Features and Overview

- Flexible clocking options
- Analog output
- Runs in background to allow system control while dialing (configuration option)
- Runs in foreground to minimize use of RAM (configuration option)
- Tones can be output continuously under program control
- Automatically adds configured tone spacing to all output tones
- Tone duration can be configured
- Output capable of -1.7 dBm un-amplified into 600 ohm load, and up to +3.1 dBm amplified drive
- Output driver capable of driving 32 ohm load at 5 Vrms

The DTMFDialer User Module is a Dual Tone Multiple Frequency signal generator. It provides a 6-bit, 2.6 volt full-scale analog output, centered around AGND. The output is a pair of simultaneously generated table sinusoids (tones) that are updated at a user-selectable update frequency. Selection of the update frequency causes a trade-off between CPU loading and signal distortion. Output tone generation is done in an interrupt routine to minimize sample skew and related distortion. Configuration options provide the ability to make design trade-off between RAM consumption and other operational features.
Functional Description

The DTMFDialer User Module is formed from one analog switched cap PSoC™ block and one digital PSoC block.

The analog block is configured as a 6-bit DAC. The digital block is used to scale the control clock and generate an interrupt at the update frequency. The update frequency corresponds to the frequency of the

\[ V_{\text{Ref}} = \text{AGND} + 1.3V \]
DTMF Clk interrupt. During the interrupt, the DAC is updated with the next value of the tone pair. The pair of tones generated for each of the tones is show in the following table.

**Tone DTMF Matrix**

<table>
<thead>
<tr>
<th>( f_r/f_c )</th>
<th>1209 Hz</th>
<th>1336 Hz</th>
<th>1477 Hz</th>
<th>1633 Hz</th>
</tr>
</thead>
<tbody>
<tr>
<td>697 Hz</td>
<td>1</td>
<td>2</td>
<td>3</td>
<td>A a</td>
</tr>
<tr>
<td>770 Hz</td>
<td>4</td>
<td>5</td>
<td>6</td>
<td>B b</td>
</tr>
<tr>
<td>852 Hz</td>
<td>7</td>
<td>8</td>
<td>9</td>
<td>C c</td>
</tr>
<tr>
<td>941 Hz</td>
<td>*</td>
<td>0</td>
<td>#</td>
<td>D d</td>
</tr>
</tbody>
</table>

The rows of the matrix shown in the table above represent the low frequencies, while the columns represent the high frequencies. For example, the digit 7 will be the combination of a low tone of 852 Hz and a high tone of 1209 Hz. The two frequencies are combined to make a continuous time DTMF signal using the following equation, where \( f_u \) is the update frequency.

\[
V_{out}(n) = 0.72V \sin\left(2\pi \frac{f_r}{f_u} n\right) + 0.54V \sin\left(2\pi \frac{f_c}{f_u} n\right)
\]

**Equation 1**

The ITU-T DTMF recommendation is that, when encoding DTMF signals, the frequency should be within 1.5% of that stated and, when decoding any signals out of tolerance by more than 3.5%, it should be ignored. When encoding, the duration of each tone should be at least 40 ms, with a signal interrupt to follow of at least 10 ms. The DTMF tone energy should exceed the levels of any other frequencies present by at least 30 dB. A summary of this is stated in the following table.

**ITU-T DTMF Recommendations**

<table>
<thead>
<tr>
<th>Parameter</th>
<th>Requirement</th>
</tr>
</thead>
<tbody>
<tr>
<td>Frequency Tolerance</td>
<td>&lt;= 1.5%</td>
</tr>
<tr>
<td>Signal Duration</td>
<td>&gt;= 40 ms</td>
</tr>
<tr>
<td>Signal Interrupt</td>
<td>&gt;= 10 ms</td>
</tr>
<tr>
<td>Signal Distortion</td>
<td>&lt;= 3%</td>
</tr>
</tbody>
</table>

A requirement that the frequency tolerance be less than 1.5% means that the control clock is one of the following:

- CPU_32_kHz using the external crystal.
- Another system clock with the PLL enabled.
- An external clock with the required accuracy brought in on a global in pin.

The input interface to the dialer is done with a string of ASCII characters. Each character defines an action. The possible actions are listed in the following table.

**Character Action Definitions**

<table>
<thead>
<tr>
<th>Character</th>
<th>Action</th>
</tr>
</thead>
<tbody>
<tr>
<td>0 1 2 3 4 5 6 7 8 9 * # A a B b C c D d</td>
<td>Output the appropriate tone pair for selected time period. Automatically adds inter-tone spacing.</td>
</tr>
<tr>
<td>, (comma)</td>
<td>900 ms tone space.</td>
</tr>
<tr>
<td>\0 (null)</td>
<td>End of phone number; dialing is finished.</td>
</tr>
<tr>
<td>All other characters</td>
<td>Ignore.</td>
</tr>
</tbody>
</table>
The default pause for a comma is 900 ms. Changing the constant iNoToneLength in `DTMF.Dialer.inc` alters the default value.

DAC data updates are interrupt driven to minimize signal distortion. The interrupt is based on the terminal count of the digital block associated with this user module. Two points to consider, since distortion is dependent on an ISR, are as follows.

- If another ISR is executing, it should not be serviced until the DTMF dialer operation has completed. If another ISR is to be serviced during DTMF dialer operation, the effect on signal distortion should be considered.
- Interrupt priority for digital blocks is based on their position. Lower numbered blocks have higher priority when more then one interrupt is pending.

## DC and AC Electrical Characteristics

### DTMFDialer DC Electrical Characteristics

<table>
<thead>
<tr>
<th>Parameter</th>
<th>Conditions and Notes</th>
<th>Minimum</th>
<th>Typical</th>
<th>Limit</th>
<th>Units</th>
</tr>
</thead>
<tbody>
<tr>
<td>Operating Voltage</td>
<td>See DC Characteristics in device family data sheet</td>
<td>3.0</td>
<td>5</td>
<td>5.5</td>
<td>Vdc</td>
</tr>
<tr>
<td>Output Voltage</td>
<td>Amplified using MDAC</td>
<td></td>
<td></td>
<td>Vdd − 0.5</td>
<td>Vdc</td>
</tr>
<tr>
<td></td>
<td>Un-amplified</td>
<td>(Vdd/2)−1.3</td>
<td></td>
<td>(Vdd/2)+1.3</td>
<td></td>
</tr>
<tr>
<td>Output Current</td>
<td>Requires use of analog buffer</td>
<td>40</td>
<td></td>
<td></td>
<td>mA</td>
</tr>
<tr>
<td>Output Reference</td>
<td>Configurable</td>
<td></td>
<td></td>
<td>Vdd/2</td>
<td>Vdc</td>
</tr>
</tbody>
</table>

### DTMFDialer AC Electrical Characteristics

<table>
<thead>
<tr>
<th>Parameter</th>
<th>Conditions and Notes</th>
<th>Minimum</th>
<th>Typical</th>
<th>Limit</th>
<th>Units</th>
</tr>
</thead>
<tbody>
<tr>
<td>DTMF Frequency Stability</td>
<td>Refer to AC Characteristics in device data sheet (External Oscillator)</td>
<td></td>
<td>1%</td>
<td></td>
<td>Hz</td>
</tr>
<tr>
<td>DTMF Noise Floor</td>
<td>0-20 kHz 31250 Hz DTMF Ck Frequency</td>
<td></td>
<td>-55</td>
<td></td>
<td>dBv</td>
</tr>
<tr>
<td></td>
<td>0 kHz-200 kHz (peak) 31250 Hz DTMF Ck Frequency</td>
<td></td>
<td>-30</td>
<td></td>
<td></td>
</tr>
<tr>
<td>DTMF Harmonic Distortion</td>
<td>31250 Hz DTMF Ck Frequency 24 MHz System Ck 0-20 kHz</td>
<td></td>
<td>&lt;1</td>
<td></td>
<td>%</td>
</tr>
<tr>
<td>High Group Pre-emphasis</td>
<td></td>
<td></td>
<td>2.5</td>
<td></td>
<td>dBv</td>
</tr>
</tbody>
</table>
Parameters and Resources
There are two system parameters and six input parameters:

- Analog Column Clock
- RefMux
- Control Clock
- DTMF Clk Period
- DTMF Clk Frequency
- Tone Duration (ms)
- Tone Spacing (ms)
- Background / Foreground Operation

Analog Column Clock

Note that this system parameter does not appear as a listed user module parameter. It is set by using the mouse to click on the appropriate column clock icons in the PSoC Device Editor display.

It is recommended that the clock for this particular column be set to 1 MHz.

The Analog Column Clock determines how fast the DAC signal is updated. It also determines how long the CPU stalls whenever a value is written to the DAC. Too slow, and the CPU stalls for a long time. Too fast, and the DAC will never settle.

RefMux

Note that this system parameter appears in the Global Resource fields of PSoC Designer.

It is recommended that the RefMux be set to \( V_{dd}/2 \pm V_{BandGap} \).

The selection of the RefMux setting controls the reference to the DAC used to generate the DTMF signals. The reference selection controls the peak-to-peak voltage of the output signal from the DAC. If the maximum voltage swing is selected \( (V_{dd}/2 + V_{dd}/2) \), and the DTMF dialer signal is output using the analog buffer, clipping may occur. This may be mitigated by running the DTMF dialer through a Multiplying DAC User Module or by selecting a lower peak-to-peak reference voltage.

Control Clock

The selection of the Control Clock sets the input frequency to the DTMF digital block.

DTMF Clk Period

The DTMF Clk Period sets the period (in clock cycles) of the DTMF interrupt. The DTMF Clk counts down from the initial (period) to terminal count (zero). At each terminal count, an interrupt is generated and the DAC output voltage is updated. This interrupt sets the DTMF Clk Frequency, which is determined using Equation 2.

\[
DTMFClkFrequency = \frac{ControlClock(Hz)}{DTMFClkPeriod + 1}
\]

Equation 2
For a Control Clock of 4 MHz and the DTMF Clk Period set to 124, the update frequency is calculated using Equation 3.

\[
DTMFClkFrequency = \frac{4MHz}{124 + 1} = 32kHz
\]

Equation 3

Given the control clock is 4 MHz and the DTMF Clk frequency is 32 kHz, Equation 4 calculates the DTMF Clk Period.

\[
DTMFClkPeriod = \frac{ControlClock}{DTMFClkFrequency} - 1 = \frac{4MHz}{32kHz} - 1 = 125 - 1 = 124
\]

Equation 4

**DTMF Clk Frequency**

The equation to calculate the DTMF Clk Frequency value is shown above. The value must be entered as a parameter. The DTMF Clk Frequency impacts the system in two ways:

- CPU loading
- Signal distortion

CPU loading and signal distortion are inversely related. Increasing the update frequency increases the CPU loading, but decreases the signal distortion. Lowering the update frequency decreases the CPU load, but increases the signal distortion.

**CPU Loading**

When operating the DTMF dialer in background mode, each update takes 244 CPU cycles and one CPU stall. The CPU loading is calculated using Equation 5.

\[
CPULoading = \left( \frac{244}{CPUClock} + \frac{4}{ColumnClock} \right) * DTMFClkFrequency
\]

Equation 5
The graph below shows the relationship between update frequency and CPU loading. CPU loading is shown for four different CPU clock frequencies. For all four plots, the column clock was set to 1 MHz.

**CPU Loading vs Update Frequency**

![Graph showing CPU loading vs update frequency for different CPU clock frequencies.](image)

For the Foreground operation mode, CPU loading is slightly higher and the following calculation applies.

\[
CPULoading = \left( \frac{269}{CPUClock} + \frac{4}{ColumnClock} \right) \times DTMFClkFrequency
\]

**Equation 6**

### Signal Distortion

Since the output is a sampled system, some distortion is introduced into the sampled signal by the reconstruction process. The number of samples per cycle is calculated using Equation 7.

\[
\text{#ofPoints} = \frac{UpdateFrequency}{ToneFrequency}
\]

**Equation 7**

The fewer number of samples per cycle, the greater the distortion. A reconstruction filter can compensate for a low number of samples per cycle. A reconstruction filter is a low-pass filter that allows the desired tones to pass while blocking the distorting harmonics. The graph below shows the relationship between distortion and update frequency.

Four different DTMF Ckl Frequency settings are shown, from 10 kHz up to 62.5 kHz. Blue traces (black if printed or seen in black and white) represent the unfiltered output of the DTMF dialer. The green trace (gray if printed or seen in black and white) shows the effect of adding an RC reconstruction filter with a single pole placed at 5.3 kHz.
Distortion components may be grouped by type, as harmonic distortion related to the dialing tones and non-harmonic noise related to the DAC reconstruction of the DTMF tones. Even at relatively low DTMF Ck Frequencies, the reconstructed waveform is significantly over-sampled, resulting in low harmonic distortion components. Addition of a simple RC reconstruction filter to the output significantly reduces reconstruction noise, as shown in the figure below. The blue trace (upper band if printed or seen in black and white) represents the raw output, while the red trace (lower band if printed or seen in black and white) represents a filtered output. In either case, all noise components are more than 30 dB below the signal levels while noise within the voice band is more than 50 dB below the DTMF signals.
DTMF Dialer Output

This selection allows the analog output to be directed to either the analog bus or to one of the analog output buffers supplied for each analog column. If the output of the DTMF dialer is used as the input to another switched capacitor block in the same column, it may be necessary to invert the phase of the output of the DTMF dialer for some configurations of the next block.

For example, if the DTMF dialer occupies block ASA10 and an MDAC6 occupies block ASB20, and the MDAC6 is to be used as an amplifier for the DTMF dialer, non-inverting gain settings on the MDAC6 will require that the output phase of the dialer is inverted. No provision is made for this configuration in the DTMFDialer User Module. All inverted gain settings of the MDAC will work for this configuration. A non-inverting configuration of this type is not necessary for proper operation of the DTMF dialer. This description is supplied to serve as an instruction example for other situations and illustrates the complex nature of clocked signals routed between switched capacitor blocks.

Tone Duration

This value defines the tone length in milliseconds. The minimum allowed value is 40 mSec while the maximum is 65535 mSec.

Tone Spacing

This value defines the inter-tone spacing in milliseconds. The minimum allowed value is 10 ms, while the maximum is 65535 ms.

Note  Internally, Tone Duration and Tone Spacing are implemented as software-timing loops. A 16-bit value is decremented each time the interrupt is entered. Given a fixed interrupt frequency, absolute time is calculated at compile time as a constant 32-bit value. Large values of DTMF Clk Frequency
and Tone Spacing or Tone Duration may result in overflow conditions when the 32-bit compiler internal calculation is applied to the 16-bit constant. The product \([\text{Tone Duration} \times \text{DTMF Clk Freq}]\) or \([\text{Tone Spacing} \times \text{DTMF Clk Freq}]\) must be less than \(65535 \times 1000\) or the maximum value of the variable used to calculate the duration of the tone (space) will be exceeded.

**Background/Foreground Operation**

Two selections are available for creating APIs: BACKGROUND (default) and FOREGROUND.

If Background is selected, all DTMFDialer processing and computation are attached to the ISR (Interrupt Service Routine) triggered by the DTMF Clk. This means that as soon as a phone number is passed to the Background algorithm, the function returns and other processing can take place while the phone number is being dialed. Background mode supports the DialDigit() function, which will begin generating the requested tone until another DialDigit() function call either terminates it or supplies a new tone to dial. An example of this would be generating a tone while a telephone key is pressed, stopping upon release of the key. A phone number can be submitted to the background process for dialing, while the foreground process repeatedly calls the bReadStatus() function to determine when the number has been completely dialed. Attaching all functions to the DTMF Clk ISR requires the use of 16 bytes of globally-defined RAM.

If Foreground mode is selected, most DTMF dialer functions are conducted outside of the DTMF Clk ISR. Only the actual tone generation is done within the ISR. The DialFromRam/Rom() function for this instance will return only after the phone number has been completely dialed. When the Foreground mode is selected, the DialDigit() function is disabled. The main advantage of using the Foreground option is that only two bytes of globally-defined RAM are used. This offers advantages if maximizing RAM memory space is critical or if dynamic re-configuration is used, since global variables assigned within user modules may be difficult to re-use, even when the module in question is unloaded.

**Interrupt Generation Control**

The following parameter is only accessible when the Enable Interrupt Generation Control check box in PSoC Designer is checked. This is available under Project >> Settings... >> Device Editor.

**IntDispatchMode**

The IntDispatchMode parameter is used to specify how an interrupt request is handled for interrupts shared by multiple user modules existing in the same block but in different overlays. Selecting “ActiveStatus” causes firmware to test which overlay is active before servicing the shared interrupt request. This test occurs every time the shared interrupt is requested. This adds latency and also produces a nondeterministic procedure of servicing shared interrupt requests, but does not require any RAM. Selecting “OffsetPreCalc” causes firmware to calculate the source of a shared interrupt request only when an overlay is initially loaded. This calculation decreases interrupt latency and produces a deterministic procedure for servicing shared interrupt requests, but at the expense of a byte of RAM.

**Placement**

The DAC block can be placed in any of the switched capacitor PSoC blocks. If the DAC output is taken off-chip, it will require use of the column analog bus. Other user modules that require use of the column comparator cannot be placed in the same column.

The Prescaler block can be placed in any digital PSoC block. If other digital block-based interrupts are used, they may interfere with the tone generation interrupt. Lower numbered digital blocks have higher priority.
Application Programming Interface

The Application Programming Interface (API) routines are provided as part of the user module to allow the designer to deal with the module at a higher level. This section specifies the interface to each function together with related constants provided by the “include” files.

Note

In this, as in all user module APIs, the values of the A and X register may be altered by calling an API function. It is the responsibility of the calling function to preserve the values of A and X prior to the call if those values are required after the call. This “registers are volatile” policy was selected for efficiency reasons and has been in force since version 1.0 of PSoC Designer. The C compiler automatically takes care of this requirement. Assembly language programmers must ensure their code observes the policy, too. Though some user module API function may leave A and X unchanged, there is no guarantee they will do so in the future.

Entry points are to initialize the Dialer, start Dialer sampling, and stop the Dialer.

**DTMF_Dialer_Start**

Description:

Sets the power level for the switched capacitor PSoC block.

C Prototype:

```c
void DTMFDialer_Start (BYTE bPowerSetting)
```

Assembly:

```assembly
mov A, bPowerSetting
call DTMFDialer_Start
```

Parameters:

bPowerSetting: One byte that specifies the power level. Following reset and configuration, the analog PSoC block assigned to DTMFDialer is powered down. Symbolic names provided in C and assembly, and their associated values, are given in the following table.

<table>
<thead>
<tr>
<th>Symbolic Name</th>
<th>Value</th>
</tr>
</thead>
<tbody>
<tr>
<td>DTMFDialer_OFF</td>
<td>0</td>
</tr>
<tr>
<td>DTMFDialer_LOWPOWER</td>
<td>1</td>
</tr>
<tr>
<td>DTMFDialer_MEDPOWER</td>
<td>2</td>
</tr>
<tr>
<td>DTMFDialer_HIGHPOWER</td>
<td>3</td>
</tr>
</tbody>
</table>

Power level will have an effect on analog performance. The correct power setting is sensitive to the sample rate of the data clock and has to be determined for each application. It is recommended that you start your development with high power selected. Testing can later be done to determine how low you can set the power setting.

Return Value:

None

Side Effects:

The A and X registers may be altered by this function.

**DTMF_Dialer_SetPower**

Description:

Sets the power level for the switched capacitor PSoC block.

C Prototype:

```c
void DTMFDialer_SetPower (BYTE bPowerSetting)
```
Assembly:
    mov    A, bPowerSetting
    call   DTMFDialer_SetPower

Parameters:
    bPowerSetting: Same as the bPowerSetting parameter used for the "Start" entry point. Allows the user to change the power level while operating the DTMFDialer.

Return Value:
    None

Side Effects:
    The A and X registers may be altered by this function.

**DTMFDialer_StartSamples**

Description:
    Enables the Prescaler counter, so that the DAC output initializes. The global interrupt must be enabled for the DTMFDialer to work.

C Prototype:
    void DTMFDialer_StartSamples (void)

Assembly:
    call   DTMFDialer_StartSamples

Parameters:
    None

Return Value:
    None

Side Effects:
    The A and X registers may be altered by this function.

**DTMFDialer_StopSamples**

Description:
    Disables the DTMF Clk counter and its interrupt. DAC output is no longer updated.

C Prototype:
    void DTMFDialer_StopSamples (void)

Assembly:
    call   DTMFDialer_StopSamples

Parameters:
    None

Return Value:
    None

Side Effects:
    The A and X registers may be altered by this function.

**DTMFDialer_Stop**

Description:
    Sets the power level on the switched capacitor PSoC block to OFF. Call this function to save power and when the DTMFDialer is not being used.
DTMF Dialer Analog Output

C Prototype:

```c
void DTMFDialer_Stop (void)
```

Assembly:

```assembly
call DTMFDialer_Stop
```

Parameters:

None

Return Value:

None

Side Effects:

The A and X registers may be altered by this function.

**DTMFDialer_DialFromRom**

Description:

Dials the phone number stored in program memory.
- Background mode: function returns immediately, while number is dialed under interrupt control.
- Foreground mode: function does not return until number has been dialed.

C Prototype:

```c
void DTMFDialer_DialFromRom (const char * sRomString)
```

Assembly:

```assembly
mov A,>sRomString
mov X,<sRomString
call DTMFDialer_DialFromRom
```

Parameters:

- Pointer to the char string in program memory. The string is assumed to be null terminated

Return Value:

None

Side Effects:

The A and X registers may be altered by this function.

**DTMFDialer_DialFromRam**

Description:

Dials phone number stored in data memory.
- Background mode: function returns immediately while number is dialed under interrupt control.
- Foreground mode: function does not return until number has been dialed.

C Prototype:

```c
void DTMFDialer_DialFromRam (char * sRamString)
```

Assembly:

```assembly
mov X,sRamString
call DTMFDialer_DialFromRam
```

Parameters:

- Pointer to the char string stored in data memory. The string is assumed to be null terminated.

Return Value:

None

Side Effects:

The A and X registers may be altered by this function.
**DTMF Dialer Analog Output**

**DTMF Dialer**

Description:

Dials a single character continuously until terminated explicitly. Available ONLY when the DTMFDialer is configured to operate in Background mode. An inter-tone delay is automatically added to the end of any dialed tone, so that successive tones will be separated. Tone are terminated by passing a null (0x00, ‘0’) to the function.

C Prototype:

```c
void DTMFDialer_DialDigit ( BYTE bDigit)
```

Assembly:

```assembly
mov A,bDigit
call DTMFDialer_DialDigit
```

Parameters:

Single character to dial.

Return Value:

None

Side Effects:

The A and X registers may be altered by this function.

**DTMF Dialer bReadStatus**

Description:

Returns the current status of the DTMFDialer. In Foreground mode, this routine is functional but meaningless, since the dialer routines return only when the called process is complete. In Background mode, the following bits may be set in the return value.

<table>
<thead>
<tr>
<th>Status Code</th>
<th>Value</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>DTMFDialer_READFRROM</td>
<td>0x1</td>
<td>Phone number is stored in ROM/Flash</td>
</tr>
<tr>
<td>DTMFDialer_DIALING</td>
<td>0x2</td>
<td>Phone (tone or number) dialing is in progress</td>
</tr>
<tr>
<td>DTMFDialer_CONTINUOUS_TONE</td>
<td>0x4</td>
<td>Tones have infinite duration</td>
</tr>
</tbody>
</table>

C Prototype:

```c
BYTE DTMFDialer_bReadStatus (void)
```

Assembly:

```assembly
call DTMFDialer_bReadStatus
mov [bStatus],A
```

Parameters:

None

Return Value:

Current status of the DTMFDialer.

Side Effects:

The A and X registers may be altered by this function.
Sample Firmware Source Code
The sample code dials the phone number stored in program memory.

; Example Assembly Code for DTMFDialer

export _main
include "DTMFDialer.inc"
include "m8c.inc"

_main:
mov A, DTMFDialer_HIGHPOWER
    call DTMFDialer_Start
    call DTMFDialer_StartSamples
M8C_EnableGInt ; enable interrupts
mov A, >pcPhoneNumber
mov X, <pcPhoneNumber
    call DTMFDialer_DialFromRom
loop:
    jmp loop ; The rest of the program

.LITERAL
pcPhoneNumber:
    ds "9,5551212"
    db 00h ; needed as a string delimiter
.ENDLITERAL

The same code is written in C as follows.

// Example C Code for DTMFDialer User Module

#include "DTMFDialer.h"
#include "m8c.h"
void main()
{
    const char * pcPhoneNumber = "9,5551212"
    M8C_EnableGInt;
    DTMFDialer_Start(DTMFDialer_HIGHPOWER);
    DTMFDialer_StartSamples();
    DTMFDialer_DialFromRom(pcPhoneNumber);
    while(1);   //rest of program
}
Configuration Registers

The DAC is a switched capacitor PSoC block. It is configured to be a 6-bit analog DAC.

Block DAC: Register CR0

<table>
<thead>
<tr>
<th>Bit</th>
<th>7</th>
<th>6</th>
<th>5</th>
<th>4</th>
<th>3</th>
<th>2</th>
<th>1</th>
<th>0</th>
</tr>
</thead>
<tbody>
<tr>
<td>Value</td>
<td>1</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>ACap</td>
</tr>
</tbody>
</table>

ACap is used by the DTMF_Clk interrupt to set the output value of the DAC.

Block DAC: Register CR1

<table>
<thead>
<tr>
<th>Bit</th>
<th>7</th>
<th>6</th>
<th>5</th>
<th>4</th>
<th>3</th>
<th>2</th>
<th>1</th>
<th>0</th>
</tr>
</thead>
<tbody>
<tr>
<td>Value</td>
<td>1</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>0</td>
</tr>
</tbody>
</table>

Block DAC: Register CR2

<table>
<thead>
<tr>
<th>Bit</th>
<th>7</th>
<th>6</th>
<th>5</th>
<th>4</th>
<th>3</th>
<th>2</th>
<th>1</th>
<th>0</th>
</tr>
</thead>
<tbody>
<tr>
<td>Value</td>
<td>ABus</td>
<td>0</td>
<td>1</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>0</td>
</tr>
</tbody>
</table>

ABus is a parameter set by the user to bring the DAC output off-chip.

Block DAC: Register CR3

<table>
<thead>
<tr>
<th>Bit</th>
<th>7</th>
<th>6</th>
<th>5</th>
<th>4</th>
<th>3</th>
<th>2</th>
<th>1</th>
<th>0</th>
</tr>
</thead>
<tbody>
<tr>
<td>Value</td>
<td>0</td>
<td>0</td>
<td>1</td>
<td>1</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>0</td>
</tr>
</tbody>
</table>

The DTMF_Clk is a digital PSoC block configured to be a counter with a selected period to divide the Control Clock to the desired DTMF Clk Frequency.

Block DTMF_Clk: Register Function

<table>
<thead>
<tr>
<th>Bit</th>
<th>7</th>
<th>6</th>
<th>5</th>
<th>4</th>
<th>3</th>
<th>2</th>
<th>1</th>
<th>0</th>
</tr>
</thead>
<tbody>
<tr>
<td>Value</td>
<td>0</td>
<td>0</td>
<td>1</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>1</td>
</tr>
</tbody>
</table>

Block DTMF_Clk: Register Input

<table>
<thead>
<tr>
<th>Bit</th>
<th>7</th>
<th>6</th>
<th>5</th>
<th>4</th>
<th>3</th>
<th>2</th>
<th>1</th>
<th>0</th>
</tr>
</thead>
<tbody>
<tr>
<td>Value</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>1</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>Clock</td>
</tr>
</tbody>
</table>

Clock selects clock input from one of 16 sources. This parameter is set in the Device Editor.
Out Enable is a parameter set by the user to enable the global output. Out Sel is a parameter set by the user to select which global output is used.

Prescaler Value is a parameter set by the user used to generate the desired DTMF Clk Frequency.

When Enable is set, DTMF_Clk is enabled. It is modified and controlled by the DTMFDialer API.