

## Lecture 9: Exam 1 Review

Mark Hasegawa-Johnson

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ECE 401: Signal and Image Analysis, Fall 2022

- 1 Topics Covered
- 2 Phasors
- 3 Spectrum
- 4 Fourier Series
- 5 Sampling and Interpolation
- 6 Summary

# Outline

- 1 Topics Covered
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# Topics Covered

- 1 HW1: Phasors
- 2 MP1: Spectrum
- 3 HW2: Fourier Series
- 4 MP2: Sampling

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# Phasors

$$\begin{aligned}x(t) &= A \cos(2\pi ft + \theta) \\&= \Re \left\{ z e^{j2\pi ft} \right\} \\&= \frac{1}{2} z^* e^{-j2\pi ft} + \frac{1}{2} z e^{j2\pi ft}\end{aligned}$$

where

$$z = A e^{j\theta}$$

# Adding Phasors

How do you add

$$z(t) = A \cos(2\pi ft + \theta) + B \cos(2\pi ft + \phi)?$$

Answer:

$$z = (A \cos \theta + B \cos \phi) + j(A \sin \theta + B \sin \phi)$$

$$z(t) = \Re \left\{ z e^{j2\pi ft} \right\}$$

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# Two-sided spectrum

The **spectrum** of  $x(t)$  is the set of frequencies, and their associated phasors,

$$\text{Spectrum}(x(t)) = \{(f_{-N}, a_{-N}), \dots, (f_0, a_0), \dots, (f_N, a_N)\}$$

such that

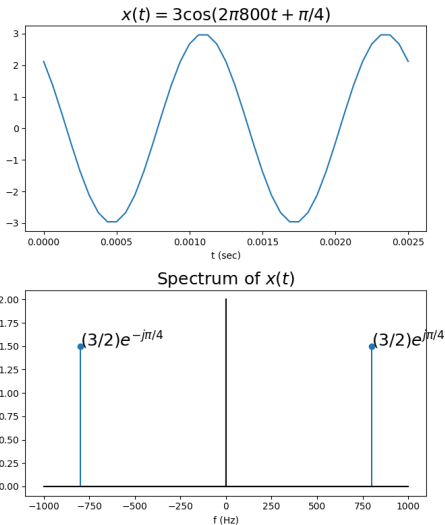
$$x(t) = \sum_{k=-N}^N a_k e^{j2\pi f_k t}$$

# Spectrum Plots

The **spectrum plot** of a periodic signal is a plot with

- frequency on the X-axis,
- showing a vertical spike at each frequency component,
- each of which is labeled with the corresponding phasor.

# Example: Cosine w/Amplitude 3, Phase $\pi/4$



# Property #1: Scaling

Suppose we have a signal

$$x(t) = \sum_{k=-N}^N a_k e^{j2\pi f_k t}$$

Suppose we scale it by a factor of  $G$ :

$$y(t) = Gx(t)$$

That just means that we scale each of the coefficients by  $G$ :

$$y(t) = \sum_{k=-N}^N (Ga_k) e^{j2\pi f_k t}$$

## Property #2: Adding a constant

Suppose we have a signal

$$x(t) = \sum_{k=-N}^N a_k e^{j2\pi f_k t}$$

Suppose we add a constant,  $C$ :

$$y(t) = x(t) + C$$

That just means that we add that constant to  $a_0$ :

$$y(t) = (a_0 + C) + \sum_{k \neq 0} a_k e^{j2\pi f_k t}$$

## Property #3: Adding two signals

Suppose we have two signals:

$$x(t) = \sum_{n=-N}^N a'_n e^{j2\pi f'_n t}$$

$$y(t) = \sum_{m=-M}^M a''_m e^{j2\pi f''_m t}$$

and we add them together:

$$z(t) = x(t) + y(t) = \sum_k a_k e^{j2\pi f_k t}$$

where, if a frequency  $f_k$  comes from both  $x(t)$  and  $y(t)$ , then we have to do phasor addition:

$$\text{If } f_k = f'_n = f''_m \text{ then } a_k = a'_n + a''_m$$

# Property #4: Time shift

Suppose we have a signal

$$x(t) = \sum_{k=-N}^N a_k e^{j2\pi f_k t}$$

and we want to time shift it by  $\tau$  seconds:

$$y(t) = x(t - \tau)$$

Time shift corresponds to a **phase shift** of each spectral component:

$$y(t) = \sum_{k=-N}^N \left( a_k e^{-j2\pi f_k \tau} \right) e^{j2\pi f_k t}$$

## Property #5: Frequency shift

Suppose we have a signal

$$x(t) = \sum_{k=-N}^N a_k e^{j2\pi f_k t}$$

and we want to shift it in frequency by some constant overall shift,  $F$ :

$$y(t) = \sum_{k=-N}^N a_k e^{j2\pi(f_k+F)t}$$

Frequency shift corresponds to amplitude modulation (multiplying it by a complex exponential at the carrier frequency  $F$ ):

$$y(t) = x(t)e^{j2\pi Ft}$$



# Property #6: Differentiation

Suppose we have a signal

$$x(t) = \sum_{k=-N}^N a_k e^{j2\pi f_k t}$$

and we want to differentiate it:

$$y(t) = \frac{dx}{dt}$$

Differentiation corresponds to scaling each spectral coefficient by  $j2\pi f_k$ :

$$y(t) = \sum_{k=-N}^N (j2\pi f_k a_k) e^{j2\pi f_k t}$$

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# Fourier Series

- **Analysis** (finding the spectrum, given the signal):

$$X_k = \frac{1}{T_0} \int_0^{T_0} x(t) e^{-j2\pi kt/T_0} dt$$

- **Synthesis** (finding the signal, given the spectrum):

$$x(t) = \sum_{k=-\infty}^{\infty} X_k e^{j2\pi kt/T_0}$$

# Discrete-Time Fourier Series

- **Analysis** (finding the spectrum, given the signal):

$$X_k = \frac{1}{N_0} \sum_0^{N_0-1} x[n] e^{-j2\pi kn/N_0}$$

- **Synthesis** (finding the signal, given the spectrum):

$$x[n] = \sum_k X_k e^{j2\pi kn/N_0}$$

where the sum is over any set of  $N_0$  consecutive harmonics.

# Spectral Properties of Fourier Series

- **Scaling:**

$$y(t) = Gx(t) \Leftrightarrow Y_k = GX_k$$

- **Add a Constant:**

$$y(t) = x(t) + C \Leftrightarrow Y_k = \begin{cases} X_0 + C & k = 0 \\ X_k & \text{otherwise} \end{cases}$$

- **Add Signals:** Suppose that  $x(t)$  and  $y(t)$  have the same fundamental frequency, then

$$z(t) = x(t) + y(t) \Leftrightarrow Z_k = X_k + Y_k$$

# Spectral Properties of Fourier Series

- **Time Shift:** Shifting to the right, in time, by  $\tau$  seconds:

$$y(t) = x(t - \tau) \Leftrightarrow Y_k = a_k e^{-j2\pi k F_0 \tau}$$

- **Frequency Shift:** Shifting upward in frequency by  $F$  Hertz:

$$y(t) = x(t) e^{j2\pi d F_0 t} \Leftrightarrow Y_k = X_{k-d}$$

- **Differentiation:**

$$y(t) = \frac{dx}{dt} \Leftrightarrow Y_k = j2\pi k F_0 X_k$$

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# How to sample a continuous-time signal

Suppose you have some continuous-time signal,  $x(t)$ , and you'd like to sample it, in order to store the sample values in a computer. The samples are collected once every  $T_s = \frac{1}{F_s}$  seconds:

$$x[n] = x(t = nT_s)$$

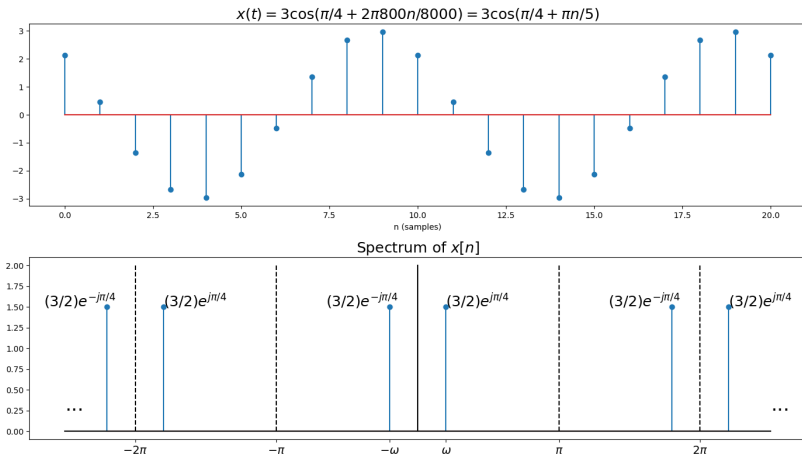


# Spectrum Plot of a Discrete-Time Periodic Signal

The spectrum plot of a **discrete-time periodic signal** is a regular spectrum plot, but with the X-axis relabeled. Instead of frequency in Hertz =  $\left[ \frac{\text{cycles}}{\text{second}} \right]$ , we use

$$\omega \left[ \frac{\text{radians}}{\text{sample}} \right] = \frac{2\pi \left[ \frac{\text{radians}}{\text{cycle}} \right] f \left[ \frac{\text{cycles}}{\text{second}} \right]}{F_s \left[ \frac{\text{samples}}{\text{second}} \right]}$$

# Example: Cosine w/Amplitude 3, Phase $\pi/4$



# Aliasing

- A sampled sinusoid can be reconstructed perfectly if the Nyquist criterion is met,  $f < \frac{F_s}{2}$ .
- If the Nyquist criterion is violated, then:
  - If  $\frac{F_s}{2} < f < F_s$ , then it will be aliased to

$$f_a = F_s - f$$

$$z_a = z^*$$

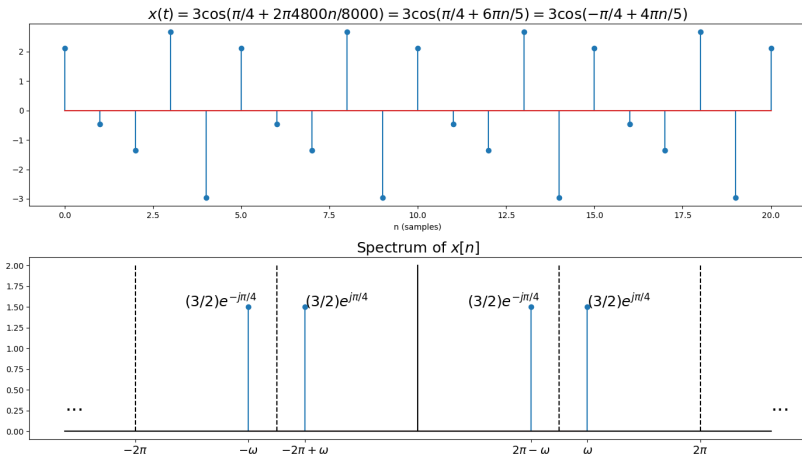
i.e., the sign of all sines will be reversed.

- If  $F_s < f < \frac{3F_s}{2}$ , then it will be aliased to

$$f_a = f - F_s$$

$$z_a = z$$

# Example: Cosine w/Amplitude 3, Phase $\pi/4$



# Interpolation

Interpolation is the general method for reconstructing a continuous-time signal from its samples. The formula is:

$$y(t) = \sum_{n=-\infty}^{\infty} y[n]p(t - nT_s)$$

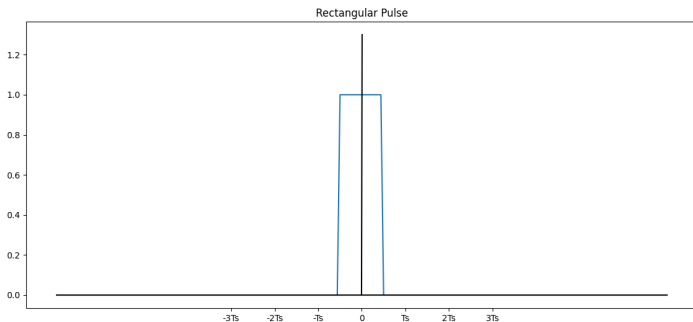
# Interpolation kernels

- Piece-wise constant interpolation = interpolate using a rectangle
- Piece-wise linear interpolation = interpolate using a triangle
- Ideal interpolation = interpolate using a sinc

# Rectangular pulses

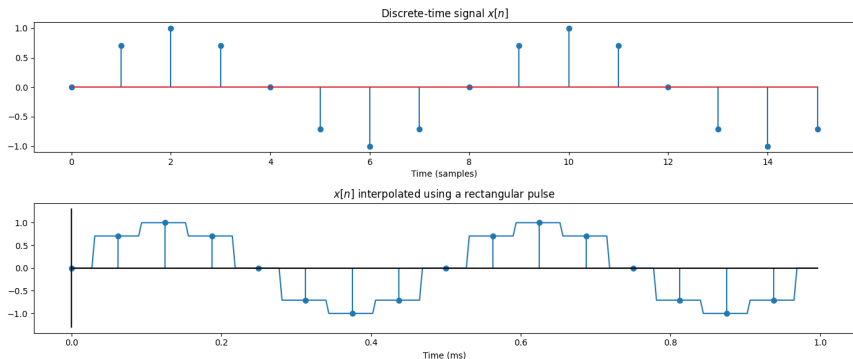
For example, suppose that the pulse is just a rectangle,

$$p(t) = \begin{cases} 1 & -\frac{T_s}{2} \leq t < \frac{T_s}{2} \\ 0 & \text{otherwise} \end{cases}$$



# Rectangular pulses = Piece-wise constant interpolation

The result is a piece-wise constant interpolation of the digital signal:

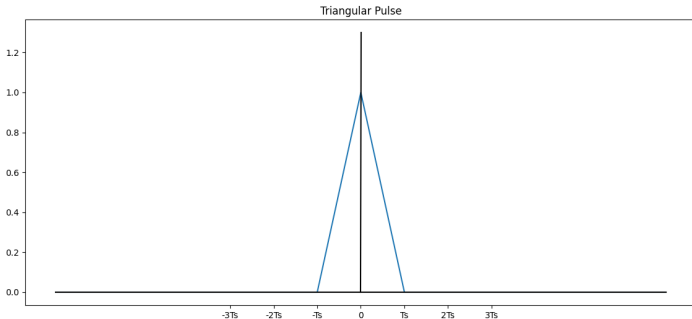




# Triangular pulses

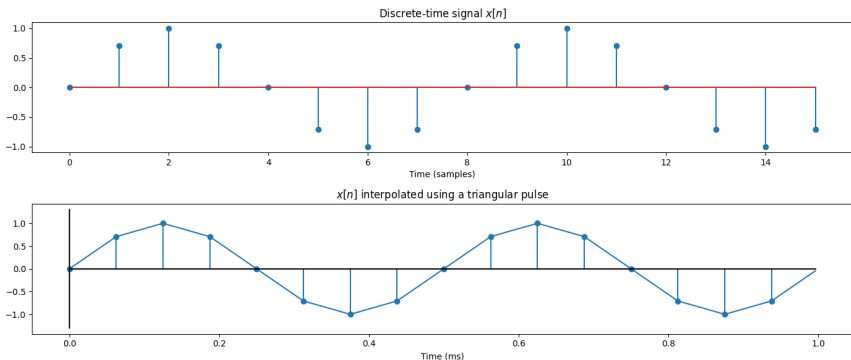
The rectangular pulse has the disadvantage that  $y(t)$  is discontinuous. We can eliminate the discontinuities by using a triangular pulse:

$$p(t) = \begin{cases} 1 - \frac{|t|}{T_S} & -T_S \leq t < T_S \\ 0 & \text{otherwise} \end{cases}$$



# Triangular pulses = Piece-wise linear interpolation

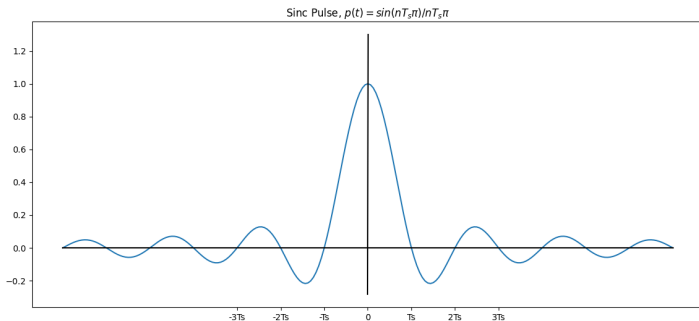
The result is a piece-wise linear interpolation of the digital signal:



# Sinc pulses

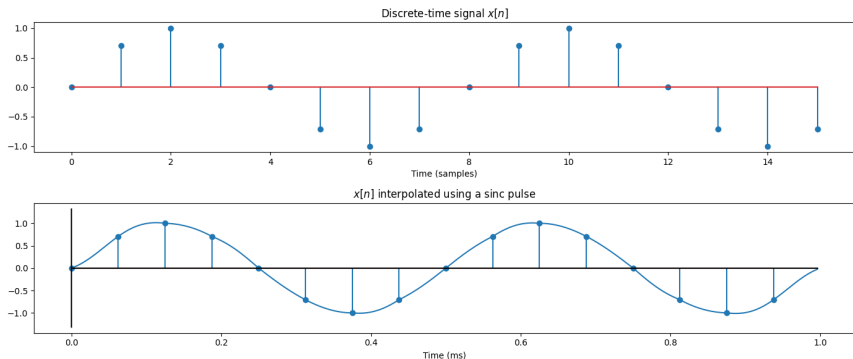
If a signal has all its energy at frequencies below Nyquist ( $f < \frac{F_s}{2}$ ), then it can be perfectly reconstructed using sinc interpolation:

$$p(t) = \frac{\sin(\pi t/T_S)}{\pi t/T_S}$$



# Sinc pulse = ideal bandlimited interpolation

If a signal has all its energy at frequencies below Nyquist ( $f < \frac{F_s}{2}$ ), then it can be perfectly reconstructed using sinc interpolation:



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