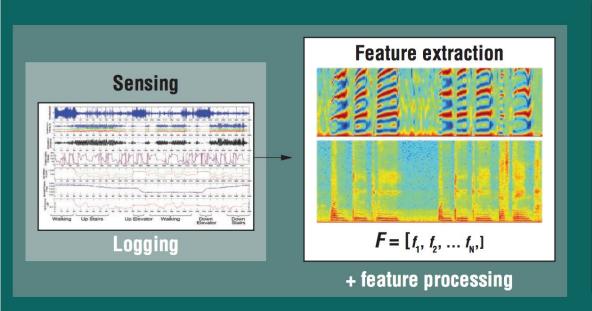
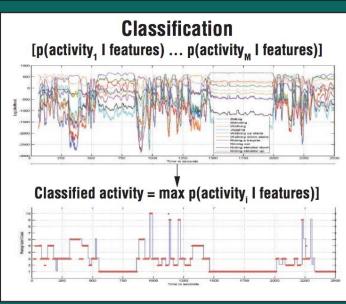
# COMPSCI 590U: Feature Engineering and Building Classifiers





+ activity recognition

# Why do Feature Engineering?

 Raw recorded data is typically not the most useful variables that characterizes what we want to detect/model.

Transformation of the raw data is required.

 The primary way to inject human knowledge into the recognition model

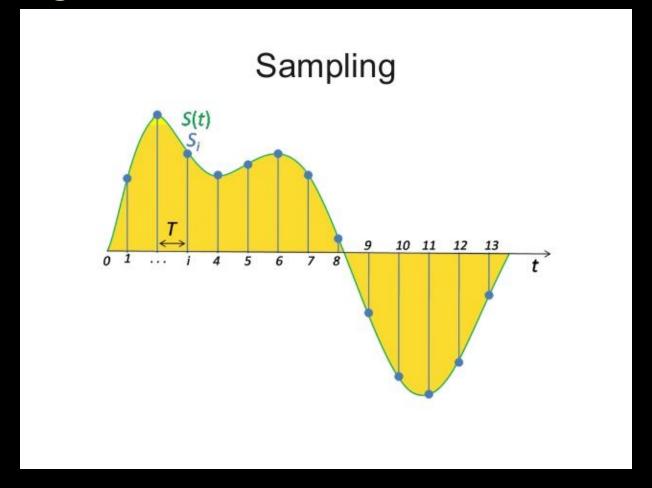
#### **Features**

 Properties of the variable that we think will help distinguish or describe the target class/label of interest

 This can be a function, transformation or combination of the raw data

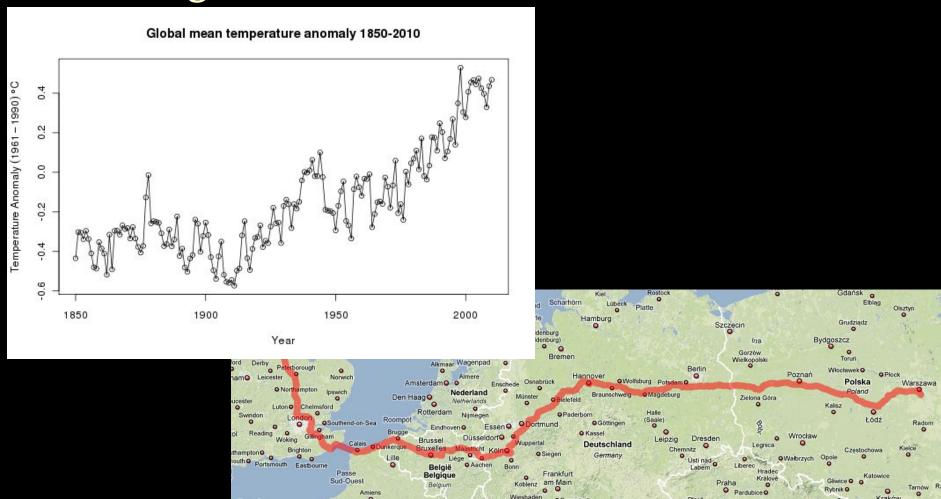
# Recording the raw data

Making an observation of a variable



# Recording the raw data

Making an observation of a variable



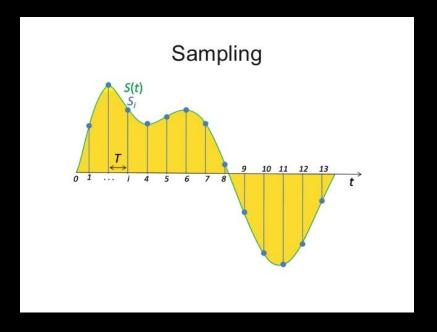
# Sampling Rate/Frequency

• The number of times per second a continuous variable is recorded.

fs = number of samples/second

• What is Sampling Period?

$$T = 1/fs$$



# Sampling Rate/Frequency

 A higher sampling rate allows a better approximation of the underlying continuous variable.

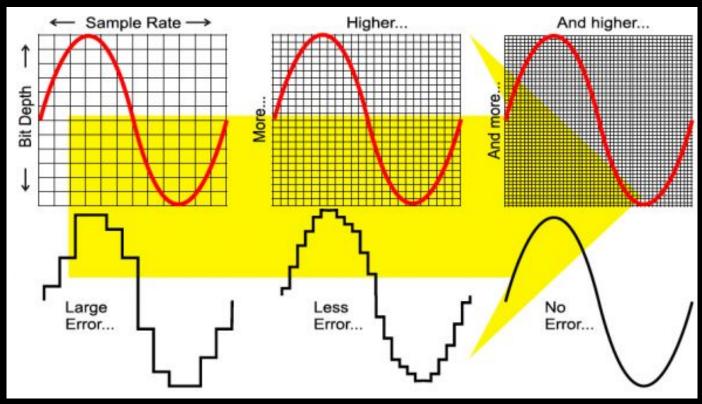


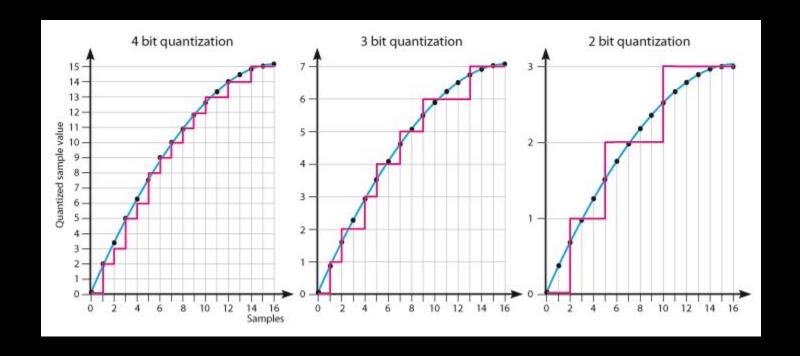
Image Source: https://www.izotope.com/en/learn/digital-audio-basics-sample-rate-and-bit-depth.html

## Quantization

- In electronics, an analog-to-digital converter (ADC) is a system that converts an analog signal, such as a sound picked up by a microphone or light entering a digital camera, into a digital signal.
- Bit depth or resolution of the ADC refers to the number bits used to convert the analog voltage or current value to a digital signal.
- The resolution also indicates the number of discrete values it can produce over the range of analog values.
- For example, an ADC with a resolution of 8 bits can encode an analog input to one in 256 different levels (2^8 = 256). The values can represent the ranges from 0 to 255 (i.e. unsigned integer) or from -128 to 127 (i.e. signed integer).

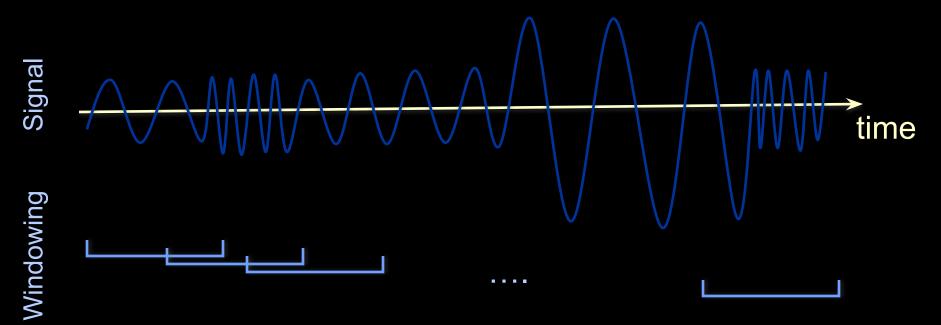
# Quantization

• With the increase of resolution or bit depth, quantization error decreases.



# Windowing

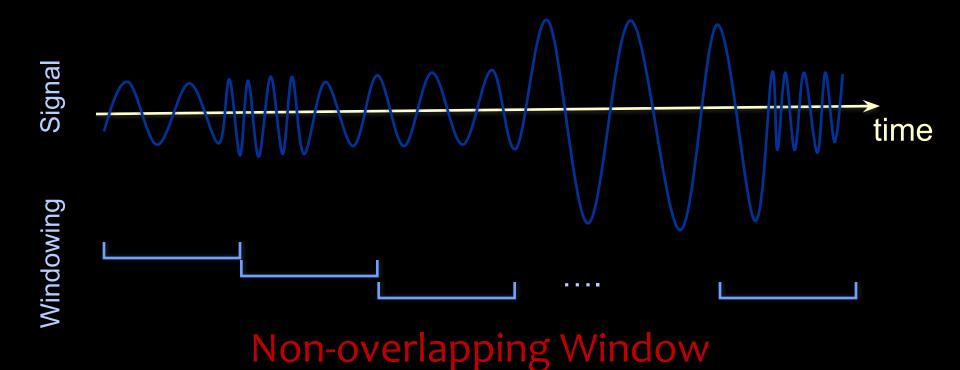
 A method of taking data from predefined intervals within the signal



Overlapping Window

# Windowing

 A method of taking data from predefined intervals within the signal



# Considerations for window size selection

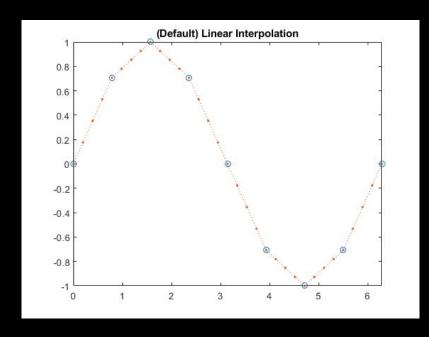
	Small Window	Large Window
Temporal Resolution	<ul> <li>High Temporal Resolution</li> <li>Can track sudden changes in time domain signal.</li> </ul>	<ul> <li>Low Temporal Resolution</li> <li>Fail to track sudden changes</li> <li>Has smearing/smoothing effect</li> </ul>
Frequency Resolution	<ul><li>Low Frequency Resolution</li><li>Fail to resolve two/more tightly spaced frequencies</li></ul>	<ul><li>High Frequency Resolution</li><li>Can resolve two/more tightly spaced frequencies</li></ul>

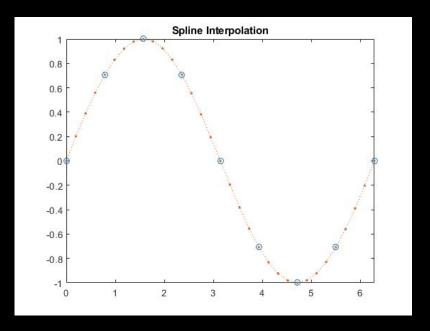
# Resampling and Interpolation

- Resampling is used to either increase the sample rate or decrease it.
- Interpolation is the process of calculating values between sample points.
- Due to device/sensor heterogeneity or due to operating system preferences, you may have a certain sensor data that has been collected with different sampling frequencies.
- Before feature extraction, you want to ensure that all the data within a certain sensor stream has the same sampling rate.

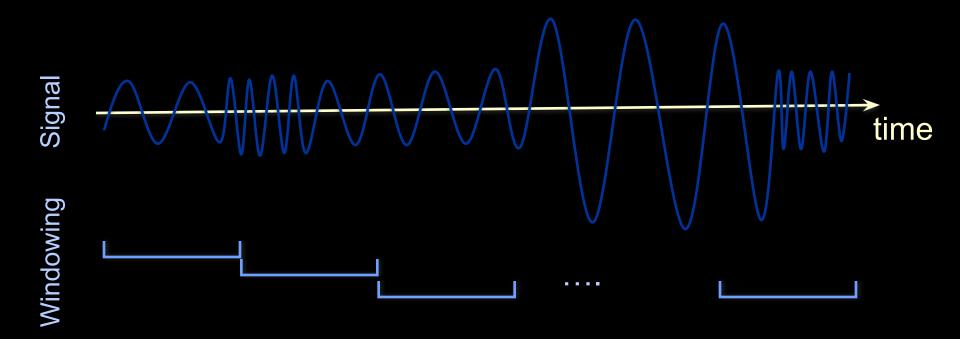
# Resampling and Interpolation

 interp1() function in Matlab allows you to achieve 1D interpolation with a wide variety of interpolation methods along with optional extrapolation criteria.





#### **Feature Extraction**



Two types of functions:

- 1. Time Domain Functions
- 2. Frequency Domain Functions

## **Examples of Time Domain Functions**

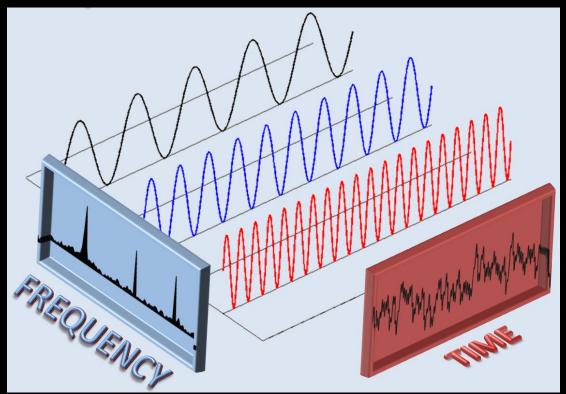
- Measure of average: Mean, mod, quartiles, percentiles, median
- Measure of spread: Variance, standard deviation, range
- Rates of change (first and second order differentials)
- Shape (e.g., slope of the raw waveform)
- Extreme values: Max, Min

# Examples of Frequency Domain Functions

- Fourier Transformation
  - raw spectra
  - Spectral features
  - filter bank features
  - Nonlinear transformations (e.g., MFCC)
- Fundamental frequency

# Fourier Analysis

- Fourier Analysis
  - All signal composed of linear combination of sinusoids.



#### **Fourier Series**

For periodic time domain signal

$$x_{T}\left(t\right)=x_{T}\left(t+T\right)$$

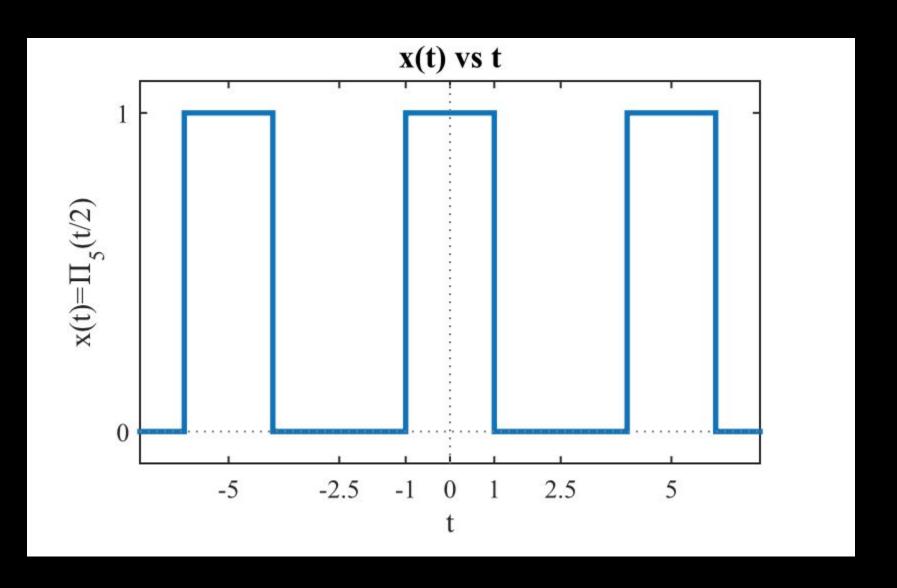
Synthesis:

$$x_T(t) = \mathcal{F}^{-1}[X[k]] = \sum_{k=-\infty}^{\infty} X[k]e^{jk\omega_0 t}$$

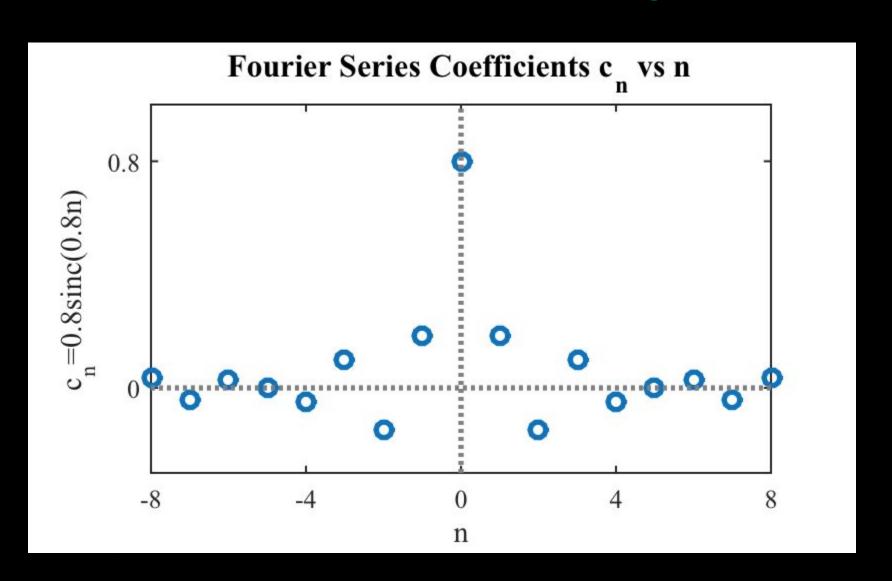
Analysis:

$$X[k] = \mathcal{F}[x_T(t)] = \frac{1}{T} \int_T x_T(t) e^{-jk\omega_0 t} dt \quad (k = 0, \pm 1, \pm 2, \cdots)$$

# Fourier Series Example



# Fourier Series Example



#### Fourier Transfom

For aperiodic time domain signal

$$x_T(t) = x_T(t+T)$$
 &  $T \to \infty \Longrightarrow \omega_0 = 2\pi/T \to 0$ 

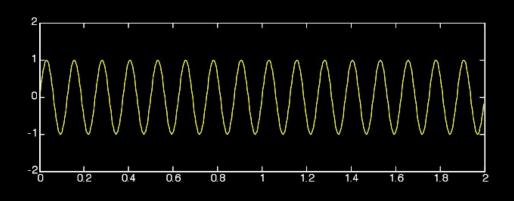
Synthesis:

$$x(t) \stackrel{\triangle}{=} \lim_{T \to \infty} x_T(t) = \lim_{\omega_0 \to 0} \frac{1}{2\pi} \sum_{k=-\infty}^{\infty} X(k\omega_0) e^{jk\omega_0 t} \omega_0 = \frac{1}{2\pi} \int_{-\infty}^{\infty} X(\omega) e^{j\omega t} d\omega$$

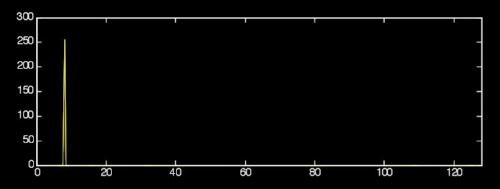
Analysis:

$$X(\omega) \stackrel{\triangle}{=} \lim_{T \to \infty} X(k\omega_0) = \lim_{T \to \infty} \int_T x_T(t) e^{-jk\omega_0 t} dt = \int_{-\infty}^{\infty} x(t) e^{-j\omega t} dt$$

### Famous Fourier Transforms

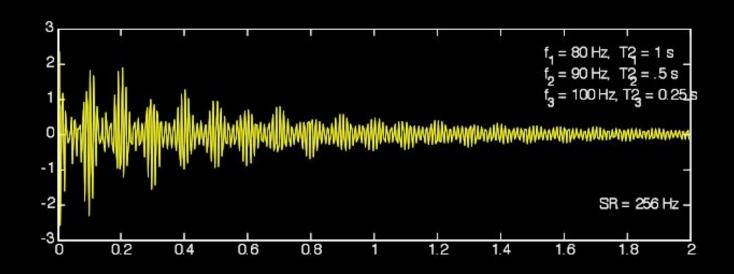


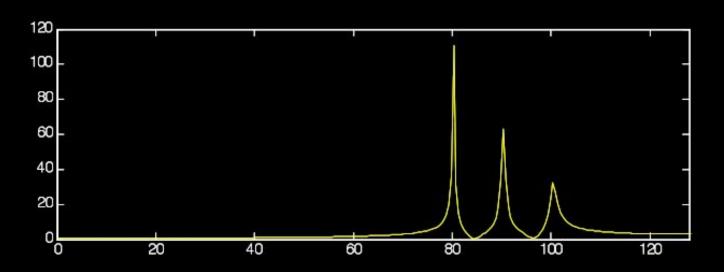
Sine wave



Delta function

### Measuring multiple frequencies





## **Fast Fourier Transformation**

# Demo

#### **Fast Fourier Transformation**

```
Matlab Code:
fs = 1000;% hz... Sampling frequency
t = 0:1/fs:10;
f1 = 100; %hz... The freq of signal 1
f2 = 250; %hz... The freq of signal 2
x1 = \sin(2^*pi^*f1^*t) + \sin(2^*pi^*f2^*t) + randn(size(t));
figure;
subplot(2,1,1)
plot(t,x1);
xlabel('time');
ylabel('signal');
FFTLen = 1024; % Length of FFT
y1 = abs(fft(x1,FFTLen));
y1 = y1 (1:FFTLen/2+1); % We will plot the first half of it as the second half is merely the reflection of the first half
y1(2:end) = y1(2:end)*2;
subplot(2,1,2);
plot([0:FFTLen/2]*fs/FFTLen,y1);
xlabel('Frequency');
ylabel('Energy');
```

#### **Fast Fourier Transformation**

```
Matlab Code:

FFTLen = 1024; % Length of FFT

y1 = abs(fft(x1,FFTLen));

y1 = y1 (1:FFTLen/2+1); % We will plot the first half of it as the second half is merely the reflection of the first half y1(2:end) = y1(2:end)*2;

subplot(2.1,2):
plot([0:FFTLen/2]*fs/FFTLen,y1);
xlabel('Frequency');
ylabel('Energy');
```

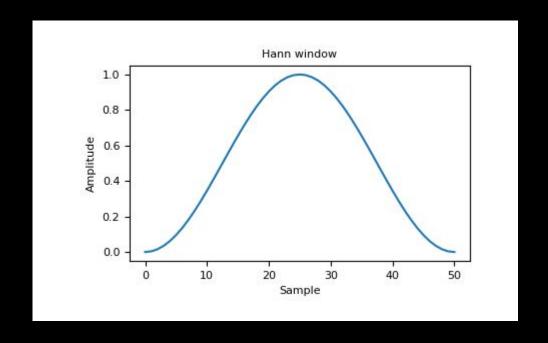
# Considerations for window size selection

	Small Window	Large Window
Temporal Resolution	<ul> <li>High Temporal Resolution</li> <li>Can track sudden changes in time domain signal.</li> </ul>	<ul> <li>Low Temporal Resolution</li> <li>Fail to track sudden changes</li> <li>Has smearing/smoothing effect</li> </ul>
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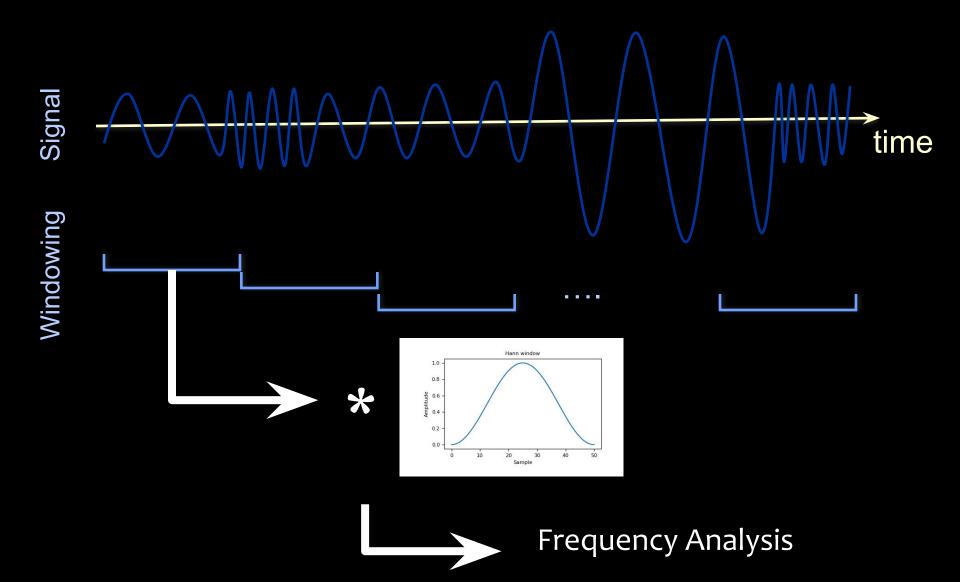
# Considerations for window function

#### **Hanning Function**

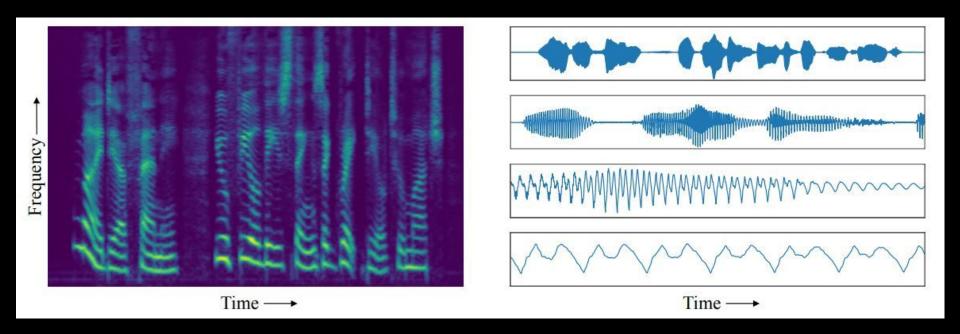
$$w(n) = 0.5 - 0.5\cos\left(\frac{2\pi n}{M-1}\right)$$
  $0 \le n \le M-1$ 



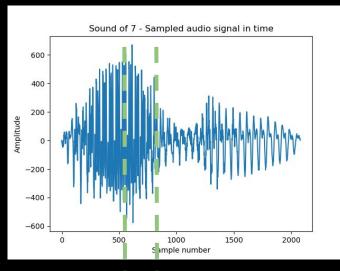
### **Feature Extraction**



# Spectrogram



# Spectrogram

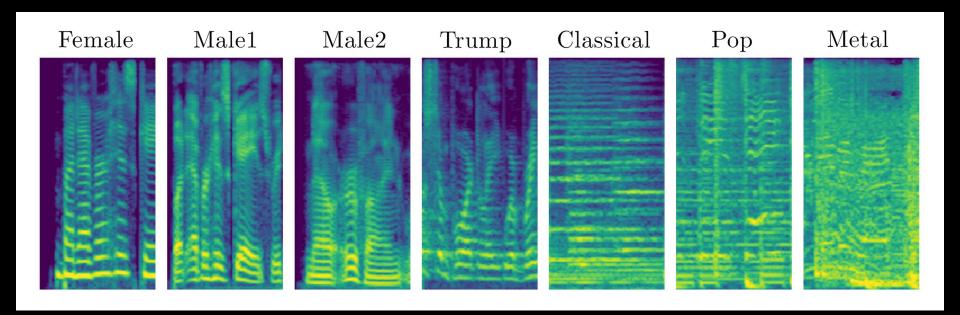


Frames: 1st 2nd ..... nth .....



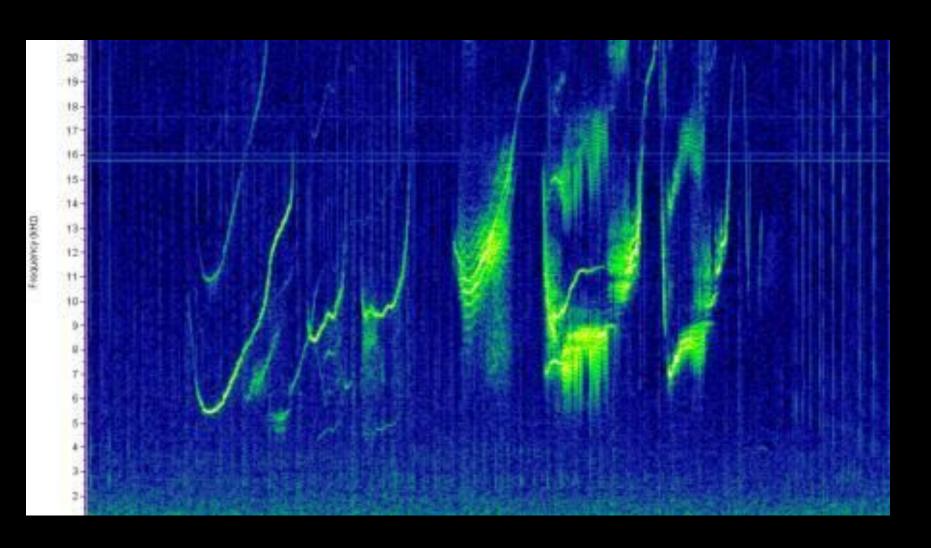
Frequencies

# Spectrogram



Frequency OHD

# Dolphin



#### Spectral Centroid

The spectral centroid is a measure that indicates where the "center of mass" of the spectrum is located.

$$ext{Centroid} = rac{\sum_{n=0}^{N-1} f(n) x(n)}{\sum_{n=0}^{N-1} x(n)}$$

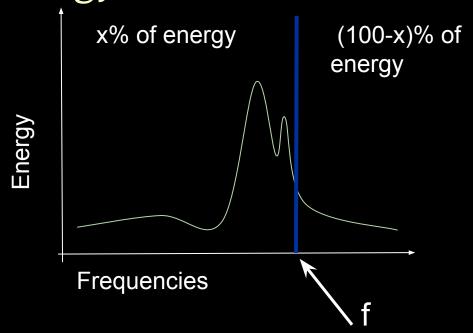
Here, f(n) refers to the nth frequency and x(n) refers to the power or weight associated with nth frequency.

#### Spectral Slope

The spectral "slope" can be quantified by applying linear regression to the Fourier magnitude spectrum of the signal, which produces a single number indicating the slope of the line-of-best-fit through the spectral data.

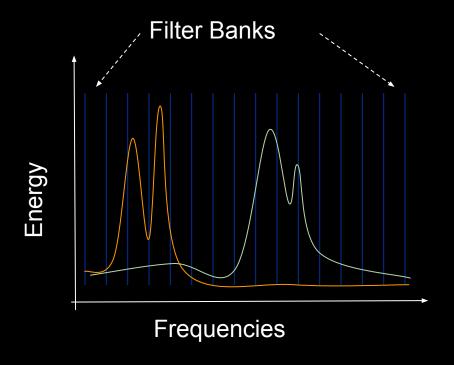
Spectral Roll off x% (e.g., 95% or 75%)

This refers to the frequency (f) below which x% of the signal energy lie.



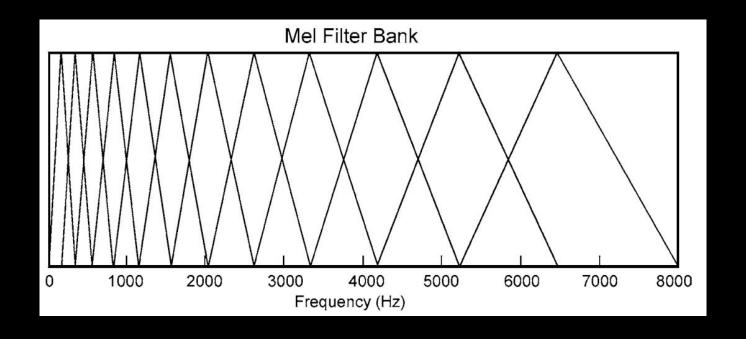
#### Filter Bank

This refers to splitting the spectra into several frequency bands and estimating the total energy in each of these bands.



#### Filter Bank

Different filters within a filter bank can have different bandwidth.

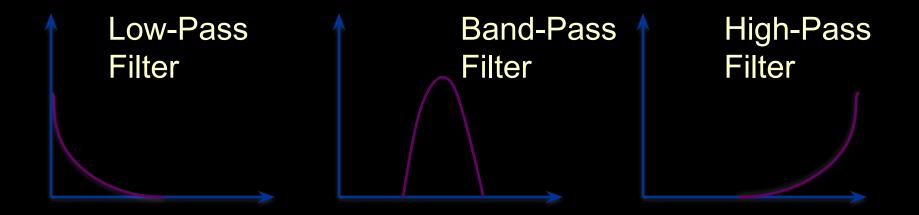


A detailed list of different spectral features can be found here.

http://docs.twoears.eu/en/latest/afe/available-processors/spectral-features/

# How to engineer good features?

- Get to know your data well (Visualize)
  - Human intuitions based on thorough observations
- Exploit current understanding
  - Talk with domain experts
  - Do a good literature review
- Begin with a large number of functions to extract a huge feature set and then use an automatic feature selection method to help you find a good feature subset.



Frequency

Original Signal

