Understanding Phone dials





The phone keypad

- The keypad is organized over four rows and three columns.
- Each row and each column is assigned a "frequency" (we will see later what this means).
- A number is then defined by its row and column.
- For example, the number 6 is defined by (770, 1477)



Dialing a number



The phone general A "sound" is a way The phone general

frequencies an **position**.

- The phone generates a digit as a sound.
- A "sound" is a wave, usually a sine function.
- The phone generates a digit as the **sum of the sine waves** whose **frequencies** are defined by its **row position** and **column**

Reminder on sine functions





The sine function is periodic (i.e. it repeats itself), with a period $T = 2\pi$



Reminder on sine functions



The sine function can be modulated to change its period.

To assign a period T to a sine function,

$$y(t) = \sin\left(\frac{2\pi}{T}t\right)$$

Example (on the right): T=1.5

$$y(t) = \sin\left(\frac{2\pi}{1.5}t\right)$$

Reminder on sine functions



When the x axis represents time, the period T is the duration of the function before it starts repeating itself.

The inverse of the period is the frequency of the signal, f:

$$f = \frac{1}{T}$$

It defines the number of time the function repeats itself over 1 second.

The frequency is expressed in Hertz (Hz)

The function can then be written in 2 ways:

$$y = \sin\left(\frac{2\pi}{T}t\right) = \sin\left(2\pi ft\right)$$



Dialing a number



The phone generates a digit as the sum of the sine waves whose frequencies are defined by its row position and column position.

For example, for the digit 6:

 $y(t) = \sin(2\pi770t) + \sin(2\pi1477t)$

Dialing a number: discrete signal

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Dialing a number: discrete signal

the digit 6:

 $y(t) = \sin(2\pi 770t) + \sin(2\pi 1477t)$

The phone transmits the sound signal "digitally": this means that it does not transmit the continuous signal, but a discrete signal in time, i.e. at only specific times. Those times are spaced uniformly, with a spacing Δ



Continuous signal

Discretization



Actual signal!

Dialing a number: discrete signal



 $y(k\Delta) = \sin(2\pi770k\Delta) + \sin(2\pi1477k\Delta)$

The phone industry has chosen the standard: **Fs** = **8192**

How to choose Δ *?*

 Δ is the **time interval between two sampled points**. It can be characterized by its inverse, Fs,

Fs is the sampling rate: it defines how many points are sampled per second.

Proper discretization requires that the sampling rate is at least twice the highest possible frequency in the signal.

For a phone keypad, the highest frequency is 1477. This means that $Fs > 2 \times 1477$



Mimicking dialing a digit on Matlab

Generating the sound for the digit 6 on a phone keypad with Matlab:

% Defining the sampling rate: Fs = 8192;% Computing the corresponding Delta in time: Delta = 1/Fs; % Defining the time for the signal: 0.2 seconds total, sampled by Delta: time= 0: Delta : 0.2 % Generating the two signals corresponding to the row and column of 6: y1 = sin (2 * pi * 770 * time); $y^2 = sin (2 * pi * 1477 * time);$ % Combining the two signals to generate the digit 6: y = y1 + y2;% Plot this signal: plot(time, y, '-r', 'LineWidth', 1.5) % We can even play the sound: sound(y)





> Given a time signal (over N points), can we find the corresponding digit that was dialed?





> Given a time signal (over N points), can we find the corresponding digit that was dialed?

We know more than the time signal!:

- As this is a signal generated by a phone,

$$\Delta = \frac{1}{Fs} = \frac{1}{8192}$$

The signal is a combination of two sines with different frequencies



What we want: A tool that:

What we want: A tool that:







What we want: A tool that:









Reverse Engineering:



Basics of Fourier transform

Any periodic function f(t), with period T can be written as a sum of sine and cosine function:

 $f(t) = a_0 + \sum_k a_k \cos\left(\frac{1}{k}\right)$ Defining the fundamental frequency $f_0 = \frac{1}{T}$

$$f(t) = a_0 + \sum_k a_k \cos\left(\frac{1}{k}\right)$$

$$\left(\frac{2\pi}{T}kt\right) + b_k \sin\left(\frac{2\pi}{T}kt\right)$$

$(2\pi k f_0 t) + b_k \sin\left(2\pi k f_0 t\right)$

Sum of cosines and sines with frequencies f_0 , $2f_0$, $3f_0$, ...,

Basics of Fourier transform

$$f(t) = a_0 + \sum_k a_k \cos\left(\frac{1}{k}\right)$$

The coefficients a_k and b_k can be computed:

$$a_k = \frac{1}{T} \int_{-T/2}^{T/2} f(t) \cos(2t)$$

$$b_k = \frac{1}{T} \int_{-T/2}^{T/2} f(t) \sin(2t)$$

 $(2\pi k f_0 t) + b_k \sin\left(2\pi k f_0 t\right)$

 $\pi k f_0 t dt$

 $(\pi k f_0 t) dt$

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 $\pi k f_0 t dt$

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This is exactly what the Fourier transform computes!

It finds how much the different frequencies kf_0 intervene in the signal!





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1) The "period" T of the signal is defined as its total length:

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3) Compute the coefficients a_k and b_k for frequency (kf_o) using Fourier transform.

If there are N values in the time signal, the Fourier transform will compute N values.

Matlab script to analyse a signal y

% Defining the phone company sampling rate: Fs = 8192;

Delta = 1/Fs;

% Length of time signal y(t) and corresponding "period' T: nval = length(y);

T = Delta*nval;

% Define the fundamental frequency identified by Fourier: $f_0 = 1/T$ f0 = 1/T;

% The Fourier is computed over a range of frequencies (0, f_0, 2^{f_0} ...) % over nval values; define this list:

freq= 0: f0 : Fs-1;

% Compute Fourier transform; get abs to combine a_k and b_k f = abs(fft(y);

% Plot

plot(freq, f, '-r', 'LineWidth', 1.5);

% Limit to frequency range of phone dials xlim([500 1500]);

Additional code to find peaks automatically

%% Find peak position %[peak_amp peak_loc]=findpeaks(f,'Minpeakheight',200); %% peak_loc is in point; convert to frequency %freq=f(peak_loc); %% Only keep values in DTMF range %peak_freq=freq(freq < 1700);</pre> peak_height = peak_amp(freq < 1700);</pre> %hold on str=num2str(int32(peak_freq)); text(peak_freq(1),peak_height(1),str(1)); text(peak_freq(2),peak_height(2),str(2));

