

Si3050 Si3018/19

GLOBAL VOICE/DATA DIRECT ACCESS ARRANGEMENT

Features

- **PCM highway data interface**
- µ-law/A-law companding
- SPI control interface
- GCI interface
- 80 dB dynamic range TX/RX
- Line voltage monitor
- Loop current monitor
- +3.2 dBm TX/RX level mode
- Parallel handset detection
- 3 µA on-hook line monitor current
- Overload detection
- Programmable line interface
- \bullet AC termination
- DC termination
- Ring detect threshold
- Ringer impedance
- TIP/RING polarity detection

Applications

Description

- Computer telephony **PBX** systems
-
- Integrated codec and 2- to 4-wire analog hybrid
- Programmable digital hybrid for near-end echo reduction
- Polarity reversal detection
- \blacksquare Programmable digital gain in 0.1 dB increments
- **Integrated ring detector**
- Type I and II caller ID support
- Pulse dialing support
- Billing tone detection
- \blacksquare 3.3 V power supply
- Daisy-chaining for up to 16 devices
- Greater than 5000 V isolation
- Patented isolation technology
- Ground start and loop start support
- Available in lead-free RoHS-compliant packages
-
- Voice mail systems
- **POTS** termination equipment
- Video conferences IP telephony

The Si3050 integrated direct access arrangement (DAA) provides a programmable line interface to meet global telephone line requirements. Available in one 20-pin TSSOP and one 16-pin TSSOP/SOIC, it eliminates the need for an analog front end (AFE), an isolation transformer, relays, opto-isolators, and a 2- to 4-wire hybrid. The Si3050 dramatically reduces discrete components and the cost required to comply with global regulatory requirements. The Si3050 interfaces directly to standard telephony PCM interfaces.

Functional Block Diagram

US Patent# 5,870,046 US Patent# 6,061,009 Other Patents Pending

Si3050 Si3018/19

TABLE OF CONTENTS

Section Page

1. Electrical Specifications

Table 1. Recommended Operating Conditions

Notes:

1. The Si3050 specifications are guaranteed when the typical application circuit (including component tolerance) and any Si3050 and any Si3018/19 are used. See ["2.Typical Application Schematic" on page 17](#page-16-0) for the typical application circuit.

2. All minimum and maximum specifications are guaranteed and apply across the recommended operating conditions. Typical values apply at nominal supply voltages and an operating temperature of 25 °C unless otherwise stated.

Table 2. Loop Characteristics

(V_D = 3.0 to 3.6 V, T_A = 0 to 70 °C for K-Grade, see [Figure 1 on page 6](#page-5-0))

Table 3. DC Characteristics, V_D = 3.3 V

(V_D = 3.0 to 3.6 V, T_A = 0 to 70 °C for F/K-Grade)

Notes:

1. V_{IH}/V_{IL} do not apply to C1A/C2A.

2. All inputs at 0.4 or V_D – 0.4 (CMOS levels). All inputs are held static except clock and all outputs unloaded (Static $I_{\text{OUT}} = 0 \text{ mA}$).

3. RGDT is not functional in this state.

Table 4. AC Characteristics

(V_D = 3.0 to 3.6 V, T_A = 0 to 70 °C for F/K-Grade, Fs = 8000 Hz, see "2. Typical Application Schematic" on page 17)

Notes:

1. Measured at TIP and RING with 600 Ω termination at 1 kHz, as shown in [Figure 1 on page 6](#page-5-0).

2. With FULL = 1, the transmit and receive full-scale level of +3.2 dBm can be achieved with a 600 Ω ac termination. While the transmit and receive level in dBm varies with reference impedance, the DAA will transmit and receive 1 dBV into all reference impedances. With FULL2 = 1, the transmit and receive full-scale level of +6.0 dBm can be achieved with a 600 Ω termination. In this mode, the DAA will transmit and receive +1.5 dBV into all reference impedances.

3. Receive full-scale level produces -0.9 dBFS at DTX.

4. DR = 20 x log (RMS V_{FS}/RMS Vin) + 20 x log (RMS V_{in}/RMS noise). The RMS noise measurement excludes harmonics. Here, V_{FS} is the 0 dBm full-scale level per Note 1 above.

5. Measurement is 300 to 3400 Hz. Applies to both transmit and receive paths.

6. $\sqrt{V} = 1$ kHz, -3 dBFS.

7. THD = 20 x log (RMS distortion/RMS signal).

8. $DR_{CID} = 20$ x log (RMS V_{CID}/RMS V_{IN}) + 20 x log (RMS V_{IN}/RMS noise). V_{CID} is the 1.5 V full-scale level with the enhanced caller ID circuit. With the typical CID circuit, the V_{CID} full-scale level is 6 V peak, and the DR_{CID} decreases to 50 dB.

9. Refer to Tables [10](#page-13-0)–11 for relative gain accuracy characteristics (passband ripple).

Table 4. AC Characteristics (Continued)

(V_D = 3.0 to 3.6 V, T_A = 0 to 70 °C for F/K-Grade, Fs = 8000 Hz, see "2. Typical Application Schematic" on page 17)

Notes:

- 1. Measured at TIP and RING with 600 Ω termination at 1 kHz, as shown in Figure 1 on page 6.
- **2.** With FULL = 1, the transmit and receive full-scale level of +3.2 dBm can be achieved with a 600 Ω ac termination. While the transmit and receive level in dBm varies with reference impedance, the DAA will transmit and receive 1 dBV into all reference impedances. With FULL2 = 1, the transmit and receive full-scale level of +6.0 dBm can be achieved with a 600 Ω termination. In this mode, the DAA will transmit and receive +1.5 dBV into all reference impedances.
- **3.** Receive full-scale level produces -0.9 dBFS at DTX.
- **4.** DR = 20 x log (RMS V_{FS}/RMS Vin) + 20 x log (RMS V_{in}/RMS noise). The RMS noise measurement excludes harmonics. Here, V_{FS} is the 0 dBm full-scale level per Note 1 above.
- **5.** Measurement is 300 to 3400 Hz. Applies to both transmit and receive paths.
- **6.** $\sqrt{2}$ $\sqrt{1}$ = 1 kHz, -3 dBFS.
- **7.** THD = 20 x log (RMS distortion/RMS signal).
- **8.** $DR_{CID} = 20 x log (RMS V_{CID}/RMS V_N) + 20 x log (RMS V_{IN}/RMS noise). V_{CID} is the 1.5 V full-scale level with the$ enhanced caller ID circuit. With the typical CID circuit, the V_{ClD} full-scale level is 6 V peak, and the DR_{CID} decreases to 50 dB.
- **9.** Refer to Tables 10–11 for relative gain accuracy characteristics (passband ripple).

Table 5. Absolute Maximum Ratings

Note: Permanent device damage can occur if the above Absolute Maximum Ratings are exceeded. Functional operation should be restricted to the conditions as specified in the operational sections of this data sheet. Exposure to absolute maximum rating conditions for extended periods might affect device reliability.

Table 6. Switching Characteristics-General Inputs

(V_D = 3.0 to 3.6 V, T_A = 0 to 70 °C for K-Grade, C_L = 20 pF)

Notes:

1. All timing (except Rise and Fall time) is referenced to the 50% level of the waveform. Input test levels are

 V_{IH} = V_{D} – 0.4 V, V_{IL} = 0.4 V. Rise and Fall times are referenced to the 20% and 80% levels of the waveform.

2. FSYNC/PCLK relationship must be fixed after RESET↑.

3. The minimum RESET pulse width is the greater of 250 ns or 10 PCLK cycle times.

Figure 2. General Inputs Timing Diagram

Table 7. Switching Characteristics-Serial Peripheral Interface

(V_{IO} = 3.0 to 3.6 V, T_A = 0 to 70 °C for K-Grade, C_L = 20 pF)

Figure 3. SPI Timing Diagram

Table 8. Switching Characteristics-PCM Highway Serial Interface

(V_D = 3.0 to 3.6 V, T_A = 0 to 70 °C for K-Grade, C_L = 20 pF)

Notes:

1. All timing is referenced to the 50% level of the waveform. Input test levels are V_{IH} = V_O – 0.4 V, V_{IL} = 0.4 V, rise and fall times are referenced to the 20% and 80% levels of the waveform.

2. FSYNC must be 8 kHz under all operating conditions.

3. Specification applies to PCLK fall to DTX tri-state when that mode is selected.

Figure 4. PCM Highway Interface Timing Diagram (RXS = TXS = 1)

Table 9. Switching Characteristics-GCI Highway Serial Interface

(V_D = 3.0 to 3.6 V, T_A = 0 to 70 °C for K-Grade, C_L = 20 pF)

Notes:

1. All timing is referenced to the 50% level of the waveform. Input test levels are V_{IH} = V_O – 0.4 V, V_{IL} = 0.4 V, rise and fall times are referenced to the 20% and 80% levels of the waveform.

2. FSYNC must be 8 kHz under all operating conditions.

3. Specification applies to PCLK fall to DTX tri-state when that mode is selected.

Figure 5. GCI Highway Interface Timing Diagram (1x PCLK Mode)

Figure 6. GCI Highway Interface Timing Diagram (2x PCLK Mode)

Table 10. Digital FIR Filter Characteristics-Transmit and Receive

(V_D = 3.0 to 3.6 V, Sample Rate = 8 kHz, T_A = 0 to 70 °C for K-Grade)

Table 11. Digital IIR Filter Characteristics-Transmit and Receive

(V_D = 3.0 to 3.6 V, Sample Rate = 8 kHz, T_A = 0 to 70 °C for K-Grade)

For Figures 7-10, all filter plots apply to a sample rate of $Fs = 8$ kHz.

For Figures 11-14, all filter plots apply to a sample rate of $Fs = 8 kHz.$

Figure 13. IIR Transmit Filter Response

 $\frac{3000}{1000}$ $\frac{4000}{1000}$ $\frac{5000}{1000}$

Figure 14. IIR Transmit Filter Passband Ripple

Figure 15. IIR Receive Group Delay

SILICON LABORATORIES

7000

8000

6000

 -60

 -70

 -80 $-$

1000

2000

2. Typical Application Schematic

3. Bill of Materials

Notes:

1. Value for C3 above is recommended for use with the Si3019 in voice applications to improve return loss performance. When using the Si3018, a value of 10 nF (250 V, 20%) can be substituted.

2. R7-R8 may be substituted for R30-R33 and C30-C31 for lower cost, but reduced CID performance.

3. Several diode bridge configurations are acceptable. Parts, such as a single HD04, a DF-04S, or four 1N4004 diodes, may be used (suppliers include General Semiconductor, Diodes Inc., etc.).

4. 56 Ω, 1/16, 1% resistors may be substituted for R12–R13 (0 Ω) to decrease emissions. (See AN81.)

5. Murata BLM18AG601SN1 may be substituted for R15–R16 (0 Ω) to decrease emissions. (See AN81.)

4. AOUT PWM Output

[Figure 18](#page-18-1) illustrates an optional circuit to support the pulse width modulation (PWM) output capability of the Si3050 for call progress monitoring purposes.To enable this mode, the INTE bit (Register 2) should be set to 0, the PWME bit (Register 1) set to 1, and the PWMM bits (Register 2) set to 00.

Figure 18. AOUT PWM Circuit for Call Progress

Component	Value	Supplier		
LS1	Speaker BRT1209PF-06	Intervox		
O6	NPN KSP13	Fairchild		
C41	0.1 µF, 16 V, X7R, ±20%	Venkel, SMEC		
R41	150 Ω , 1/10 W, ±5%	Venkel, SMEC, Panasonic		

Table 12. Component Values-AOUT PWM

Registers 20 and 21 allow the receive and transmit paths to be attenuated linearly. When these registers are set to all 0s, the transmit and receive paths are muted. These registers affect the call progress output only and do not affect transmit and receive operations on the telephone line.

The PWMM[1:0] bits (Register 1, bits 5:4) select one of three different PWM output modes for the AOUT signal, including a delta-sigma data stream, a 32 kHz return to 0 PWM output, and a balanced 32 kHz PWM output.

5. Functional Description

The Si3050 is an integrated direct access arrangement (DAA) providing a programmable line interface that meets global telephone line requirements. The Si3050 implements Silicon Laboratories' patented isolation capacitor technology, which offers the highest level of integration by replacing an analog front end (AFE), an isolation transformer, relays, opto-isolators, and a 2- to 4-wire hybrid with two thin small shrink outline packages (TSSOPs).

The Si3050 DAA is fully software programmable to meet global requirements and is compliant with FCC, TBR21, JATE, and other country-specific PTT specifications as shown in [Table 13.](#page-20-0) In addition, the Si3050 meets the most stringent global requirements for out-of-band energy, emissions, immunity, high-voltage surges, and safety, including FCC Parts 15 and 68, EN55022, EN55024, and many other standards.

5.1. Line-Side Device Support

Two different line-side devices can be used with the Si3050 system-side device. Both line-side devices support the following features:

- Global compliance.
- Selectable 5 Hz or 200 Hz RX low-pass filter pole.
- -16.5 to 13.5 dB digital gain/attenuation adjustment in 0.1 dB increments for the transmit and receive paths.

5.1.1. Si3018

- Globally-compliant line-side device-targets global DAA requirements for voice/telephony applications. This line-side device supports both FCC-compliant countries and non-FCC-compliant countries.
	- Selectable dc terminations.
	- Four selectable ac terminations to increase return loss and trans-hybrid loss performance.

5.1.2. Si3019

- Globally-compliant, enhanced features line-side device—targets global DAA requirements for voice/ telephony applications. The Si3019 contains all the features available on the Si3018, plus the following additional features/enhancements:
	- Sixteen selectable ac terminations to further increase return loss and trans-hybrid loss performance.
	- Line voltage monitoring in on- and off-hook modes to enable line in-use/parallel handset detection.
	- Programmable line current / voltage threshold interrupt.
	- Polarity reversal interrupt.
	- +3.2 dBm TX/RX level mode (600 Ω)
	- Higher resolution (1.1 mA/bit) loop current measurement.

Table 13. Country Specific Register Settings

Note:

1. Supported for loop current ≥ 20 mA.

2. TBR21 includes the following countries: Austria, Belgium, Denmark, Finland, France, Germany, Greece, Iceland, Ireland, Italy, Luxembourg, Netherlands, Norway, Portugal, Spain, Sweden, Switzerland, and the United Kingdom.

3. See ["5.16.DC Termination" on page 30](#page-29-0) for DCV and MINI settings.

4. ACIM is 0000 for data applications and 1010 for voice applications.

Register	16	31	16	16	26	26	26	30
Country	OHS	OHS ₂	RZ	RT	ILIM	DCV[1:0]	MINI[1:0]	ACIM[3:0]
Indonesia	$\mathbf 0$	$\pmb{0}$	$\pmb{0}$	$\boldsymbol{0}$	$\pmb{0}$	11	00	0000
Ireland	$\mathbf 0$	$\mathbf 1$	$\mathbf 0$	$\mathbf 0$	1	11	00	0010
Israel	$\pmb{0}$	$\mathbf 1$	$\mathbf 0$	$\mathbf 0$	1	11	00	0010
Italy	$\mathbf 0$	$\mathbf 1$	$\pmb{0}$	$\mathbf 0$	1	11	00	0010
Japan	$\mathbf 0$	$\mathbf 0$	$\pmb{0}$	$\mathbf 0$	$\mathbf 0$	01	01	0000
Jordan	$\mathbf 0$	$\pmb{0}$	$\pmb{0}$	$\mathbf 0$	$\pmb{0}$	01	01	0000
Kazakhstan	$\mathbf 0$	$\pmb{0}$	$\mathbf 0$	$\mathbf 0$	$\pmb{0}$	11	00	0000
Kuwait	$\mathbf 0$	$\pmb{0}$	$\pmb{0}$	$\mathbf 0$	$\pmb{0}$	11	00	0000
Latvia	$\pmb{0}$	$\mathbf 1$	$\pmb{0}$	$\mathbf 0$	1	11	00	0010
Lebanon	$\pmb{0}$	$\mathbf 1$	$\pmb{0}$	$\mathbf 0$	1	11	00	0010
Luxembourg	$\pmb{0}$	$\mathbf 1$	$\pmb{0}$	$\mathbf 0$	1	11	00	0010
Macao	$\mathbf 0$	$\pmb{0}$	$\pmb{0}$	$\mathbf 0$	$\pmb{0}$	11	00	0000
Malaysia ¹	$\mathbf 0$	$\pmb{0}$	$\mathbf 0$	$\mathbf 0$	$\pmb{0}$	01	01	0000
Malta	$\pmb{0}$	$\mathbf 1$	$\pmb{0}$	$\mathbf 0$	1	11	00	0010
Mexico	$\mathbf 0$	$\pmb{0}$	$\pmb{0}$	$\mathbf 0$	$\pmb{0}$	11	00	0000
Morocco	$\pmb{0}$	$\mathbf 1$	$\pmb{0}$	$\mathbf 0$	1	11	00	0010
Netherlands	$\pmb{0}$	$\mathbf 1$	$\pmb{0}$	$\mathbf 0$	1	11	00	0010
New Zealand	$\pmb{0}$	$\pmb{0}$	$\pmb{0}$	$\mathbf 0$	$\pmb{0}$	11	00	0100
Nigeria	$\mathbf 0$	$\mathbf 1$	$\mathbf 0$	$\mathbf 0$	1	11	00	0010
Norway	$\pmb{0}$	1	$\mathbf 0$	$\mathbf 0$	1	11	00	0010
Oman	$\mathbf 0$	$\pmb{0}$	$\pmb{0}$	$\mathbf 0$	$\pmb{0}$	01	01	0000
Pakistan	$\pmb{0}$	$\pmb{0}$	$\pmb{0}$	$\pmb{0}$	$\pmb{0}$	01	01	0000
Peru	$\mathbf 0$	$\mathbf 0$	$\mathsf{O}\xspace$	$\mathbf 0$	0	11	$00\,$	0000
Philippines	$\mathbf 0$	$\mathbf 0$	$\pmb{0}$	$\pmb{0}$	$\pmb{0}$	01	01	0000
Poland	$\pmb{0}$	$\mathbf{1}$	$\mathsf 0$	$\mathbf 0$	1	11	00	0010
Portugal	$\mathbf 0$	$\mathbf{1}$	$\pmb{0}$	$\mathbf 0$	1	11	$00\,$	0010
Romania	$\pmb{0}$	$\mathbf{1}$	$\pmb{0}$	$\mathbf 0$	$\overline{1}$	11	$00\,$	0010
Russia	$\pmb{0}$	$\pmb{0}$	$\pmb{0}$	$\pmb{0}$	$\pmb{0}$	11	$00\,$	0000

Table 13. Country Specific Register Settings (Continued)

Note:

1. Supported for loop current ≥ 20 mA.

2. TBR21 includes the following countries: Austria, Belgium, Denmark, Finland, France, Germany, Greece, Iceland, Ireland, Italy, Luxembourg, Netherlands, Norway, Portugal, Spain, Sweden, Switzerland, and the United Kingdom.

3. See "5.16.DC Termination" on page 30 for DCV and MINI settings.

4. ACIM is 0000 for data applications and 1010 for voice applications.

Register	16	31	16	16	26	26	26	30
Country	OHS	OHS ₂	RZ	RT	ILIM	DCV[1:0]	MINI[1:0]	ACIM[3:0]
Saudi Arabia	$\mathbf 0$	$\mathbf 0$	0	0	0	11	00	0000
Singapore	$\mathbf 0$	0	0	$\mathbf 0$	0	11	00	0000
Slovakia	$\overline{0}$	1	0	$\mathbf 0$	1	11	00	0010
Slovenia	$\mathbf 0$	1	0	$\mathbf 0$	1	11	00	0010
South Africa	$\overline{0}$	0	1	$\mathbf 0$	$\mathbf 0$	11	00	0011
South Korea	$\mathbf 0$	0	1	$\mathbf 0$	0	11	00	0000
Spain	$\mathbf 0$	1	0	$\mathbf 0$	1	11	00	0010
Sweden	$\mathbf 0$	1	0	$\mathbf 0$	1	11	00	0010
Switzerland	$\mathbf 0$	1	0	$\mathbf 0$	1	11	00	0010
Taiwan	$\mathbf 0$	0	0	$\mathbf 0$	0	11	00	0000
TBR21 ²	$\mathbf 0$	$\mathbf 0$	0	$\mathbf 0$	1	11	00	0010
Thailand	$\mathbf 0$	$\mathbf 0$	0	$\mathbf 0$	$\overline{0}$	01	01	0000
UAE	$\mathbf 0$	$\mathbf 0$	0	$\mathbf 0$	$\mathbf 0$	11	00	0000
United Kingdom	$\overline{0}$	1	0	$\mathbf 0$	1	11	00	0101
USA	$\mathbf 0$	$\mathbf 0$	$\mathbf 0$	$\mathbf 0$	$\mathbf 0$	11	00	0000
Yemen	$\mathbf 0$	0	0	$\mathbf 0$	0	11	00	0000

Table 13. Country Specific Register Settings (Continued)

Note:

1. Supported for loop current ≥ 20 mA.

2. TBR21 includes the following countries: Austria, Belgium, Denmark, Finland, France, Germany, Greece, Iceland, Ireland, Italy, Luxembourg, Netherlands, Norway, Portugal, Spain, Sweden, Switzerland, and the United Kingdom.

3. See "5.16.DC Termination" on page 30 for DCV and MINI settings.

4. ACIM is 0000 for data applications and 1010 for voice applications.

5.2. Power Supplies

The Si3050 operates from a $3.0-3.6$ V power supply. The Si3050 input pins can only accept 3.3 V CMOS signal levels. If support of 5 V signal levels is necessary, a level shifter is required. The Si3018/19 derives its power from two sources: the Si3050 and the telephone line. The Si3050 supplies power over the patented isolation capacitor link between the two devices, allowing the Si3019 to communicate with the Si3050 while on-hook and perform other on-hook functions, such as line voltage monitoring. When off-hook, the Si3018/19 also derives power from the line current supplied from the telephone line. This feature is exclusive to DAAs from Silicon Labs and allows the most cost-effective implementation for a DAA while still maintaining robust performance over all line conditions.

5.3. Initialization

Each time the Si3050 is powered up, assert the RESET pin. When the RESET pin is deasserted, the registers have default values to guarantee the line-side device (Si3018/19) is powered down without the possibility of loading the line (i.e., off-hook). An example initialization procedure follows:

- 1. Supply the appropriate PCLK and FSYNC signals to the Si3050.
- 2. Wait until the PLL is locked. This time is less than 1 ms from the application of PCLK.
- 3. Write an 00H into Register 6 to power up the line-side device (Si3018/19).
- 4. Set the desired line interface parameters (i.e., DCV[1:0], MINI[1:0], ILIM, DCR, ACIM[3:0], OHS, RT, RZ, TGA2, and TXG2[3:0]) shown in [Table 13 on](#page-20-0) [page 21](#page-20-0).
- 5. For voice applications, set the FULL (or FULL2) + IIRE bits.

When this procedure is complete, the Si3018/19 is ready for ring detection and off-hook operation.

5.4. Isolation Barrier

The Si3050 achieves an isolation barrier through low-cost, high-voltage capacitors in conjunction with Silicon Laboratories' patented signal processing techniques. Differential capacitive communication eliminates signal degradation from capacitor mismatches, common mode interference, or noise coupling. As shown in the ["2.Typical Application](#page-16-0) [Schematic" on page 17,](#page-16-0) the C1, C2, C8, and C9 capacitors isolate the Si3050 (system-side) from the Si3018/19 (line-side). Transmit, receive, control, ring detect, and caller ID data are passed across this barrier.

The communications link is disabled by default. To enable it, the PDL bit (Register 6, bit 4) must be cleared. No communication between the Si3050 and Si3018/19 can occur until this bit is cleared. Allow the PLL to lock to the PCLK and FSYNC input signals before clearing the PDL bit.

5.5. Power Management

The Si3050 supports four basic power management operation modes. The modes are normal operation, reset operation, sleep mode, and full powerdown mode. The power management modes are controlled by the PDN and PDL bits (Register 6).

On powerup, or following a reset, the Si3050 is in reset operation. The PDL bit is set, and the PDN bit is cleared. The Si3050 is operational, except for the communications link. No communication between the Si3050 and line-side device (Si3018/19) can occur during reset operation. Bits associated with the line-side device are invalid in this mode.

In typical applications, the DAA will predominantly be operated in normal mode. In normal mode, the PDL and PDN bits are cleared. The DAA is operational and the communications link passes information between the Si3050 and the Si3018 or Si3019.

The Si3050 supports a low-power sleep mode that supports ring validation and wake-up-on-ring features. To enable the sleep mode, the PDN bit must be set. When the Si3050 is in sleep mode, the PCLK signal must remain active. In low-power sleep mode, the Si3050 is non-functional except for the communications link and the RGDT signal. To take the Si3050 out of sleep mode, pulse the reset pin (RESET) low.

In summary, the powerdown/up sequence for sleep mode is as follows:

- 1. Ensure the PDL bit (Register 6, bit 4) is cleared.
- 2. Set the PDN bit (Register 6, bit 3).
- 3. The device is now in sleep mode. PCLK must remain active.
- 4. To exit sleep mode, reset the Si3050 by pulsing the RESET pin.
- 5. Program registers to desired settings.

The Si3050 also supports an additional Powerdown mode. When both the PDN (Register 6, bit 3) and PDL (Register 6, bit 4) bits are set, the chipset enters a complete powerdown mode and draws negligible current (deep sleep mode). In this mode, the Si3050 is non-functional. The RGDT pin does not function and the Si3050 will not detect a ring. Normal operation can be restored using the same process for taking the Si3050 out of sleep mode.

5.6. Calibration

The Si3050 initiates two auto-calibrations by default when the device goes off-hook or experiences a loss of line power. A 17 ms resistor calibration is performed to allow circuitry internal to the DAA to adjust to the exact line conditions present at the time of going off-hook. This resistor calibration can be disabled by setting the RCALD bit (Register 25, bit 5). A 256 ms ADC calibration is also performed to remove offsets that might be present in the on-chip A/D converter, which could affect the A/D dynamic range. The ADC auto-calibration is initiated after the DAA dc termination stabilizes and the resistor calibration completes. Due to the large variation in line conditions and line card behavior presented to the DAA, it might be beneficial to use manual ADC calibration instead of auto-calibration.

Manual ADC calibration should be executed as close as possible to 256 ms before valid transmit/receive data is expected.

The following steps should be taken to implement manual ADC calibration:

- 1. The CALD bit (auto-calibration disable—Register 17) must be set to 1.
- 2. The MCAL bit (manual calibration) must be toggled to one and then 0 to begin and complete the calibration.
- 3. The calibration is completed in 256 ms.

5.7. In-Circuit Testing

The Si3050ís advanced design provides the designer with an increased ability to determine system functionality during production line tests and support for end-user diagnostics. Six loopback modes allow increased coverage of system components. For four of the test modes, a line-side power source is needed. While a standard phone line can be used, the test circuit in [Figure 1 on page 6](#page-5-0) is adequate. In addition, an off-hook sequence must be performed to connect the power source to the line-side device.

For the start-up loopback test mode, no line-side power is necessary, and no off-hook sequence is required. The start-up test mode is enabled by default. When the PDL bit (Register 6, bit 4) is set (the default case), the line side is in a powerdown mode, and the system-side is in a digital loopback mode. In this mode, data received on DRX passes through the internal filters and is transmitted on DTX. This path introduces approximately 0.9 dB of attenuation on the DRX signal received. The group delay of both transmit and receive filters exists between DRX and DTX. Clearing the PDL bit disables this mode, and the DTX data switches to the receive data from the line side. When the PDL bit is cleared, the

FDT bit (Register 12, bit 6) becomes active to indicate that successful communication between the line side and system side is established. This provides verification that the communications link is operational.

The digital data loop-back mode offers a way to input data on the DRX pin and have the identical data output on the DTX pin through bypassing the transmit and receive filters. Setting the DDL bit (Register 10, bit 0) enables this mode, which provides an easy way to verify communication between the host processor/DSP and the DAA. No line-side power or off-hook sequence is required for this mode.

The remaining test modes require an off-hook sequence to operate. The following sequence lists the off-hook requirements:

- 1. Powerup or reset.
- 2. Allow the internal PLL to lock on PCLK and FSYNC.
- 3. Enable line-side by clearing PDL bit.
- 4. Issue an off-hook command.
- 5. Delay 402.75 ms for calibration to occur.
- 6. Set desired test mode.

The communications link digital loopback mode allows the host processor to provide a digital input test pattern on DRX and receive that digital test pattern back on DTX. To enable this mode, set the IDL bit (Register 1, bit 1). The communications link is tested in this mode. The digital stream is delivered across the isolation capacitors, C1 and C2, of the ["2.Typical Application](#page-16-0) [Schematic" on page 17,](#page-16-0) to the line-side device and returned across the same path. In this digital loopback mode, the 0.9 dB attenuation and filter group delays also exist.

The PCM analog loopback mode extends the signal path of the analog loopback mode. In this mode, an analog signal is driven from the line into the line-side device. This analog signal is converted to digital data and then passed across the communications link to the system-side device. The data passes through the receive filter, through the transmit filter, and is then passed across the communications link and sent back out onto the line as an analog signal. Set the PCML bit (Register 33, bit 7) to enable this mode.

With the final testing mode, internal analog loopback, the system can test the operation of the transmit and receive paths on the line-side device and the external components in the ["2.Typical Application Schematic" on](#page-16-0) [page 17](#page-16-0). The host provides a digital test waveform on DRX. Data passes across the isolation barrier, is transmitted to and received from the line, passes back across the isolation barrier, and is presented to the host on DTX. Clear the HBE bit (Register 2, bit 1) to enable this mode.

When the HBE bit is cleared, it produces a dc offset that affects the signal swing of the transmit signal. Silicon Laboratories recommends that the transmit signal be 12 dB lower than normal transmit levels. A lower level eliminates clipping from the dc offset that results from disabling the hybrid. It is assumed in this test that the line ac impedance is nominally 600 Ω .

Note: All test modes are mutually exclusive. If more than one test mode is enabled concurrently, the results are unpredictable.

5.8. Exception Handling

The Si3050 can determine if an error occurs during operation. Through the secondary frames of the serial link, the controlling DSP can read several status bits. The bit of highest importance is the frame detect bit (FDT, Register 12, bit 6) which indicates that the system-side (Si3050) and line-side (Si3018 or Si3019) devices are communicating. During normal operation, the FDT bit can be checked before reading the bits that indicate information about the line side. If FDT is not set, the following bits related to the line side are invalid—RDT, RDTN, RDTP, LCS[4:0], LSID[1:0], REVB[3:0], LVS[7:0], LCS2[7:0], ROV, BTD, DOD, and OVL; the RGDT operation is also non-functional.

Following powerup and reset, the FDT bit is not set because the PDL bit (Register 6 bit 4) defaults to 1. In this state, the ISOcap is not operating and no information about the line side can be determined. The user must provide a valid PCLK and FSYNC to the system and clear the PDL bit to activate the ISOcap link. Communication with the line-side device takes less than 10 ms to establish.

The FDT bit also can indicate if the line side executes an off-hook request successfully. If the line side is not connected to a phone line, the FDT bit remains cleared. The controlling DSP must provide sufficient time for the line side to execute the off-hook request. The maximum time for FDT to be valid following an off-hook request is 10 ms. If the FDT bit is high, the LCS[4:0] bits indicate the amount of loop current flowing. If the FDT is not set following an off-hook request, the PDL bit (Register 6, bit 4) must be set high for at least 1 ms to reset the line side.

5.9. Revision Identification

The Si3050 provides information to determine the revision of the Si3050 and/or the Si3018/19. The REVA[3:0] bits (Register 11) identify the revision of the Si3050. The REVB[3:0] bits (Register 13) identify the revision of the line-side device. [Table 14](#page-25-4) lists revision values and might contain future revisions not yet in existence.

Table 14. Revision Values

Revision	Si3050	Si3018/19
A	0000	0001
R	0010	0010
C	0011	0011
	0100	0100

5.10. Transmit/Receive Full-Scale Level (Si3019 Line-Side Only)

The Si3050 supports programmable maximum transmit and receive levels. The default signal level supported by the Si3050 is 0 dBm into a 600 Ω load. Two additional modes of operation offer increased transmit and receive level capability to enable use of the DAA in applications that require higher signal levels. The full-scale mode is enabled by setting the FULL bit in Register 31. With FULL = 1, the full-scale signal level increases to +3.2 dBm into a 600 Ω load or 1 dBV into all reference impedances. The enhanced full-scale mode (or 2x full scale) is enabled by setting the FULL2 bit in Register 30. With FULL2 = 1, the full-scale signal level increases to +6.0 dBm into a 600 Ω load or 1.5 dBV into all reference impedances. The full-scale and enhanced full-scale modes provide the ability to trade off TX power and TX distortion for a peak signal. By using the programmable digital gain registers in conjunction with the enhanced full-scale signal level mode, a specific power level (+3.2 dBm for example) can be achieved across all ACT settings.

5.11. Parallel Handset Detection

The Si3050 can detect a parallel handset going off-hook. When the Si3050 is off-hook, the loop current can be monitored with the LCS or LCS2 bits. A significant drop in loop current signals a parallel handset going off-hook. If a parallel handset going off-hook causes the loop current to drop to 0, the LCS and LCS2 bits will read all 0s. Additionally, the Drop-Out Detect (DOD) bit will fire (and generate an interrupt if the DODM bit is set) indicating that the line-derived power supply has collapsed.

With the Si3019 line side, the LVS bits also can be read when on- or off-hook to determine the line voltage. Significant drops in line voltage can signal a parallel handset. For the Si3050 to operate in parallel with another handset, the parallel handset must have a sufficiently high dc termination to support two off-hook DAAs on the same line. Improved parallel handset operation can be achieved by changing the dc

impedance from 50 Ω to 800 Ω and reducing the DCT pin voltage with the DCV[1:0] bits.

5.12. Line Voltage/Loop Current Sensing

The Si3050 can measure loop current with either the Si3018 or the Si3019 line-side device. The 5-bit LCS[4:0] register reports loop current measurements when off-hook. The Si3019 offers an additional register to report loop current to a finer resolution (LCS2[7:0]). The Si3050 can only measure line voltage when used with the Si3019 line-side device. The LVS[7:0] register available with the Si3019 monitors voltage both on and off-hook. These registers can be used to help determine the following line conditions:

- When on-hook, detect if a line is connected.
- When on-hook, detect if a parallel phone is off-hook.
- When off-hook, detect if a parallel phone goes on or off-hook.
- Detect if enough loop current is available to operate.
- When used in conjunction with the OPD bit, detect if an overload condition exists. (See ["5.27.Overload](#page-35-0) [Detection" on page 36.](#page-35-0))

5.12.1. Line Voltage Measurement (Si3019 Line Side Only)

The Si3050 reports line voltage with the LVS[7:0] bits (Register 29) in both on- and off-hook states with a resolution of 1 V per bit. The accuracy of these bits is approximately ±10%. Bits 0 through 7 of this 8-bit signed number indicate the value of the line voltage in 2s complement format. Bit 7 indicates the polarity of the TIP/RING voltage.

If the INTE bit (Register 2, bit 7) and the POLM bit (Register 3, bit 0) are set, a hardware interrupt is generated on the AOUT/INT pin when Bit 7 of the LVS register changes state. The edge-triggered interrupt is cleared by writing 0 to the POLI bit (Register 4, bit 0). The POLI bit is set each time bit 7 of the LVS register changes state, and must be written to 0 to clear it. The default state of the LVS register forces the LVS[7:0] bits to 0 when the line voltage is 3 V or less. The LVFD bit (Register 31, bit 0) disables this force-to-zero function and allows the LVS register to display non-zero values of 3 V and below. This register may display unpredictable values at line voltages between 0 to 2 V. At 0 V, the LVS register displays all 0s.

Figure 19. Typical Loop Current LCS Transfer Function (ILIM = 0)

5.12.2. Loop Current Measurement

When the Si3050 is off-hook, the LCS[4:0] bits measure loop current in 3.3 mA/bit resolution. With the LCS[4:0] bits, a user can detect another phone going off-hook by monitoring the dc loop current. The line current sense transfer function is shown in [Figure 19](#page-26-1) and is detailed in [Table 15.](#page-27-2) The LCS and LCS2 bits report loop current down to the minimum operating loop current for the DAA. Below this threshold, the reported value of loop current is unpredictable. The minimum operating loop current of the DAA is set by the MINI[1:0] bits in Register 26.

When the LCS bits reach max value, the Loop Current Sense Overload Interrupt bit (Register 4) fires. LCSOI firing however, does not necessarily imply that an overcurrent situation has occurred. An overcurrent situation in the DAA is determined by the status of the OPD bit (Register 19). After the LCSOI interrupt fires, the OPD bit should be checked to determine if an overcurrent situation exists. The OPD bit indicates an overcurrent situation when loop current exceeds either 160 mA (ILIM = 0) or 60 mA (ILIM = 1), depending on the setting of the ILIM bit (Register 26).

Table 15. Loop Current Transfer Function

The LCS2 register also reports loop current in the off-hook state. This register has a resolution of 1.1 mA per bit.

5.13. Off-Hook

The communication system generates an off-hook command by setting the OH bit (Register 5, bit 0). This off-hook state seizes the line for incoming/outgoing calls. It also can be used for pulse dialing.

With the OH bit at logic 0, negligible dc current flows through the hookswitch. When a logic 1 is written to the OH bit, the hookswitch transistor pair, Q1 and Q2, turn on. A termination impedance across TIP and RING is applied and causes dc loop current to flow. The termination impedance has both an ac and a dc component.

Several events occur in the DAA when the OH bit is set. There is a 250 μ s latency for the off-hook command to be communicated to the line-side device. When the line-side device goes off-hook, an off-hook counter forces a delay to allow line transients to settle before transmission or reception can occur. The off-hook counter time is controlled by the FOH[1:0] bits (Register 31, bits 6:5). The default setting for the off-hook counter time is 128 ms, but can be adjusted up to 512 ms or down to 64 or 8 ms.

After the off-hook counter expires, a resistor calibration is performed for 17 ms to allow the DAA internal circuitry to adjust to the exact conditions present at the time of going off-hook. This resistor calibration can be disabled by setting the RCALD bit (Register 25, bit 5).

After the resistor calibration is performed, an ADC calibration is performed for 256 ms. This calibration helps to remove offset in the A/D sampling the telephone line. ADC calibration can be disabled by setting the CALD bit (Register 17, bit 5). See ["5.6.Calibration" on page 25](#page-24-0) for more information on automatic and manual calibration.

Silicon Laboratories recommends that the resistor and the ADC calibrations not be disabled except when a fast response is needed after going off-hook, such as when responding to a Type II Caller-ID signal. See ["5.26.Caller ID" on page 34](#page-33-2) for detailed information.

To calculate the total time required to go off-hook and start transmission or reception, include the digital filter delay (typically 1.5 ms with the FIR filter) in the calculation.

5.14. Ground Start Support

The Si3050 DAA supports loop-start applications by default. It can also support ground-start applications with the RG, TGD, and TGDE pins and the schematic shown in [Figure 20](#page-28-1). The component values are listed in [Table 16](#page-28-2).

Figure 20. Typical Application Circuit for Ground Start Support on the SI3050

5.14.1. Ground Start Idle

Ensure the relay in series with TIP is closed by clearing the TGOE bit (Register 32, bit 1). This enables the DAA to sense if the CO grounds TIP. Set RG to 1 (Register 32, bit 0) so that no current flows through the relay connecting RING to ground.

5.14.2. DAA Requests Line Seizure

With TGOE set to zero, seize the line by closing the relay in series with RING (clear the RG bit, Register 32, bit 0). The CO detects this current flowing on RING and grounds TIP. This sets the TGD bit (Register 32, bit 2). The DAA may then be taken

off-hook and the relay in series with RING opened (clear the RG bit). The call continues as in loop-start mode.

5.14.3. CO Requests Line Seizure

In a normal on-hook state, the relay in series with TIP should be closed, connecting the -24 V isolated supply. The CO grounds TIP to request line seizure, causing current to flow. The opto-isolator U3 (see [Figure 20 on](#page-28-1) [page 29](#page-28-1)) detects this current and sets the TGD bit (Register 32, bit 2). This bit remains high as long as current is detected. The TGDI bit (Register 4, bit 1) is a sticky bit, and remains high until cleared. A hardware interrupt on the AOUT/INT can be made to occur when TIP current begins to flow by enabling the TGDM bit (Register 3, bit 1). Clear the interrupt by writing 0 to the TGDI bit (Register 4 bit 1). The DAA may then be taken off-hook and the call continued as in loop-start mode.

5.15. Interrupts

The AOUT/INT pin can be used as a hardware interrupt pin by setting the INTE bit (Register 2, bit 7). When this bit is set, the analog output used for call progress monitoring is not available. The default state of this interrupt output pin is active low, but active high operation can be enabled by setting the INTP bit (Register 2, bit 6). This pin is an open-drain output when the INTE bit is set and requires a 4.7 kΩ pullup or pulldown for correct operation. If multiple INT pins are connected to a single input, the combined pullup or pulldown resistance should equal 4.7 k Ω . Bits 7-0 in Register 3 and bit 1 in Register 44 can be set to enable hardware interrupt sources (bit 0 is available with the Si3019 line-side only). (When one or more of these bits are set, the AOUT/INT pin goes into an active state and stays active until the interrupts are serviced. If more than one hardware interrupt is enabled in Register 3, use software polling to determine the cause of the interrupts. Register 4 and bit 3 of Register 44 contain sticky interrupt flag bits. Clear these bits after servicing the interrupt.

Registers 43 and 44 contain the line current/voltage threshold interrupt. These line current/voltage registers and interrupt are only available with the Si3019 line-side device. This interrupt is triggered when the measured line voltage or current in the LVS or LCS2 registers, as selected by the CVS bit (Register 44, bit 2), crosses the threshold programmed into the CVT[7:0] bits. With the CVP bit, the interrupt can be programmed to occur when the measured value rises above or falls below the threshold. Only the magnitude of the measured value is used for comparison to the threshold programmed into the CVT[7:0] bits. Therefore, only positive numbers should be used as a threshold.

5.16. DC Termination

The DAA has programmable settings for the dc impedance, current limiting, minimum operational loop current and TIP/RING voltage. The dc impedance of the DAA is normally represented with a 50 Ω slope as shown in [Figure 21,](#page-29-3) but can be changed to an 800 Ω slope by setting the DCR bit. This higher dc termination presents a higher resistance to the line as loop current increases.

For applications requiring current limiting per the TBR21 standard, the ILIM bit may be set to select this mode. In this mode, the dc I/V curve is changed to a 2000 Ω slope above 40 mA, as shown in [Figure 22.](#page-29-4) This allows the DAA to operate with a 50 V, 230 Ω feed, which is the maximum linefeed specified in the TBR21 standard.

The MINI[1:0] bits select the minimum operational loop current for the DAA, and the DCV[1:0] bits adjust the DCT pin voltage, which affects the TIP/RING voltage of the DAA. These bits allow important trade-offs to be made between signal headroom and minimum operational loop current. Increasing TIP/RING voltage increases signal headroom, whereas decreasing the TIP/RING voltage allows compliance to PTT standards in low-voltage countries, such as Japan. Increasing the minimum operational loop current above 10 mA also increases signal headroom and prevents degradation of the signal level in low-voltage countries.

Finally, Australia has separate dc termination requirements for line seizure versus line hold. Japan mode may be used to satisfy both requirements. However, if a higher transmit level for modem operation is desired, switch to FCC mode 500 ms after the initial off-hook. This satisfies the Australian dc termination requirements.

5.17. AC Termination (Si3018 Line-Side Only)

The Si3050 has four ac termination impedances when used with the Si3018 line-side device. The ACIM[3:0] bits in Register 30 are used to select the ac impedance setting on the Si3018. The four available settings for the Si3018 are listed in [Table 17](#page-29-5). If an ACIM[3:0] setting other than the four listed in [Table 17](#page-29-5) is selected, the ac termination is forced to 600 Ω (ACIM[3:0] = 0000). The programmable digital hybrid can be used to further reduce near-end echo for each of the four listed ac termination settings. See ["5.29.Transhybrid Balance" on](#page-36-0) [page 37](#page-36-0) for details.

Table 17. AC Termination Settings for the Si3018 Line-Side Device

5.18. AC Termination (Si3019 Line-Side Only)

The Si3050 has sixteen real and complex ac termination impedances when used with the Si3019 line-side device. The ACIM[3:0] bits in Register 30 are used to select the ac impedance setting on the Si3019. The sixteen available settings for the Si3019 are listed in [Table 18.](#page-30-1)

The most widely used ac terminations are available as register options to satisfy various global PTT requirements. The real 600 Ω impedance satisfies the requirements of FCC Part 68, JATE, and other country requirements. The 270 Ω + (750 Ω || 150 nF) satisfies the requirements of TBR21.

There are two selections useful for satisfying non-standard ac termination requirements. The 350 Ω + (1000 Ω || 210 nF) impedance selection in Register 30 is the ANSI/EIA/TIA 464 compromise impedance network for trunks. The last ac termination selection, $ACIM[3:0] = 1111$, is designed to satisfy minimum return loss requirements for every country that requires a complex termination. By selecting this setting, the system is ensured to meet minimum PTT requirements.

For each of the sixteen ac termination settings, the programmable digital hybrid can be used to further reduce near-end echo. See ["5.29.Transhybrid Balance"](#page-36-0) [on page 37](#page-36-0) for details.

Table 18. AC Termination Settings for the Si3019 Line-Side Device

5.19. Ring Detection

The ring signal is resistively coupled from TIP and RING to the RNG1 and RNG2 pins. The Si3050 supports either full- or half-wave ring detection. With full-wave ring detection, the designer can detect a polarity reversal of the ring signal. See "5.26.Caller ID" on [page 34.](#page-33-2) The ring detection threshold is programmable with the RT bit (Register 16, bit 0) and RT2 bit (Register 17, bit 4). The ring detector output can be monitored in three ways. The first method uses the RGDT pin. The second method uses the register bits, RDTP, RDTN, and RDT (Register 5). The final method uses the DTX output.

The ring detector mode is controlled by the RFWE bit (Register 18, bit 1). When the RFWE bit is 0 (default mode), the ring detector operates in half-wave rectifier mode. In this mode, only positive ring signals are detected. A positive ring signal is defined as a voltage greater than the ring threshold across RNG1-RNG2. Conversely, a negative ring signal is defined as a voltage less than the negative ring threshold across RNG1-RNG2. When the RFWE bit is 1, the ring detector operates in full-wave rectifier mode. In this mode, both positive and negative ring signals are detected.

The first method to monitor ring detection output uses the RGDT pin. When the RGDT pin is used, it defaults to active low, but can be changed to active high by setting the RPOL bit (Register 14, bit 1). This pin is an open-drain output, and requires a 4.7 kΩ pullup or pulldown for correct operation. If multiple RGDT pins are connected to a single input, the combined pullup or pulldown resistance should equal 4.7 kΩ.

When the RFWE bit is 0, the RGDT pin is asserted when the ring signal is positive, which results in an output signal frequency equal to the actual ring frequency. When the RFWE bit is 1, the RGDT pin is asserted when the ring signal is positive or negative. The output then appears to be twice the frequency of the ring waveform.

The second method to monitor ring detection uses the ring detect bits (RDTP, RDTN, and RDT). The RDTP and RDTN behavior is based on the RNG1-RNG2 voltage. When the signal on RNG1-RNG2 is above the positive ring threshold, the RDTP bit is set. When the signal on RNG1-RNG2 is below the negative ring threshold, the RDTN bit is set. When the signal on RNG1-RNG2 is between these thresholds, neither bit is set.

Si3050 Si3018/19

The RDT behavior is also based on the RNG1-RNG2 voltage. When the RFWE bit is 0, a positive ring signal sets the RDT bit for a period of time. When the RFWE bit is 1, a positive or negative ring signal sets the RDT bit.

The RDT bit acts like a one shot. When a new ring signal is detected, the one shot is reset. If no new ring signals are detected prior to the one shot counter reaching 0, then the RDT bit clears. The length of this count is approximately 5 seconds. The RDT bit is reset to 0 by an off-hook event. If the RDTM bit (Register 3, bit 7) is set, a hardware interrupt occurs on the AOUT/INT pin when RDT is triggered. This interrupt can be cleared by writing to the RDTI bit (Register 4, bit 7). When the RDI bit (Register 2, bit 2) is set, an interrupt occurs on both the beginning and end of the ring pulse as defined by the RTO bits (Register 23, bits 6:3). Ring validation may be enabled when using the RDI bit.

The third method to monitor detection uses the DTX data samples to transmit ring data. If the ISOcap is active (PDL=0) and the device is not off-hook or in on-hook line monitor mode, the ring data is presented on DTX. The waveform on DTX depends on the state of the RFWE bit.

When RFWE is 0, DTX is -32768 (0x8000) while the RNG1-RNG2 voltage is between the thresholds. When a ring is detected, DTX transitions to +32767 when the ring signal is positive, then goes back to -32768 when the ring is near 0 and negative. Thus a near square wave is presented on DTX that swings from -32768 to +32767 in cadence with the ring signal.

When RFWE is 1, DTX sits at approximately +1228 while the RNG1-RNG2 voltage is between the thresholds. When the ring becomes positive, DTX transitions to +32767. When the ring signal goes near 0, DTX remains near 1228. As the ring becomes negative, the DTX transitions to -32768 . This repeats in cadence with the ring signal.

To observe the ring signal on DTX, watch the MSB of the data. The MSB toggles at the same frequency as the ring signal independent of the ring detector mode. This method is adequate for determining the ring frequency.

5.20. Ring Validation

Ring validation prevents false triggering of a ring detection by validating the ring parameters. Invalid signals, such as a line-voltage change when a parallel handset goes off-hook, pulse dialing, or a high-voltage line test are ignored. Ring validation can be enabled during normal operation and in low-power sleep mode

when a valid external PCLK signal is supplied.

The ring validation circuit operates by calculating the time between alternating crossings of positive and negative ring thresholds to validate that the ring frequency is within tolerance. High and low frequency tolerances are programmable in the RAS[5:0] and RMX[5:0] fields. The RCC[2:0] bits define how long the ring signal must be within tolerance.

Once the duration of the ring frequency is validated by the RCC bits, the circuitry stops checking for frequency tolerance and begins checking for the end of the ring signal, which is defined by a lack of additional threshold crossings for a period of time configured by the RTO[3:0] bits. When the ring frequency is first validated, a timer defined by the RDLY[2:0] bits is started. If the RDLY[2:0] timer expires before the ring timeout, then the ring is validated and a valid ring is indicated. If the ring timeout expires before the RDLY[2:0] timer, a valid ring is not indicated.

Ring validation requires the following five parameters:

- Timeout parameter to place a lower limit on the frequency of the ring signal (the RAS[5:0] bits in Register 24). The frequency is measured by calculating the time between crossings of positive and negative ring thresholds.
- Minimum count to place an upper limit on the frequency (the RMX[5:0] bits in Register 22).
- \blacksquare Time interval over which the ring signal must be the correct frequency (the RCC[2:0] bits in Register 23).
- \blacksquare Timeout period that defines when the ring pulse has ended based on the most recent ring threshold crossing.
- Delay period between when the ring signal is validated and when a valid ring signal is indicated to accommodate distinctive ringing.

The RNGV bit (Register 24, bit 7) enables or disables the ring validation feature in both normal operating mode and low-power sleep mode.

Ring validation affects the behavior of the RDT status bit, the RDTI interrupt, the INT pin, and the RGDT pin.

- 1. When ring validation is enabled, the status bit seen in the RDT read-only bit (r5.2), represents the detected envelope of the ring. The ring validation parameters are configurable so that this envelope may remain high throughout a distinctive-ring sequence.
- 2. The RDTI interrupt fires when a validated ring occurs. If RDI is zero (default), the interrupt occurs on the rising edge of RDT. If RDI is set, the interrupt occurs on both rising and falling edges of RDT.

- 3. The INT pin follows the RDTI bit with configurable polarity.
- 4. The RGDT pin can be configured to follow the ringing signal envelope detected by the ring validation circuit by setting RFWE to 0. If RFWE is set to 1, the RGDT pin follows an unqualified ring detect one-shot signal initiated by a ring-threshold crossing and terminated by a fixed counter timeout of approximately 5 seconds. (This information is shown in [Register 18\)](#page-72-0).

5.21. Ringer Impedance and Threshold

The ring detector in a typical DAA is ac coupled to the line with a large 1 μ F, 250 V decoupling capacitor. The ring detector on the Si3018/19 is resistively coupled to the line. This coupling produces a high ringer impedance to the line of approximately 20 M Ω to meet the majority of country PTT specifications including FCC and TBR21.

Several countries including Poland, South Africa, and Slovenia require a maximum ringer impedance that can be met with an internally-synthesized impedance by setting the RZ bit (Register 16). Certain countries also specify ringer thresholds differently. The RT and RT2 bits (Register 16 and Register 17, respectively) select between three different ringer thresholds: 15 V ±10%, 21 V ±10%, and 45 V ±10%. These three settings enable satisfaction of global ringer threshold requirements. Thresholds are set so that a ring signal is guaranteed to not be detected below the minimum, and a ring signal is guaranteed to be detected above the maximum.

5.22. Pulse Dialing and Spark Quenching

Pulse dialing is accomplished by going off- and on-hook to generate make and break pulses. The nominal rate is 10 pulses per second. Some countries have strict specifications for pulse fidelity including make and break times, make resistance, and rise and fall times. In a traditional, solid-state dc holding circuit, there are a number of issues in meeting these requirements.

The Si3050 dc holding circuit has active control of the on- and off-hook transients to maintain pulse dialing fidelity.

Spark quenching requirements in countries, such as Italy, the Netherlands, South Africa, and Australia, deal with the on-hook transition during pulse dialing. These tests provide an inductive dc feed resulting in a large voltage spike. This spike is caused by the line inductance and the sudden decrease in current through the loop when going on-hook. The traditional way of dealing with this problem is to put a parallel RC shunt across the hookswitch relay. The capacitor is large

(~1 µF, 250 V) and relatively expensive. In the Si3050, loop current can be controlled to achieve three distinct on-hook speeds to pass spark quenching tests without additional BOM components. Through the settings of four bits in three registers, OHS (Register 16), OHS2 (Register 31), SQ0, and SQ1 (Register 59), a slow ramp down of loop current can be achieved which induces a delay between the time the OH bit is cleared and the time the DAA actually goes on-hook.

To ensure proper operation of the DAA during pulse dialing, disable the automatic resistor calibration that is performed each time the DAA enters the off-hook state by setting the RCALD bit (Register 25, bit 5).

5.23. Billing Tone Detection and Receive Overload

"Billing tones" or "metering pulses" generated by the central office can cause modem connection difficulties. The billing tone is typically either a 12 or 16 kHz signal and is sometimes used in Germany, Switzerland, and South Africa. Depending on line conditions, the billing tone might be large enough to produce major errors in the line data. The Si3050/19 chipset can provide feedback indicating the beginning and end of a billing tone.

Billing tones less than 1.1 V_{PK} on the line are filtered out by the low-pass digital filter on the Si3050/19. The ROV bit is set when a billing tone is greater than 1.1 V_{PK} , indicating a receive overload condition. The BTD bit is set when a billing tone is large enough to excessively reduce the line-derived power supply of the line-side device (Si3018/19).

The OVL bit (Register 19) can be polled following billing tone detection. After the OVL bit returns to 0, this indicates that the billing tone has passed. The ROV bit is sticky and must be written to 0 to be reset. After the billing tone passes, the DAA initiates an auto-calibration sequence that must complete before data can be transmitted.

Certain line events, such as an off-hook event on a parallel phone or a polarity reversal, can trigger the ROV or the BTD bits, after which the ROV bit must be reset. The user should look for multiple events before qualifying whether billing tones are actually present.

Although the DAA remains off-hook during a billing tone event, the received data from the line is corrupted when a large billing tone occurs. To receive data through a billing tone, an external LC filter must be added. A manufacturer can provide this filter to users in the form of a dongle that connects on the phone line before the DAA. This relieves the manufacturer of providing a costly internal LC filter when it may only be necessary to

support a few countries/customers.

Alternatively, when a billing tone is detected, the system software can notify the user that a billing tone has occurred. This notification is used to prompt the user to contact the telephone company to disable the billing tones or to purchase an external LC filter.

Disturbances on the line other than billing tones can also cause a receive overload. Some conditions may result in a loop current collapse to a level below the minimum required operating current of the DAA. When this occurs, the dropout detect bit (DOD) is set, and an interrupt will be generated if the dropout detect interrupt mask bit (DODM) is set.

5.24. Billing Tone Filter (Optional)

To operate without degradation during billing tones in Germany, Switzerland, and South Africa, an external LC notch filter is required. The Si3018/19 can remain off-hook during a billing tone event, but line data is lost in the presence of large billing tone signals. The notch filter design requires two notches, one at 12 kHz and one at 16 kHz. Because these components are expensive and few countries utilize billing tones, this filter is typically placed in an external dongle or added as a population option. [Figure 23](#page-33-3) shows a billing tone filter example. [Table 19](#page-33-4) gives the component values.

L1 must carry the entire loop current. The series resistance of the inductors is important to achieve a narrow and deep notch. This design has more than 25 dB of attenuation at both 12 kHz and 16 kHz.

Figure 23. Billing Tone Filter

Table 19. Component Values-Optional Billing Tone Filters

Component	Value
C1, C2	$0.027 \mu F$, 50 V, ±10%
C ₃	0.01 µF, 250 V, $\pm 10\%$
l 1	3.3 mH, >120 mA, <10 Ω , ±10%
12	10 mH, >40 mA, <10 Ω , ±10%

The billing tone filter affects the DAAís ac termination and return loss. The global compromise complex ac termination as selected by ACIM[3:0] = 1111 passes global return loss specifications with and without the billing tone filter by at least 3 dB. This ac termination is optimized for frequency response and hybrid cancellation and has greater than 4 dB of margin with or without the dongle for South Africa, Australia, TBR21, Germany, and Switzerland country-specific specifications.

5.25. On-Hook Line Monitor

The on-hook line monitor mode allows the Si3050 to receive line activity when in an on-hook state. This mode is typically used to detect caller ID data (see the [ì5.26.Caller IDî](#page-33-2) section) and is enabled by setting the ONHM bit (Register 5, bit 3). Caller ID data can be gained up or attenuated using the receive gain control bits in Registers 39 and 41.

5.26. Caller ID

The Si3050 can pass caller ID data from the phone line to a caller ID decoder connected to the DAA.

5.26.1. Type I Caller ID

Type I Caller ID sends the CID data when the phone is on-hook.

In systems where the caller ID data is passed on the phone line between the first and second rings, utilize the following method to capture the caller ID data:

- 1. After identifying a ring signal using one of the methods described in ["5.19.Ring Detection" on page](#page-30-0) [31,](#page-30-0) determine when the first ring is complete.
- 2. Assert the ONHM bit (Register 5, bit 3) to enable caller ID data detection. The caller ID data is passed across the RNG 1/2 pins and presented to the host via the DTX pin.
- 3. Clear the ONHM bit after the caller ID data is received.

In systems where the caller ID data is preceded by a line polarity (battery) reversal, use the following method to capture the caller ID data:

- 1. Enable full wave rectified ring detection (RFWE, Register 18, bit 1).
- 2. Monitor the RDTP and RDTN register bits (or the POLI bit with the Si3019 line-side) to identify if a polarity reversal or a ring signal has occurred. A polarity reversal trips either the RDTP or RDTN ring detection bits, therefore the full-wave ring detector must be used to distinguish a polarity reversal from a ring. The lowest specified ring frequency is 15 Hz; so, if a battery reversal occurs, the DSP should wait a minimum of 40 ms to verify that the event is a battery reversal and not a ring signal. This time is greater than half the period of the longest ring signal. If another edge is detected during this 40 ms pause, this event is characterized as a ring signal and not a battery reversal.
- 3. Assert the ONHM bit (Register 5, bit 3) to enable caller ID data detection. The caller ID data is passed across the RNG 1/2 pins and presented to the host via the DTX pin.
- 4. Clear the ONHM bit after the caller ID data is received.

5.26.2. Type II Caller ID (Si3019 Line-Side Device Only)

Type II Caller ID sends the CID data while the phone is off-hook. This mode is often referred to as caller ID/ call waiting (CID/CW). To receive the CID data when off-hook, use the following procedure (also see [Figure 24](#page-35-2)):

- 1. The Caller Alert Signal (CAS) tone is sent from the central office (CO) and is digitized along with the line data. The host processor detects the presence of this tone.
- 2. The DAA must check if there is another parallel device on the same line, which is accomplished by briefly going on-hook, measuring the line voltage, and returning to an off-hook state.
	- a. Set the CALD bit (Register 17, bit 5) to disable the calibration that automatically occurs when going off-hook.
	- b. Set the RCALD bit (Register 25, bit 5) to disable the resistor calibration that automatically occurs when going off-hook
	- c. Set the FOH[1:0] bits (Register 31 bits 6:5) to 11 to reduce the time period for the off-hook counter to 8 ms allowing compliance to the Type II CID timing requirements.
- d. Clear the OH bit (Register 5, bit 0). This puts the DAA into an on-hook state. The RXM bit (Register 15, bit 3) also can be set to mute the receive path.
- e. Read the LVS bits to determine the state of the line. If the LVS bits read the typical on-hook line voltage, then there are no parallel devices active on the line, and CID data reception can be continued. If the LVS bits read well below the typical on-hook line voltage, then there are one or more devices present and active on the same line that are not compliant with Type II CID. Do not continue CID data reception.
- f. Set the OH bit to return to an off-hook state. Immediately after returning to an off-hook state, the off-hook counter must be allowed to expire. This allows the line voltage to settle before transmitting or receiving data. After 8 ms normal data transmission and reception can begin. If a non-compliant parallel device is present, then a reply tone is not sent by the host tone generator and the CO does not send the CID data. If all devices on the line are Type II CID compliant, then the host must mute its upstream data output to avoid the propagation of its reply tone and the subsequent CID data. When muting its upstream data output, the host processor should return an acknowledgement (ACK) tone to the CO requesting transmission of CID data.
- 3. The CO then responds with CID data after receiving the CID data, the host processor unmutes the upstream data output and continues with normal operation.
- 4. The muting of the upstream data path by the host processor mutes the handset in a telephone application so the user cannot hear the acknowledgement tone and CID data being sent.
- 5. The CALD and the RCALD bits can be cleared to re-enable the automatic calibrations when going off-hook. The FOH[1:0] bits also can be programmed to 01 to restore the default off-hook counter time.

Because of the nature of the low-power ADC, the data presented on DTX can have up to a 10% dc offset. The caller ID decoder must either use a high-pass or a band-pass filter to accurately retrieve the caller ID data.

Si3050 Si3018/19

Notes:

- **1.** The off-hook counter and calibrations prevent transmission or reception of data for 402.75 ms (default) for the line voltage to settle.
- **2.** The caller alert signal (CAS) tone transmits from the CO to signal an incoming call.
- **3.** The device is taken on-hook to read the line voltage in the LVS bits to detect parallel handsets. In this mode, no data is transmitted on the DTX pin.
- **4.** When the device returns off-hook, the normal off-hook counter is reduced to 8 ms. If the CALD and RCALD bits are set, then the automatic calibrations are not performed.
- **5.** After allowing the off-hook counter to expire (8 ms), normal transmission and reception can continue. If CID data reception is required, send the appropriate signal to the CO at this time.

Figure 24. Implementing Type II Caller ID on the Si3050/19

5.27. Overload Detection

The Si3050 can be programmed to detect an overload condition that exceeds the normal operating power range of the DAA circuit. To use the overload detection feature, the following steps should be followed:

- 1. Set the OH bit (Register 5, bit 0) to go off-hook, and wait 25 ms to allow line transients to settle.
- 2. Enable overload detection by then setting the OPE bit (Register 17, bit 3).

If the DAA senses an overload situation it automatically presents an 800 Ω impedance to the line to reduce the hookswitch current. At this time, the DAA also sets the OPD bit (Register 19, bit 0) to indicate that an overload condition exists. The line current detector within the DAA has a threshold that is dependent on the ILIM bit (Register 26). When ILIM = 0, the overload detection threshold equals 160 mA. When $ILIM = 1$, the overload detection threshold equals 60 mA. The OPE bit should always be cleared before going off-hook.

5.28. Gain Control

The Si3050 supports multiple levels of gain and attenuation for the transmit and receive paths.

The TXG2 and RXG2 bits (Registers 38-39) enable gain or attenuation in 1 dB increments for the transmit and receive paths (up to 12 dB gain and 15 dB attenuation). The TGA2 and RGA2 bits select either gain or attenuation. The TXG3 and RXG3 bits (Registers $40-41$) enable gain or attenuation in 0.1 dB increments up to 1.5 dB for the transmit and receive paths. The TGA3 and RGA3 bits select either gain or attenuation. The transmit and receive paths can be individually muted with the TXM and RXM bits (Register 15). The signal flow through the Si3050 and the Si3018/19 is shown in Figures $25-26$.

Figure 26. Si3050 Signal Flow Diagram

5.29. Transhybrid Balance

The Si3050 contains an on-chip analog hybrid that performs the 2- to 4-wire conversion and near-end echo cancellation. This hybrid circuit is adjusted for each ac termination setting selected to achieve a minimum transhybrid balance of 20 dB.

The Si3050 also offers a digital hybrid stage for additional near-end echo cancellation. For each ac termination setting, the eight programmable hybrid registers (Registers 45-52) can be programmed with coefficients to increase cancellation of real-world line anomalies. This digital filter can produce 10 dB or greater of near-end echo cancellation in addition to the 20 dB from the analog hybrid circuitry. See AN84 for a more detailed description of the digital hybrid and how to use it.

5.30. Filter Selection

The Si3050 supports additional filter selections for the receive and transmit signals as defined in Tables [10](#page-13-0) and [11](#page-13-1). The IIRE bit (Register 16, bit 4) selects between the IIR and FIR filters. The IIR filter provides a shorter, but non-linear, group delay alternative to the default FIR

filter, and only operates with an 8 kHz sample rate. The FILT bit (Register 31, bit 1) selects $a -3$ dB low frequency pole of 5 Hz when cleared and $a - 3$ dB low frequency pole of 200 Hz (per EIA/TIA 464) when set. The FILT bit affects the receive path only.

5.31. Clock Generation

The Si3050 generates the necessary internal clock frequencies from the PCLK input. PCLK must be synchronous to the 8 kHz FSYNC clock and run at one of the following rates: 256 kHz, 512 kHz, 768 kHz, 1.024 MHz, 1.53 MHz, 2.048 MHz, 4.09 MHz, or 8.192 MHz. The ratio of the PCLK rate to the FSYNC rate is determined internally by the DAA and is transferred into internal registers after a reset. These internal registers are not accessible through register reads or writes. [Figure 27](#page-37-0) shows the operation of the Si3050 clock circuitry.

The PLL clock synthesizer settles quickly after powerup. However, the settling time depends on the PCLK frequency and it can be approximately predicted by the following equation:

$$
T_{\text{settle}} = 64/F_{\text{PCLK}}
$$

For all valid PCLK frequencies listed above, the default line sample rate is 8 kHz. This sample rate can be increased to 16 kHz by setting the HSSM bit (Register 7, bit 3). Regardless of the sample rate frequency, the serial data communication rate of the PCM and GCI highways remains 8 kHz. When the 16 kHz sample rate is selected, additional timeslots in the PCM or GCI highway are used to transfer the additional data.

5.32. Communication Interface Mode Selection

The Si3050 supports two communication interface protocols:

■ PCM/SPI mode where data and control information transmission/reception occurs across separate buses (PCM highway for data, and SPI port for

control).

GCI mode where data and control information is multiplexed and transmission/reception occurs across the GCI highway bus.

A pin-strapping method (specifically, the state of SCLK on power-up [reset]) is used to select between the two communication interface protocols. Tables [19](#page-33-0) and [20](#page-37-1) specify how to select a communication mode, and how the various pins are used in each mode.

When operating in PCM/SPI mode, the GCI control register should not be written (i.e., Register 42 must each remain set at 0000 0000 when using the PCM/ SPI highway mode). Similarly, when operating in GCI highway mode the PCM registers should not be written $(i.e.,$ Registers 33-37 must remain set to 0000 0000 when using the GCI highway mode).

Figure 27. PLL Clock Synthesizer

Table 20. PCM or GCI Highway Mode Selection

Pin Name	PCM Mode	GCI Mode
SDI THRU	SPI Data Throughput pin for Daisy Chaining Operation (Connects to the SDI pin of the subsequent device in the daisy chain)	Sub-frame Selector, bit 2
SCLK	SPI Clock Input	PCM/GCI Mode Selector
SDI	SPI Serial Data Input	B1/B2 Channel Selector
SDO	SPI Serial Data Output	Sub-frame Selector, bit 1
CS	SPI Chip Select	Sub-frame Selector, bit 0
FSYNC	PCM Frame Sync Input	GCI Frame Sync Input
PCLK	PCM Input Clock	GCI Input Clock
DTX.	PCM Data Transmit	GCI Data Transmit
DRX	PCM Data Receive	GCI Data Receive
Note: This table denotes pin functionality after the rising edge of RESET and mode selection.		

Table 21. Pin Functionality in PCM or GCI Highway Mode

5.33. PCM Highway

The Si3050 contains a flexible programmable interface for the transmission and reception of digital PCM samples. PCM data transfer is controlled via the PCLK and FSYNC inputs, the PCM Transmit and Receive Start Count registers (Registers 34-37), and the PCM Mode Select register (Register 33). The interface can be configured to support from 4 to 128 8-bit timeslots in each frame, which corresponds to PCLK frequencies of 256 kHz to 8.192 MHz in power of 2 increments. Time slot assignment and data delay from FSYNC edge are handled via the TXS and RXS registers. These 10-bit values are programmed with the number of PCLK cycles following the rising edge of FSYNC until the data transfer begins. Because the Si3050 looks for the rising edge of FSYNC, both long and short FSYNC pulse widths can be accommodated. A value of 0 in the PCM Transmit and Receive Start Count registers signifies that the MSB of the data should occur in the same cycle as the rising edge of FSYNC.

By setting the correct starting point of the data, the Si3050 can operate with buses having multiple devices requiring different time slots. The DTX pin is high impedance except during transmission of an 8-bit PCM sample. DTX returns to high impedance either on the negative edge of PCLK during the LSB or on the positive edge of PCLK following the LSB. This behavior is based on the setting of the TRI bit in the PCM Mode Select register. Tristating on the negative edge allows the transmission of data by multiple sources in adjacent timeslots without the risk of driver contention. In addition to 8-bit data modes, a 16-bit linear mode is also provided. This mode can be activated via the PCMF bits in the PCM Mode Select register. Double-clocked timing also is supported in which the duration of a data bit is two PCLK cycles. This mode is activated via the PHCF bit in the PCM Mode Select register. Setting the TXS or RXS registers greater than the number of PCLK cycles in a sample period stops data transmission or reception. Figures [28](#page-39-0)–[31](#page-40-0) illustrate the usage of the PCM highway interface to adapt to common PCM standards.

Figure 28. PCM Highway Transmission, Short FSYNC, Single Clock Cycle Delayed Transmission (TXS = **RXS** = **0, PHCF** = **0, TRI** = **1)**

Figure 29. PCM Highway Transmission, Long FSYNC (TXS = **RXS** = **0, PHCF** = **0, TRI** = **1)**

5.34. Companding in PCM Mode

The Si3050 supports both µ-Law and A-Law companding formats in addition to 16-bit linear data. The 8-bit companding schemes follow a segmented curve formatted as a sign bit, three chord bits, and four step bits. µ-Law is commonly used in North America and Japan, while A-Law is primarily used in Europe. Data format is selected via the PCMF bits (Register 33). [Table 22 on page 43](#page-42-0) and [Table 23 on page 44](#page-43-0) define the µ-Law and A-Law encoding formats. If linear mode is used the resulting 16-bit data is transmitted in two consecutive 8-bit PCM highway timeslots as shown in [Figure 32](#page-41-0).

5.35. 16 kHz Sampling Operation in PCM Mode

The Si3050 can be configured to support a 16 kHz sampling rate and transmit the data on an 8 kHz PCM or GCI highway bus. By setting the HSSM bit (Register 7, bit 3) to 1, the DAA changes its sampling rate, Fs, to 16 kHz if it was originally configured for an 8 kHz sampling rate. If μ -law or A-law companding is used, the resulting 8-bit samples are transmitted in two consecutive 8-bit PCM highway timeslots. If linear mode is used, the resulting 16-bit samples are transmitted in four consecutive 8-bit PCM highway timeslots as shown in [Figure 33](#page-41-1).

Notes:

1. Characteristics are symmetrical about analog 0 with sign bit = 1 for negative analog values.

2. Digital code includes inversion of both sign and magnitude bits.

Notes:

1. Characteristics are symmetrical about analog 0 with sign bit = 1 for negative analog values.

2. Digital code includes inversion of all even numbered bits.

5.36. SPI Control Interface

The control interface to the Si3050 is a 4-wire interface modeled on commonly available micro-controller and serial peripheral devices. The interface consists of four pins: clock (SCLK), chip select (CS), serial data input (SDI), and serial data output (SDO). In addition, the Si3050 includes a serial data through output pin (SDITHRU) to support daisy chain operation of up to 16 devices. The device can operate with 8-bit and 16-bit SPI controllers. Each SPI operation consists of a control byte, an address byte (of which only the six LSBs are used internally), and either one or two data bytes depending on the width of the controller. Bytes are transmitted MSB first.

There are a number of variations of usage on this four-wire interface as follows:

■ Continuous clocking. During continuous clocking, assertion of the \overline{CS} pin controls the data transfers. The CS pin must be asserted before the falling edge of SCLK on which the first bit of data is expected during a read cycle, and must remain low for the

duration of the 8-bit transfer (command/address or data), going high after the last rising edge of SCLK after the transfer.

- Clock only during transfer. The clock is active during the actual byte transfers only. Each byte transfer consists of eight clock cycles in a return to 1 format.
- SDI/SDO wired operation. Independent of the clocking options described, the SDI and SDO pins can be treated as two separate lines or wired together if the master can tri-state its output during the data byte transfer of a read operation.

The SPI state machine resets when the \overline{CS} pin is asserted during an operation on an SCLK cycle that is not a multiple of eight. This provides a mechanism for the controller to force the state machine to a known state in the case where the controller and the device are not synchronized.

The control byte has the following structure and is presented on the SDI pin MSB first.

The bits are defined as follows:

Figure 34. SPI Daisy Chain Control Architecture

SPI Control Byte

Figure 35. Sample SPI Control Byte to Access Channel 0

Figure 36. Sample SPI Control Byte for Broadcast Mode (Write Only)

In [Figure 35](#page-46-0) the CID field is 0. As this field is decremented in LSB to MSB order, the value decrements for each SDI down the line. The BRCT and R/W bits remain unchanged as the control word passes through the entire chain. A unique CID is presented to each device, and the device receiving a CID value of 0 is the target of the operation (channel 0 in this case). [Figure 36](#page-46-1) illustrates that in broadcast mode, all bits pass through the chain without permutation.

Figure 38. Read Operation via an 8-bit SPI Port

[Figure 37](#page-46-2) and [Figure 38](#page-47-0) illustrate WRITE and READ operations via an 8-bit SPI controller. Each of these operations are performed as a 3-byte transfer. The \overline{CS} pin is asserted between each byte. The \overline{CS} pin must be asserted before the first falling edge of SCLK after the DATA byte to indicate to the state machine that only one byte should be transferred. The state of the SDI pin is ignored during the DATA byte of a read operation.

Figure 40. Read Operation via a 16-bit SPI Port

Figures [39](#page-47-1) and [40](#page-47-2) illustrate WRITE and READ operations via a 16-bit SPI controller. These operations require a 4-byte transfer arranged as two 16-bit words. The CS pin does not go high when the eighth bit of data is received, which indicates to the SPI state machine that eight more SCLK pulses follow to complete the operation. In the case of a WRITE operation, the last eight bits are ignored. In a read operation, the 8-bit data value is repeated so that the data may be captured during the last half of a data transfer if required by the controller. The Si3050 autodetects the SPI mode (16-bit or 8-bit mode).

5.37. GCI Highway

The Si3050 contains an alternate communication interface to the SPI and PCM highway control and data interface. The general circuit interface (GCI) can be used for the transmission and reception of control and data information onto a GCI highway bus. The PCM and GCI highways are 4-wire interfaces and share the same pins. The SPI control interface is not used as a communication interface in the GCI highway mode, but rather as hardwired channel selector pins.

When GCI mode is selected, the sub-frame selection pins must be tied to the correct state to select one of eight sub-frame timeslots in the GCI frame ([Table 24](#page-48-0)). These pins must remain in this state when the Si3050 is operating. Selecting a particular subframe automatically causes that individual Si3050 to transmit and receive on the appropriate sub-frame in the GCI frame, which is initiated by an FSYNC pulse. No more register settings are needed to select which sub-frame a device uses, and the sub-frame for a particular device cannot be changed when in operation. Only one Si3050 DAA can be assigned per sub-frame, which allows a total of eight DAAs to be connected to the same GCI highway bus.

GCI mode supports a 1x and a 2x PCLK rate as shown in Figures [5](#page-12-0) and [6](#page-13-2) on pages [13](#page-12-0) and [14,](#page-13-2) respectively. The PCLK rate is autodetected and no internal register settings are needed to support either 1x or 2x PCLK mode.

Table 24. GCI Mode Sub-Frame Selection

The GCI highway requires either a 2.048 or 4.096 MHz clock frequency on the PCLK pin, and an 8 kHz frame sync input on the FSYNC pin. The overall unit of data used to communicate on the GCI highway is a frame, which is 125 µs in length. Each frame is initiated by a pulse on the FSYNC pin and the rising edge signifies the beginning of the next frame. In 2x PCLK mode, there are twice as many PCLK cycles during each 125 µs frame versus 1x PCLK mode. Each frame consists of eight fixed timeslot sub-frames that are assigned using the Sub-Frame Select pins as described in [Table 21 on page 39](#page-38-0) (SDI_THRU, SDO, and CS). Within each sub-frame are four channels (bytes) of data, including the two voice data channels (B1 and B2), one Monitor channel (M) for initialization and setup of the device, and one Signaling and Control channel (SC) for communicating status of the device and for initiating commands. Within the SC channel are six Command/Indicate (C/I) bits and two handshaking bits (MR and MX). The C/I bits are used for status and command communication, whereas the handshaking bits Monitor Receive (MR) and Monitor Transmit (MX) are used for data exchanges in the Monitor channel. [Figure 41](#page-49-0) illustrates the contents of a GCI highway frame.

5.38. Companding in GCI Mode

The Si3050 supports μ -Law and A-Law companding formats in addition to 8-bit or 16-bit linear data. The 8-bit companding schemes are described in ["5.34.Companding in PCM Mode" on page 42](#page-41-3) and are shown in [Table 22](#page-42-0) and [Table 23.](#page-43-0) If 16-bit linear mode is used, the resulting 16-bit samples are transmitted in both the B1 and B2 channels of a single subframe. For proper operation, select all Si3050 DAAs to use the B1 channel with only one DAA per subframe.

5.39. 16 kHz Sampling Operation in GCI Mode

The Si3050 can be configured to support a 16 kHz sampling rate (as described in "5.35.16 [kHz Sampling](#page-41-2) [Operation in PCM Mode" on page 42](#page-41-2)) and transmit the data on an 8 kHz GCI Highway bus. If 8-bit samples are used with a 16 kHz sample rate, the samples are transmitted in both the B1 and B2 channels of a single subframe. If 16-bit linear mode is used, the resulting 16-bit samples are transmitted in both the B1 and B2 channels of two consecutive subframes. In this case, assign one DAA per two subframes.

5.40. Monitor Channel

The Monitor channel is used for initialization and setup of the Si3050. It also can be used for general communication with the Si3050 by allowing read and

write access to the Si3050's registers. Use of the monitor channel requires manipulation of the MR and MX handshaking bits, located in bits 1 and 0 of the SC channel described below. For purposes of this specification, "downstream" is identified to be the data sent by a host to the Si3050. "Upstream" is identified to be the data sent by the Si3050 to a host.

The following diagram illustrates the Monitor channel communication protocol. For successful communication with the Si3050, the transmitter should anticipate the falling edge of the receiver's acknowledgement. This also maximizes communication speed. Because of the handshaking protocol required for successful communication, the data transfer rate using the Monitor channel is less than 8 kbytes/second.

The Idle state is achieved by the MX and MR bits being held inactive (signal is high) for two or more frames. When a transmission is initiated by a host device, an active state (signal is low) is present on the downstream MX bit. This signals to the Si3050 that a transmission has begun on the Monitor channel and the Si3050 should begin accepting data from host device. The Si3050, after reading the data on the Monitor channel, acknowledges the initial transmission by placing the upstream MR bit in an active state. The data is received and the upstream MR becomes active in the frame immediately following the downstream MX becoming active. The upstream MR then remains active until either the next byte is received or an end of message is detected. The end of message is signaled by the downstream MX being held inactive for two or more consecutive frames. Receipt of initial data is signaled by the upstream MR bit's transitioning from an inactive to an active state. Upon receiving acknowledgement from the Si3050 that the initial data is received, the host device places the downstream MX bit in the inactive state for one frame and then either transmit another byte by placing the downstream MX bit in an active state again, or signal an end of message by leaving the downstream MX bit inactive for a second frame.

When the host is performing a write command, the host only manipulates the downstream MX bit, and the Si3050 only manipulates the upstream MR bit. If a read command is performed, the host initially manipulates the downstream MX bit to communicate the command, but then manipulates the downstream MR bit in response to the Si3050 responding with the requested data. Similarly, the Si3050 initially manipulates its upstream MR bit to receive the read command, and then manipulates its upstream MX bit to respond with the requested data. If the host is transmitting data, the Si3050 always transmits a \$FF value on its Monitor data byte. While the Si3050 is transmitting data, the host should always transmit a \$FF value on its Monitor byte. If the Si3050 is transmitting data and detects a value other than a \$FF on the downstream Monitor byte, the Si3050 signals an Abort.

For read and write commands, an initial address must be specified. The Si3050 responds to a read or a write command at this address, and then subsequently increment this address after every register access.

In this manner, multiple consecutive registers can be read or written in one transmission sequence. By correctly manipulating the MX and MR bits, a transmission sequence can continue from the beginning specified address until an invalid memory location is reached. To end a transmission sequence, the host processor must signal an end-of-message (EOM) by placing the downstream MX and MR bits inactive for two consecutive frames. The transmission also can be stopped by the Si3050 by signaling an Abort. This is signaled by placing the upstream MR bit inactive for at least two consecutive cycles in response to the downstream MX bit going active. An abort is signaled by the Si3050 for the following reasons:

- A read or write to an invalid memory address is attempted
- An invalid command sequence is received
- A data byte was not received for at least two consecutive frames
- A collision occurs on the Monitor data bytes while the Si3050 is transmitting data

When the Si3050 aborts because of an invalid command sequence, the state of the Si3050 does not change. If a read or write to an invalid memory address is attempted, all previous reads or writes in that transmission sequence are valid up to the read or write to the invalid memory address. If an EOM is detected before a valid command sequence is communicated, the Si3050 returns to the idle state and remains unchanged.

The data presented to the Si3050 in the downstream Monitor bits must be present for two consecutive frames to be considered valid data. The Si3050 checks to ensure it receives the same data in two consecutive frames. If not, it does not acknowledge receipt of the data byte and waits until it does receive two consecutive identical data bytes before acknowledging to the transmitter that it received the data. If the transmitter attempts to signal transmission of a subsequent data byte by placing the downstream MX bit in an inactive state while the Si3050 is still waiting to receive a valid data byte transmission of two consecutive identical data bytes, the Si3050 signals an abort and ends the transmission. [Figure 43](#page-51-0) shows a state diagram for the Receiver Monitor channel for the Si3050. [Figure 44 on](#page-52-0) [page 53](#page-52-0) shows a state diagram for the Transmitter Monitor channel for the Si3050.

Si3050 Si3018/19

MR: MR bit calculated and transmitted on DTX line. MX: MX bit received data downstream (DRX line). LL: Last look of monitor byte received on DRX line. ABT: Abort indication to internal source.

Figure 43. Si3050 Monitor Receiver State Diagram

Figure 44. Si3050 Monitor Transmitter State Diagram

Si3050 Si3018/19

5.41. Summary of Monitor Channel Commands

Communication with the Si3050 should be in the following format:

Byte 1: Device Address Byte

Byte 2: Command Byte

Byte 3: Register Address Byte

Bytes 4-n: Data Bytes

Bytes n+1, n+2: EOM

5.42. Device Address Byte

The Device Address byte identifies which device connected to the GCI highway receives the particular message. This address should be the first byte sent to the Si3050 at the beginning of every transmission sequence. For Read commands, the address sent to the Si3050 is the first byte transmitted in response to the Read command before register data is transmitted. This Device Address byte has the following structure:

The lowest programmable bit, C, has a special function. This bit enables a register read or write, or enables a special Channel Identification Command (CID).

C = 1: Normal command follows.

C = 0: Channel Identification Command.

The CID is a special command to identify themselves by software. For this special command, the subsequent command byte transmitted by the host processor must be \$00 (binary), and have no address or data bytes. The Si3050 in turn responds with a fixed 2-byte identification code:

Upon sending the 2-byte identification code, the Si3050 sends an EOM ($MR = MX = 1$) for two consecutive frames. When $A = 0$, B must be 0 or the Si3050 signals an abort due to an invalid command. In this mode, bit C is the only other programmable bit.

 $A = 0$: Response to CID command from the device using channel B1 is placed in Monitor Data.

 $A = 1$: Response to CID command from the device using channel B2 is placed in Monitor Data.

When $C = 1$, bits A and B are channel enable bits. When these bits are set to 1, the individual corresponding channels receives the command in the next command byte. The channels whose corresponding bits are set to 0 ignores the subsequent command byte.

A = 1: Channel B1 receives the command.

A = 0: Channel B1 does not receive the command.

- B = 1: Channel B2 receives the command.
- B = 0: Channel B2 does not receive the command.

5.43. Command Byte

The Command byte has the following structure:

The RW bit is a register read/write bit.

 $RW = 0$: A write is performed to the Si3050's register.

 $RW = 1$: A read is performed on the Si3050's register.

The CMD[6:0] bits specify the actual command to be performed.

 $CMD[6:0] = 0000001$: Read or write a register on the Si3050.

 $CMD[6:0] = 0000010 - 11111111$: Reserved.

5.44. Register Address Byte

The Register Address byte has the following structure:

ADDRESS[7:0]

This byte contains the actual 8-bit address of the register to be read or written.

5.45. SC Channel

The SC channel consists of six C/I bits and two handshaking bits, MR and MX. One of these channels is contained in every 4-byte sub-frame and is transmitted every frame. The handshaking bits are described in the above Monitor Channel section. The definition of the six C/I bits depends on the direction the bits are being sent, either transmitted to the GCI highway bus via the DTX pin or received from the GCI highway bus via the DRX pin.

5.46. Receive SC Channel

These bits are defined as follows:

CIR6: Reserved

CIR5: Reserved

CIR4: ONHM

CIR3: TGDE

CIR2: RG

CIR1: OH

Data that is received must be consistent and match for at least two consecutive frames to be considered valid. When a new command or status is communicated via the C/I bits, the data must be sent for at least two consecutive frames to be recognized by the Si3050. The following steps describe the protocol of how C/I bits are stored, detected, and validated. This is illustrated in [Figure 47](#page-57-0).

- 1. The current state of the C/I bits are stored in a primary register P. If the received C/I bits are identical to this current state, no action is taken.
- 2. Upon receipt of an SC channel with C/I bits that differ from the current state, these new C/I bits are immediately latched into a secondary register S.
- 3. The C/I bits in the SC channel received in the frame immediately after the SC channel just stored in S are compared with the C/I bits in the S register.
	- a. If the C/I bits in these two channels are identical, then the C/I bits in the S register are loaded into the P register and are considered a valid change of C/I bits. The Si3050 then responds accordingly to the changed C/I bits.
	- b. If a set of C/I bits is latched into the S register and the subsequent set of C/I bits received does not match either the S or P registers, then the newly received set of C/I bits are latched into the S register. This continues to occur as long as the subsequent set of C/I bits received differs from the C/I bits in the S and P registers.
	- c. If the C/I bits in the SC channel received immediately after the SC channel just stored in S do not match the C/I bits stored in S, but DO match the C/I bits stored in P, then the single set of C/I bits stored in the S latch are invalidated, and the current state of the C/I bits in P remains unchanged.

Figure 47. Protocol for Receiving C/I Bits in the Si3050

5.47. Transmit SC Channel

The following diagram shows the definition of the transmitted SC channel, which is transmitted MSB first:

These bits are defined as follows:

CIT6: Reserved

CIT5: CVI

CIT4: DOD

CIT3: INT (represents the state of the INT pin)

CIT2: Battery Reversal (represents the state of bit 7 of the LVS register)

CIT1: TGD

6. Control Registers

Note: Registers not listed here are reserved and must not be written.

Table 25. Register Summary

Register 1. Control 1

Register 2. Control 2

Register 3. Interrupt Mask

Register 4. Interrupt Source

Si3050 Si3018/19

Register 5. DAA Control 1

Register 6. DAA Control 2

Reset settings = 0001_0000

Register 7. Sample Rate Control

Register 8-9. Reserved

Reset settings = 0000_0000

Register 10. DAA Control 3

Register 11. System-Side and Line-Side Device Revision

Reset settings = xxxx_xxxx

Register 12. Line-Side Device Status

Register 13. Line-Side Device Revision

Reset settings = xxxx_xxxx

Register 14. DAA Control 4

Register 15. TX/RX Gain Control 1

Register 16. International Control 1

Register 17. International Control 2

Register 18. International Control 3

Register 19. International Control 4

Register 20. Call Progress RX Attenuation

Reset settings = 0000_0000

Register 21. Call Progress TX Attenuation

Register 22. Ring Validation Control 1

Register 23. Ring Validation Control 2

Register 24. Ring Validation Control 3

Reset settings = 0001_1001

Register 25. Resistor Calibration

Reset settings = xx0x_xxxx

Register 26. DC Termination Control

Register 27. Reserved

Reset settings = xxxx_xxxx

Register 28. Loop Current Status

Reset settings = 0000_0000

Register 29. Line Voltage Status

Register 30. AC Termination Control

Register 31. DAA Control 5

Register 32. Ground Start Control

Register 33. PCM/SPI Mode Select

Register 34. PCM Transmit Start Count-Low Byte

Reset settings = 0000_0000

Register 35. PCM Transmit Start Count-High Byte

Reset settings = 0000_0000

Register 36. PCM Receive Start Count-Low Byte

Register 37. PCM Receive Start Count-High Byte

Reset settings = 0000_0000

Register 38. TX Gain Control 2

Register 39. RX Gain Control 2

Register 40. TX Gain Control 3

Register 41. RX Gain Control 3

Register 42. GCI Control

Register 43. Line Current/Voltage Threshold Interrupt (Si3019 line-side only)

Reset settings = 0000_0000

Register 44. Line Current/Voltage Threshold Interrupt Control (Si3019 line-side only)

Register 45. Programmable Hybrid Register 1

Reset settings = 0000_0000

Register 46. Programmable Hybrid Register 2

Register 47. Programmable Hybrid Register 3

Reset settings = 0000_0000

Register 48. Programmable Hybrid Register 4

Register 49. Programmable Hybrid Register 5

Reset settings = 0000_0000

Register 50. Programmable Hybrid Register 6

Register 51. Programmable Hybrid Register 7

Reset settings = 0000_0000

Register 52. Programmable Hybrid Register 8

Reset settings = 0000_0000

Register 53-58. Reserved

Reset settings = xxxx_xxxx

Register 59. Spark Quenching Control

Reset settings = xxxx_xxxx

APPENDIXóUL1950 3RD EDITION

Introduction

Although designs using the Si3018 and Si3019 comply with UL1950 3rd Edition and pass all overcurrent and overvoltage tests, there are still several issues to consider.

[Figure 48](#page-96-0) shows two designs that can pass the UL1950 overvoltage tests and electromagnetic emissions. The top schematic shows the configuration in which the ferrite beads (FB1, FB2) are on the unprotected side of the sidactor (RV1). For this configuration, the current rating of the ferrite beads needs to be 6 A. However, the higher current ferrite beads are less effective in reducing electromagnetic emissions.

The bottom schematic of [Figure 48](#page-96-0) shows the

configuration in which the ferrite beads (FB1, FB2) are on the protected side of the sidactor (RV1). For this design, the ferrite beads can be rated at 200 mA.

In a cost optimized design, it is important to remember that compliance to UL1950 does not always require overvoltage tests. It is best to plan ahead and know which overvoltage tests applies to your system. System-level elements in the construction, such as fire enclosure and spacing requirements, must be considered during the design stages. Consult with your professional testing agency during the design of the product to determine which tests apply to your system.

Figure 48. Circuits that Pass all UL1950 Overvoltage Tests

7. Pin Descriptions: Si3050

Table 26. Si3050 Pin Descriptions (Continued)

8. Pin Descriptions: Si3018/19

9. Ordering Guide

Notes:

1. Many Silicon Labs devices are available in lead-free packages. For lead-free parts, the "K" in the part number suffix is replaced with an "F".

2. The ordered part number for Silicon Labs devices may include the revision letter (example Si3050-D-KS, where "D" is the die revision letter). Refer to "10. Product Identification" for more information on part naming conventions.

10. Product Identification

The product identification number is a finished goods part number or is specified by a finished goods part number, such as a special customer part number.

Example:

11. Package Outline: 20-Pin TSSOP

[Figure 49](#page-102-0) illustrates the package details for the Si3050. [Table 27](#page-102-1) lists the values for the dimensions shown in the illustration.

Figure 49. 20-Pin Thin Shrink Small Outline Package (TSSOP)

Table 27. Package Diagram Dimensions

12. Package Outline: 16-Pin SOIC

[Figure 50](#page-103-0) illustrates the package details for the Si3018/19. [Table 28](#page-103-1) lists the values for the dimensions shown in the illustration.

Figure 50. 16-pin Small Outline Integrated Circuit (SOIC) Package

Symbol	Millimeters	
	Min	Max
Α	1.35	1.75
$\overline{A1}$.10	.25
В	.33	.51
Ĉ	.19	.25
D	$\overline{9.80}$	10.00
E	3.80	4.00
e	1.27 BSC	
H	5.80	6.20
h	.25	.50
	.40	1.27
γ	$\overline{0.10}$	
θ	0°	8°
aaa	0.25	
bbb	0.25	

Table 28. Package Diagram Dimensions

SILICON LABS Si3050 SUPPORT DOCUMENTATION

- AN16: Multiple Device Support
- AN17: Designing for International Safety Compliance
- AN30: Ground Start Implementation with Silicon Laboratoriesí DAAs
- AN67: Layout Guidelines
- AN72: Ring Detection/Validation with the Si305x DAAs
- AN77: Silicon DAA Software Guidelines (Si3050)
- AN81: Emissions Design Considerations
- AN84: Digital Hybrid with the Si305x DAAs
- Si3050PPT-EVB Data Sheet

Note: Refer to www.silabs.com for a current list of support documents for this chipset.

DOCUMENT CHANGE LIST

Revision 0.81 to Revision 1.0

- [Table 3 on page 7](#page-6-0)
	- Power Supply Current values updated.
	- Total Supply Current values updated.
	- AOUT Low Level Current added.
	- AOUT High Level Current added.
	- Note 1 added.
- [Table 4 on page 8](#page-7-0)
	- Caller ID Full-Scale Level typical updated.
	- AOUT Low Level Current removed.
	- AOUT High Level Current removed.
	- Note 5 updated.
- [Table 6 on page 10](#page-9-0)
	- Added Note 2.
- [Table 8 on page 12](#page-11-0)
	- Added Note 3.
- [Table 9 on page 13](#page-12-0)
	- Added Note 3.
- "3. Bill of Materials" on page 18 \bullet Note 3 updated.
- "Register 17. International Control 2" on page 72 • BTE updated.
- ["Register 19.International Control 4" on page 74](#page-73-0) • DOD updated.
- ["Register 33. PCM/SPI Mode Select" on page 84](#page-83-0)
	- \bullet PCMF updated.

Revision 1.0 to Revision 1.01

- Updated "2.Typical Application Schematic" on page [17.](#page-16-0)
- Updated "3. Bill of Materials" on page 18.
	- Changed recommended component suppliers.
	- Listed values for enhanced optional caller ID circuit.
	- Removed R14 (0 Ω resistor).
- Updated [Table 13 on page 21.](#page-20-0)

• Changed recommended country settings for Australia, Austria, Bahrain, Brazil, Bulgaria, China, Croatia, Cyprus, Czech Republic, Egypt, Germany, Hungary, Israel, India, Japan, Jordan, Kazakhstan, Latvia, Lebanon, Malaysia, Malta, Morocco, Nigeria, Oman, Pakistan, Philippines, Poland, Romania, Russia, Slovakia, Slovenia, South Africa, South Korea, Syria, Taiwan, Thailand.

- Updated ["5.3.Initialization" on page 24](#page-23-0).
- Updated [Table 18 on page 31.](#page-30-0)
	- Changed AC termination value for $ACIM[3:0] = 1010.$
- Updated Register 30, "AC Termination Control," on [page 81.](#page-80-0)
	- Updated ACIM bit definition.
- Updated [Figure 19 on page 27](#page-26-0).
- Updated [Figure 21 on page 30](#page-29-0).
- Updated [Figure 22 on page 30](#page-29-1).
- Updated [Figure 27 on page 38](#page-37-0).
- Updated [Table 2 on page 6.](#page-5-0)
- Updated [Table 4 on page 8.](#page-7-0)
- Updated [Table 6 on page 10](#page-9-0)
	- Added PCLK jitter tolerance
- Updated [Table 8 on page 12](#page-11-0)
	- Added PCLK jitter tolerance
- Updated [Table 9 on page 13](#page-12-0)
	- Added PCLK jitter tolerance
- Updated "Transmit/Receive Full Scale Level" functional description.
	- \bullet Includes new enhanced full-scale mode with Revision E or later line-side.
- The following bits have been added, but will only be supported with Si3018/19/10 Revision E or later line-side devices.

• Added FULL2 bit in Register 30, "AC Termination Control," on page 81.

- Added RG1 and GCE bits in Register 59, "Spark Quenching Control," on page 96.
- Updated ["9.Ordering Guide" on page 102](#page-101-1).
- Added "10. Product Identification" on page 102.

Si3050 Si3018/19

NOTES:

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