
Section 66. 16-Bit Sigma-Delta A/D Converter

HIGHLIGHTS

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66.1 INTRODUCTION

The PIC24F 16-bit Sigma-Delta Analog-to-Digital (SD A/D) Converter has the following key features:

- Sigma-Delta Conversion
- Programmable Data Rate up to 62.5 ksp/s
- Differential inputs with Gain Settings
- Dithering Option and Adjustable Oversampling Ratios
- Independent Module Reset Option

The SD A/D Converter module accepts the analog signal and converts it to a corresponding digital value. The 16-bit SD A/D Converter can have up to eight differential analog inputs (channels). Some of the channels can be configured as single-ended inputs or can be used for the external voltage reference connection. The actual number of analog inputs and the external voltage reference input configuration can vary with the device. For more information on the actual number and type of channels available in the device, refer to the specific device data sheet.

The SD A/D Converter contains the following sections:

- Analog Input Selection
- Output Buffer
- Timing and Control Functions

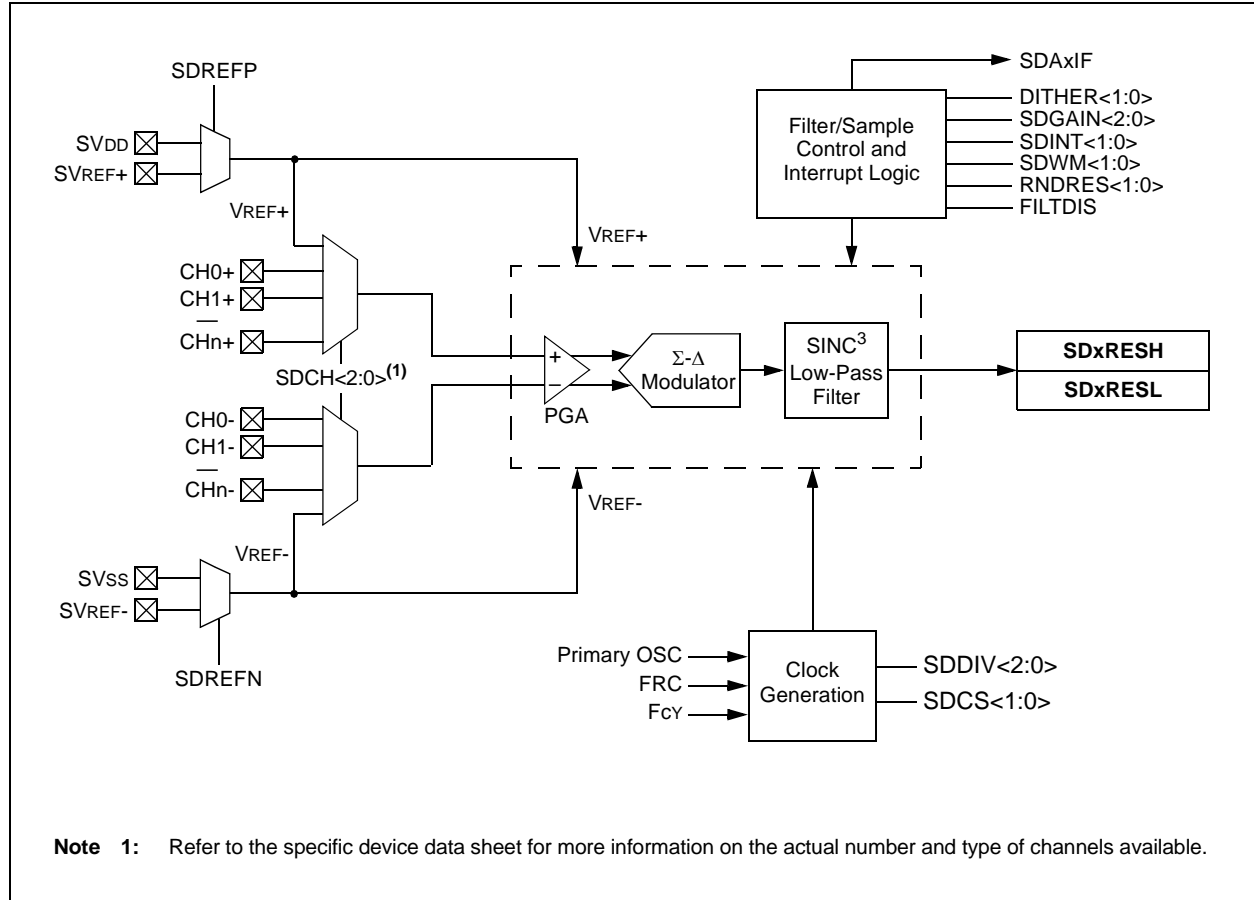
The 16-bit SD A/D Converter samples an input signal less than Nyquist Frequency. The analog input multiplexer selects the signal to be converted from multiple analog input pins. The differential analog input signal can be positive or negative (i.e., bipolar). The sampled voltage is converted to a digital value, which represents the ratio of input voltage to a reference voltage. The reference voltages can be configured to either external references or SV_{DD} and SV_{SS} .

The input signal can be amplified up to 32 times using the gain settings. Dithering averages the nonlinearity of the Digital-to-Analog Converter (DAC) and thereby reduces harmonic distortion. The bandwidth of the input amplifier can be selected using the power level option.

A simplified block diagram for the SD A/D Converter is shown in [Figure 66-1](#).

Note: The Nyquist Frequency is half the sampling frequency of the SD A/D Converter.
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Figure 66-1: 16-Bit Sigma-Delta A/D Converter Block Diagram



66.2 REGISTERS

The 16-bit SD A/D Converter uses five registers for its operation:

- SDxCON1: SD A/Dx Control Register 1
- SDxCON2: SD A/Dx Control Register 2
- SDxCON3: SD A/Dx Control Register 3
- SDxRESH: SD A/Dx Result Register High Word
- SDxRESL: SD A/Dx Result Register Low Word

All registers are mapped in the data memory space. The SDxCON1, SDxCON2 and SDxCON3 registers ([Register 66-1](#), [Register 66-2](#) and [Register 66-3](#)) control the overall operation of the SD A/D module. This includes enabling the module, configuring the conversion clock and voltage reference sources, setting dithering scheme, gain of the input amplifier, oversampling frequency and so on.

Register 66-1: SDxCON1: SD A/Dx Control Register 1

R/W-0	U-0	R/W-0	R/W-0	r-0	R/W-0	R/W-0	R/W-0
SDON	—	SDSIDL	SDRST	r	SDGAIN2	SDGAIN1	SDGAIN0
bit 15				bit 8			

R/W-0	R/W-0	U-0	R/W-0	U-0	R/W-0	R/W-0	R/W-0
DITHER1	DITHER0	—	VOSCAL	—	SDREFN	SDREFP	PWRLVL
bit 7				bit 0			

Legend:	r = Reserved bit
R = Readable bit	W = Writable bit
U = Unimplemented bit, read as '0'	
-n = Value at POR	'1' = Bit is set
	'0' = Bit is cleared
	x = Bit is unknown

- bit 15 **SDON:** SD A/D Module Enable bit
1 = Sigma-Delta A/D module is enabled
0 = Sigma-Delta A/D module is disabled
- bit 14 **Unimplemented:** Read as '0'
- bit 13 **SDSIDL:** SD A/D Stop in Idle Mode bit
1 = Discontinues module operation when device enters Idle mode
0 = Continues module operation in Idle mode
- bit 12 **SDRST:** SD A/D Reset bit
1 = Resets all SD A/D module circuitry (analog section remains in Reset as long as SDRST bit is set)
0 = Releases from Reset (Run mode)
- bit 11 **Reserved:** Always maintain as '0' for proper A/D operation
- bit 10-8 **SDGAIN<2:0>:** SD A/D Gain Control bits
11x = Reserved; do not use
101 = 1:32
100 = 1:16
011 = 1:8
010 = 1:4
001 = 1:2
000 = 1:1
- bit 7-6 **DITHER<1:0>:** Dithering Mode Select bits
11 = High Dither mode (preferred with higher Oversampling Ratio (OSR) and positive reference below SVDD)
10 = Medium Dither mode (preferred for low-to-medium OSR and positive reference below SVDD)
01 = Low Dither mode (preferred when the positive reference is at or near SVDD)
00 = No Dither mode
- bit 5 **Unimplemented:** Read as '0'
- bit 4 **VOSCAL:** Internal Offset Measurement Enable bit
1 = Converter is configured to sample its own internal offset error
0 = Converter is configured for normal operation
- bit 3 **Unimplemented:** Read as '0'
- bit 2 **SDREFN:** SD A/D Negative Voltage Reference Configuration bit
1 = SVREF- pin
0 = SVSS
- bit 1 **SDREFP:** SD A/D Positive Voltage Reference Configuration bit
1 = SVREF+ pin
0 = SVDD
- bit 0 **PWRLVL:** Analog Amplifier Bandwidth Select bit
1 = 2x bandwidth (higher power consumption compared to normal bandwidth)
0 = Normal bandwidth

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Register 66-2: SDxCON2: SD A/Dx Control Register 2

R/W-0	R/W-0	R/W-0	R/W-0	U-0	U-0	R/W-0	R/W-0
CHOP1	CHOP0	SDINT1	SDINT0	—	—	SDWM1	SDWM0
bit 15						bit 8	

U-0	U-0	U-0	R/W-0	R/W-0	U-0	U-0	HS/C-0
—	—	—	RNDRES1	RNDRES0	—	—	SDRDY
bit 7						bit 0	

Legend:	HS = Hardware Settable bit	C = Clearable bit
R = Readable bit	W = Writable bit	U = Unimplemented bit, read as '0'
-n = Value at POR	'1' = Bit is set	'0' = Bit is cleared
		x = Bit is unknown

bit 15-14 **CHOP<1:0>:** Amplifier Chopping Enable bits

- 11 = Chopping is enabled
- 10 = Reserved
- 01 = Reserved
- 00 = Chopping is disabled

bit 13-12 **SDINT<1:0>:** SD A/D Interrupt Event Generation Select bits

- 11 = Interrupt on every data output
- 10 = Interrupt on every fifth data output
- 01 = Interrupt when the new result is less than the old result
- 00 = Interrupt when the new result is greater than the old result

bit 11-10 **Unimplemented:** Read as '0'

bit 9-8 **SDWM<1:0>:** SD A/D Output Result Register Write Mode bits

- 11 = Reserved; do not use
- 10 = SDxRESH/SDxRESL are never updated (used for > or < threshold compare)
- 01 = SDxRESH/SDxRESL are updated on every interrupt
- 00 = SDxRESH/SDxRESL are updated on every interrupt after SDRDY is cleared to '0' by the software

bit 7-5 **Unimplemented:** Read as '0'

bit 4-3 **RNDRES<1:0>:** Round Data Control bits

- 11 = Round result to 8 bits
- 10 = Round result to 16 bits
- 01 = Round result to 24 bits
- 00 = No Rounding (32 bits)

bit 2-1 **Unimplemented:** Read as '0'

bit 0 **SDRDY:** SD A/D Filter Data Ready bit (set by hardware)

- 1 = SINC filter delay is satisfied (clear this bit in software)
- 0 = SINC filter delay is not satisfied

Register 66-3: SDxCON3: SD A/Dx Control Register 3

R/W-0	R/W-0	R/W-0	R/W-0	R/W-0	R/W-0	R/W-0	R/W-0
SDDIV2 ⁽¹⁾	SDDIV1 ⁽¹⁾	SDDIV0 ⁽¹⁾	SDOSR2	SDOSR1	SDOSR0	SDCS1	SDCS0
bit 15							bit 8
U-0	U-0	U-0	U-0	U-0	R/W-0	R/W-0	R/W-0
—	—	—	—	—	SDCH2 ⁽³⁾	SDCH1 ⁽³⁾	SDCH0 ⁽³⁾
bit 7							bit 0

Legend:

R = Readable bit

W = Writable bit

U = Unimplemented bit, read as '0'

-n = Value at POR

'1' = Bit is set

'0' = Bit is cleared

x = Bit is unknown

- bit 15-13 **SDDIV<2:0>**: SD A/D Input Clock Divider/Postscaler Ratio bits⁽¹⁾
- 111 = Reserved
 - 110 = 64
 - 101 = 32
 - 100 = 16
 - 011 = 8
 - 010 = 4
 - 001 = 2
 - 000 = 1 (No divider, clock selected by SDCS<1:0> is provided directly to A/D)
- bit 12-10 **SDOSR<2:0>**: SD A/D Oversampling Ratio (OSR) Selection bits
- 111 = Reserved
 - 110 = 16 (fastest conversion results, low quality)
 - 101 = 32
 - 100 = 64
 - 011 = 128
 - 010 = 256
 - 001 = 512
 - 000 = 1024 (slowest conversion results, best quality)
- bit 9-8 **SDCS<1:0>**: SD A/D Clock Source Select bits
- 11 = Reserved
 - 10 = Primary Oscillator (OSCI/CLKI)
 - 01 = FRC (8 MHz)⁽²⁾
 - 00 = System clock (FOSC/2)
- bit 7-3 **Unimplemented**: Read as '0'
- bit 2-0 **SDCH<2:0>**: SD A/D Analog Channel Input Select bits⁽³⁾

- Note 1:** To avoid overclocking or underclocking the module, set SDDIV<2:0> to obtain an A/D clock frequency (input frequency selected by SDCS<1:0> is divided by the SDDIVx postscaler) at or between 1 MHz and 4 MHz.
- 2:** The 8 MHz FRC output is used directly, prior to the FRCDIV<2:0> postscaler.
- 3:** Refer to the specific device data sheet for more information on the actual number and type of channels available.

66.3 SD A/D MODULE CONFIGURATION

All of the registers described in the previous section must be configured for the module operation. The various configuration and control functions of the module are distributed throughout the module's three control registers. Control functions can be sorted into four groups: input, timing, conversion and output. [Table 66-1](#) shows the location of the control or status bits by register.

Table 66-1: SD A/D Module Functions by Registers and Bits

SD A/D Function	Register(s)	Specific Bits
Input	SDxCON1	PWRLVL, SDREFF, SDREFN, VOSCAL, DITHER<1:0>, SDGAIN<2:0>
	SDxCON2	CHOP<1:0>
	SDxCON3	SDCH<2:0>
Timing	SDxCON3	SDCS<1:0>, SDDIV<2:0>
Conversion	SDxCON1	SDON, SDSIDL
	SDxCON2	SDINT<1:0>
Output	SDxCON2	RNDRES<1:0>, SDWM<1:0>
	SDxCON3	SDOSR<2:0>

The following steps should be followed for performing an SD A/D conversion.

1. Configure the SD A/D module:
 - a) Select the power level
 - b) Select the voltage reference source to match the expected range on the analog inputs
 - c) Set the dither value
 - d) Set the gain
 - e) Select the output resolution
 - f) Select the Data Overwrite mode
 - g) Select the Chopping Clock mode
 - h) Select the input channel
 - i) Select the SD A/D module clock source
 - j) Select the desired output data rate
 - k) Select the sampling frequency
2. Configure the SD A/D interrupt (if required):
 - a) Select the Interrupt mode
 - b) Clear the SDAxIF bit
 - c) Select the SD A/D interrupt priority
3. Turn on the SD A/D module.

The options for each configuration step are described in the subsequent sections. [Example 66-1](#) shows the possible initialization sequence.

Example 66-1: Channel Measurement

```
signed short int count;
signed short int result;

// Configure the SD A/D module
SD1CON1bits.PWRLVL = 0; // Low power, normal bandwidth
SD1CON1bits.SDREFP = 0; // Positive Voltage Reference is SVDD
SD1CON1bits.SDREFN = 0; // Negative Voltage Reference is SVSS
SD1CON1bits.VOSCAL = 0; // Internal Offset Measurement is disabled
SD1CON1bits.DITHER = 1; // Low Dither
SD1CON1bits.SDGAIN = 0; // Gain is 1:1

SD1CON2bits.RNDRES = 2; // Round result to 16-bit
SD1CON2bits.SDWM = 1; // SDxRESH/SDxRESL updated on every Interrupt
SD1CON2bits.CHOP = 3; // Chopping should be enabled

SD1CON3bits.SDCH = 0; // Channel 0 (see the specific device data sheet)
SD1CON3bits.SDCS = 1; // Clock Source is a 8 MHz FRC
SD1CON3bits.SDOSR = 0; // Oversampling Ratio (OSR) is 1024 (best quality)
SD1CON3bits.SDDIV = 1; // Input Clock Divider is 2 (SD ADC clock is 4MHz)

// Configure SD A/D interrupt
SD1CON2bits.SDINT = 3; // Interrupt on every data output
IFS6bits.SDA1IF = 0; // Clear interrupt flag

// Turn on the SD A/D module
SD1CON1bits.SDON = 1;

// Wait for a minimum of five interrupts to be generated. Need to throw at least
// the first four away when using interrupt every period option, since the
// low pass SINC filter needs to be flushed with new data when we change
// ADC channel or initialize the ADC.
for(count=0; count<8; count++)
{
    //Clear interrupt flag.
    IFS6bits.SDA1IF = 0;
    //Wait for the result ready.
    while(IFS6bits.SDA1IF == 0);
}

// Channel #0 conversion result.
result = SD1RESH;
```

66.3.1 Selecting the Voltage Reference Source

The voltage references for SD A/D conversions are selected using the SDREFN and SDREFP control bits (SDxCON1<2:1>). The upper voltage reference (VREF+) can be SVDD or the voltage applied on the SVREF+ pin. The lower voltage reference (VREF-) can be SVSS or the voltage applied on the SVREF- pin.

66.3.2 Gain Control

The Programmable Gain Amplifier (PGA) block can be used to amplify very low signals, but the differential input range of the SD A/D module should not be exceeded to avoid a saturated output result code. The PGA is controlled by the SDGAIN<2:0> bits in the SDxCON1 register. [Table 66-2](#) shows the PGA configuration settings.

Table 66-2: PGA Configuration Setting

SDGAIN<2:0>	Gain (V/V)	Gain (dB)	Differential VIN Range (V)
000	1	0	$\pm \text{Reference} $
001	2	6	$\pm (\text{Reference} / 2)$
010	4	12	$\pm (\text{Reference} / 4)$
011	8	18	$\pm (\text{Reference} / 8)$
100	16	24	$\pm (\text{Reference} / 16)$
101	32	30	$\pm (\text{Reference} / 32)$

66.3.2.1 SELECTION OF ANALOG AMPLIFIER BANDWIDTH

Setting the PWRLVL bit in the SDxCON1 register improves the bandwidth and response time of the input amplifier, but at the cost of consuming more power. Smaller response time decreases the settling error in amplifier stages, and thereby, improves harmonic distortions. The PWRLVL bit can be cleared to save power.

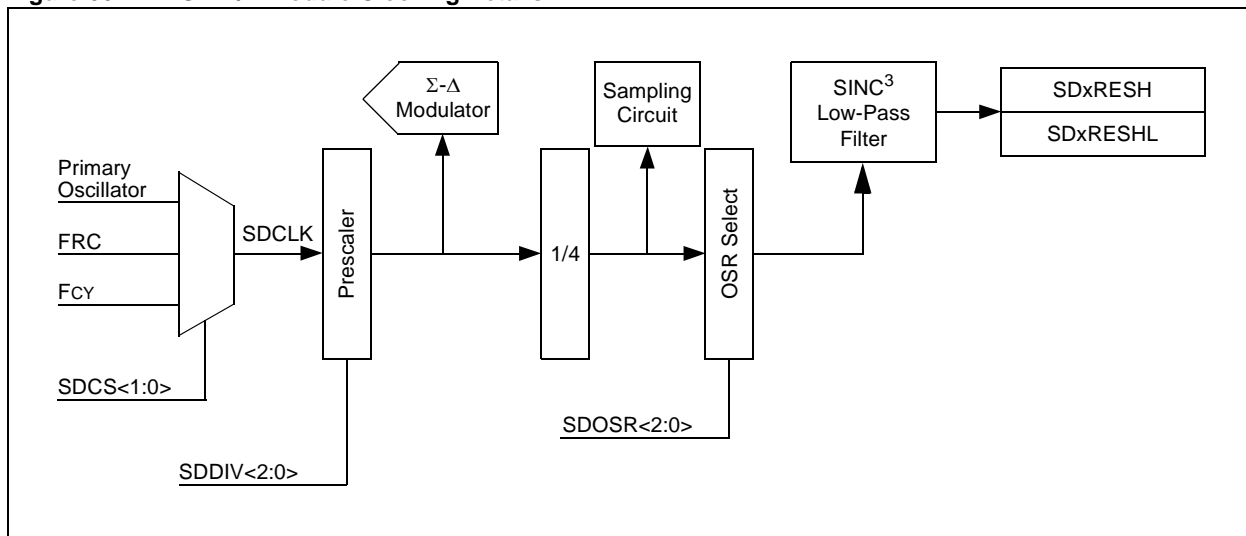
66.3.3 Selecting the SD A/D Module Clock Source

The clock source to the SD A/D Converter (SDCLK) can be one of the following three possible clocks:

- Primary Oscillator
- FRC (8 MHz)
- System Clock (FCY)

The selection can be done using the SDCS<1:0> bits in the SDxCON3 register. The A/D Converter source clock is divided inside of the module, as shown in [Figure 66-2](#).

Figure 66-2: SD A/D Module Clocking Details



66.3.3.1 SELECTING THE SD A/D CONVERTER SAMPLING FREQUENCY AND OUTPUT DATA RATE

The selected source clock (see SDCLK in [Figure 66-2](#)) is divided by the selected ratio determined by the SDDIV<2:0> bits in the SDxCON3 register. The resulting clock must be in the range of 1 MHz to 4 MHz.

The module clock is further divided by the selected ratio determined by the SDOSR<2:0> bits in the SDxCON3 register. The result is the rate at which the SD A/D outputs new data. To determine the SDOSR<2:0> setting for the required output data rate, see [Table 66-3](#).

Table 66-3: Data Rate Settings

Data Rate (ksp/s)	A/D Clock Frequency (MHz)	SDOSR<2:0>
62.500	4	110
31.2500	4	101
	2	110
15.6250	4	100
	2	101
	1	110
7.8125	4	011
	2	100
	1	101
3.90625	4	010
	2	011
	1	100
1.953125	4	001
	2	010
	1	011
0.9765625	4	000
	2	001
	1	010
0.4883	2	000
	1	001
0.2441	1	000

66.3.4 Selection of Dither

Dithering helps in suppressing the Idle tones present in the SD A/D module. Dithering is the process of intentionally adding a small error to the SD A/D feedback loop in order to “decorrelate” the outputs and “break” the Idle tones’ behavior. The dithering process scrambles the Idle tones into baseband white noise and ensures that dynamic parameters (SNR, SINAD, THD, SFDR) are less signal dependent.

The baseband white noise can then be filtered and averaged through the internally provided hardware Oversampling Ratio (OSR) to achieve better result quality. It is therefore, generally recommended to enable dithering in most applications. However, the optimum dithering setting for a given application depends on the OSR selected and the relative voltage of the applied A/D reference to the V_{DD} voltage.

The higher OSR settings (e.g., 128 to 1024) are better at averaging the baseband white noise introduced by dithering. At the higher OSR settings, enabling dithering improves overall performance and is recommended. At lower OSR settings (e.g., 16 to 64), the averaging is not as effective and the extra noise injected by dithering can potentially decrease the SNR, even though it may help to suppress Idle tones. At the very low OSR settings, this results in a performance trade-off, and in some cases (at very low OSR), it may be preferable to disable dithering. The dithering can be enabled by the DITHER<1:0> bits in the SDxCON1 register.

66.3.5 Chop Clock Select Mode

Chopping Select bits (CHOP<1:0> bits in the SDxCON2 register) are mainly used to remove the $1/f$ noise from the signal band in the modulator. To obtain 16-bit resolution, chopping should always be enabled.

66.3.6 SINC³ Filter and Rounding of Result

The SD A/D module includes a decimation filter that is a third-order SINC (or notch) filter. This filter processes the multibit stream into 8, 16, 24 or 32-bit words (depending on the RNDRES<1:0> bits in the SDxCON2 register). The settling time of the filter is three data rate periods. It is recommended to discard unsettled data. This can be done by polling the SDRDY bit in the SDxCON2 register. The output of the SINC filter is padded with Least Significant zeros for any resolution less than 32 bits.

66.3.7 Data Write Modes

The Write mode bits, SDWM<1:0> (SDxCON2) determine when the Result register is updated.

There are three options in Write mode:

- Result register is never updated (useful for the Threshold Compare feature)
- Result register is updated on every interrupt
- Result register is updated on every interrupt, after the SDRDY bit (SDxCON2<0>) is cleared to ‘0’ in software

66.3.7.1 RESULT DATA READY

The conversion result value is valid only when the SDRDY bit (SDxCON2<0>) reads as ‘1’ and the A/D module interrupt is generated. The SDRDY bit can only be cleared by software or on a device Reset. If the user sets the SDWM<1:0> bits (SDxCON2<9:8>) to ‘00’, then the SD A/D Converter does not update the Result register with the result until the SDRDY bit is cleared in software.

If SDWM<1:0> = 01, then the SD A/D Converter updates the Result register, regardless of the SDRDY bit.

66.3.7.2 SD A/D RESULT REGISTER

The SD A/D Converter result is stored in the SDxRESH:SDxRESL registers as a 32-bit 2's complement signed number. When rounding is enabled (RNDRES<1:0> bits in the SDxCON2 register), the Least Significant bits (LSbs) which are rounded off will read as '0'. Additionally, when operating the A/D at OSR values less than 1024, not enough data will be internally generated by the hardware to fill all 32 bits of the SDxRESH/SDxRESL register pair. Therefore, especially when operating at OSR values less than 1024, some of the LSbs will be fully meaningless and may read back as static values. The Most Significant bits (MSbs) of the 32-bit result will remain meaningful, but will be restricted by the result quality for a given OSR setting.

66.3.8 Input Channel Selection

The input analog channel can be selected using the SDCH<2:0> bits in the SDxCON3 register. The SD A/D Converter comprises both single-ended and differential channels. The number of single-ended and differential channels available is device-specific. For more information, refer to the specific device data sheet. Each differential channel has inverting (CHn-) and non-inverting inputs (CHn+). If any channel is not used, both CHn- and CHn+ should be connected to SVSS. Single-ended channels are available as the positive input to the SD A/D Converter. The negative input for these channels will always be SVSS. This performs a single-ended conversion on the selected channel. When performing single-ended conversions, the A/D module still produces two's complement signed numbers. It is therefore possible for slightly negative number results to occur due to offset error. The application firmware should check for negative results, and treat all negative numbers as '0', if true unsigned single-ended behavior is desired. The module also facilitates a dedicated differential channel used for gain error measurements. For more information on gain error measurements, see [Section 66.6 "Gain Error Calibration"](#).

66.3.9 Channel Switching

When the channel is switched, five data rate clocks are required to flush the previous contents from the SINC³ filter (low-pass filter). During this time, the results generated may not be valid. The results can have residual voltage information from the previous selected channel due to the low-pass filtering effect of the SINC filter. Therefore, it is recommended to select an interrupt on every fifth data output by setting the SDINT<1:0> bits to '10', and upon first interrupt, the required interrupt can be selected.

66.3.10 Enabling the Module

The SD A/D is enabled by setting the SDON bit (SDxCON1<15>) to '1'. After SDON is set to '1', the module requires a power-up analog settling time of about 10 µs, during which time, the A/D Converter may not generate accurate results. The application firmware should therefore throw away the initial results associated with the analog settling time and the time it takes to flush the low-pass SINC filter with new data. When SDON is set to '0', the module is disabled; the digital and analog portions of the module are turned off for maximum current savings.

The module should be configured to the required settings and the SDRDY bit (SDxCON2<0>) should be cleared before turning on the SD A/D.

Note: When the module is enabled (SDON = 1), the firmware may change the channel selection (SDCH<2:0> or VOSCAL bit settings). If other A/D module settings must be changed, the module should be disabled first (SDON = 0).

66.4 INTERRUPT MODES

There are four Interrupt modes:

- Interrupt on Every Data Ready mode
- Interrupt on every Fifth Data Ready mode
- Interrupt on Threshold Current Value Lesser than the Previous Value mode
- Interrupt on Threshold Current Value Greater than the Previous Value mode

The four modes are controlled by the SDINT<1:0> bits in the SDxCON2 register.

66.4.1 Interrupt on Every Data Ready

This mode is useful after the first data is read. Note that the frequency of the data rate clock must be less than $F_{CY}/2$ to avoid missing the samples.

66.4.2 Interrupt on Every Fifth Data Ready

This mode is useful immediately after the module is enabled, when the input channel is switched or when the device wakes up from Sleep. The module takes five data rate clocks for flushing old values from the SINC³ filter.

66.4.3 Interrupt on Threshold Compare

The SDINT<1:0> bits allow the user to have interrupts, only if the data coming out of the filter is greater than, or less than, the value currently in the Result register. If the user wishes to retain the threshold value, the SDWM<1:0> bits must be set to '10'. If SDWM<1:0> are set to '01' or '00', the threshold compare function turns into a "find the lowest/highest sampled value" type of operation.

66.5 OFFSET CALIBRATION

The SD A/D module includes a feature for measuring internal offset error. Application firmware can enable this feature by fully configuring the A/D module and setting the VOSCAL bit in the SDxCON1 register.

When the VOSCAL bit is set, the A/D module ignores the currently selected SDCH<2:0> channel select bits, and instead, performs conversions on an internal channel that generates SDxRESH/SDxRESL results which represent the offset error of the A/D Converter. After the SINC filter delay has been satisfied, the application firmware should save the result value (which will be formatted as a signed two's complement value, the same as normal A/D module results). After the offset error measurement result data has been saved, the application firmware can clear the VOSCAL bit to switch the A/D module back into Normal Channel Measurement mode.

The application firmware can then correct for offset error in subsequent analog channel measurements by subtracting the offset error value from all future results. [Equation 66-1](#) shows the method for correcting offset error.

Equation 66-1: Correcting Offset Error

$\text{Offset Corrected Result} = (\text{New SDxRES Result}) - (\text{VOSCAL Result})$
--

[Example 66-2](#) shows an example C code for the offset calibration.

66.6 GAIN ERROR CALIBRATION

The SD A/D module has a dedicated channel for gain error measurements. Selecting this channel allows the module to internally sample the reference voltages as an input to the A/D Converter. In theory, if the A/D module has no offset or gain error, this always generates an SDxRESH result of 32767 (maximum full-scale value). In practice, it is normal for a real world Sigma-Delta A/D to have a significant, but highly correctable, gain error. In this design, the A/D Converter has a gain error, which will cause the A/D Converter to generate a result that is somewhat less than the theoretical 32767 full-scale result value. Gain error varies depending on the SDGAIN<2:0> bits (SDxCON1<10:8>) selection, and some part, to part variations. [Equation 66-2](#) shows the method for calculating a floating-point “gain error correction factor”, which can be multiplied against subsequent result values to compensate for gain error.

Equation 66-2: Gain Error Correction Factor

$$G_{ECF} = \frac{ADRES_{EXPECTED}}{ADRES_{RAW} - ADRES_{VOSCAL}}$$

Legend:

G_{ECF} = Floating-point gain error correction factor

$ADRES_{EXPECTED}$ = The expected A/D result (with no offset or gain error) based on the applied input voltage. When the A/D is configured to measure the references in 1x Gain mode, the expected result is 32767.

$ADRES_{RAW}$ = The raw, uncorrected 16-bit signed result read from the SDxRESH register.

$ADRES_{VOSCAL}$ = The 16-bit value representing the offset error measured with the A/D “VOSCAL” feature.

[Example 66-2](#) shows the process to perform a channel measurement, which uses offset correction and gain error correction to compute a fully calibrated result value.

66.6.1 Considerations for Gain Error

The following needs to be considered for gain error correction:

1. Gain error of the A/D Converter can be measured by intentionally inputting a well-known analog voltage and comparing the output result code with the theoretically expected result code.
2. If the VREF+/VREF- reference sources are measured (with a gain setting of 1), the result code should theoretically be full scale (e.g., 32767).
3. Due to the internal gain error of the A/D design, the actual value will be less than full scale.
4. A floating-point gain error correction factor can be computed using the raw A/D Converter result (e.g., when measuring the references) and the expected result, as shown in [Example 66-2](#).
5. Once the gain error correction factor is known, it can be saved in the firmware. While carrying out A/D Converter measurements on other channels/input voltage sources, the gain error correction factor can be multiplied by the raw result (after offset correction through VOSCAL) to compensate for the A/D Converter gain error.

Example 66-2: Channel Conversion with Offset and Gain Correction

```
unsigned char    count;
signed short int offset;
signed long int  maxValue;
const double     expectedMaxValue = 32767; // 0x7FFF
double          gain;
signed long int  channelValue;
double          calibratedResult;

// ADC initialization.
SD1CON1bits.PWRLVL = 0; // Low power, normal bandwidth
SD1CON1bits.SDREFP = 0; // Positive Voltage Reference is SVDD
SD1CON1bits.SDREFN = 0; // Negative Voltage Reference is SVSS
SD1CON1bits.VOSCAL = 1; // Internal Offset Measurement Enable
SD1CON1bits.DITHER = 1; // Low Dither
SD1CON1bits.SDGAIN = 0; // Gain is 1:1

SD1CON2bits.RNDRES = 2; // Round result to 16-bit
SD1CON2bits.SDWM   = 1; // SDxRESH/SDxRESL updated on every Interrupt
SD1CON2bits.SDINT   = 3; // Interrupt on every data output
SD1CON2bits.CHOP    = 3; // Chopping should be enabled

SD1CON3bits.SDCH    = 0; // Channel 0 (see the specific device data sheet)
SD1CON3bits.SDCS    = 1; // Clock Source is a 8 MHz FRC
SD1CON3bits.SDOSR   = 0; // Oversampling Ratio (OSR) is 1024 (best quality)
SD1CON3bits.SDDIV   = 1; // Input Clock Divider is 2 (SD ADC clock is 4MHz)

// Enable ADC.
SD1CON1bits.SDON    = 1;

// Wait for a minimum of five interrupts to be generated. Need to throw at least
// the first four away when using interrupt every period option, since the
// low pass SINC filter needs to be flushed with new data when we change
// ADC channel or initialize the ADC.
for(count=0; count<8; count++)
{
    //Clear interrupt flag.
    IFS6bits.SDA1IF = 0;
    //Wait for the result ready.
    while(IFS6bits.SDA1IF == 0);
}

// Save the initial VOSCAL offset value from the ADC.
// This value will be used to correct the offset error for all
// subsequent ADC measurements.
offset = SD1RESH;

// Configure ADC for normal operation.
SD1CON1bits.VOSCAL = 0;

// Measures the reference selected by SDREFP/SDREFN
// (see the specific device data sheet).
SD1CON3bits.SDCH    = 3;
```


Example 66-2: Channel Conversion with Offset and Gain Correction (Continued)

```
// Wait for a minimum of five interrupts to be generated.
for(count=0; count<8; count++)
{
    //Clear interrupt flag.
    IFS6bits.SDA1IF = 0;
    //Wait for the result ready.
    while(IFS6bits.SDA1IF == 0);
}

// Save the maximum value to calculate the gain.
maxValue = (signed short int) SD1RESH;

// Calculate gain.
gain = expectedMaxValue/((double)(maxValue-offset));

// Select channel #0 for the measurement.
// (see the specific device data sheet).
SD1CON3bits.SDCH = 0;

// Adjust loop end value for desired settling time (value must be >= 5).
for(count=0; count<8; count++)
{
    // Clear interrupt flag.
    IFS6bits.SDA1IF = 0;
    // Wait for the result ready.
    while(IFS6bits.SDA1IF == 0);
}

// Not corrected channel #0 result.
channelValue = (signed short int) SD1RESH;

// Correct offset.
channelValue -= offset;

// Correct gain.
calibratedResult = gain*channelValue;
```

66.6.2 Gain Error Correction Using Integer Math

Floating-point math is easy and convenient to use, but is often undesirable during actual execution because it generally consumes a large amount of CPU execution time and code space to perform. Fortunately, it is possible to correct for A/D gain error using pure integer-based math.

In mathematical theory, if an expression is multiplied by a constant and then divided by the same constant, the expression stays the same ([Equation 66-3](#)).

Equation 66-3: Math Example

$$1.2345 = (1.2345 \cdot 2^{18}) / 2^{18}$$

In this example, multiplying and dividing by a constant number (2^{18} or 262144) does not change the value of the expression. Using this approach, it is possible to convert floating-point numbers into integers with minimal truncation error ([Equation 66-4](#)).

Equation 66-4: Converting Floating Point Number into Integer Example

$$1.23456 = 1.23456 \cdot 262144 / 262144 \cong 323623 / 262144$$

The final expression (323623/262144) is very easy and quick to evaluate by the microcontroller during run time, as it is equivalent to taking a 32-bit integer and arithmetically shifting it to the right by 18 bit positions (equivalent to dividing by 262144). This mathematical principle can therefore be used to perform traditional floating-point math operations, using integer math with very little error, provided that the constant number being multiplied and divided by is a very large number (ex: 2^{18} or larger). [Example 66-3](#) shows how this principle can be used to correct for the Sigma-Delta A/D gain error quickly during application run time.

Example 66-3: Channel Conversion with Offset and Gain Correction Using Integer Math

```
// This is fixed point location
#define FIXED_POINT_POS          15

unsigned char    count;
signed long int  offset;
signed long int  maxValue;
signed long int  expectedMaxValue;
signed long int  gain;
signed long int  channelValue;
signed short int calibratedResult;

// ADC initialization.
SD1CON1bits.PWRLVL = 0; // Low power, normal bandwidth
SD1CON1bits.SDREFP = 0; // Positive Voltage Reference is SVDD
SD1CON1bits.SDREFN = 0; // Negative Voltage Reference is SVSS
SD1CON1bits.VOSCAL = 1; // Internal Offset Measurement Enable
SD1CON1bits.DITHER = 1; // Low Dither
SD1CON1bits.SDGAIN = 0; // Gain is 1:1

SD1CON2bits.RNDRES = 2; // Round result to 16-bit
SD1CON2bits.SDWM   = 1; // SDxRESH/SDxRESL updated on every Interrupt
SD1CON2bits.SDINT   = 3; // Interrupt on every data output
SD1CON2bits.CHOP    = 3; // Chopping should be enabled

SD1CON3bits.SDCH    = 0; // Channel 0 (see the specific device data sheet)
SD1CON3bits.SDCS    = 1; // Clock Source is a 8 MHz FRC
SD1CON3bits.SDOSR    = 0; // Oversampling Ratio (OSR) is 1024 (best quality)
SD1CON3bits.SDDIV   = 1; // Input Clock Divider is 2 (SD ADC clock is 4MHz)
```

Example 66-3: Channel Conversion with Offset and Gain Correction Using Integer Math (Continued)

```
// Enable ADC
SD1CON1bits.SDON = 1;

// Wait for a minimum of five interrupts to be generated. Need to throw at least
// the first four away when using interrupt every period option, since the
// low pass SINC filter needs to be flushed with new data when we change
// ADC channel or initialize the ADC.
for(count=0; count<8; count++)
{
    //Clear interrupt flag.
    IFS6bits.SDA1IF = 0;
    //Wait for the result ready.
    while(IFS6bits.SDA1IF == 0);
}

// Save the initial VOSCAL offset value from the ADC.
// This value will be used to correct the offset error for all
// subsequent ADC measurements.
offset = (signed short int)SD1RESH;

// Configure ADC for normal operation.
SD1CON1bits.VOSCAL = 0;

// Measures the reference selected by SDREFP/SDREFN
// (see the specific device data sheet).
SD1CON3bits.SDCH = 3;

// Wait for a minimum of five interrupts to be generated.
for(count=0; count<8; count++)
{
    //Clear interrupt flag.
    IFS6bits.SDA1IF = 0;
    //Wait for the result ready.
    while(IFS6bits.SDA1IF == 0);
}

// Save the maximum value to calculate the gain.
maxValue = (signed short int)SD1RESH;

// Correct offset.
maxValue -= offset;

// Calculate gain.
expectedMaxValue = ((signed long int)32767)<<FIXED_POINT_POS;
gain = expectedMaxValue/maxValue;

// Select channel #0 for the measurement.
// (see the specific device data sheet).
SD1CON3bits.SDCH = 0;
```

Example 66-3: Channel Conversion with Offset and Gain Correction Using Integer Math (Continued)

```
// Adjust loop end value for desired settling time (value must be >= 5).
for(count=0; count<8; count++)
{
    // Clear interrupt flag.
    IFS6bits.SDA1IF = 0;
    // Wait for the result ready.
    while(IFS6bits.SDA1IF == 0);
}

// Not corrected channel #0 result.
channelValue = (signed short int)SD1RESH;

// Correct offset.
channelValue -= offset;

// Correct gain.
channelValue = gain*channelValue;

// 16-bit signed result
calibratedResult = channelValue>>FIXED_POINT_POS;
```

66.7 OPERATION DURING SLEEP AND IDLE MODES

The SD A/D module supports the following power modes:

- Idle mode
- Sleep mode

66.7.1 Idle Mode

The SDSIDL bit (SDxCON1<13>) determines whether the module stops or continues operation in Idle mode. If SDSIDL = 0, the module will continue normal operation when the device enters Idle mode. If the SD A/D interrupt is enabled (SDAxIE = 1), the device will wake-up from Idle mode when the SD A/D Interrupt Flag (SDAxIF = 1) occurs. Program execution will resume at the SD A/D Interrupt Service Routine (ISR) if the SD A/D interrupt is greater than the current CPU priority. Otherwise, execution will continue from the instruction after the `PWRSV #IDLE` instruction (or `Idle()`) that placed the device in Idle mode.

If SDSIDL = 1, the module will stop in Idle. If the device enters Idle mode in the middle of a conversion, the conversion is aborted. The converter will not resume a partially completed conversion on exiting from Idle mode.

66.7.2 Sleep Mode

If the device enters Sleep mode while the module is operating, the module will be suspended in its current state until clock execution resumes. The output value will retain the last value before Sleep.

66.8 EFFECTS OF A RESET

A device Reset forces all registers to their Reset state. This forces the SD A/D module to be turned off and any conversion in progress to be aborted. All pins that are multiplexed with analog inputs will be configured as analog inputs. The corresponding TRIS bits will be set to '1'.

66.8.1 Use of SDRST

With the SDON bit set, the SDRST bit can be used for the module to switch between active and inactive modes quickly, without any delay that is associated with the SDON bit.

66.9 REGISTER MAP

Table 66-4: SD A/D Converter Register Map

File Name	Bit 15	Bit 14	Bit 13	Bit 12	Bit 11	Bit 10	Bit 9	Bit 8	Bit 7	Bit 6	Bit 5	Bit 4	Bit 3	Bit 2	Bit 1	Bit 0	All Resets
SDxCON1	SDON	—	SDSIDL	SDRST	r	SDGAIN2	SDGAIN1	SDGAIN0	DITHER1	DITHER0	—	VOSCAL	—	SDREFN	SDREFP	PWRLVL	0000
SDxCON2	CHOP1	CHOP0	SDINT1	SDINT0	—	—	SDWM1	SDWM0	—	—	—	RNDRES1	RNDRES0	—	—	SDRDY	0000
SDxCON3	SDDIV2	SDDIV1	SDDIV0	SDOSR2	SDOSR1	SDOSR0	SDCS1	SDCS0	—	—	—	—	—	SDCH2	SDCH1	SDCH0	0000
SDxRESH	Sigma-Delta A/D Result Register (bits 31-16)																0000
SDxRESL	Sigma-Delta A/D Result Register (bits 15-0)																0000

Legend: — = unimplemented, read as '0'. Reset values are shown in hexadecimal.

66.10 DESIGN TIPS

Question 1: *Can we scan more than one channel?*

Answer: Yes. However, it is recommended that you “flush” the pipeline before moving to the next channel or the result will be erroneous.

66.11 RELATED APPLICATION NOTES

This section lists application notes that are related to this section of the manual. These application notes may not be written specifically for the PIC24F device family, but the concepts are pertinent and could be used with modification and possible limitations. The current application notes related to the 16-Bit A/D Converter module are:

Title	Application Note #
Designing with the MCP3901 Dual Channel Analog-to-Digital Converters	AN1300

Note: Please visit the Microchip web site (www.microchip.com) for additional application notes and code examples for the PIC24F family of devices.
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66.12 REVISION HISTORY

Revision A (April 2013)

This is the initial released revision of this document.

NOTES:

Note the following details of the code protection feature on Microchip devices:

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ISBN: 978-1-62077-164-8

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