## ELEC1200: A System View of Communications: from Signals to Packets Lecture 12

- Motivation
  - Music as a combination of sounds at different frequencies
- Cosine waves
  - Continuous time
  - Discrete time
- Decomposing signals into a sum of cosines
  - Amplitude Spectrum

### **Motivation:** Music

• Music is a combination of sounds with different frequencies:



• In this lecture, we see that this is true of any waveform ELEC1200

# Visualizing the frequency content of music

- Windows Media Player
- Right click->Visualizations->Bars and Waves->Bars



## Sinusoidal Waves



### We only need cosines

• A sine wave is just a cosine with a phase shift of  $-\frac{\pi}{2}$ 



• A negative phase shift introduces a delay of  $d = \frac{-\phi}{2\pi\tilde{f}} = \frac{-\phi\tilde{T}}{2\pi}$ 

$$\cos(2\pi \tilde{f}t + \phi) = \cos\left(2\pi \tilde{f}\left(t - \frac{-\phi}{2\pi \tilde{f}}\right)\right) = \cos\left(2\pi \tilde{f}\left(t - d\right)\right)$$

## **General Form**

• Any sinusiodal signal can be expressed as



 $x(t) = A \cos(2\pi \tilde{f}t + \phi)$  A = amplitude $\tilde{f} = \text{frequency}$ 

$$\phi = \mathsf{phase}$$

#### **Discrete Time Cosines**

- Consider a discrete time cosine waveform with N samples:  $cos(2\pi fn + \phi)$  for n = 0, 1, ... (N - 1)
- $\cdot$  In the following, we assume N is even.
- If we have only N samples, it turns out that we only need to consider N/2+1 frequencies:

$$f_k = \frac{k}{N} \text{ for } k \in \left\{0, 1, \dots \frac{N}{2}\right\} \longleftarrow \begin{array}{c} f_k \text{ is called a normalized frequency} \\ \text{It has units of cycles/sample} \end{array}$$

- For all N, it is always true that  $0 \le f_k \le 0.5$
- k indicates how many times the cosine repeats in N samples  $cos(2\pi f_k n) = cos(2\pi k \frac{n}{N})$
- The larger k, the higher the frequency

### Discrete time cosines with varying $f_k$



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## Normalized vs continuous frequency

Given a continuous cosine of frequency  $\tilde{f}$  and sampling at ٠ frequency  $F_s$ , we get a set of samples x(n):



# Sinusoidal Representation of Sampled Data

• Any sampled data waveform x(n) with N samples:

x(n) for n = 0, 1, ... (N - 1)

can be expressed as the the sum of a N/2+1 cosine waves with frequencies L

$$f_k = \frac{\kappa}{N}$$
 for  $k \in \left\{0, 1, \dots, \frac{N}{2}\right\}$ 

using the equation  $x(n) = A_0 + \sum_{k=1}^{N/2} A_k \cos(2\pi f_k n + \phi_k)$ each term in the summation is called a <u>frequency component</u>

- This equation is called the Fourier series expansion of x(n)
- Different waveforms are obtained by changing the values of  $A_k$  and  $\varphi_k$

# **Interpretation of x(n)** = $\sum_{k=0}^{N/2} A_k \cos(2\pi f_k n + \phi_k)$

- $A_k$  controls the amplitude of the cosine with frequency  $f_k$ .
- $\phi_k$  controls the phase of the cosine with frequency  $f_k$ .



# Amplitude, Phase and Frequency

- Amplitude  $A_k$ 
  - tells us how much of each frequency there is.
  - $A_0$  controls the average level of the signal and is sometimes called the DC component.
- Phase  $\phi_k$ 
  - Not so important, just shifts each cosine left or right.
- Frequency  $f_k$ 
  - tells us about how quickly the signal changes
  - Low frequencies (small k) indicate slow changes.
  - High frequencies (large k) indicate fast changes.
- We are most interested in the amplitude and how it changes with k.
  - The frequency components with the largest amplitude are the "most important."

### Amplitude Spectrum

- A plot of A<sub>k</sub> versus k
- Gives a graphical representation of which frequencies are most important





# More Complex Example





# Transforms

- The Fourier Series is only one of a set of representations (transforms) you will study later:
  - Fourier Transform
  - Laplace Transform
  - Z Transform
- Transforms are merely a different way of expressing the same data. No information is lost or gained when taking a transform.
  - Loose analogy: transform ~ translation
- We use transforms because
  - It gives us a different way of viewing or understanding the signal.
  - Some operations on signals are easier to understand/analyze after taking the transform.

# Summary

- Any signal can be expressed as the sum of cosines with different frequencies.
- Cosines in discrete time use normalized frequency, which has units of cycles/sample
- The amplitude spectrum
  - Is the weightings (sizes) of the different frequency components,  $A_k$ , plotted as a function of k
  - Tells us important information about what the signal looks like.
  - Frequency components with the largest amplitudes are the "most important"
- Knowledge of both the amplitude and phase spectrum is equivalent to knowledge of the signal.

## Appendix: Discrete Fourier Transform (DFT)

- The DFT takes a waveform x(n) and returns a set of N complex valued coefficients, called <u>Fourier coefficients</u>.
  - The Fourier coefficients are denoted by  $X_k$  for k=0,1...(N-1)
  - The formula (not important for this course) is

$$X_{k} = \sum_{n=0}^{N-1} x(n) \underbrace{e^{-j\frac{2\pi kn}{N}}}_{\text{This is the complex exponential}} \text{ For more details wait until you take ELEC2100}$$

- For this course, we use MATLAB's "fft" function to compute the Fourier coefficients.
- From the first N/2+1 Fourier coefficients, we can compute  $A_k$  and  $\phi_k$  for k=0,1,...(N/2)

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### Appendix: complex numbers

- Define  $j = \sqrt{-1}$
- A complex number is given by z = a+jb, where
  - a is the "real" part = Re{z}
  - b is the "imaginary" part = Im{z}
- Since a complex number has two parts, we can represent it as a point on a 2D plane (the complex plane)
- By analogy with polar coordinates, we can define the magnitude and phase.



# Appendix: The DFT in MATLAB

• The MATLAB function "fft" implements the "Fast Fourier Transform," a fast way to compute the DFT.

>> Xdft = fft(x);

returns an N dimensional vector containing the DFT coefficients.

- We need to be a bit careful with MATLAB's indexing
  - The first index of x(n) in our notation is n=0, but the first index of a vector in MATLAB starts with 1.
  - Thus, if we represent a waveform x(n) for n=0,..(N-1) in a MATLAB vector x. Then,

$$x(1) = x(0), x(2) = x(1), ..., x(N) = x(N-1)$$
  
MATLAB signal

- Similarly,

$$Xdft(1) = X_0$$