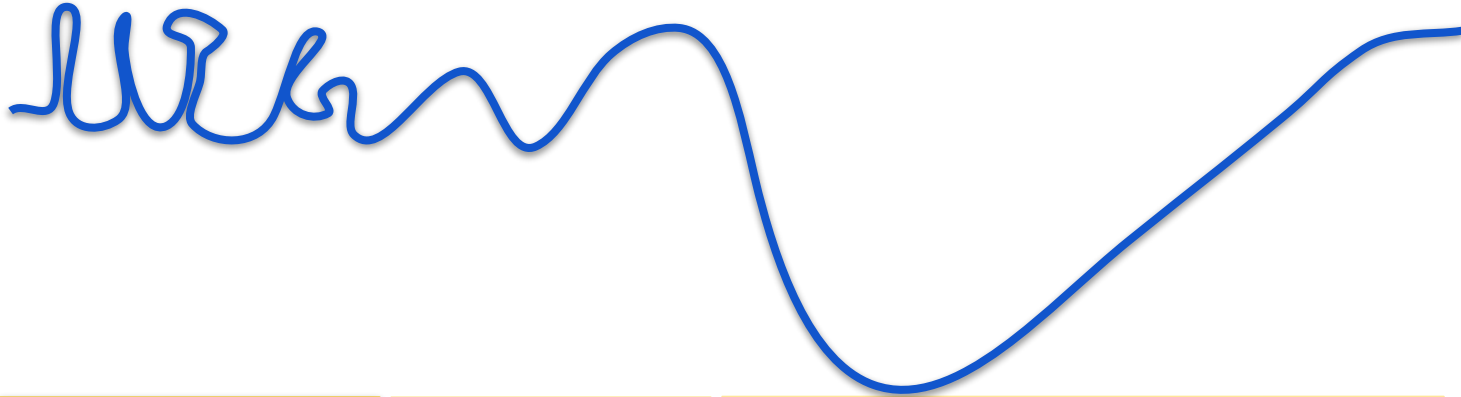


# Computing with Signals



**DA 623**

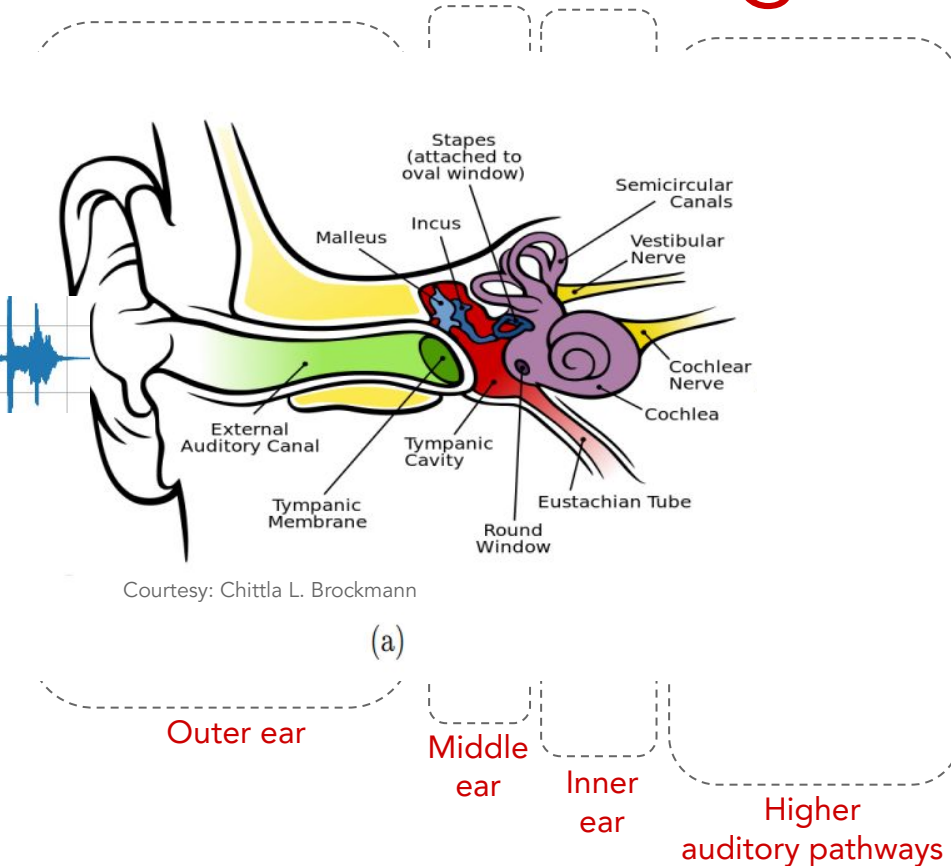
Jan - May 2024

IIT Guwahati

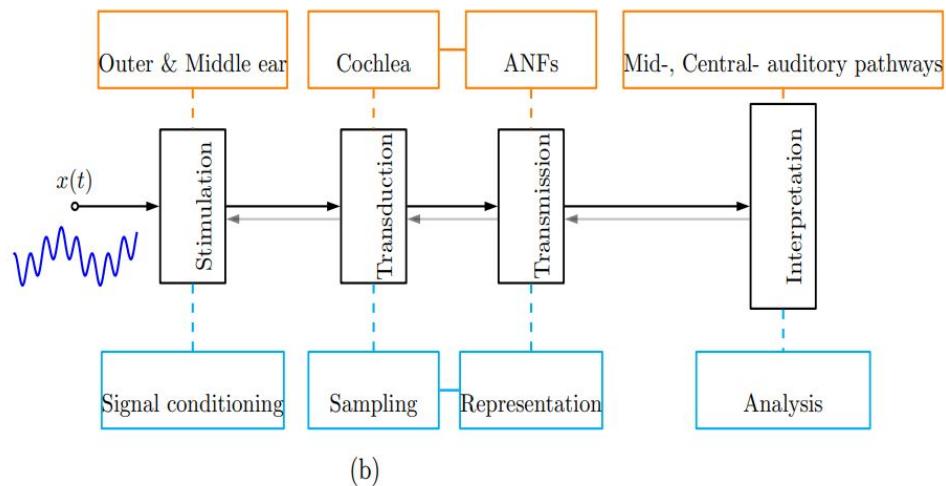
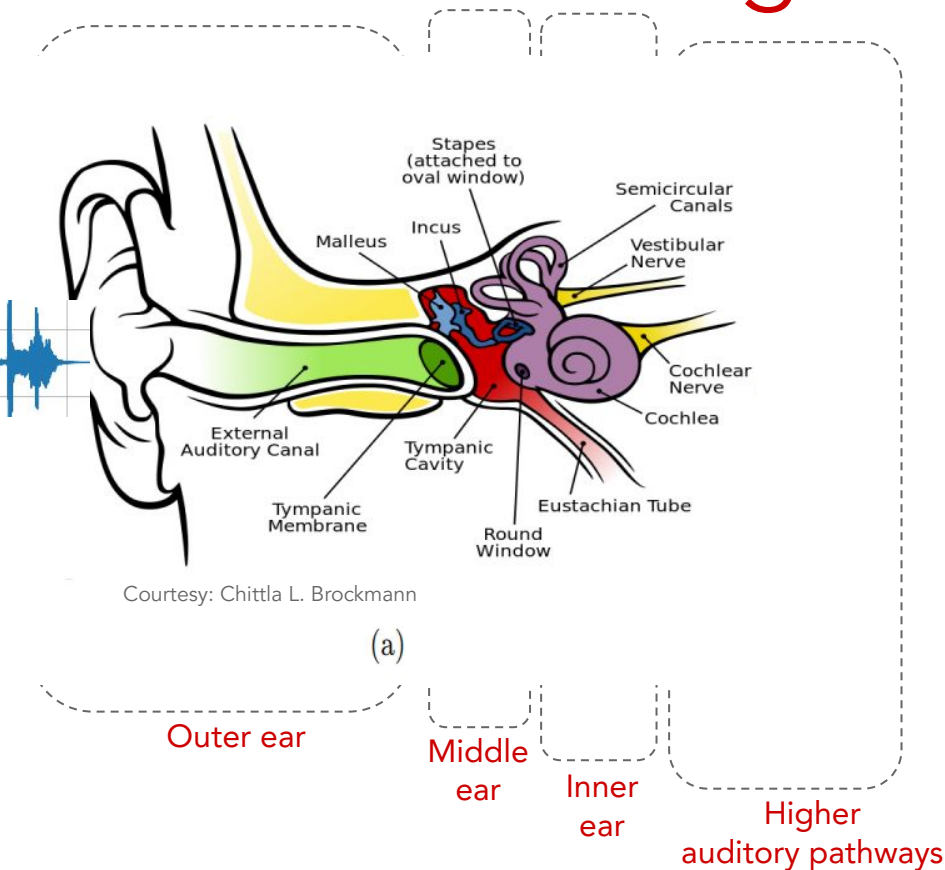
Instructors: Neeraj Sharma

Lecture-21

# Human hearing mechanism



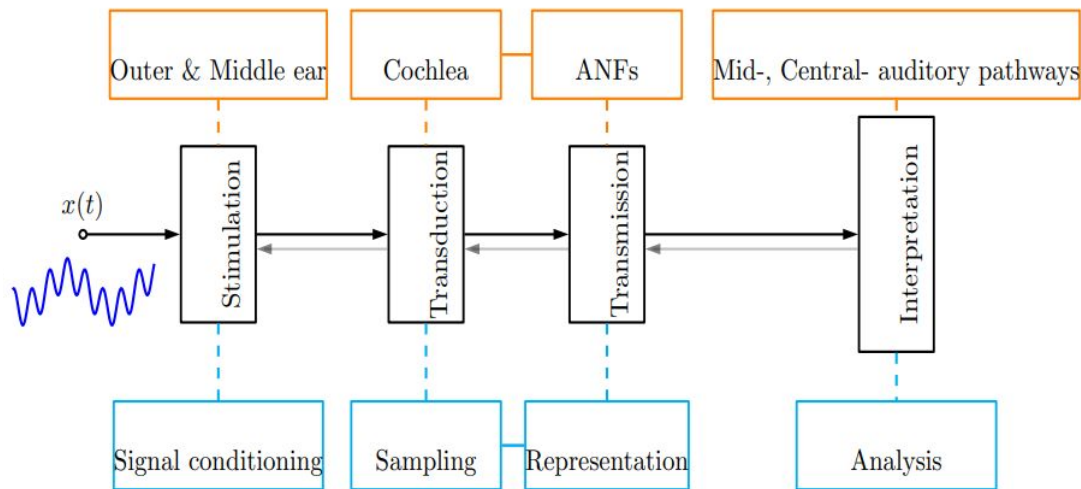
# Human hearing mechanism



## Data processing pipeline modeling hearing

Ref: N.K. Sharma, Information-rich sampling of time-varying signals, PhD Thesis, Indian Institute of Science, 2018

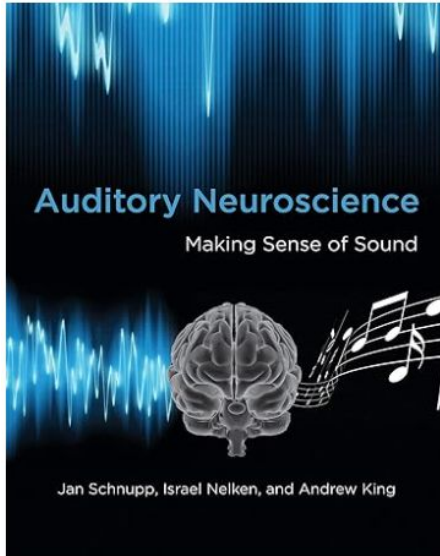
# Human hearing mechanism



(b)

Data processing pipeline modeling hearing

# Perceiving sound



Chap. 1

Why do things sound the way they do

[https://auditoryneuroscience.com/sites/default/files/Schnupp\\_FM\\_Ch1.pdf](https://auditoryneuroscience.com/sites/default/files/Schnupp_FM_Ch1.pdf)

# First commercial speech recognizer

- The first machine that recognized speech was a toy from the 1920s
- Radio Rex was a celluloid dog that moved when a spring was released by 500 Hz acoustic energy
- Since 500 Hz is roughly the first formant of the vowel [eh] in "Rex", Rex seemed to come when he was called (David, Jr. and Selfridge, 1962)



# Sound signal as a file

SoXI – Sound eXchange Information, display sound file metadata

```
$ soxi file_name.wav
```

```
Input File      : 'nMIOAh7qRFf3pqbchc1OLKbPD0m1_heavy_cough.wav'  
Channels        : 1  
Sample Rate     : 16000  
Precision       : 16-bit  
Duration        : 00:00:03.65 = 58378 samples ~ 273.647 CDDA sectors  
File Size       : 117k  
Bit Rate        : 256k  
Sample Encoding : 16-bit Signed Integer PCM
```

# Sound signal as a file

SoXI – Sound eXchange Information, display sound file metadata

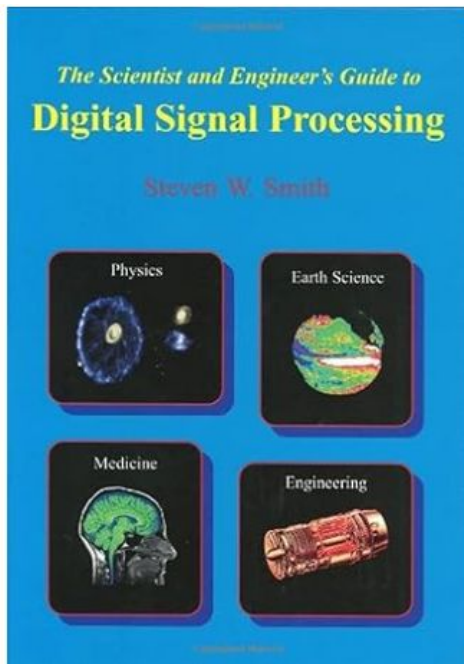
```
$ soxi file_name.wav
```

**note:** a sound signal captured and stored is referred as **Audio**

```
Input File      : 'nMIOAh7qRFf3pqbchc1OLKbPD0m1_heavy_cough.wav'  
Channels       : 1  
Sample Rate    : 16000  
Precision      : 16-bit  
Duration       : 00:00:03.65 = 58378 samples ~ 273.647 CDDA sectors  
File Size      : 117k  
Bit Rate       : 256k  
Sample Encoding: 16-bit Signed Integer PCM
```



# Audio Processing



## Chap. 22

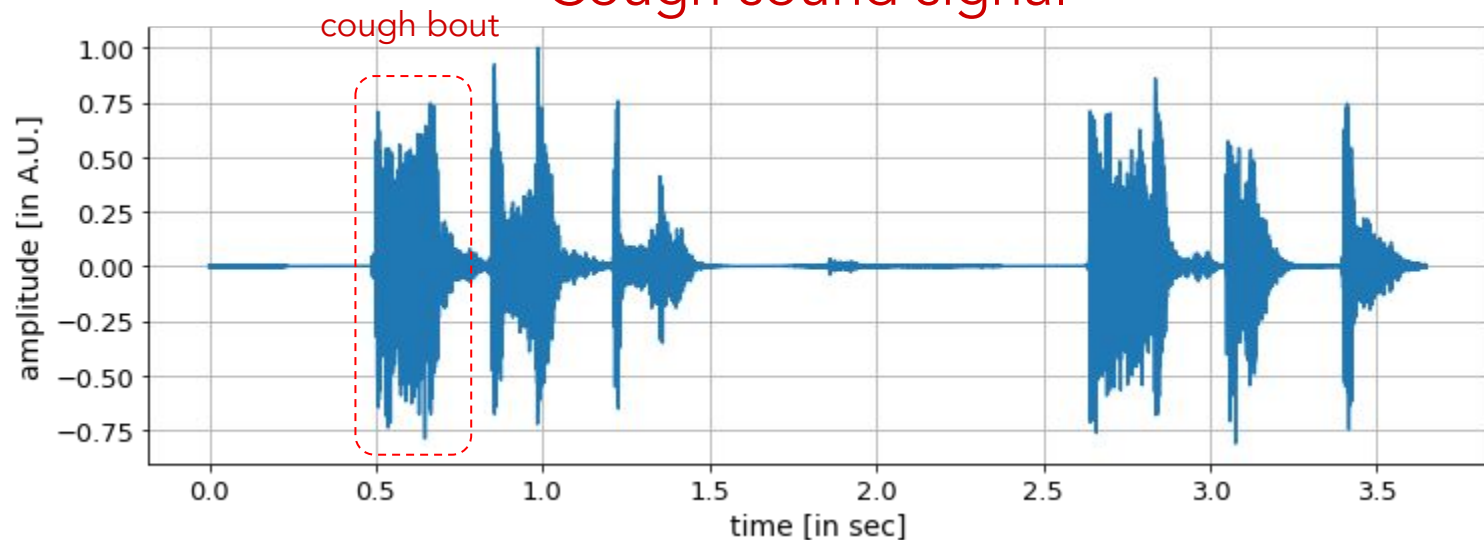
Audio Processing in The Scientist and Engineer's Guide to Digital Signal Processing By Steven W. Smith

<http://www.dspguide.com/CH22.PDF>

# Audio as a time-series

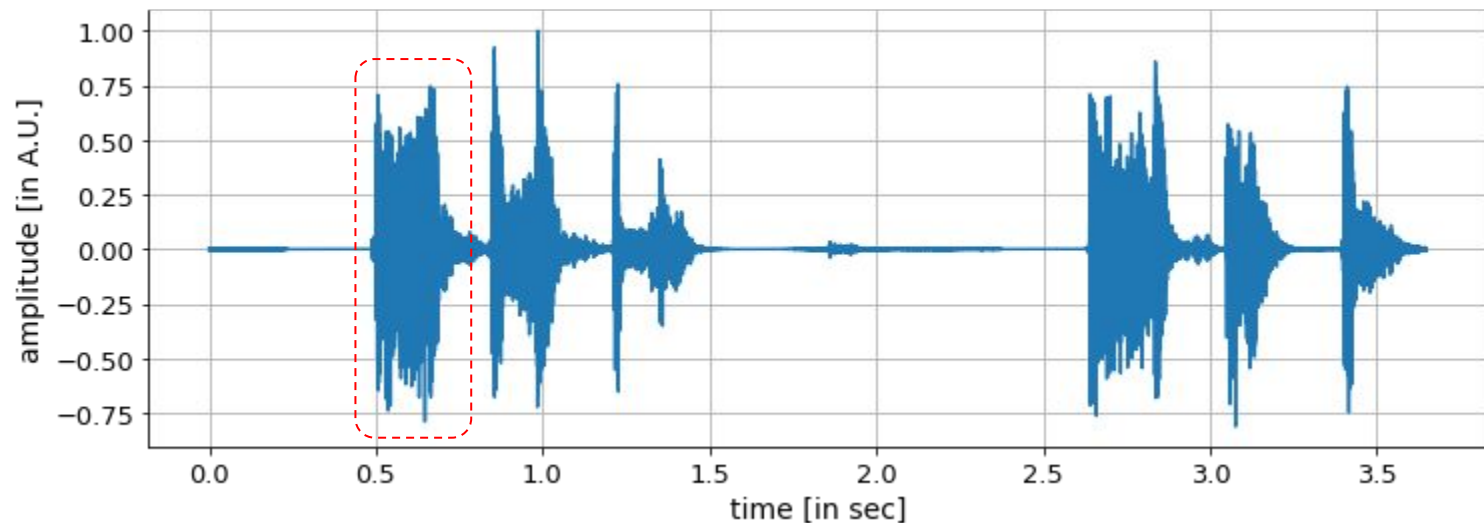
```
fs = 16000
fname = 'nMIOAh7qRFf3pqbchc1OLKbPD0m1_heavy_cough.wav'
dname = './my_data/'
# Load
x, sr = librosa.load(dname+fname, sr=16000)
x = x/max(np.abs(x))
times = np.arange(0, len(x))/fs
```

Cough sound signal

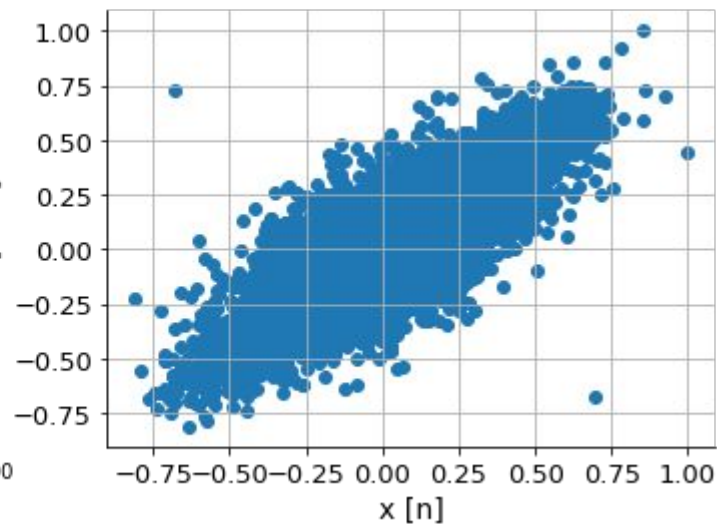
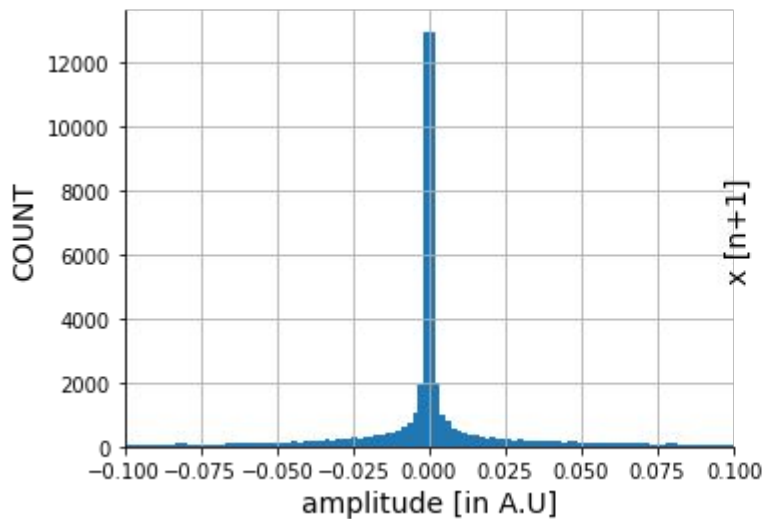
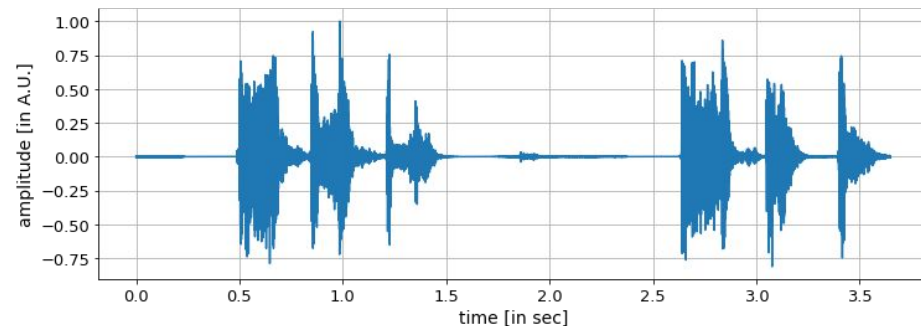


# Audio as a time-series

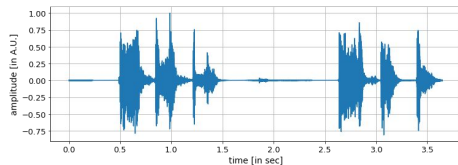
```
fs = 16000
fname = 'nMIOAh7qRFf3pqbchc1OLKbPD0m1_heavy_cough.wav'
dname = './my_data/'
# Load
x, sr = librosa.load(dname+fname, sr=16000)
x = x/max(np.abs(x))
times = np.arange(0, len(x))/fs
```



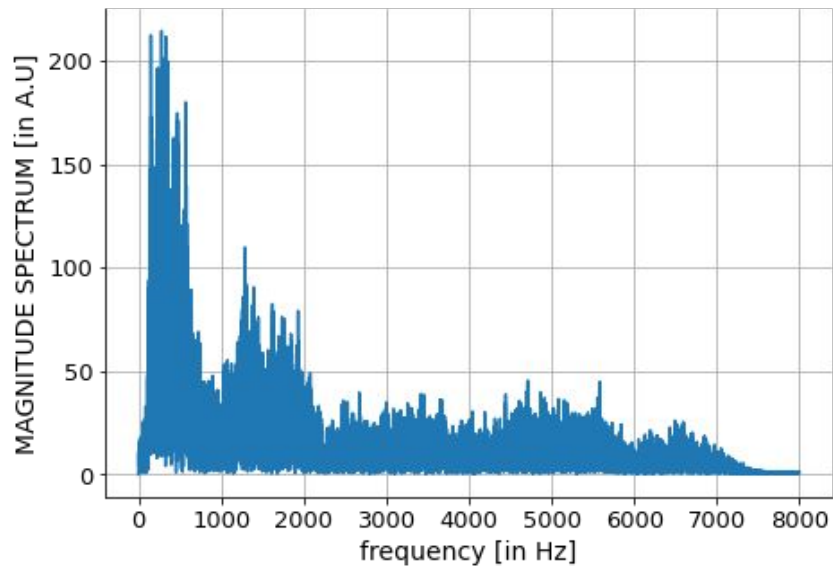
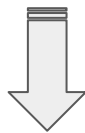
# Audio as a time-series



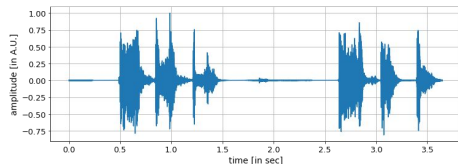
# Audio as a frequency spectrum



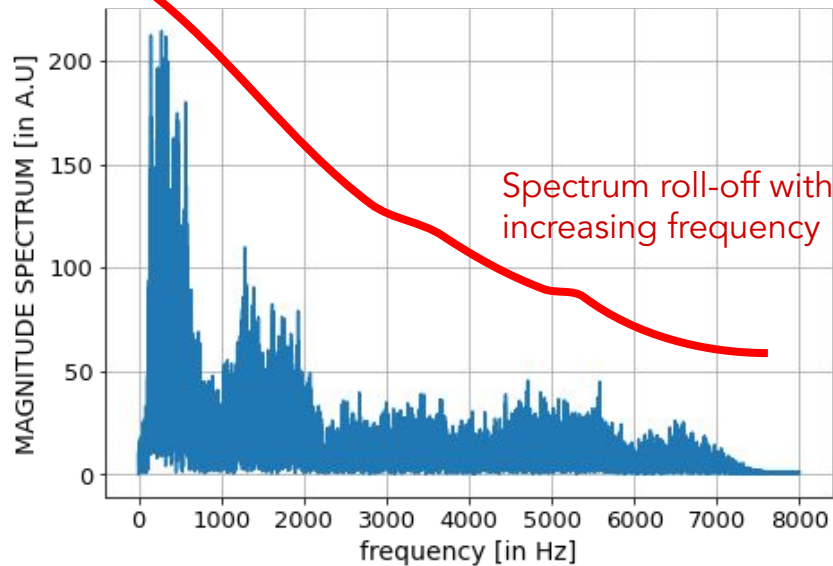
Magnitude spectrum  
obtained using DFT



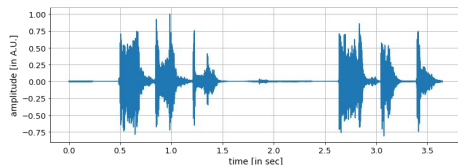
# Audio as a frequency spectrum



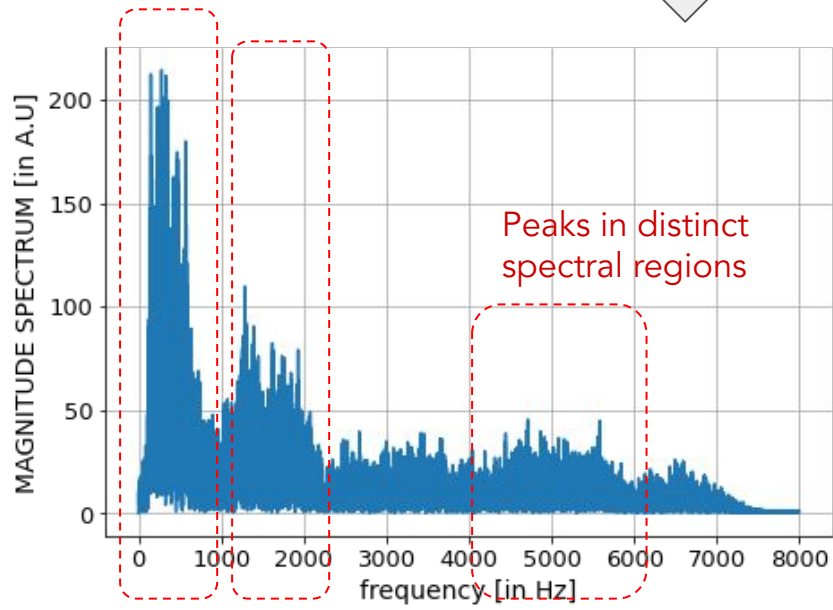
DFT



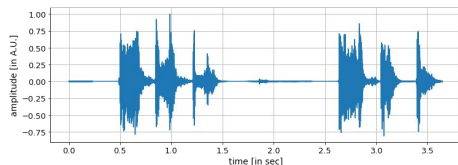
# Audio as a frequency spectrum



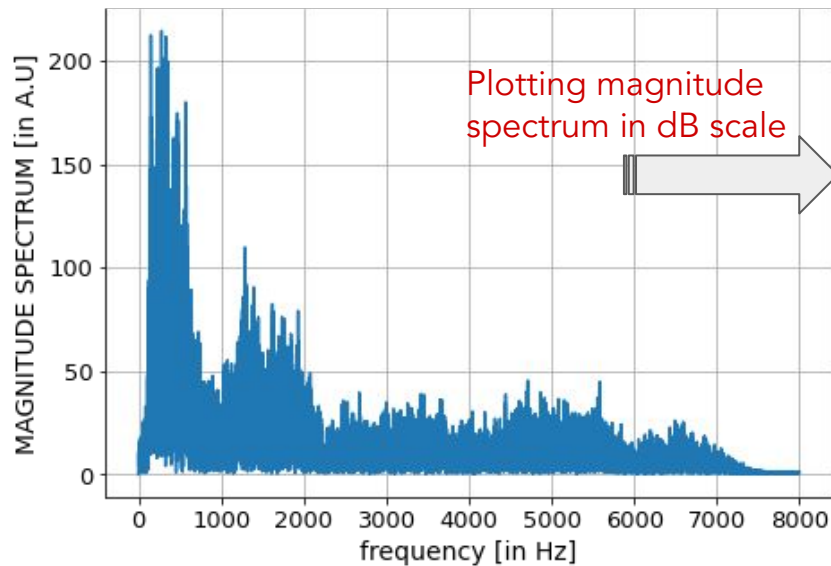
DFT



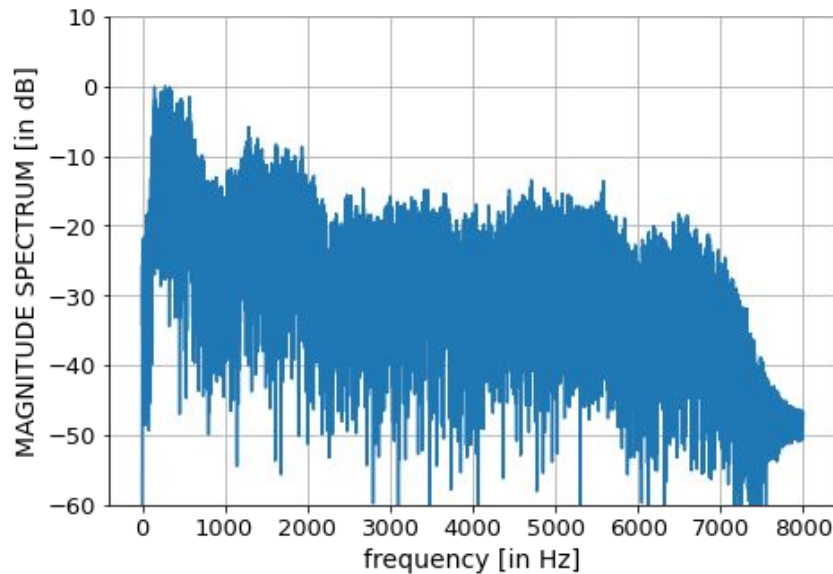
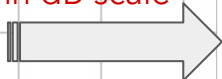
# Audio as a frequency spectrum



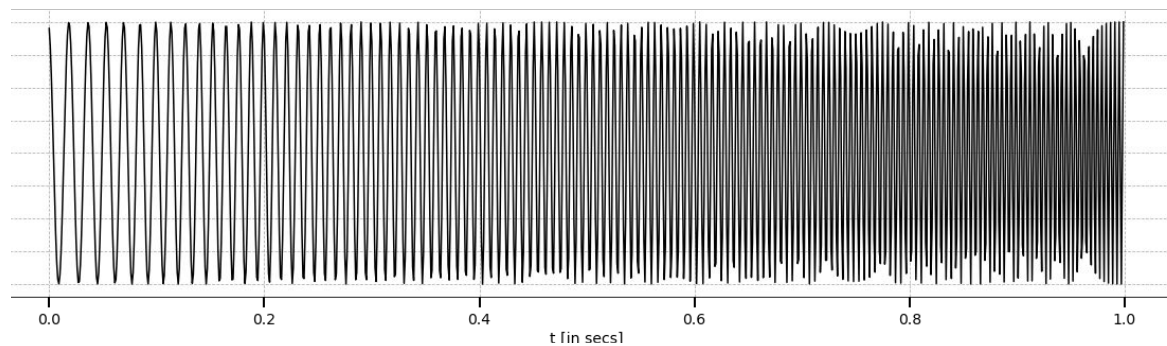
DFT



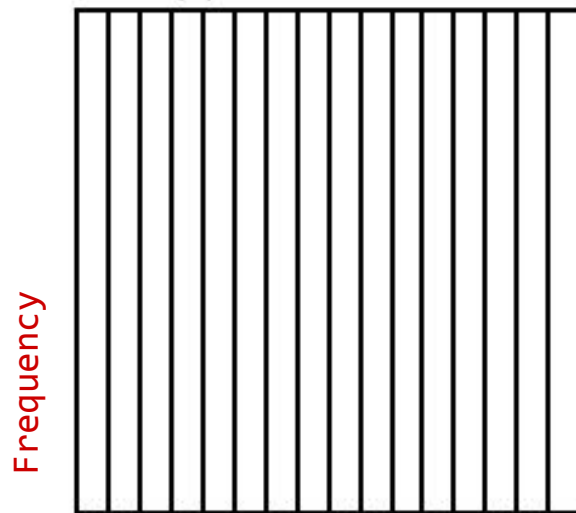
Plotting magnitude  
spectrum in dB scale



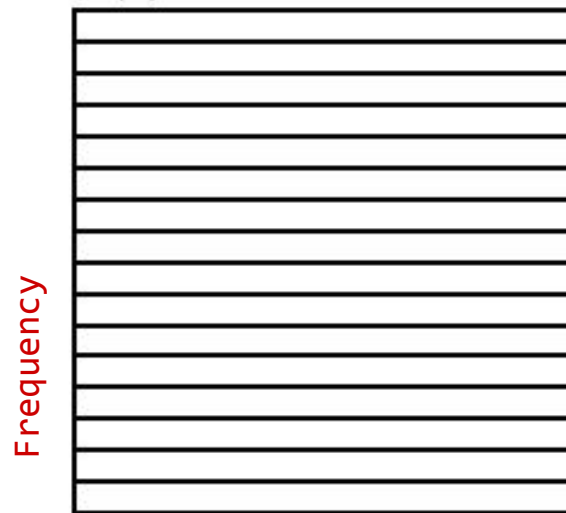


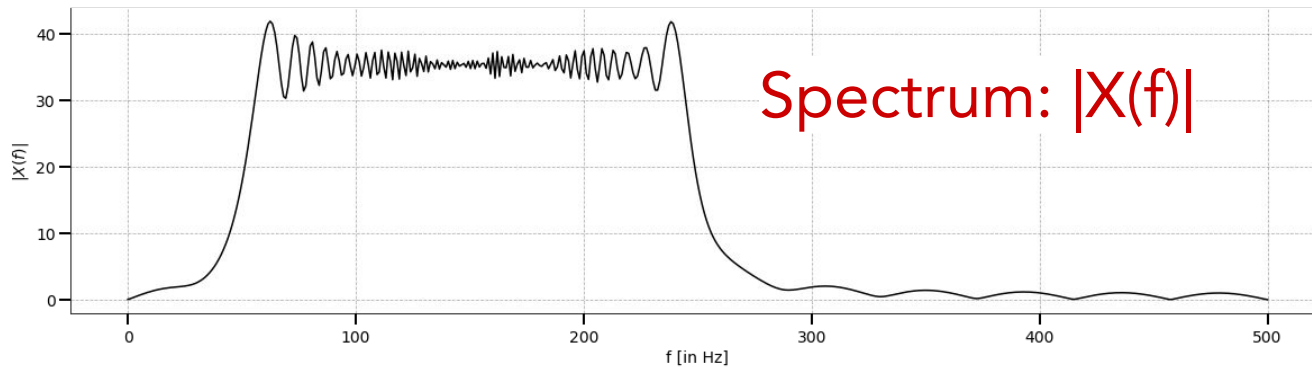
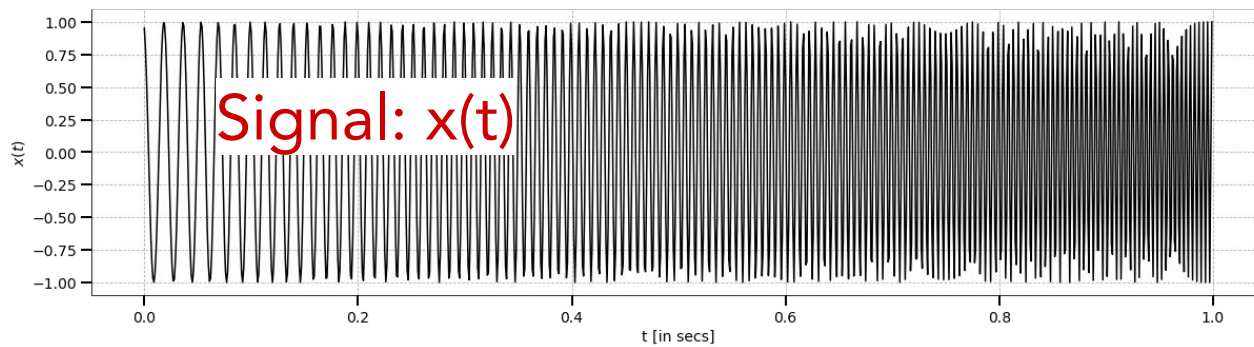


(a) Time series

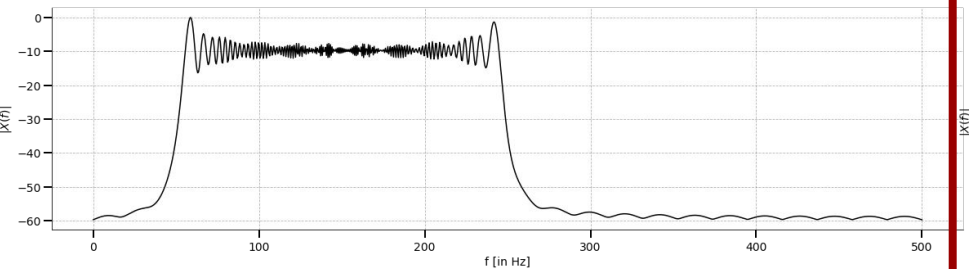
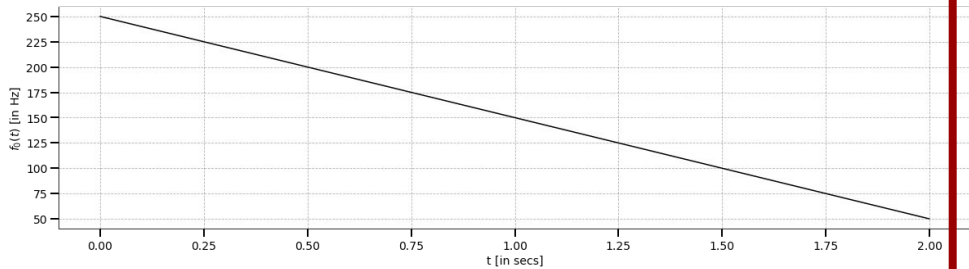
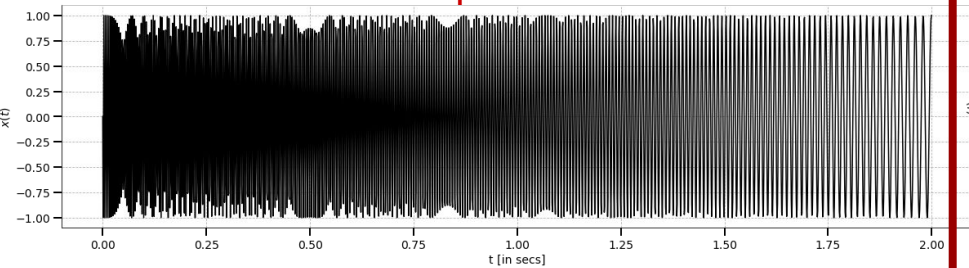


(b) Fourier transform

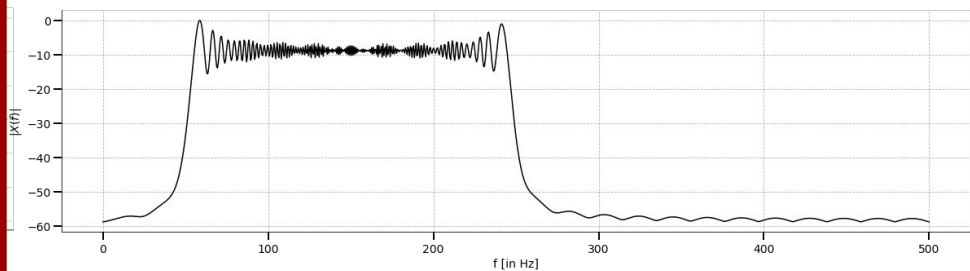
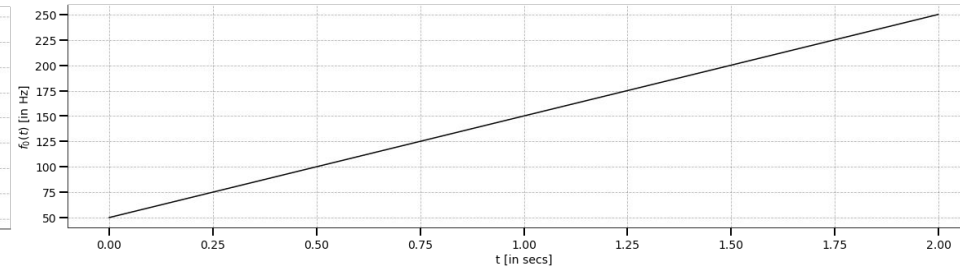
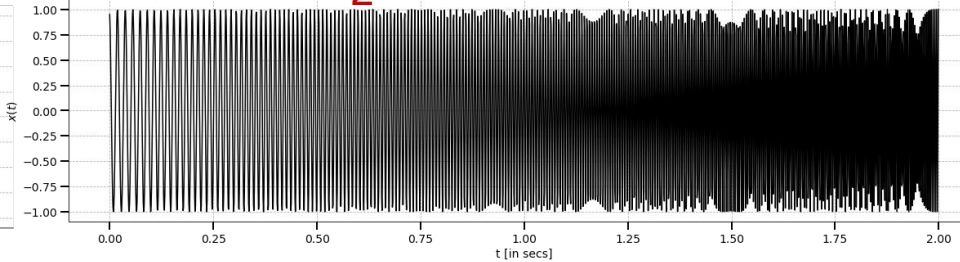




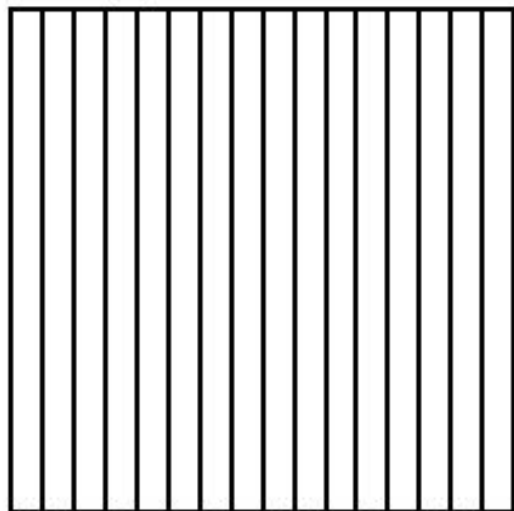
Signal:  $x_1(t)$



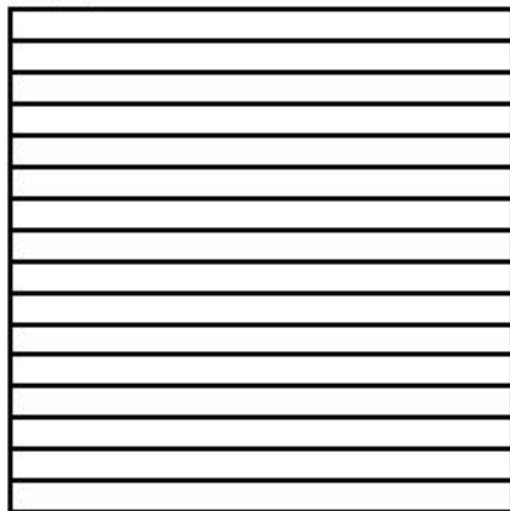
Signal:  $x_2(t)$



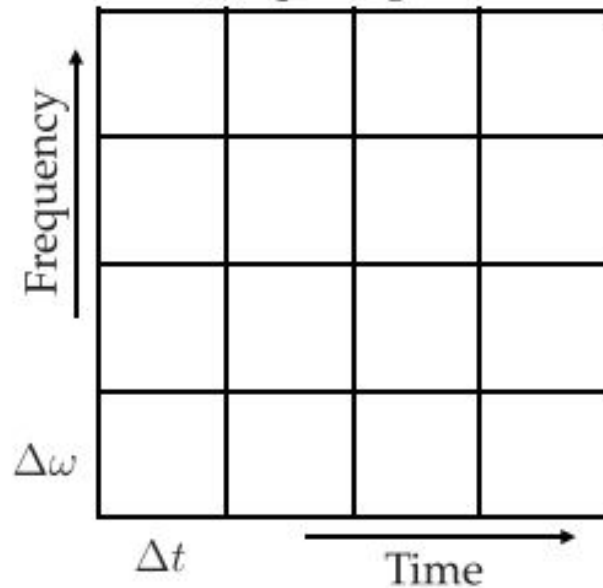
(a) Time series

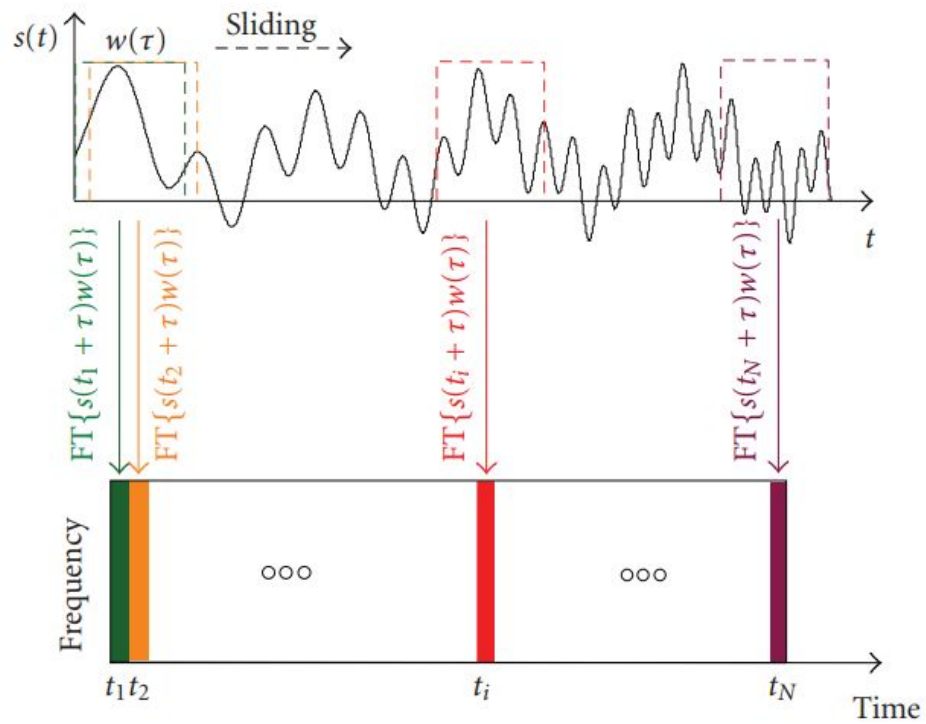


(b) Fourier transform



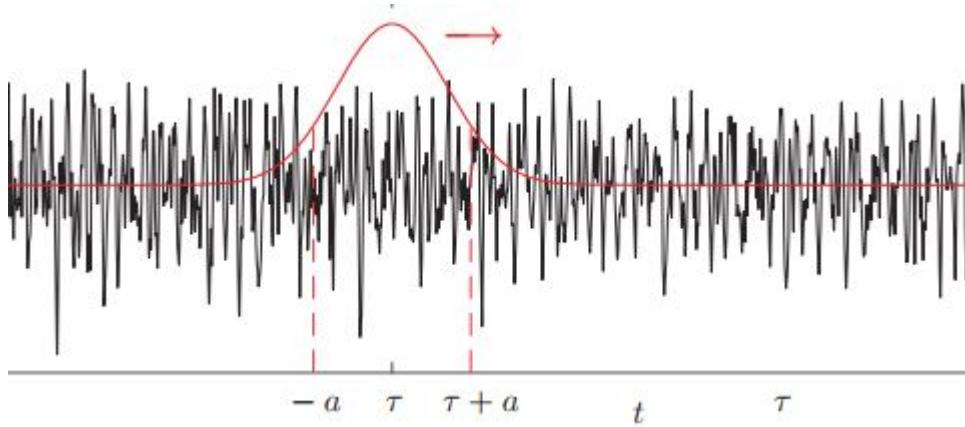
(c) Spectrogram

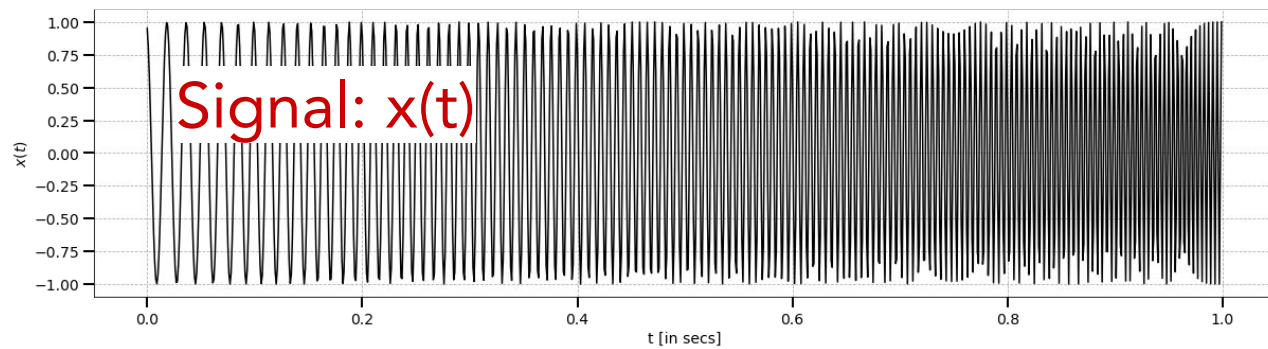


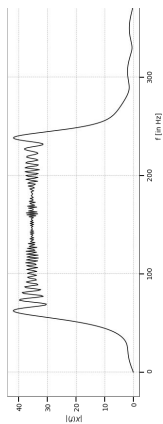
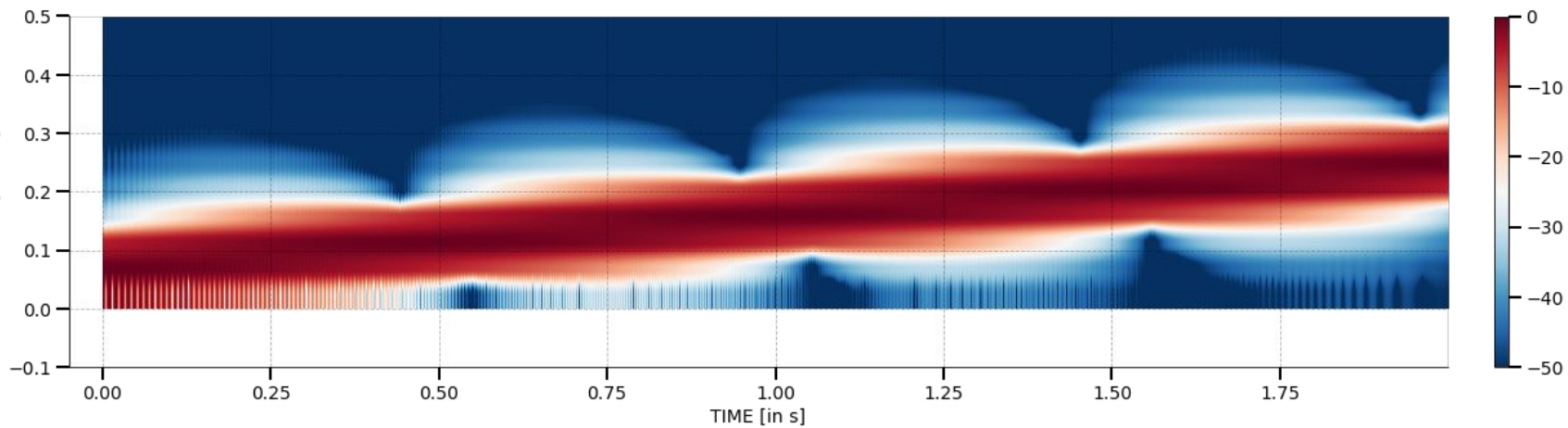
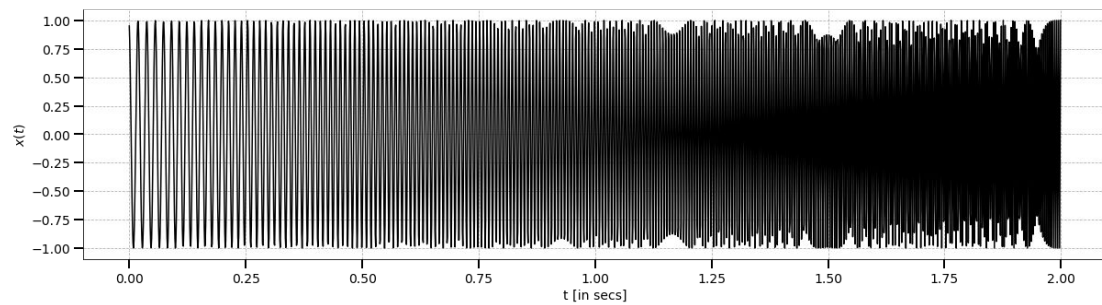


# Short-time Fourier Transform (STFT)

$$X(t, f) = \int_{-\infty}^{\infty} w(t - \tau)x(\tau)e^{-j2\pi f\tau} d\tau$$



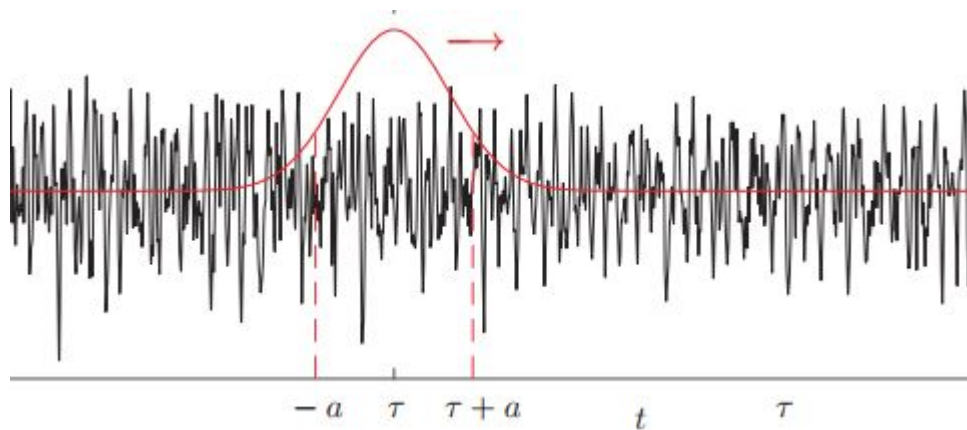






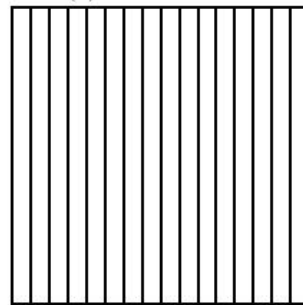
# Short-time Fourier Transform (STFT)

$$X(t, f) = \int_{-\infty}^{\infty} w(t - \tau)x(\tau)e^{-j2\pi f\tau} d\tau$$

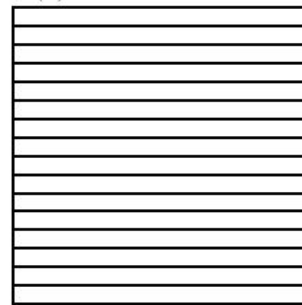


Resolution

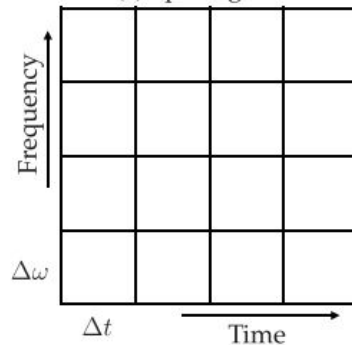
(a) Time series



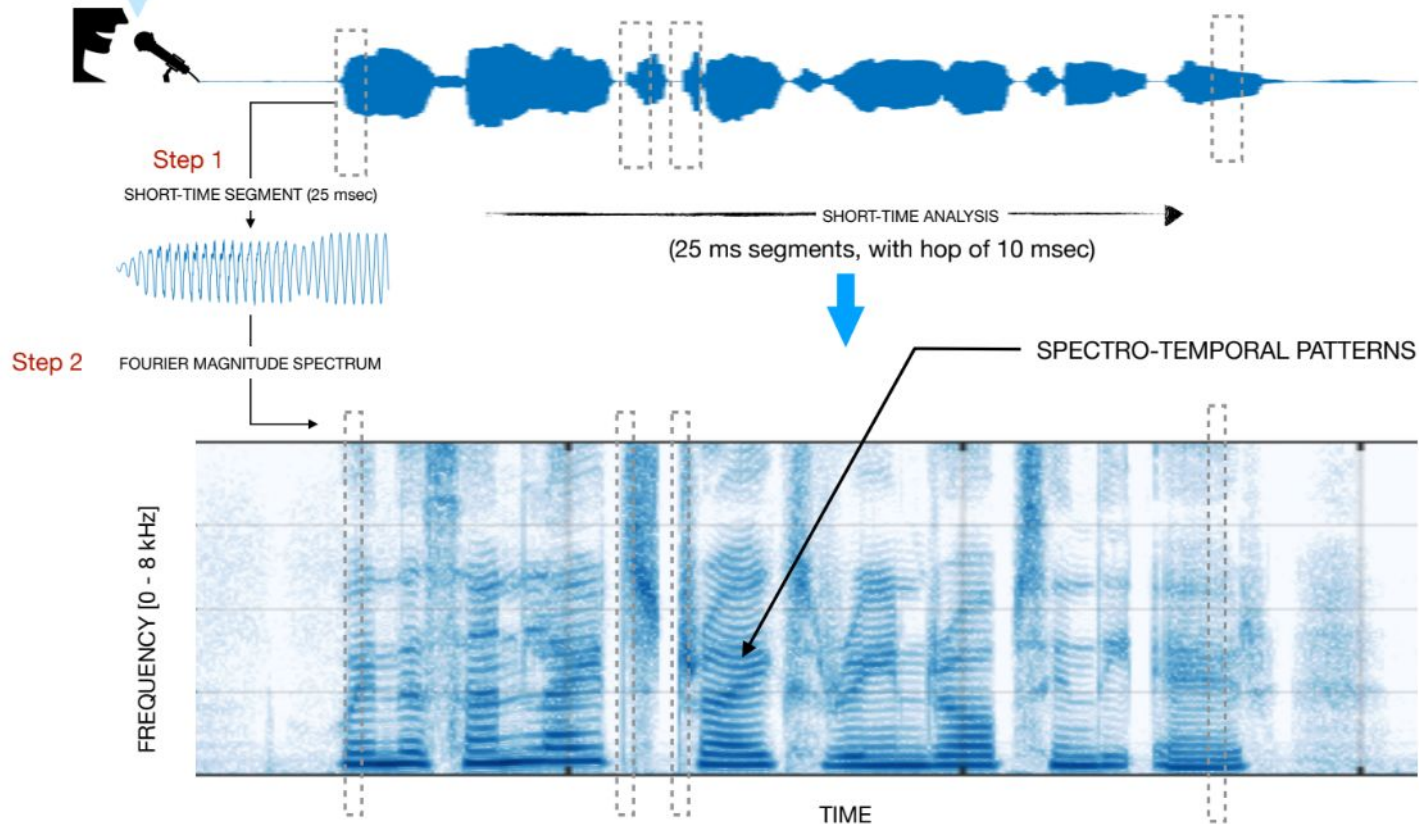
(b) Fourier transform



(c) Spectrogram



when sunlight strikes raindrops in the air



SPECTROGRAM: A 2-D VISUALIZATION OF SPEECH

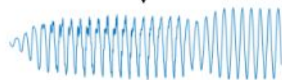
Each column is a FRAME

when sunlight strikes raindrops in the air



Step 1

SHORT-TIME SEGMENT (25 msec)

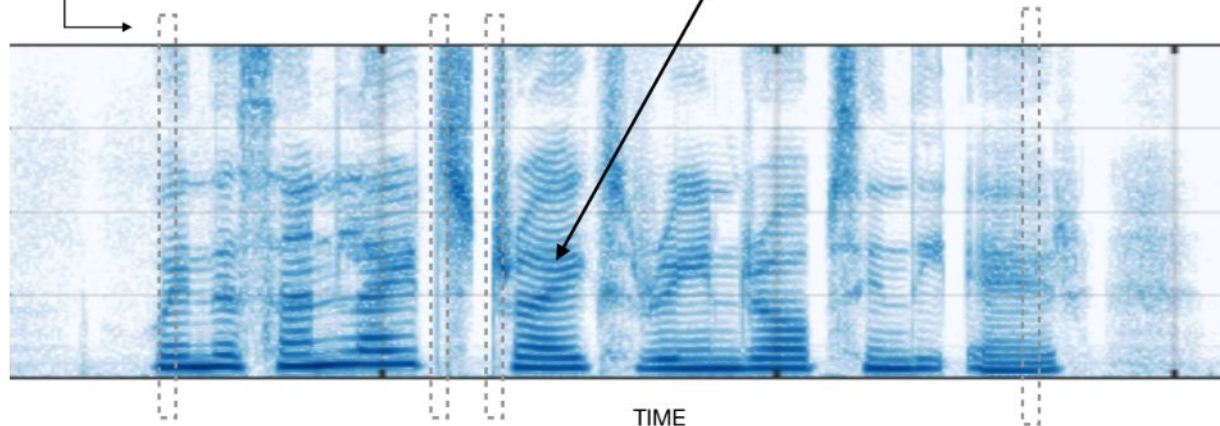


Step 2

FOURIER MAGNITUDE SPECTRUM

FREQUENCY [0 - 8 kHz]

Spectrogram



TIME

SPECTROGRAM: A 2-D VISUALIZATION OF SPEECH

Each column is a FRAME

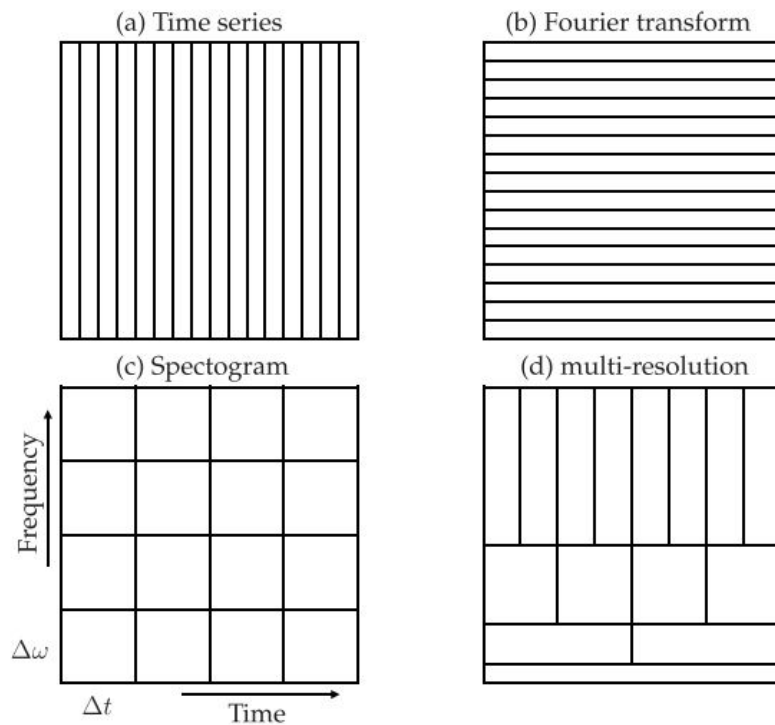
# Multi-resolution analysis

## Resolution

- Wavelets

$$\psi_{a,b}(t) = \frac{1}{\sqrt{a}} \psi \left( \frac{t-b}{a} \right)$$

$$\mathcal{W}_{\psi}(f)(a, b) = \langle f, \psi_{a,b} \rangle = \int_{-\infty}^{\infty} f(t) \bar{\psi}_{a,b}(t) dt$$



## Reading Material

### **Estimating and Interpreting The Instantaneous Frequency of a Signal—Part 1: Fundamentals**

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BOUALEM BOASHASH, SENIOR MEMBER, IEEE

### **THEORY OF COMMUNICATION\***

By D. GABOR, Dr. Ing., Associate Member.†

Thank you

