel 2 leads that of Channel 1 by 22.5°.

The I and Q outputs are provided as currents to facilitate summation. The summed current outputs are converted to voltages by a high dynamic range, current-to-voltage (I-V) converter, such as the [AD8021,](http://www.analog.com/AD8021?doc=AD8333.pdf) configured as a transimpedance amplifier. The resultant signal is then applied to a high resolution ADC, such as the [AD7665](http://www.analog.com/AD7665?doc=AD8333.pdf) (16 bit/570 kSPS).

The two I/Q demodulators can be used independently in other nonbeamforming applications. In that case, a transimpedance amplifier is needed for each of the I and Q outputs, four in total for the dual I/Q demodulator.

The dynamic range is 159 dB/Hz at the I and Q outputs, but the following transimpedance amplifier is an important element in maintaining the overall dynamic range, and attention needs to be paid to optimal component selection and design.

The [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) is available in a 32-lead LFCSP (5 mm \times 5 mm) package for the industrial temperature range of −40°C to +85°C.

Rev. F [Document Feedback](https://form.analog.com/Form_Pages/feedback/documentfeedback.aspx?doc=AD8333.pdf&product=AD8333&rev=F)

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REVISION HISTORY

8/2012—Rev. D to Rev. E

9/2010—Rev. C to Rev. D

9/2008—Rev. B to Rev. C

5/2007—Rev. A to Rev. B

5/2006—Rev. 0 to Rev. A

10/2005—Revision 0: Initial Version

SPECIFICATIONS

 $V_s = \pm 5$ V, $T_A = 25$ °C, $f_{4LO} = 20$ MHz, $f_{RF} = 5.01$ MHz, $f_{BB} = 10$ kHz, $P_{LO} \ge 0$ dBm, single-ended, sine wave; per channel performance, dBm (50 Ω), unless otherwise noted (se[e Figure 41\)](#page-14-1).

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ABSOLUTE MAXIMUM RATINGS

Table 2.

¹ 4-layer JEDEC board no airflow (exposed pad soldered to PCB).

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.

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

05543-002 NOTES
1. THE EXPOSED PAD IS NOT CONNECTED INTERNALLY.
FOR INCREASED RELIABILITY OF THE SOLDER
JOINTS AND MAXIMUM THERMAL CAPABILITY, IT IS
RECOMMENDED THAT THE PADDLE BE SOLDERED
TO THE GROUND PLANE.

Figure 2. 32-Lead LFCSP Pin Configuration

Table 3. Pin Function Descriptions

EQUIVALENT INPUT CIRCUITS

Figure 3. Logic Inputs

Figure 4. Local Oscillator Inputs

Figure 5. Local Oscillator Decoupling Pin

Figure 7. Output Drivers

TYPICAL PERFORMANCE CHARACTERISTICS

 $V_S = \pm 5$ V, T_A = 25°C, f_{4LO} = 20 MHz, f_{LO} = 5 MHz, f_{RF} = 5.01 MHz, f_{BB} = 10 kHz, P_{LO} ≥ 0 dBm (50 Ω); single-ended sine wave; per channel performance, differential voltages, dBm (50 Ω), phase select code = 0000, unless otherwise noted (see [Figure 41\)](#page-14-1).

Figure 8. Normalized Vector Plot of Phase, Channel 2 with Respect to Channel 1; Channel 1 Is Fixed at 0°, Channel 2 Stepped 22.5°/Step, All Codes Displayed

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Figure 10. Amplitude Error of Channel 2 with Respect to Channel 1 vs. Code at 1 MHz and 5 MHz

Figure 11. Phase Error of Channel 2 with Respect to Channel 1 vs. Code at 1 MHz and 5 MHz

Figure 12. I or Q Output of Channel 2 with Respect to Channel 1, First Quadrant Shown

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Figure 18. Typical I2xO/I1xOor Q2xO/Q1xOAmplitude Match vs. RF Frequency, First Quadrant, at Three Temperatures

Figure 19. I2xO/I1xO or Q2xO/Q1xO Phase Error vs. RF Frequency, Baseband Frequency = 10 kHz, at Three Temperatures

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Figure 38. Phase Shifting Response—Channel 2 Leads Channel 1 by 180°, Top: Input to PH23 Select Code = 1000; Bottom (Red): Reference Channel 1 I_{OUT} ; Bottom (Gray): Channel 2 I_{OUT} Phase Shifted 180°, Channel 1 Reference Phase Code = 0000

Figure 39. PSRR vs. Frequency (se[e Figure 51\)](#page-16-2)

Figure 40. Quiescent Supply Current vs. Temperature

TEST CIRCUITS

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Figure 44. Enable Response

SIGNAL GENERATOR

50Ω

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50Ω

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SIGNAL GENERATOR

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AD8021

05543-035

Figure 48. OIP3 vs. Baseband Frequency

THEORY OF OPERATION

The [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) is a dual I/Q demodulator with a programmable phase shifter for each channel. The primary applications are phased array beamforming in medical ultrasound, phased array radar, and smart antennae for mobile communications. The [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) can also be used in applications that require two wellmatched I/Q demodulators.

[Figure 52](#page-17-3) shows the block diagram and pinout of th[e AD8333.](http://www.analog.com/AD8333?doc=AD8333.pdf) Three analog and nine quasilogic level inputs are required. Two RF inputs accept signals from the RF sources and a local oscillator (applied to the differential input pins marked 4LOx) common to both channels constitute the analog inputs. Four logic inputs per channel define one of 16 delay states/360° (or 22.5°/step), selectable with PHx0 to PHx3. The reset input is used to synchronize [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) devices used in arrays.

Figure 52. Block Diagram and Pinout

Each of the current formatted I and Q outputs sum together for beamforming applications. Multiple channels are summed and converted to a voltage using a transimpedance amplifier. If desired, channels can also be used individually.

QUADRATURE GENERATION

The internal 0° and 90° LO phases are digitally generated by a divide-by-4 logic circuit. The divider is dc-coupled and inherently broadband; the maximum LO frequency is limited only by its switching speed. The duty cycle of the quadrature LO signals is intrinsically 50% and is unaffected by the asymmetry of the externally connected 4LOx inputs. Furthermore, the divider is implemented such that the 4LOx signals reclock the final flipflops that generate the internal LO signals and thereby minimizes noise introduced by the divide circuitry.

For optimum performance, the 4LOx inputs are driven differentially but can also be driven in a single-ended fashion. A good choice for a drive is an LVDS device. The common-mode range on each pin is approximately 0.2 V to 3.8 V with nominal \pm 5 V supplies.

The minimum LO level is frequency dependent (see [Figure 22\)](#page-10-0). For optimum noise performance, it is important to ensure that the LO source has very low phase noise (jitter) and adequate input level to ensure stable mixer-core switching. The gain through the divider determines the LO signal level vs. RF frequency. The [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) can be operated to very low frequencies at the LO inputs if a square wave is used to drive the LO.

Beamforming applications require a precise channel-to-channel phase relationship for coherence among multiple channels. A reset pin (RSET) is provided to synchronize the 4LOx divider circuits when [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) devices are used in arrays. The RSET pin resets the counters to a known state after power is applied to multiple [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) devices. A logic input must be provided to the RSET pin when using more than one [AD8333.](http://www.analog.com/AD8333?doc=AD8333.pdf) See the [Reset](#page-24-2) [Input](#page-24-2) section for more details.

I/Q DEMODULATOR AND PHASE SHIFTER

The I/Q demodulators consist of double-balanced Gilbert cell mixers. The RF input signals are converted into currents by transconductance stages that have a maximum differential input signal capability of 2.8 V p-p. These currents are then presented to the mixers, which convert them to baseband: RF − LO and RF + LO. The signals are phase shifted according to the code applied to Pin PHx0 to Pin PHx3 (se[e Table 4\)](#page-18-1). The phase shift function is an integral part of the overall circuit (patent pending). The phase shift listed in Column 1 of [Table 4 i](#page-18-1)s defined as being between the baseband I or Q channel outputs. As an example, for a common signal applied to the RF inputs of an [AD8333,](http://www.analog.com/AD8333?doc=AD8333.pdf) the baseband outputs are in phase for matching phase codes. However, if the phase code for Channel 1 is 0000 and that of Channel 2 is 0001, Channel 2 leads Channel 1 by 22.5°.

Following the phase shift circuitry, the differential current signal is converted from differential to single ended via a current mirror. An external transimpedance amplifier is needed to convert the I and Q outputs to voltages.

Table 4. Phase Nibble Select Codes

DYNAMIC RANGE AND NOISE

[Figure 53](#page-18-2) is an interconnection block diagram of th[e AD8333.](http://www.analog.com/AD8333?doc=AD8333.pdf) For optimum system noise performance, the RF input signal is provided by a very low noise amplifier, such as the LNA of a[n AD8332](http://www.analog.com/AD8332?doc=AD8333.pdf) or the preamplifier of an [AD8335.](http://www.analog.com/AD8335?doc=AD8333.pdf) In beamformer applications, the I and Q outputs of a number of receiver channels are summed (for example, the two channels illustrated i[n Figure 53\)](#page-18-2). The dynamic range of the system increases by the factor $10 \log_{10}(N)$, where N is the number of channels (assuming random uncorrelated noise). The noise in the two-channel example of [Figure 53](#page-18-2) is increased by 3 dB while the signal doubles (6 dB), yielding an aggregate SNR improvement of $(6 dB – 3 dB) = 3 dB$.

Judicious selection of the RF amplifier ensures the least degradation in dynamic range. The input-referred spectral voltage noise density (e_n) of the [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) is nominally 9 nV/ \sqrt{Hz} to 10 nV/√Hz. For the noise of th[e AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) to degrade the system noise figure (NF) by 1 dB, the combined noise of the source and the LNA must be about twice that of the [AD8333,](http://www.analog.com/AD8333?doc=AD8333.pdf) or 18 nV/√Hz. If the noise of the circuitry before the [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) is <18 nV/ \sqrt{Hz} , the system NF degrades more than 1 dB. For example, if the noise contribution of the LNA and source is equal to th[e AD8333,](http://www.analog.com/AD8333?doc=AD8333.pdf) or 9 nV/ \sqrt{Hz} , the degradation is 3 dB. If the circuit noise preceding the [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) is 1.3× as large as that of th[e AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) (or about 11.7 nV/ \sqrt{Hz}), the degradation is 2 dB. For a circuit noise of 1.45 \times that of the [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) (13.1 nV/ \sqrt{Hz}), the degradation is 1.5 dB.

To determine the input-referred noise, it is important to know the active low-pass filter (LPF) values R_{FILT} and C_{FILT} , shown in [Figure 53.](#page-18-2) Typical filter values (for example, those used on the evaluation board) are 787 Ω and 2.2 nF and implement a 90 kHz single-pole LPF. If the RF and LO are offset by 10 kHz, the demodulated signal is 10 kHz and is passed by the LPF. The single-channel mixing gain from the RF input to the [AD8021](http://www.analog.com/AD8021?doc=AD8333.pdf) output (for example, ΣI, ΣQ) is approximately 1.7 × 4.7 dB. This together with the 9 nV/ \sqrt{Hz} AD8333 noise results in about 15.3 nV/ \sqrt{Hz} at the [AD8021](http://www.analog.com/AD8021?doc=AD8333.pdf) output. Because the [AD8021,](http://www.analog.com/AD8021?doc=AD8333.pdf) including the 787 Ω feedback resistor, contributes another $4.4 \text{ nV}/\sqrt{\text{Hz}}$, the total output-referred noise is about 16 nV/√Hz. This value can be adjusted by increasing the filter resistor while maintaining the corner frequency, thereby increasing the gain. The factor limiting the magnitude of the gain is the output swing and drive capability of the operational amplifier selected for the I-to-V converter, in this instance the [AD8021.](http://www.analog.com/AD8021?doc=AD8333.pdf)

SUMMATION OF MULTIPLE CHANNELS (ANALOG BEAMFORMING)

Beamforming, as applied to medical ultrasound, is defined as the phase alignment and summation of signals generated from a common source but received at different times by a multielement ultrasound transducer. Beamforming has two functions: it imparts directivity to the transducer, enhancing its gain, and it defines a focal point within the body from which the location of the returning echo is derived. The primary application for the [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) is in analog beamforming circuits for ultrasound.

PHASE COMPENSATION AND ANALOG BEAMFORMING

Modern ultrasound machines used for medical applications employ a $2ⁿ$ binary array of receivers for beamforming, with typical array sizes of 16 or 32 receiver channels phase-shifted and summed together to extract coherent information. When used in multiples, the desired signals from each of the channels can be summed to yield a larger signal (increased by a factor N, where N is the number of channels), while the noise is increased by the square root of the number of channels. This technique enhances the signal-to-noise performance of the machine. The critical elements in a beamformer design are the means to align the incoming signals in the time domain and the means to sum the individual signals into a composite whole.

In traditional analog beamformers incorporating Doppler, a V-to-I converter per channel and a crosspoint switch precede passive delay lines used as a combined phase shifter and summing circuit. The system operates at the receive frequency (RF) through the delay line, and then the signal is down-converted by a very large dynamic range I/Q demodulator.

The resultant I and Q signals are filtered and sampled by two high resolution ADCs. The sampled signals are processed to extract the relevant Doppler information.

Alternatively, the RF signal can be processed by downconversion on each channel individually, phase shifting the downconverted signal and then combining all channels. Th[e AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) provides the means to implement this architecture. The downconversion is done by an I/Q demodulator on each channel, and the summed current output is the same as in the delay line approach. The subsequent filters after the I-to-V conversion and the ADCs are similar.

The [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) integrates the phase shifter, frequency conversion, and I/Q demodulation into a single package and directly yields the baseband signal. To illustrate this, [Figure 54](#page-19-2) is a simplified diagram showing two channels. The ultrasound wave (USW) is received by two transducer elements, TE1 and TE2, in an ultrasound probe and generates the E1 and E2 signals. In this example, the phase at TE1 leads the phase at TE2 by 45°.

Figure 54. Simplified Example of th[e AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) Phase Shifter

In a real application, the phase difference depends on the element spacing, λ (wavelength), speed of sound, angle of incidence, and other factors. The ES1 and ES2 signals are amplified 19 dB by the low noise amplifiers in th[e AD8332.](http://www.analog.com/AD8332?doc=AD8333.pdf) For optimum signal-tonoise performance, the output of the LNA is applied directly to the input of th[e AD8333.](http://www.analog.com/AD8333?doc=AD8333.pdf) To sum the ES1 and ES2 signals, ES2 is shifted 45° relative to ES1 by setting the phase code in Channel 2 to 0010. The phase-aligned current signals at the output of the [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) are summed in an I-to-V converter to provide the combined output signal with a theoretical improvement in dynamic range of 3 dB for the sum of two channels.

In a beamformer using the [AD8333,](http://www.analog.com/AD8333?doc=AD8333.pdf) the bipolar currents at the I and Q outputs are summed directly[. Figure 55](#page-20-1) illustrates 16 summed channels (for clarity, these channels are shown as current sources) as an example of an active current summing circuit using th[e AD8333.](http://www.analog.com/AD8333?doc=AD8333.pdf) This figure also illustrates th[e AD8021](http://www.analog.com/AD8021?doc=AD8333.pdf) as first-order current summing circuits an[d AD797 d](http://www.analog.com/AD797?doc=AD8333.pdf)evices as low noise second-order summing circuits. Beginning with the operational amplifiers, there are a few important considerations in the circuit shown i[n Figure 55.](#page-20-1)

The operational amplifiers selected for the first-order summing amplifiers must have good frequency response over the full operating frequency range of the [AD8333 d](http://www.analog.com/AD8333?doc=AD8333.pdf)evices and be able to source the current required at the [AD8333 I](http://www.analog.com/AD8333?doc=AD8333.pdf) and Q outputs.

The total current of eac[h AD8333 i](http://www.analog.com/AD8333?doc=AD8333.pdf)s 6.6 mA for the multiples of the 45° phase settings (Code 0010, Code 0110, Code 1010, and Code 1110) and is divided nearly equally between the baseband frequencies (including a dc component) and the second harmonic of the local oscillator frequency. The desired CW signal tends to be much less (<40 dB) than the unwanted interfering signals.

When determining the large signal requirements of the firstorder summing amplifiers and low-pass filters, the very small CW signal can be ignored. The number of channels that can be summed is limited by the output drive current capacity of the operational amplifier selected: 60 mA to 70 mA for a linear output current for ±5 V and ±12 V, respectively, for the [AD8021.](http://www.analog.com/AD8021?doc=AD8333.pdf) Because the [AD8021 i](http://www.analog.com/AD8021?doc=AD8333.pdf)mplements an active LPF together with R1x and C1x, it must absorb the worst-case current provided by the [AD8333,](http://www.analog.com/AD8333?doc=AD8333.pdf) for example, 6.6 mA. Therefore, the maximum number of channels that the [AD8021 c](http://www.analog.com/AD8021?doc=AD8333.pdf)an sum is 10 for ±12 V or eight for ±5 V supplies. In practical applications, CW channels are used in powers of two, thus the maximum number per [AD8021 i](http://www.analog.com/AD8021?doc=AD8333.pdf)s eight.

Another consideration for the operational amplifier selected as an I-to-V converter is the compliance voltage of the [AD8333 I](http://www.analog.com/AD8333?doc=AD8333.pdf) and Q outputs. The maximum compliance voltage is 0.5 V, and a dc bias must be provided at these pins. The [AD8021](http://www.analog.com/AD8021?doc=AD8333.pdf) active LPF satisfies these requirements; it keeps the outputs at 0 V via the virtual ground at the operational amplifier inverting input while providing any needed dc bias current.

Figure 55. A 16-Channel Beamformer

As previously noted, a typical CW signal has a large dc and very low frequency component compared with its desired low CW Doppler baseband frequency, and another unwanted component at the $2 \times$ LO. The dc component flows through the gain resistors R1x, and the $2 \times$ LO flows through the capacitors C1x. The smaller desired CW Doppler baseband signal is in the frequency range of 1 kHz to 50 kHz.

Because the output current of th[e AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) contains the baseband frequency, a dc component, and the $2 \times$ LO frequency voltages, the desired small amplitude baseband signal must be extracted after a series of filters. These are shown in [Figure 55](#page-20-1) as LPFnA, HPFnA, and gain stages.

Before establishing the value of CLPF1, the resistor RLPF1 is selected based on the peak operating current and the linear range of the operational amplifier. Because the peak current for eac[h AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) is 6.6 mA and there are eight channels to be summed, the total peak current required is 52.8 mA. Approximately half of this current is dc, and the other half is at a frequency of $2 \times$ LO. Therefore, about 26.4 mA flows through the resistor, and the remaining 26.4 mA flows through the capacitor. R1 was selected as 100 Ω and, after filtering, generates a peak dc and very low frequency voltage of 2.64 V at the [AD8021 o](http://www.analog.com/AD8021?doc=AD8333.pdf)utput. For power supplies of \pm 5 V, 100 Ω is a good choice for R1.

However, because the CW signal needs to be amplified as much as possible and the noise degradation of the signal path minimized, the value of R1 must be as large as possible. A larger supply helps in this regard, and the only factor limiting the largest supply voltage is the required power.

For a ± 10 V supply on th[e AD8021,](http://www.analog.com/AD8021?doc=AD8333.pdf) R1 can be increased to 301 Ω to realize the same headroom as with a \pm 5 V supply. If a higher value of R1 is used, C1 must be adjusted accordingly (in this example, 1/3 the value of the original value) to maintain the desired LPF roll-off. The principal advantage of a higher supply is greater dynamic range, and the trade-off is power consumption. The user must weigh the trade-offs associated with the supply voltage, R1, C1, and the following circuitry. A suggested design sequence is as follows:

Select a low noise, high speed operational amplifier. The spectral density noise (e_n) must be <2 nV/ \sqrt{Hz} , and the 3 dB bandwidth must be $\geq 3 \times$ the expected maximum 2 \times LO frequency.

Divide the maximum linear output current by 6.6 mA to determine the maximum number of [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) channels that can be summed.

Select the largest value of R1 that permits the output voltage swing within the power supply rails.

Calculate the value of C1 to implement the LPF corner that allows the CW Doppler signal to pass with maximum attenuation of the $2 \times$ LO signal.

The filter LPF1A establishes the upper frequency limit of the baseband frequency and is selected well below the $2 \times LO$ frequency, typically 100 kHz or less (for example, 88 kHz in [Figure 55\)](#page-20-1).

A useful equation for calculating C1 is

$$
CI = \frac{1}{2\pi R I f_{LPF1}} \qquad (1)
$$

As previously mentioned, the [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) output current contains a dc current component. This dc component is converted to a large dc voltage by the [AD8021](http://www.analog.com/AD8021?doc=AD8333.pdf) LPF. Capacitor C2 filters this dc component and, with R2 + R3, establishes a high-pass filter with a low frequency cutoff of about 100 Hz. Capacitor C3 is much smaller than C2 and, consequently, can be neglected. C2 can be calculated by

$$
C2 = \frac{1}{2\pi (R2 + R3)f_{HPF1}}\tag{2}
$$

To achieve maximum attenuation of the $2 \times$ LO frequency, a second low-pass filter, LPF2, is established using the parallel combination of R2 and R3, and C3. Its −3 dB frequency is

$$
f_{LPF2} = \frac{1}{2\pi (R2 || R3)C3} \tag{3}
$$

In the example shown i[n Figure 55,](#page-20-1) $f_{LPF2} = 81$ kHz.

Finally, the feedback resistor of th[e AD797](http://www.analog.com/AD797?doc=AD8333.pdf) must be calculated. This is a function of the input current (number of channels) and the supply voltage.

The second-order summing amplifier requires a very low noise operational amplifier, such as th[e AD797,](http://www.analog.com/AD797?doc=AD8333.pdf) with 0.9 nV/ \sqrt{Hz} , because the amplifier gain is determined by Feedback Resistor R4 divided by the parallel combination of the LPF2A resistors seen looking back toward the [AD8021](http://www.analog.com/AD8021?doc=AD8333.pdf) devices. Referring to [Figure 55,](#page-20-1) the [AD797](http://www.analog.com/AD797?doc=AD8333.pdf) in-band (100 Hz to 88 kHz) gain is expressed as

$$
\frac{R4}{[(R2A + R3A)][(R2B + R2B)]}
$$
(4)

Th[e AD797](http://www.analog.com/AD797?doc=AD8333.pdf) noise gain can increase to unacceptable levels because the denominator of the gain equation is the parallel resistance of all the R2 + R3 resistors in th[e AD8021](http://www.analog.com/AD8021?doc=AD8333.pdf) outputs. For example, for a 64-channel beamformer, the resistance seen looking back toward the [AD8021](http://www.analog.com/AD8021?doc=AD8333.pdf) devices is about 1.4 kΩ/8 = 175 Ω. For this reason, the value of $(R2x + R3x)$ must be as large as possible to minimize the noise gain of th[e AD797.](http://www.analog.com/AD797?doc=AD8333.pdf) (Note that this is the case for the [AD8021](http://www.analog.com/AD8021?doc=AD8333.pdf) stages because they look back into the high impedance current sources of th[e AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) devices.)

Due to these considerations, it is advantageous to increase the gain of th[e AD8021](http://www.analog.com/AD8021?doc=AD8333.pdf) devices as much as possible because the value of (R2x + R3x) can be increased proportionally. Resistors (R2x + R3x) convert the CW voltages to currents that are summed at the inverting inputs of th[e AD797](http://www.analog.com/AD797?doc=AD8333.pdf) operational amplifier, and then amplified and converted to voltages by R4.

The value of R4 needs to be chosen iteratively as follows:

- 1. Determine the number o[f AD8021](http://www.analog.com/AD8021?doc=AD8333.pdf) first-order summing amplifiers. I[n Figure 55,](#page-20-1) there are two; for a 32-channel beamformer, there must be four, and for a 64-channel beamformer, there must be eight.
- 2. Determine the output noise from the [AD8021](http://www.analog.com/AD8021?doc=AD8333.pdf) devices. A first-order calculation can be based on a value of [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) output current noise of about 20 pA/√Hz. For the values in [Figure 55,](#page-20-1) this results in about 6 nV/ \sqrt{Hz} for eight channels after th[e AD8021](http://www.analog.com/AD8021?doc=AD8333.pdf) devices. Adding the noise of the [AD8021](http://www.analog.com/AD8021?doc=AD8333.pdf) and the 100 Ω feedback resistor results in about 6.5 nV/ \sqrt{Hz} total noise after the [AD8021](http://www.analog.com/AD8021?doc=AD8333.pdf) LPF in the CW Doppler band.
- 3. Determine the noise of the circuitry after the [AD797](http://www.analog.com/AD797?doc=AD8333.pdf) and determine the desired signal level.
- 4. Determine the voltage and current noise of the secondorder summing amplifiers.
- 5. Choose a value for $(R2x + R3x)$ and for R4. Determine the resulting output noise after th[e AD797](http://www.analog.com/AD797?doc=AD8333.pdf) for one channel, and then multiply this value by the square root of the number of summed [AD8021](http://www.analog.com/AD8021?doc=AD8333.pdf) devices. Next, check [AD797](http://www.analog.com/AD797?doc=AD8333.pdf) output noise (both current and voltage noise). Ideally, the sum of the noise of the resistors and th[e AD797](http://www.analog.com/AD797?doc=AD8333.pdf) must be less than a factor of 3 than the noise due to th[e AD8021](http://www.analog.com/AD8021?doc=AD8333.pdf) outputs.
- 6. Check the following stages output noise against the calculated noise from the combiner circuit an[d AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) devices. Ideally, the noise from the following stage must be less than 1/3 of the calculated noise.
- 7. If the combined noise is too large, experiment with increasing/decreasing values for $(R2x + R3x)$ and R4.

To simplify, the user can also simulate or build a combiner circuit for optimum performance. It must be noted that the ~20 pA/ \sqrt{Hz} output from th[e AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) is for th[e AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) with shorted RF inputs. In an actual system, the current noise output from the [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) is most likely dominated by the noise from th[e AD8332](http://www.analog.com/AD8332?doc=AD8333.pdf) LNA and the noise from the source and other circuitry before the LNA. This helps ease the design of the combiner. The preceding procedures for determining the optimum values for the combiner are based on the noise floor of th[e AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) only.

As an example, for a 32-channel beamformer using four low-pass filters, as shown in [Figure 55,](#page-20-1) $(R2x + R3x) = 1.4 k\Omega$ and R4 = 6.19 kΩ. The theoretical noise increase of \sqrt{N} is degraded by only about 1 dB.

DYNAMIC RANGE INFLATION

Although all 64 channels can theoretically be summed together at a single amplifier, it is important to realize that the dynamic range of the summed output increases by $10 \log_{10}(N)$ if all channels have uncorrelated noise, where N is the number of channels to be summed.

The summed signal level increases by a factor of N, whereas the noise increases only as \sqrt{N} . In the case of 64 channels, this is an increase in dynamic range of 18 dB. Note that the [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) dynamic range is already about 160 dB/Hz; the summed dynamic range is 178 dB/Hz (equivalent to about 29.5 bits/Hz). In a 50 kHz noise bandwidth, this is 131 dB (21.7 bits).

DISABLING THE CURRENT MIRROR AND DECREASING NOISE

The noise contribution of th[e AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) can potentially be reduced if the current mirrors that convert the internal differential signals to single-ended signals are bypassed (se[e Figure 56\)](#page-22-2). Current mirrors interface to the [AD8021 I](http://www.analog.com/AD8021?doc=AD8333.pdf)-V converters shown in [Figure 53,](#page-18-2) and output capacitors across the positive and negative outputs provide low-pass filtering. Th[e AD8021](http://www.analog.com/AD8021?doc=AD8333.pdf) devices force th[e AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) output voltage to 0 V and then process the bipolar output current; however, the internal current mirrors introduce a significant amount of noise. This noise can be reduced if the mirrors are disabled and the outputs are externally biased.

The mirrors are disabled by connecting VNEG to ground and providing external bias networks, as shown i[n Figure 56.](#page-22-2) The larger the drop across the resistors, the less noise they contribute to the output; however, the voltage on the I and Q output nodes cannot exceed 0.5 V. Voltages exceeding approximately 0.7 V turn on the PNP devices and forward bias the ESD protection diodes. Inductors provide an alternative to resistors, enabling reduced static power by eliminating the power dissipation in the bias resistors.

Figure 56. Bypassing the Internal Current Mirrors

With inductors, the main limitation might be low frequency operation, as is the case in CW Doppler in ultrasound where the frequency range of interest goes from a few hundred hertz to about 30 kHz. In addition, it is still important to provide enough gain through the I-to-V circuitry to ensure that the bias resistor and I-to-V converter noise do not contribute significantly to the noise from th[e AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) outputs. Another approach is to provide a single external current mirror that combines all channels; it is also possible to implement a high-pass filter with this circuit to help with offset and low frequency reduction.

The main disadvantage of the external bias approach is that two I-V amplifiers are needed because of the differential output (see [Figure 56\)](#page-22-2). For beamforming applications, the outputs are still summed, but there is twice the number of lines. Only two bias resistors are needed for all outputs that are connected together. The resistors are scaled by dividing the value of a single output bias resistor through N, the number of channels connected in parallel. The bias current depends on the phase selected: for phase 0°, it is about 2.5 mA per side, whereas in the case of 45°, it is about 3.5 mA per side. The bias resistors must be chosen based on the larger bias current value of 3.5 mA and the chosen VNEG. VNEG must be at least −5 V and can be larger for additional noise reduction.

Excessive noise or distortion at high signal levels degrades the dynamic range of the signal. Transmitter leakage and echoes from slow moving tissue generate the largest signal amplitudes in ultrasound CW Doppler mode and are largest near dc and at low frequencies. A high-pass filter introduced immediately following the [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) reduces the dynamic range. This is shown by the two coupling capacitors after the external bias resistors i[n Figure 56.](#page-22-2) Users have to determine what is acceptable for a particular application. Care must be taken in designing the external circuitry to avoid introducing noise via the external bias and low frequency reduction circuitry.

APPLICATIONS INFORMATION

The [AD8333 i](http://www.analog.com/AD8333?doc=AD8333.pdf)s the key component of a phase-shifter system that aligns time-skewed information contained in RF signals. Combined with a variable gain amplifier (VGA) and low noise amplifier (LNA), the [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) forms a complete analog receiver for a high performance ultrasound system[. Figure 57 i](#page-24-5)s a block diagram of a complete receiver using th[e AD8333,](http://www.analog.com/AD8333?doc=AD8333.pdf) [AD8331,](http://www.analog.com/AD8331?doc=AD8333.pdf) [AD8332,](http://www.analog.com/AD8332?doc=AD8333.pdf) an[d AD8334.](http://www.analog.com/AD8334?doc=AD8333.pdf)

As a major element of an ultrasound system, it is important to consider the many input/output options of th[e AD8333 t](http://www.analog.com/AD8333?doc=AD8333.pdf)hat are necessary to perform its intended function. [Figure 61 s](#page-25-3)hows the basic connections.

LOGIC INPUTS AND INTERFACES

The logic inputs of th[e AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) are all bipolar-level sensitive inputs. They are not edge triggered, nor are they to be confused with classic TTL or other logic family input topologies. The voltage threshold for these inputs is $VPOS \times 0.3$, so for a 5 V supply the threshold is 1.5 V, with a hysteresis of \pm 0.2 V. Although the inputs are not of themselves logic inputs, any 5 V logic family can drive them.

RESET INPUT

The RSET pin is used to synchronize the LO dividers in [AD8333 a](http://www.analog.com/AD8333?doc=AD8333.pdf)rrays. Because they are driven by the same internal LO, the two channels in an[y AD8333 a](http://www.analog.com/AD8333?doc=AD8333.pdf)re inherently synchronous. However, when multipl[e AD8333 d](http://www.analog.com/AD8333?doc=AD8333.pdf)evices are used, it is possible that their dividers wake up in different phase states. The function of the RSET pin is to phase align all the LO signals in multipl[e AD8333 d](http://www.analog.com/AD8333?doc=AD8333.pdf)evices.

The $4 \times$ LO divider of each [AD8333 c](http://www.analog.com/AD8333?doc=AD8333.pdf)an initiate in one of four possible states: 0°, 90°, 180°, or 270°. The internally generated I/Q signals of eac[h AD8333 L](http://www.analog.com/AD8333?doc=AD8333.pdf)O are always at a 90° angle relative to each other, but a phase shift can occur during power-up between the internal LOs of the differen[t AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) devices.

The RSET pin provides an asynchronous reset of the LO dividers by forcing the internal LO to hang. This mechanism also allows the measurement of nonmixing gain from the RF input to the output.

The rising edge of the active high RSET pulse can occur at any time, but the duration must be \geq 300 ns minimum (t_{PW-MIN}). When the RSET pulse transitions from high to low, the LO dividers are reactivated; however, there is a short delay until the divider recovers to a valid state. To guarantee synchronous operation of an array of $AD8333$ devices, the $4 \times LO$ clock must be disabled when the RSET transitions high, and then remain disabled for at least 300 ns after RSET transitions low.

Synchronization of multipl[e AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) devices can be checked as follows:

- 1. Set the phase code of all [AD8333 c](http://www.analog.com/AD8333?doc=AD8333.pdf)hannels to the same setting, for example, 0000.
- 2. Apply a test signal to a single channel that generates a sine wave in the baseband output, and then measure the output.
- 3. Apply the same test signal to all channels simultaneously, and then measure the output.

Because all the phase codes of th[e AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) devices are the same, the combined signal must be N times bigger than the single channel. The combined signal is less than N times one channel if any of the LO phases of individua[l AD8333 d](http://www.analog.com/AD8333?doc=AD8333.pdf)evices are in error.

CONNECTING TO THE LNA OF THE [AD8331/](http://www.analog.com/AD8331?doc=AD8333.pdf)[AD8332](http://www.analog.com/AD8332?doc=AD8333.pdf)[/AD8334/](http://www.analog.com/AD8334?doc=AD8333.pdf)[AD8335](http://www.analog.com/AD8335?doc=AD8333.pdf) VGAs

Figure 59. Connecting th[e AD8333 t](http://www.analog.com/AD8333?doc=AD8333.pdf)o the LNA of a[n AD8332](http://www.analog.com/AD8332?doc=AD8333.pdf)

The RFxx inputs (Pin 12, Pin 13, Pin 28, and Pin 29) are optimized for maximum dynamic range when dc-coupled to the differential output pins of the LNA of the [AD8331](http://www.analog.com/AD8331?doc=AD8333.pdf)[/AD8332/](http://www.analog.com/AD8332?doc=AD8333.pdf) [AD8334 o](http://www.analog.com/AD8334?doc=AD8333.pdf)r th[e AD8335 s](http://www.analog.com/AD8335?doc=AD8333.pdf)eries of VGAs and can be connected directly, as shown in Figure 59.

INTERFACING TO OTHER AMPLIFIERS

If amplifiers other than th[e AD8332](http://www.analog.com/AD8332?doc=AD8333.pdf) LNA are connected to the input, attention must be paid to their bias and drive levels. For maximum input signal swing, the optimum bias level is 2.5 V, and the RF input must not exceed 5 V to avoid turning on the ESD protection circuitry. If ac coupling is used, a bias circuit, such as that illustrated in Figure 60, is recommended. An internal bias network is provided; however, additional external biasing can center the RF input at 2.5 V.

Figure 60. AC Coupling th[e AD8333 R](http://www.analog.com/AD8333?doc=AD8333.pdf)F Input

To realize the full range of performance, the [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) must be driven from a differential source. Using a single-ended source is strongly discouraged because of internal supply headroom constraints.

LO INPUT

The LO input is a high speed, fully differential analog input that responds to differences in the input levels, not in the logic levels. The LO inputs can be driven with a low common-mode voltage amplifier, such as the National Semiconductor DS90C401 LVDS driver.

[Figure 22 a](#page-10-0)n[d Figure 23](#page-10-1) show the range of common-mode voltages and useable LO levels when the LO input is driven with a singleended sine wave. Logic families, such as TTL or CMOS, are unsuitable for direct coupling to the LO input.

***OPTIONAL BIAS NETWORK. THESE COMPONENTS CAN BE DELETED IF THE LO IS DC-COUPLED FROM AN LVDS SOURCE BIASED AT 1.2V.**

Figure 61[. AD8333 B](http://www.analog.com/AD8333?doc=AD8333.pdf)asic Connections

EVALUATION BOARD

The [AD8333-EVALZ](http://www.analog.com/AD8333-EVALZ?doc=AD8333.pdf) evaluation board provides a platform for test and evaluation of th[e AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) I/Q demodulator and phase shifter. The board is shipped fully assembled and tested and is signal ready. A pair of [AD8332](http://www.analog.com/AD8332?doc=AD8333.pdf) low-noise amplifiers (LNA) provide input matching and amplification for the differential input of the [AD8333.](http://www.analog.com/AD8333?doc=AD8333.pdf) A photograph of the board is shown in [Figure 62](#page-26-2) and a schematic diagram is shown i[n Figure 64.](#page-28-1) The board requires dual 5 V supplies capable of supplying 300 mA or greater. Except for the optional components shown in grayscale, the board is completely built and tested.

Figure 62. Evaluation Board (Actual Size)

FEATURES AND OPTIONS

The evaluation board has several user-configurable features and options. [Table 5 l](#page-26-3)ists the configuration switches and their functions.

Table 5. Switch Functions

Phase Nibble

The phase nibble configures the phase delay for each channel in sixteen 22.5° increments from 0° to 337.5°. The increments increase proportionally in a simple binary format from 0H (hexadecimal) to FH[. Table 4](#page-18-1) lists the phase shift and corresponding code for each bit. The bits are labeled 0 and 1, corresponding to low and high, respectively, on the silkscreen. The switches select the desired state.

Enable and Reset Switches

For normal operation, place a switch in the upper position of ENBL. To disable the [AD8333,](http://www.analog.com/AD8333?doc=AD8333.pdf) move the switch to the lower position. For normal operation, the switch for RST is in its right position. When the switch is in the left position, the device counter is held in reset and no mixing occurs.

Fixed Options

Several options can be realized by adding or changing resistors.

LNA Input Impedance

The shipping configuration of the input impedance of the LNA is 50 Ω to match the output impedance of most signal generators. Input impedances up to 6 kΩ are obtained by selecting the R9 and R10 values. Details concerning this circuit feature are found in th[e AD8332](http://www.analog.com/AD8332?doc=AD8333.pdf) data sheet. For reference[, Table 6](#page-26-4) lists common values of input impedance and corresponding feedback resistor values.

Current Summing

The output transimpedance amplifiers, A1 through A4, are configured as I-to-V converters to convert the output current of the [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) to a voltage. The low-pass filters formed by the feedback components are designed for single-channel operation with ±5 V supplies.

Optional Resistors R4 and R5 sum the two channels. With R4 and R5 installed, R2 and R3 are removed, and then the sum of the outputs is seen at the I1xO and Q1xO output SMA connectors.

The user has the option to adjust the values of R39, R40, R41, or R42 according to the power supply voltages and expected input current levels. For the same supply voltages, if two channels are summed together, the feedback resistors are halved and the filter capacitor values doubled to optimize the output swing.

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Filter Capacitors C26, C29, C31, and C 32 establish the roll-off characteristic according to the following well-known equation:

$$
f = \frac{1}{\omega RC}
$$

where R is the value of R39, R40, R41, or R42, and C is the value of C26, C29, C31, or C32.

Reset Input

For normal operation, the reset input is high (no reset). To drive the reset with a dynamic signal, a provision is made to connect a signal generator at the RST input. A 49.9 Ω , 0603 surface-mount resistor can be installed at R15 to terminate the reset input for pulsed experiments. In this configuration, the switch at RST is not used and must be removed to avoid loading the power supply.

MEASUREMENT SETUP

[Figure 63 i](#page-27-1)s a layout of th[e AD8333-EVALZ](http://www.analog.com/AD8333-EVALZ?doc=AD8333.pdf) showing the connectors and switches[. Figure 65](#page-29-0) shows a typical board and test equipment setup with two signal generators, a power splitter, and a \pm 5 V, 300 mA (minimum) power supply.

For ease in observing waveforms, the signal generators can be synchronized. Remember that the f4LO signal generator frequency is four times that of the nominal frequency of the RF source. For example, to detect signals with a nominal center frequency of 5 MHz, an f4LO frequency of 20 MHz is applied to the oscillator input. For an applied RF signal of 5.01 MHz, the mix frequencies are 10 kHz and 10.01 MHz. Because of the lowpass active filter of the transconductance amplifiers (A1 through A4), the 10.01 MHz component is suppressed, and only the 10 kHz is observed at the output.

Take care to avoid overdriving the LNA input of the [AD8332.](http://www.analog.com/AD8332?doc=AD8333.pdf) The LNA gain is 19 dB (9.5 \times) and the maximum output swing must not be exceeded; -10 dBm suffices for many experiments. The f_{4LO} input is ac-coupled to a 5 V LVDS buffer to provide an ideal interface to th[e AD8333.](http://www.analog.com/AD8333?doc=AD8333.pdf)

The f_{4LO} level is frequency dependent; refer t[o Figure 22](#page-10-0) for minimum signal levels, and then adjust the generator output level accordingly.

Figure 63. Evaluation Board Assembly

EVALUATION BOARD SCHEMATIC AND ARTWORK

Figure 64. Evaluation Board Schematic

Figure 65. Typical Board Test Connections (One Channel Shown)

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BOARD LAYOUT

The [AD8333](http://www.analog.com/AD8333?doc=AD8333.pdf) evaluation board has four layers. The interconnecting circuitry is located on the outer layers with the inner layers dedicated as power and ground planes[. Figure 66,](#page-30-1) [Figure 67,](#page-30-2) [Figure 69,](#page-30-3) an[d Figure 70](#page-30-4) illustrate the copper patterns.

Figure 70. Power Plane Copper

OUTLINE DIMENSIONS

COMPLIANT TO JEDEC STANDARDS MO-220-WHHD.

Figure 71. 32-Lead Lead Frame Chip Scale Package [LFCSP] 5 mm × 5 mm Body and 0.75 mm Package Height (CP-32-7) Dimensions shown in millimeters

ORDERING GUIDE

 $1 Z =$ RoHS Compliant Part.

 2 WP = waffle pack.

