



## Dual-Tone Multi-Frequency (DTMF) Decoder

In this project, the concept of *dual-tone multi-frequency (DTMF)* signaling will be explored. As the name implies, DTMF signals are mixtures of two sinusoids at distinct frequencies. They are used in communications over analog telephone lines.

A particular version of DTMF signaling is utilized in dialing a number with push-button telephone handsets, a scheme known as touch-tone dialing. When the caller dials a number, the DTMF generator produces a dual-tone signal for each digit dialed.

The synthesized signal is in the form

$$x(t) = \sin(2\pi f_1 t) + \sin(2\pi f_2 t), \quad 0 \leq t \leq T_{digit}$$

Frequency assignments for the digits on a telephone keypad are shown in [Fig. 1](#).

$f_2$	1209 Hz	1336 Hz	1477 Hz
697 Hz	1	2	3
770 Hz	4	5	6
852 Hz	7	8	9
941 Hz	*	0	#

**Figure 1**

The goal of this project is to develop a DTMF synthesizer function using MATLAB.

**a.** For exploratory data analysis, **write** a MATLAB script to accomplish the following:

**Express** the signal  $x(t) = \sin(2\pi f_1 t) + \sin(2\pi f_2 t)$  through the use of a MATLAB anonymous function with  $f_1 = 852$  Hz and  $f_2 = 1477$  Hz.

**Compute** the signal in the time interval  $0 \leq t \leq 0.2$  with a time increment of  $\Delta t = 1/8000$  s.

**Play back** the resulting sound  $x(t)$  using the `sound` function of MATLAB.

If the signal was computed properly, you should hear a clean short tone.

**Graph** the resulting signal  $x(t)$  using the `plot` function.

**Repeat** the same experiment with different frequencies and different time increments based on [Fig. 1](#). You should hear different short tones.



- b. Develop** a function named `ss_dtmf1` to produce the signal for one digit.

The syntax of the function should be

$$x = \text{ss\_dtmf1}(n, t)$$

The first argument “*n*” is the digit for which the DTMF signal is to be generated.

Let values  $n = 0$  through  $n = 9$  represent the corresponding keys on the keypad.

Map the remaining two keys “\*” and “#” to values  $n = 10$  and  $n = 11$  respectively.

Finally, the value  $n = 12$  should represent a pause, that is, a **silent period** ( $f_1 = 0, f_2 = 0$ ).

The vector “*t*” contains the time instants at which the DTMF signal  $x(t)$  is evaluated and returned in vector “*x*”.

- c. Develop** a function named `ss_dtmf` with the syntax

$$x = \text{ss\_dtmf}(\text{number}, \text{delta}, n_d, n_p)$$

The arguments for the function `ss_dtmf` are defined as follows:

**number**: The phone number to be dialed, entered as a vector.

For example, to dial the number 5551212, the vector “*number*” would be entered as

$$\text{number} = [5, 5, 5, 1, 2, 1, 2]$$

**delta**: The time increment  $\Delta t$  to be used in computing the amplitudes of the DTMF signal.

***n<sub>d</sub>***: Parameter to control the duration of the DTMF signal for each digit.

The duration of each digit should be

$$T_{\text{digit}} = n_d \Delta t$$

***n<sub>p</sub>***: Parameter to control the duration of pause between consecutive digits.

The duration of pause should be

$$T_{\text{pause}} = n_p \Delta t$$

The function `ss_dtmf` should use the function `ss_dtmf1` to produce the signals for each digit (and the pauses between digits) and append them together to create the signal  $x(t)$ .

- d. Write** a script to test the function `ss_dtmf` with the number 5551212.

Use a time increment of 125 microseconds ( $\text{delta} = 1/8000$ ) corresponding to 8000 values per second. The duration of each digit should be 200 milliseconds ( $n_d = 1600$ ) with 80 millisecond pauses between digits ( $n_p = 640$ ).

**Play back** the resulting signal  $x(t)$  using the `sound` function.

Also, **graph** the resulting signal  $x(t)$  using the `plot` function.



- e. Write a MATLAB script to accomplish the following tasks:

**Express** the following three signals through the use of MATLAB anonymous functions with  $f_1 = 852$  Hz and  $f_2 = 1477$  Hz.

$$x_1(t) = \sin(2\pi f_1 t)$$

$$x_2(t) = \sin(2\pi f_2 t)$$

$$x(t) = \sin(2\pi f_1 t) + \sin(2\pi f_2 t)$$

**Compute** each signal for  $n = 1600$  samples with a sampling frequency of  $f_s = 8000$ .

You can compute the time vector ( $t$ ) using the following MATLAB code:

```
fs = 8000;
```

```
n = 1600;
```

```
t = (0:(n-1)) / fs;
```

**Listen** to each resulting signal **separately** using the sound function of MATLAB.

**Graph** the resulting signals  $x_1(t)$ ,  $x_2(t)$ ,  $x(t)$  in **one graph** using the `subplot` function.

The Fourier transform of the sine function is

$$\mathcal{F}\{\sin(2\pi f_0 t)\} = \frac{1}{2j} [\delta(f - f_0) - \delta(f + f_0)]$$

**Compute** the discrete Fourier transform (DFT) of each signal using the function `fft`, then **shift** the zero-frequency component to center of spectrum using the function `fftshift`.

**Graph** the **magnitude of the DFT** of each signal in **one graph** using the `subplot` function.

You can compute the frequency vector ( $f$ ) using the following MATLAB code:

```
f = (-n/2:n/2-1) * (fs/n);
```

## Report

Deliver a report, show the main function, and discuss your implementation.



## MATLAB Useful References

You may need the following MATLAB functions and references.

- plot
- subplot
- sound
- cat
- fft: [How to Do FFT in MATLAB](#)
- fftshift
- Plot FFT using MATLAB

## Example

[How to take FFT in Matlab | FFT Matlab Plot Frequency | FFT Matlab Easy Tutorial - YouTube](#)

## Bonus

**Develop** a GUI-based MATLAB program to model the operation of the DTMF decoder.

Your program should include an interactive telephone keypad are shown in [Fig. 1](#).

GOOD LUCK,  
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