Using Hardware Controls with SigmaDSP GPIO Pins
by Brett Gildersleeve

INTRODUCTION
Several devices in the SigmaDSP® family include general-purpose input/output (GPIO) pins. These pins can be connected to external hardware elements, such as LEDs, push-buttons, or rotary encoders. Some of the GPIO pins can be multiplexed to an auxiliary ADC, allowing direct connection of a potentiometer or an analog control voltage to the IC. Each pin can operate in a variety of modes, including open-collector output, inputs with debounce, and outputs driven directly by the control port.

Within the SigmaStudio™ graphical programming software, GPIO pins can be assigned to control or be controlled by various parts of the audio signal processing program. The functionality of the pins can be changed during run-time operation of the device. Use of the GPIO pins for user interface functionality can reduce or eliminate the need for a microcontroller, which can greatly reduce the cost of a simple audio system.

This application note begins by describing the hardware interfacing necessary to effectively use the GPIOs. In addition, this application note discusses several cases that could greatly reduce the complexity of a SigmaDSP audio system.
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# REVISION HISTORY

9/10—Rev. 0 to Rev. A
Changes to Figure 16 and ADC Section and Figure 18: 6
Changes to Push-Button Volume Control—Up/Down Mute: 8
Additional GPIO Expansion Board Schematics Section: 15

12/08—Revision 0: Initial Version
HARDWARE INTERFACING

GPIO pins can be interfaced to several types of hardware controls. This section details several examples of these controls and their associated circuits. Unless otherwise mentioned, IOVDD = 3.3 V.

MOMENTARY PUSH-BUTTON INPUT

One of the simplest forms of a control input to the SigmaDSP is a push-button. In Figure 1, a momentary switch is used in an active-low configuration with a 10 kΩ pull-up resistor. More information about pin current limits can be found in the relevant device data sheet.

![Figure 1. Active-Low Push-Button Input Circuit Example](image1)

SigmaDSP GPIO pins feature a debounce circuit, which should be activated in SigmaStudio to avoid errors from contact chatter on switching. For active-low operation, as in this example circuit, the corresponding GPIO register should be set as an inverting input. For active-high operation, the register can be set as noninverting and the location of the circuit switch and resistor should be switched. For devices without hardware invert bits, a software logic inverter is available in SigmaStudio.

SINGLE-POLE SWITCH INPUT

Switches can be used in a similar fashion to momentary push-buttons. In the example shown in Figure 2, the switch is shown in the active-low configuration with a pull-up resistor. This switch is not momentary, in contrast to the example shown in Figure 1.

![Figure 2. Active-Low Switch Input Circuit Example](image2)

ROTARY ENCODER INPUT

A GPIO conditioning cell is available in SigmaStudio for incremental rotary encoders, also called relative rotary encoders. An incremental rotary encoder is essentially a knob that produces output pulses depending upon which direction it is turned. This should not be confused with an absolute rotary encoder, which has a set of binary codes corresponding to all possible angular positions. To use an incremental rotary encoder, connect the output pins to two GPIO pins. The corresponding block in SigmaStudio calculates the phase difference between pulses on the two pins to recognize clockwise and counter-clockwise turns.

![Figure 3. Rotary Encoder Input Circuit Example](image3)

POTENTIOMETER INPUT

A potentiometer can be used as an analog control, most often to adjust volume. Because the auxiliary ADC has linearly spaced steps, a linear potentiometer should be used for best results. A logarithmic lookup table can be implemented within software if a logarithmic control is desired. For some SigmaDSPs, the full-scale input voltage for the ADC is less than IOVDD. Because of this, a voltage divider should be used to appropriately scale the signal, as in the example circuit in Figure 4.

![Figure 4. Potentiometer Input Circuit Example](image4)

LED OUTPUT

An LED is the simplest form of GPIO output. A transistor is necessary to avoid excessive loading of the SigmaDSP. An example transistor is the MMBT2222A. The SigmaDSP can source or sink the current, as long as it does not exceed the current limitations as specified in the device data sheet. See the relevant data sheet for more details. For an active-high, current-sourcing implementation, the transistor should be NPN in a common-emitter (or N-channel in a common-source) configuration. For an active-low, current-sinking implementation, the transistor should be PNP in a common-collector (or P-channel in a common-drain) configuration.

![Figure 5. Current Sourcing LED Output Circuit Example](image5)
USING GPIO IN A SIGMASTUDIO PROJECT

GPIO pins are accessible in SigmaStudio via the General Purpose Input and General Purpose Output cells in the IO section of the ToolBox.

Click the drop-down menu for access to all available GPIO pins. These cells may be wired just as any other cell in Sigma-Studio, with control signal inputs and outputs shown in red. Auxiliary ADC input cells work in the same way.

SETTING UP GPIO PINS IN THE SIGMASTUDIO REGISTER WINDOW

GPIO pins must be independently configured using the Register Control window in SigmaStudio. Examples of GPIO register controls are shown in Figure 9, Figure 10, and Figure 11.

Here, the GPIO pins can be set appropriately depending on interface circuitry. Possible settings include Input GPIO Debounce, Input GPIO No Debounce, Output GPIO, Output GPIO Open Collector, and ADC. In addition, some devices can input or output digital audio data or clocks on these pins. On some devices, GPIO pins are labeled MP (for multipurpose), but they are used in the same way.

Consult the relevant device and evaluation board data sheets, available from Analog Devices, Inc., for more details about GPIO register settings.
Input GPIO Debounce

When connecting a switch or push-button to a GPIO pin, a common problem that arises is contact bounce (also called chatter). Due to various mechanical and electrical factors, a series of random oscillations may appear during switching. To mitigate these effects, a timing-based debounce circuit is integrated into the GPIO circuits of certain SigmaDSPs. The debounce time can generally be set in the GPIO section of the register control window. The core reads the input value from its associated register once per audio frame.

![Figure 12. Input GPIO with Debounce Data Flow](image1)

Input GPIO No Debounce

For inputs not subject to contact bounce effects, such as outputs from external logic ICs, the debounce circuit can be bypassed with this setting. The core reads the input value from its associated register once per audio frame.

![Figure 13. Input GPIO with No Debounce Data Flow](image2)

Output GPIO

This setting allows the pin to be used as a digital output. Typically, each pin can drive a maximum of a few milliamps. See the associated SigmaDSP data sheet for more information. The GPIO pin reads the output value from its associated register once per audio frame.

![Figure 14. Output GPIO Data Flow](image3)

Output GPIO Open Collector

This setting puts the pin in an open-collector or open-drain output mode (depending on the device’s internal circuitry) and requires an external pull-up resistor. The pull-up resistor can connect to a different IOVDD supply than that of the DSP; therefore, this mode is useful when interfacing with ICs at different logic levels. The GPIO pin reads the output value from its associated register once per audio frame.

![Figure 15. Output GPIO Open Collector Data Flow](image4)
**Input Driven by Control Port**

In this mode, the GPIO pin is bypassed, and the core reads its value from the associated register. This register's value can be written or read via the control port. This mode is useful for controlling elements of the signal flow with an external host controller. The core reads the input value from its associated register once per audio frame.

![Figure 16. Input Driven by Control Port Data Flow](image)

**Output Driven by Control Port**

In this mode, the signal flow in the core does not affect the output of the associated GPIO pin, and the pin reads its output value from the associated register. This register's value can be written or read via the control port.

This mode is useful for directly controlling circuitry, such as an LED, connected to the GPIO pins with an external host controller. The GPIO pin reads the output value from its associated register once per audio frame.

![Figure 17. Output Driven by Control Port Data Flow](image)

**ADC**

When the pin is set in ADC mode, it is used as one of the multiplexed inputs to the auxiliary ADC. On the ADAU170x, the invert bit should be activated for proper ADC function. The core reads the input value from the ADC once per audio frame, although the ADC sampling rate is dependent on the particular SigmaDSP being used.

![Figure 18. ADC Data Flow](image)
USING THE EVALUATION BOARD WITH THE GPIO BOARD

Some SigmaDSP evaluation packages include a daughterboard with reference GPIO interface circuits. It may be helpful to evaluate GPIO applications or algorithms by using this board.

The evaluation package includes:

- Four potentiometers
- Four LEDs with transistor drivers
- Eight momentary push-buttons
- Six slide switches
- One incremental rotary encoder
- Header connections for serial data I/O
- Prototyping area for user-created interfacing circuitry

Documentation for the daughterboard is included in the associated evaluation board kits.
EXAMPLE IMPLEMENTATIONS OF GPIO IN SOFTWARE

This section provides examples of some commonly used GPIO conditioning signal flows in the context of a SigmaStudio project.

The SigmaStudio software is continually updated with new algorithms, thus the appearance and function of some functional blocks may change over time. However, the ideas presented in this application note should be applicable in all future versions of the software.

The information in this application note corresponds to the Version 3.1 release of SigmaStudio.

PUSH-BUTTON VOLUME CONTROL—UP/DOWN/MUTE

This example uses two push-buttons to control the pushhold GPIO cell, which in turn controls volume with a lookup table. Pressing both buttons simultaneously mutes the audio output. This example is shown in Figure 20.

Pressing each push-button once increments or decrements the index. If a push-button is depressed for a duration defined by hold (ms), the index increments repeatedly at an interval defined by repeat (ms).

Two GPIO pins, GPIO_0 and GPIO_1, are used as inputs to the pushhold U1 cell, running the push/hold with a two-button mute algorithm. The three output pins of this cell, Up Pulse, Down Pulse, and Mute Pulse, are used as inputs to the look-up table, UpDownLUT1. Click Table to display the table values.

The table in Figure 19 shows an example of a volume curve ranging from 0 to 1. Press the Up push-button for the cell to output the next number in the table. Press the Down push-button for the cell to output the previous number in the table. Note that the Up and Down push-buttons are not visible in Figure 19.

In addition to control inputs and outputs (shown in red in Figure 20), the lookup table cell has a yellow interface register input and output, which are linked to the Interface Read and Interface Write cells. These cells enable writeback of parameter values from the ADAU1701 to an external EEPROM on power-down. For more information, see the ADAU1701 data sheet. For some SigmaDSPs, the interface read and interface write cells are implemented in software.

The lookup table output is sent as a control input to the SW volume cell. This cell also has two audio inputs (green) and two audio outputs (blue). The control input takes a value from the table and scales the audio accordingly. When a push-button is pressed, the table outputs a new value, and the volume cell slews to this new value based on the SW slew rate entered in the cell.
Table 1 shows several 33-point index lists of ascending and descending values for linear and logarithmic volume controls. A logarithmic index table is recommended for audio volume control applications.

### Table 1. Common 33-Point Index Table Values

<table>
<thead>
<tr>
<th>Logarithmic (Exponential) Curve</th>
<th>Linear Curve</th>
</tr>
</thead>
<tbody>
<tr>
<td>Ascending, -96 dB to 0 dB (3 dB per Step)</td>
<td>Descending, 0 dB to -96 dB (3 dB per Step)</td>
</tr>
<tr>
<td>1.5849E-05</td>
<td>1.0000</td>
</tr>
<tr>
<td>2.2387E-05</td>
<td>0.7079</td>
</tr>
<tr>
<td>3.1623E-05</td>
<td>0.5012</td>
</tr>
<tr>
<td>4.4668E-05</td>
<td>0.3548</td>
</tr>
<tr>
<td>6.3096E-05</td>
<td>0.2512</td>
</tr>
<tr>
<td>8.9125E-05</td>
<td>0.1778</td>
</tr>
<tr>
<td>1.2589E-04</td>
<td>0.1259</td>
</tr>
<tr>
<td>1.7783E-04</td>
<td>0.0891</td>
</tr>
<tr>
<td>2.5119E-04</td>
<td>0.0631</td>
</tr>
<tr>
<td>3.5481E-04</td>
<td>0.0447</td>
</tr>
<tr>
<td>5.0119E-04</td>
<td>0.0316</td>
</tr>
<tr>
<td>7.0795E-04</td>
<td>0.0224</td>
</tr>
<tr>
<td>0.0010</td>
<td>0.0158</td>
</tr>
<tr>
<td>0.0014</td>
<td>0.0112</td>
</tr>
<tr>
<td>0.0020</td>
<td>0.0079</td>
</tr>
<tr>
<td>0.0028</td>
<td>0.0056</td>
</tr>
<tr>
<td>0.0040</td>
<td>0.0040</td>
</tr>
<tr>
<td>0.0056</td>
<td>0.0028</td>
</tr>
<tr>
<td>0.0079</td>
<td>0.0020</td>
</tr>
<tr>
<td>0.0112</td>
<td>0.0014</td>
</tr>
<tr>
<td>0.0158</td>
<td>0.0010</td>
</tr>
<tr>
<td>0.0224</td>
<td>7.0795E-04</td>
</tr>
<tr>
<td>0.0316</td>
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<td>0.0447</td>
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<td>0.0631</td>
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<tr>
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<tr>
<td>0.1259</td>
<td>1.2589E-04</td>
</tr>
<tr>
<td>0.1778</td>
<td>8.9125E-05</td>
</tr>
<tr>
<td>0.2512</td>
<td>6.3096E-05</td>
</tr>
<tr>
<td>0.3548</td>
<td>4.4668E-05</td>
</tr>
<tr>
<td>0.5012</td>
<td>3.1623E-05</td>
</tr>
<tr>
<td>0.7079</td>
<td>2.2387E-05</td>
</tr>
<tr>
<td>1.0000</td>
<td>1.5849E-05</td>
</tr>
</tbody>
</table>

The number of points in a volume curve is not fixed at 33. It can be changed as required in a given application. Although values can be entered in the table in floating-point format, they are stored in the SigmaDSP in 5.23 decimal format. The calculation of gain values for logarithmic and linear curves is shown in Equation 1 and Equation 2.

For an ascending, x-point linear curve, the gain value $g$ for index $n$ can be calculated with Equation 1. The descending curve can be derived by reversing the indices.

$$g_n = \frac{n}{x}$$

where $n = 0$ to $x$.

For a descending, x-point exponential curve with gain steps of $s$ dB, the gain value ($g$) for index $n$ can be calculated with Equation 2. The ascending curve can be derived by reversing the indices.

$$g_n = 10^{-\frac{n 	imes s}{20}}$$

where $n = 0$ to $x$.

The cell names as they appear in the software and the number of each used in this example are as follows: GPIO Input (2), Interface Read (1), Push and Hold (1), Interface Write (1), Input (1), Up/Down Control with Look-up Table (1), Single Slew Ext Vol (1), Output (2).

In SigmaStudio Version 3.1 and later, a push-button volume cell is available. This cell combines the functionality of the Push and Hold, Up/Down Control, Index Lookup Table, and Single Slew Ext Vol cells. A simplified implementation using this block is shown in Figure 21.
ROTARY ENCODER VOLUME CONTROL

This example project uses the rotary encoder cell as a volume control. The PushHold cell is replaced by the RotEnc cell, with the remainder of the schematic equivalent to the previous example.

The top control input (red) to the RotEnc cell is the up pulse input (see Figure 23). The bottom control input is the down pulse input. The text entry block in the center of the cell sets the time constant in samples, which is 20 in this example.

Note that the back end of this signal flow (everything after the rotary encoder block) is the same as in the push-button volume control example described in the Push-Button Volume Control—Up/Down/Mute section.

The cells used in this example are as follows: GPIO Input (2), Interface Read (1), Rotary Encoder (1), Interface Write (1), Input (1), Single Slew Ext Vol (1), and Output (2).

In SigmaStudio Version 3.1 and later, a cell is available that combines the functionality of the Rotary Encoder, Up/Down Control, Index Lookup Table, and Single Slew Ext Vol cells. It is called the Rotary Volume cell. A simplified implementation using this block is shown in Figure 22.

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**Figure 22. Rotary Volume Simplified Signal Flow**

**Figure 23. Rotary Encoder Volume Control Signal Flow**
PUSH-BUTTON-CONTROLLED DEMULTIPLEXER

This example uses a set of two push-buttons to control a demultiplexer, which allows a 1 kHz sine wave to be output on any of eight possible outputs.

The table entries in this case are slightly different because of the number formats used in the DSP. There are two main formats used by SigmaDSP: integer format (28.0) and decimal format (5.23). All audio data and most parameters are represented in 5.23 decimal format within the DSP. Most GPIO-related signals are represented in 28.0 integer format because it increases the range of allowable values they can take.

In SigmaStudio index tables, the index value is represented in 28.0 integer format, whereas the output is in 5.23 decimal format. However, because this 5.23 output signal is used to control the demultiplexer (which accepts a 28.0 input format), number formatting is required within the index table.

To convert between 28.0 integer format and 5.23 decimal format, use Equation 3 and Equation 4. Equation 3 converts 28.0 to 5.23 while Equation 4 converts 5.23 to 28.0.

\[ n_{28.0} = n_{5.23} \times 2^{23} \]  

\[ n_{5.23} = n_{28.0} \times 2^{-23} \]  

The first eight indices, converted to 5.23 number format, are shown in Figure 24.

The cell names as they appear in the software and the number of each used in this example are as follows: GPIO Input (2), Interface Read (1), Up/Down Control with Lookup Table (1), Interface Write (1), Tone (lookup sine) (1), Index Selectable Demultiplexer (1), and Output (8).

![Figure 24. Push-Button-Controlled Demultiplexer Index Table](image)

![Figure 25. Push-Button-Controlled Demultiplexer Signal Flow](image)
PUSH-BUTTON-CONTROLLED FILTER
This example uses the push-buttons as described in the previous examples, although it now uses the index value to select from one of four equalizer curves.

The cell names as they appear in the software and the number of each used in this example are as follows: GPIO Input (2), Interface Read (1), Up/Down Control w/Lookup Table (1), Interface Write (1), Input (1), General (2nd Order/Lookup) (1), and Output (2).

Figure 26. Push-Button-Controlled Filter Index Table

Figure 27. Push-Button-Controlled Filter Signal Flow

Figure 28. Push-Button-Controlled Filter Tone Control Window
AUXILIARY ADC VOLUME CONTROL
Sometimes it is desirable to have analog control of one or more parameters in a system. The most common case is an analog master volume control.

In this example, a SW Slew Volume Control is inserted between the input and output of the audio signal flow. The auxiliary ADC input is connected directly to the control pin on the volume control.

The cell names as they appear in the software and the number of each used in this example are as follows: Auxiliary ADC Input (1), Input (1), Single Slew Ext Vol (1), and Output (1).

AUXILIARY ADC VOLTAGE-CONTROLLED OSCILLATOR
In this example, an analog input voltage is used to control the frequency of an oscillator.

As the input voltage increases, the frequency of the oscillator increases accordingly.

The cell names as they appear in the software and the number of each used in this example are as follows: Auxiliary ADC Input (1), Voltage Controlled Oscillator (1), T Connection (1), and Output (2).
AUXILIARY ADC SLEW MULTIPLEXER

An analog input voltage can also be used to select between sources (see Figure 31). In this example, an input voltage is broken down into five equally-sized zones, each associated with a different tone generator.

By multiplying the auxiliary ADC input signal by 5 in 28.0 format (implemented here as a DC input entry cell), the full scale of the input is mapped to five index values: 0, 1, 2, 3, and 4 in 28.0 format, which are appropriate input values for the multiplexer. The multiplexer cell uses this index to output the appropriate sine tone.

The cell names as they appear in the software and the number of each used in this example are as follows: ADC Input (1), DC Input Entry (1), Multiply (1), Tone (lookup/sine) (5), Index Selectable Slewing Mux (1), T Connection (1), and Output (2).

BLINKING LED

In the example shown in Figure 32, a square wave generator is used to drive an output LED. Output GPIO pins turn off when their associated register is equal to zero and turn on when equal to any other value. By taking a square wave with a minimum value of −1.23 and a maximum of 1.23, then adding 1.23, the result is a waveform with a minimum value of 0.23, and maximum value of 2.46. This causes the LED to blink at a frequency equal to that of the square wave generator (3 Hz in this example).

The cell names as they appear in the software and the number of each used in this example are as follows: DC Input Entry (1), Signal Add (1), Square Wave (1), and GPIO Output (1).

Figure 31. Auxiliary ADC Slew Multiplexer Signal Flow

Figure 32. Square Wave LED Driver Signal Flow
GPIO EXPANSION BOARD SCHEMATICS

Figure 33. GPIO Extension Board, Connectors, and Power
Figure 34. GPIO Extension Board, GPIO Circuitry
Figure 35. GPIO Extension Board, LED Driver (Not Populated)