

Digital Signal Processing

Class 13
03/04/2025

ENGR 71

- Class Overview
 - Frequency Analysis of Discrete Signals
- Assignments
 - Reading:
Chapter 4: Frequency Analysis of Signals

- New Lab
 - Using frequency domain features for classification

Frequency Analysis for Discrete Signals

- Frequency analysis for discrete signals:
 - Three transforms to consider:
 - Discrete Time Fourier Transform - DTFT
 - Fourier transform of sampled signal
 - Discrete Time Fourier Series - DTFS
 - Fourier series of sampled periodic signal
 - Discrete Fourier Transform – DFT
 - Create periodic extension of finite sequence
 - Then find the Fourier series.
 - This is the transform that is most often used
 - Fast algorithm to compute: Fast Fourier Transform (FFT)

Discrete-time Fourier transform

- Discrete Time Fourier Transform (DTFT)

- Fourier transform of sampled signal

$$x_s(t) = \sum_n x(nT_s) \delta(t - nT_s)$$

$$\mathcal{F}\{x_s(t)\} = \sum_n x(nT_s) \mathcal{F}\{\delta(t - nT_s)\} = \sum_n x(nT_s) e^{-jn\Omega T_s}$$

$$\left[\text{Using } \mathcal{F}\{\delta(t)\} = 1 \text{ and shift property } \mathcal{F}\{x(t - \tau)\} = X(\Omega) e^{-j\Omega\tau} \right]$$

Ω is the analog frequency variable

Note that: $\mathcal{F}\{x_s(t)\}$ is periodic:

$$\sum_n x(nT_s) e^{-jn\Omega T_s} = \sum_n x(nT_s) e^{-jn\left(\Omega + \frac{2\pi k}{T_s}\right) T_s}$$

So, the spectrum of a sampled signal is periodic

Discrete-time Fourier transform

– Discrete Time Fourier Transform (DTFT)

Define $\omega = \Omega T_s$ as the frequency of the discrete signal (in radians)
and define $x[n] = x(nT_s)$ as samples of the sampled signal

- Fourier transform of sampled signal

$$X(e^{j\omega}) = \sum_n x[n] e^{-j\omega n} \quad -\pi \leq \omega < \pi$$
$$x[n] = \frac{1}{2\pi} \int_{-\pi}^{\pi} X(e^{j\omega}) e^{j\omega n} d\omega$$

- Note that this is a continuous function in the variable ω
- Measures frequency content of discrete signal
(Discrete frequency is in radians)
- DTFT is periodic in frequency ω

$$X(e^{j(\omega+2\pi k)}) = \sum_n x[n] e^{-j(\omega+2\pi k)n} = \sum_n x[n] e^{-j\omega n} e^{-j2\pi k n} = \sum_n x[n] e^{-j\omega n} = X(e^{j\omega})$$

Discrete-time Fourier transform

- DTFT exists if sequence is absolutely summable

$$\left| X(e^{j\omega}) \right| \leq \sum_n |x[n]| \left| e^{-j\omega n} \right| = \sum_n |x[n]| < \infty$$

- Relationship of z-transform to DTFT:

$$X(z) \Big|_{z=e^{j\omega}} = \sum_n x[n] z^{-n} \Big|_{e^{j\omega}} \Rightarrow \sum_n x[n] e^{-j\omega n} = X(e^{j\omega})$$

- i.e. Z-transform computed on unit circle.
(Region of Convergence (ROC) must include unit circle.)

Discrete-time Fourier transform

- Eigenfunctions and the DTFT

- Suppose input to system is $x[n] = e^{j\omega_0 n}$
- Output is

$$\begin{aligned} y[n] &= \sum_k h[k] x[n-k] = \sum_k h[k] e^{j\omega_0(n-k)} \\ &= e^{j\omega_0 n} \sum_k h[k] e^{-j\omega_0 k} = H(e^{j\omega_0}) e^{j\omega_0 n} \end{aligned}$$

- Output is same as input multiplied by DTFT of the impulse response
- That is to say, $x[n] = e^{j\omega_0 n}$ are eigenvectors of systems with eigenvalues of $H(e^{j\omega_0})$, the DTFT evaluated at ω_0

Discrete-time Fourier transform

- Since DTFT can be obtained from z-transform
 - Has same properties for time shifts, convolution, etc.
 - Expressed in terms of $e^{-j\omega}$ instead of z

Discrete-time Fourier transform

Discrete-time Fourier Transforms (DTFT)

	Discrete-time signal	DTFT $X(e^{j\omega})$, periodic of period 2π
(1)	$\delta[n]$	$1, -\pi \leq \omega < \pi$
(2)	A	$2\pi A\delta(\omega), -\pi \leq \omega < \pi$
(3)	$e^{j\omega_0 n}$	$2\pi \delta(\omega - \omega_0), -\pi \leq \omega < \pi$
(4)	$\alpha^n u[n], \alpha < 1$	$\frac{1}{1 - \alpha e^{-j\omega}}, -\pi \leq \omega < \pi$
(5)	$n \alpha^n u[n], \alpha < 1$	$\frac{\alpha e^{-j\omega}}{(1 - \alpha e^{-j\omega})^2}, -\pi \leq \omega < \pi$
(6)	$\cos(\omega_0 n) u[n]$	$\pi [\delta(\omega - \omega_0) + \delta(\omega + \omega_0)], -\pi \leq \omega < \pi$
(7)	$\sin(\omega_0 n) u[n]$	$-j\pi [\delta(\omega - \omega_0) - \delta(\omega + \omega_0)], -\pi \leq \omega < \pi$
(8)	$\alpha^{ n }, \alpha < 1$	$\frac{1 - \alpha^2}{1 - 2\alpha \cos(\omega) + \alpha^2}, -\pi \leq \omega < \pi$
(9)	$p[n] = u[n + N/2] - u[n - N/2]$	$\frac{\sin(\omega(N+1)/2)}{\sin(\omega/2)}, -\pi \leq \omega < \pi$
(10)	$\alpha^n \cos(\omega_0 n) u[n]$	$\frac{1 - \alpha \cos(\omega_0) e^{-j\omega}}{1 - 2\alpha \cos(\omega_0) e^{-j\omega} + \alpha^2 e^{-2j\omega}}, -\pi \leq \omega < \pi$
(11)	$\alpha^n \sin(\omega_0 n) u[n]$	$\frac{\alpha \sin(\omega_0) e^{-j\omega}}{1 - 2\alpha \cos(\omega_0) e^{-j\omega} + \alpha^2 e^{-2j\omega}}, -\pi \leq \omega < \pi$

Discrete-time Fourier transform

Properties of the DTFT

Z-transform:	$x[n], X(z), z = 1 \in ROC$	$X(e^{j\omega}) = X(z) _{z=e^{j\omega}}$
Periodicity:	$x[n]$	$X(e^{j\omega}) = X(e^{j(\omega+2\pi k)}), k \text{ integer}$
Linearity:	$\alpha x[n] + \beta y[n]$	$\alpha X(e^{j\omega}) + \beta Y(e^{j\omega})$
Time-shifting:	$x[n - N]$	$e^{-j\omega N} X(e^{j\omega})$
Frequency-shift:	$x[n]e^{j\omega_0 n}$	$X(e^{j(\omega-\omega_0)})$
Convolution:	$(x * y)[n]$	$X(e^{j\omega}) Y(e^{j\omega})$
Multiplication:	$x[n]y[n]$	$\frac{1}{2\pi} \int_{-\pi}^{\pi} X(e^{j\theta}) Y(e^{j(\omega-\theta)}) d\theta$
Symmetry:	$x[n]$ real-valued	$ X(e^{j\omega}) $ even function of ω $\angle X(e^{j\omega})$ odd function of ω
Parseval's relation:	$\sum_{n=-\infty}^{\infty} x[n] ^2 = \frac{1}{2\pi} \int_{-\pi}^{\pi} X(e^{j\omega}) ^2 d\omega$	

Discrete-time Fourier series

- Consider the frequency representation of a periodic sequence where N is the period. $x[n + kN] = x[n]$
 - A periodic sequence can be represented in terms of a sum over basis functions:
$$\phi[k, n] = e^{j2\pi kn/N} \quad (\text{Different notation, but same as } s_k(n) \text{ in Proakis and Manolakis})$$
 - These basis functions are periodic in k and n with period N
 - Easy to show. Substitute $k = k + rN$; substitute $n = n + rN$ where r is an integer
 - Basis functions are orthogonal over period N

$$\sum_{n=0}^{N-1} \phi[k, n] \times \phi^*[l, n] = \sum_{n=0}^{N-1} e^{j\frac{2\pi}{N}kn} \times e^{-j\frac{2\pi}{N}ln} = \begin{cases} N & k = l \\ 0 & k \neq l \end{cases}$$

Discrete-time Fourier series

- You can show orthogonality using our old friend, the geometric series, but not consider the finite geometric series:

$$1 + r + r^2 + r^3 + \dots + r^{N-1} = \sum_{n=0}^{N-1} r^n = \frac{1 - r^N}{1 - r} \text{ for } r \neq 1$$

$$\begin{aligned} \sum_{n=0}^{N-1} \phi[k, n] \times \phi^*[l, n] &= \sum_{n=0}^{N-1} e^{j\frac{2\pi}{N}kn} \times e^{-j\frac{2\pi}{N}ln} = \sum_{n=0}^{N-1} e^{j\frac{2\pi}{N}(k-l)n} \\ &= \sum_{n=0}^{N-1} \left(e^{j\frac{2\pi(k-l)}{N}} \right)^n = \frac{1 - e^{j\frac{2\pi(k-l)N}{N}}}{1 - e^{j\frac{2\pi(k-l)}{N}}} = \frac{1 - e^{j2\pi(k-l)}}{1 - e^{j\frac{2\pi(k-l)}{N}}} = 0 \text{ if } k \neq l \end{aligned}$$

If $k = l$,

$$\sum_{n=0}^{N-1} e^{j\frac{2\pi}{N}(k-l)n} = \sum_{n=0}^{N-1} \left(e^{j\frac{2\pi}{N}n} \right)^0 = \sum_{n=0}^{N-1} 1 = N$$

Discrete-time Fourier series

- The orthogonality of $\phi[k, n] = e^{j2\pi kn/N}$ can be used to represent a periodic sequence (of period N) as:

$$x[n] = \sum_{k=0}^{N-1} X[k] e^{j\frac{2\pi}{N}kn} \quad \text{where} \quad X[k] = \frac{1}{N} \sum_{n=0}^{N-1} x[n] e^{-j\frac{2\pi}{N}kn}$$

The nomenclature in the book is different than that shown here. $c_k \equiv X[k]$

which is the Fourier Series of $x[n]$.

The fundamental frequency is $\omega_o = 2\pi/N$

Notice that the frequency components for $X[k]$ are discrete

Both signal and Fourier series are discrete sequences.

(In contrast to Discrete Time Fourier Transform)

Discrete-time Fourier series

– Power spectrum

$$P_x = \frac{1}{N} \sum_{k=0}^{N-1} |x[n]|^2$$

Also

$$P_x = \frac{1}{N} \sum_{k=0}^{N-1} x[n]x^*[n] = \sum_{k=0}^{N-1} X^*[k] \left(\frac{1}{N} \sum_k x[n] e^{-j\frac{2\pi}{N}kn} \right)$$

$$P_x = \sum_{k=0}^{N-1} X^*[k]X[k]$$

$$P_x = \sum_{k=0}^{N-1} |X[k]|^2$$

$$P_x = \frac{1}{N} \sum_{k=0}^{N-1} |x[n]|^2 = \sum_{k=0}^{N-1} |X[k]|^2$$

Discrete-time Fourier series

– Energy spectrum

$$E_x = \sum_{k=0}^{N-1} |x[n]|^2$$

Also

$$E_x = \sum_{k=0}^{N-1} x[n]x^*[n] = \sum_{k=0}^{N-1} X^*[k] \left(\frac{N}{N} \sum_k x[n] e^{-j\frac{2\pi}{N}kn} \right)$$

$$E_x = N \sum_{k=0}^{N-1} X^*[k]X[k]$$

$$E_x = N \sum_{k=0}^{N-1} |X[k]|^2$$

$$E_x = \sum_{k=0}^{N-1} |x[n]|^2 = N \sum_{k=0}^{N-1} |X[k]|^2$$

Discrete-time Fourier series

- Symmetry for real signals

$$X^*[k] = \frac{1}{N} \sum_{n=0}^{N-1} x[n] e^{+j\frac{2\pi}{N}kn} = X[-k]$$

$$|X[-k]| = |X[k]|$$

$$-\angle X[-k] = \angle X[k]$$

$X[k]$ is also periodic

$$X[k + N] = X[k] \Rightarrow X[N - k] = X[-k]$$

$$|X[k]| = |X[N - k]|$$

$$\angle X[k] = -\angle X[N - k]$$

$$|X[0]| = |X[N]|$$

$$|X[1]| = |X[N - 1]|$$

$$|X[N/2]| = |X[N/2]| \quad N \text{ even}$$

$$|X[(N - 1)/2]| = |X[(N + 1)/2]| \quad N \text{ odd}$$

$$\angle X[0] = -\angle X[N]$$

$$\angle X[1] = -\angle X[N - 1]$$

$$\angle X[N/2] = 0 \quad N \text{ even}$$

$$\angle X[(N - 1)/2] = -\angle X[(N + 1)/2] \quad N \text{ odd}$$

Discrete-time Fourier series

- Obtaining Fourier series coefficients for discrete sequences from the z-transform is similar to what you do for continuous signals from the Laplace transform.

For $x_1[n] = x[n](u[n] - u[n - N])$

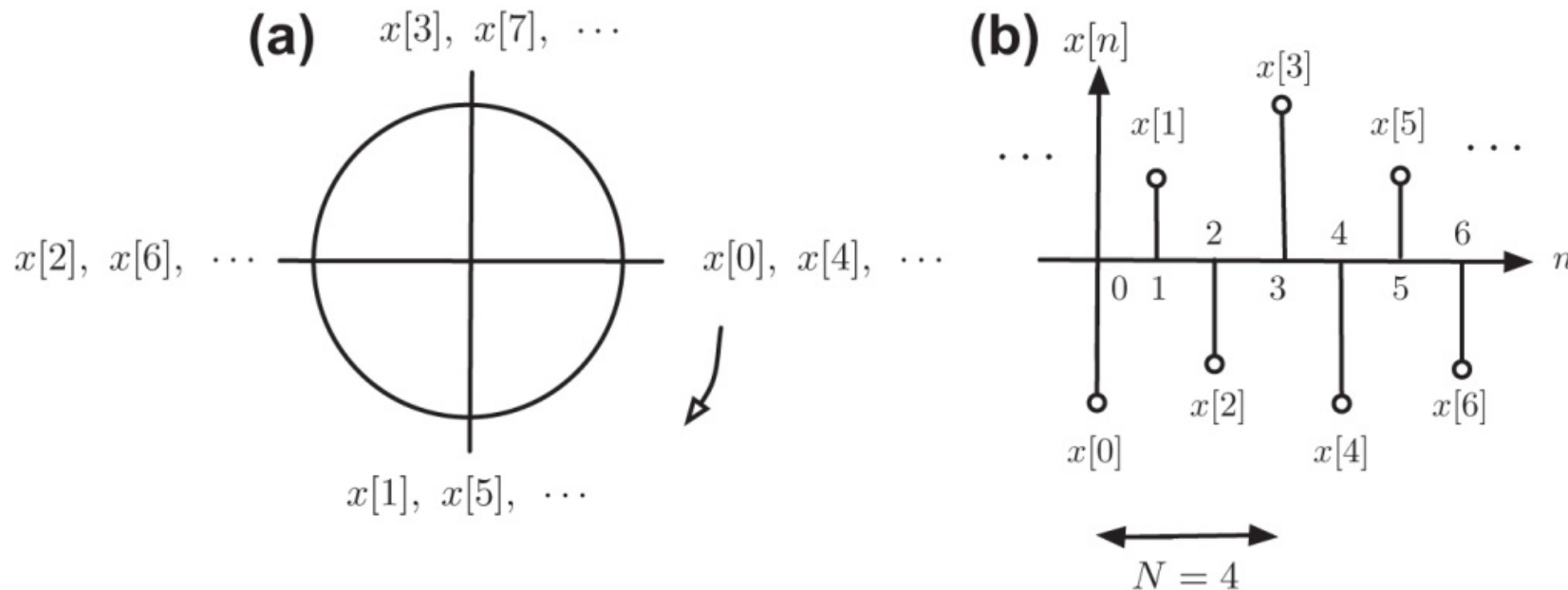
(i.e., one period of the periodic sequence $x[n]$)

$$Z\{x_1[n]\} = \sum_{n=0}^{N-1} x[n] z^{-n}$$

$$X[k] = \frac{1}{N} \sum_{n=0}^{N-1} x[n] e^{-j\frac{2\pi}{N}kn} = \frac{1}{N} Z\{x_1[n]\} \Big|_{z=e^{j\frac{2\pi}{N}k}}$$

Discrete-time Fourier series

- For periodic sequences, it is convenient to think of the sequence values as being on a circle



Discrete-time Fourier series

- Periodic convolution
 - For periodic sequence, convolution is a bit different
 - The product of two periodic sequences is also periodic
 - Periodic convolution:

$$v[n] = \sum_{m=0}^{N-1} x[m] y[n-m] \quad \Leftrightarrow \quad V[k] = NX[k]Y[k]$$

$$w[n] = x[n]y[n] \quad \Leftrightarrow \quad W[k] = \sum_{m=0}^{N-1} X[m]Y[n-m]$$

All are periodic with period N

Discrete-time Fourier series

Fourier Series of Discrete-time Periodic signals

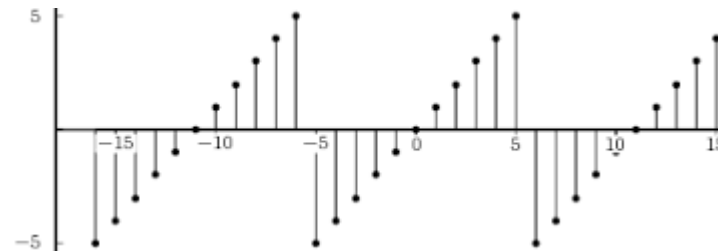
	$x[n]$ periodic signal of period N	$X[k]$ periodic FS coefficients of period N
Z-transform	$x_1[n] = x[n](u[n] - u[n - N])$	$X[k] = \frac{1}{N} \mathcal{Z}(x_1[n]) _{z=e^{j2\pi k/N}}$
DTFT	$x[n] = \sum_k X[k] e^{j2\pi nk/N}$	$X(e^{j\omega}) = \sum_k 2\pi X[k] \delta(\omega - 2\pi k/N)$
LTI response	input $x[n] = \sum_k X[k] e^{j2\pi nk/N}$	output: $y[n] = \sum_k X[k] H(e^{jk\omega_0}) e^{j2\pi nk/N}$ $H(e^{j\omega})$ (frequency response of system)
Time-shift (circular shift)	$x[n - M]$	$X[k] e^{-j2\pi kM/N}$
Modulation	$x[n] e^{j2\pi Mn/N}$	$X[k - M]$
Multiplication	$x[n] y[n]$	$\sum_{m=0}^{N-1} X[m] Y[k - m]$ periodic convolution
Periodic convolution	$\sum_{m=0}^{N-1} x[m] y[n - m]$	$NX[k] Y[n]$

Discrete Fourier Transform (DFT)

- The step from the Discrete Fourier Series to the Discrete Fourier Transform is a short one.
 - Consider a periodic sequence $x[n]$ (period N)
 - It has a Fourier series
 - Consider a finite length sequence $x[n]$, $0 \leq n \leq N-1$
 - One can think of making a periodic extension of this sequence and then take it's Fourier series.
 - This is essentially the Discrete Fourier transform
 - Except ... traditionally, the $1/N$ goes with the sum over the DFT coefficients.

Discrete Fourier Transform

- Discrete Fourier Transform (DFT)
 - Signals may not be periodic, but are generally finite in length
 - In practice, all signals are finite.
 - If you are working with really long signals, you can always break it up into shorter length sections.
 - Although signal is not periodic, you can create a periodic extension of the signal by repeating the signal before and after real signal.
 - Create a periodic signal
 - You can then find the Discrete Time Fourier Series of the periodic extension of the signal



Discrete Fourier Transform

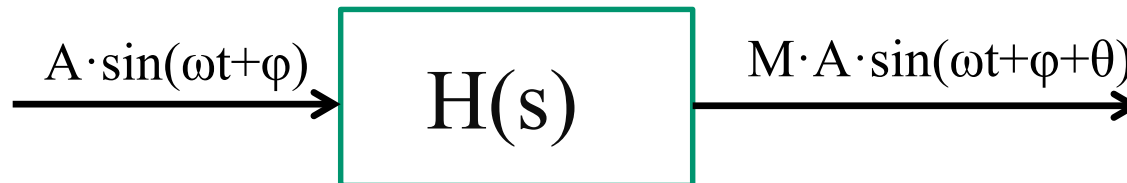
- Discrete Fourier Transform (DFT)
 - The DFT is usually written a little differently than the DTFS
 - For a finite length signal of length L , one often pads it out to a larger number of samples, N , that is L or greater:
 - The factor of $1/N$ is usually put with the “inverse” transform

$$\begin{aligned} X[k] &= \sum_{n=0}^{N-1} x[n] e^{\frac{-j2\pi nk}{N}} & 0 \leq k \leq N-1 \\ x[n] &= \frac{1}{N} \sum_{k=0}^{N-1} X[k] e^{\frac{j2\pi nk}{N}} & 0 \leq n \leq N-1 \end{aligned}$$

We will discuss the Discrete Fourier Transform in more detail later

Frequency-Domain Analysis of LTI Systems

- Key concept behind frequency decomposition of signals:
 - Basis functions of sines and cosines (and complex exponential)
 - **Frequency components of signal are unchanged when passed through Linear Time Invariant systems**
 - Only amplitude and phase change



Changes magnitude and phase

Frequency-Domain Analysis of LTI Systems

- Consider the Discrete-Time Fourier Transform of signals (Proakis & Manolakis refer to this just the Discrete Fourier transform in Chapter 5)
 - Frequency response completely characterizes LTI system
- We obtained the DTFT by taking the Fourier transform of sampled signal
- Previously, we considered the general expression for the DTFT for a signal, $x(n)$.

$$X(e^{j\omega}) = \sum_n x(n)e^{-j\omega n} \quad -\pi \leq \omega < \pi$$
$$x(n) = \frac{1}{2\pi} \int_{-\pi}^{\pi} X(e^{j\omega})e^{j\omega n} d\omega$$

Frequency-Domain Analysis of LTI Systems

- Now, we concentrate on the frequency response of an LTI system
 - The response of an LTI system to any input is:

$$y(n) = \sum_{k=-\infty}^{\infty} h(k)x(n-k)$$

- Considering a complex exponential input $x(n) = Ae^{j\omega n}$

$$y(n) = \sum_{k=-\infty}^{\infty} h(k)Ae^{j\omega(n-k)} = A \left[\sum_{k=-\infty}^{\infty} h(k)e^{-j\omega k} \right] e^{j\omega n}$$

$$H(\omega) = \sum_{k=-\infty}^{\infty} h(k)e^{-j\omega k}$$

$$y(n) = AH(\omega)e^{j\omega n}$$

Frequency-Domain Analysis of LTI Systems

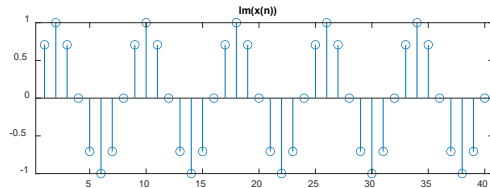
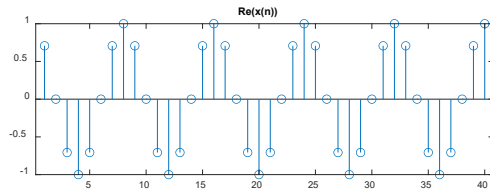
- This shows that complex exponentials are the eigenfunctions and $H(\omega)$ are the eigenvalues of an LTI system.
- Since any signal can be decomposed into complex exponentials, $H(\omega)$ completely characterizes the LTI system.
- Example of how the system modifies the amplitude and phase of a sinusoidal input but not the frequency:
 - Impulse response of system is

$$h(n) = \left(\frac{1}{2}\right)^n u(n)$$
$$H(\omega) = \sum_{n=-\infty}^{\infty} h(n)e^{-j\omega n} = \sum_{n=-\infty}^{\infty} \left(\frac{1}{2}\right)^n u(n)e^{-j\omega n} = \sum_{k=0}^{\infty} \left(\frac{1}{2}\right)^k e^{-j\omega k} = \sum_{k=0}^{\infty} \left(\frac{1}{2}e^{-j\omega}\right)^k = \frac{1}{1 - \frac{1}{2}e^{-j\omega}}$$

Frequency-Domain Analysis of LTI Systems

- What does the system do to an complex exponential input (i.e. and input at some particular frequency)
 - Consider an input with a frequency of $\pi/4$

$$x(n) = Ae^{j\omega n} = Ae^{jn\pi/4}$$



$$H(\omega) = \frac{1}{1 - \frac{1}{2}e^{-j\omega}}, \quad H(\pi/4) = \frac{1}{1 - \frac{1}{2}e^{-\frac{j\pi}{4}}}$$

$$|H(\pi/4)| = 1.3572, \quad \phi = -28.68^\circ$$

- Example in book shows:

$$|H(\pi/2)| = 0.8944 \quad \phi = -26.6^\circ$$

$$|H(\pi)| = 0.6667 \quad \phi = 0^\circ$$

Frequency-Domain Analysis of LTI Systems

- If an LTI system changes the magnitude and phase of an input
 - You can begin to see how filtering works
 - Consider what the LTI does to each frequency component

Example of a moving average filter:

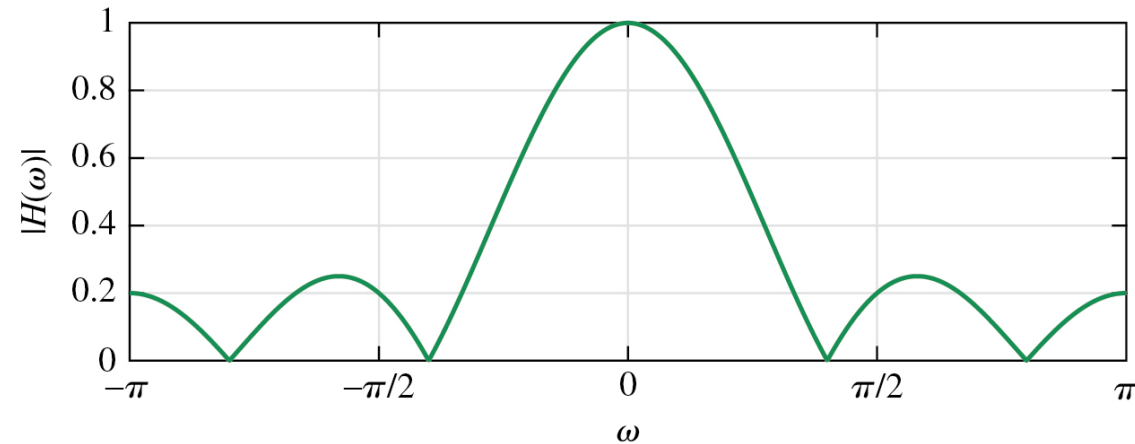
$$y(n) = \frac{1}{M+1} \sum_{k=1}^M x(n-k)$$

Frequency response is (using the finite geometric series sum)

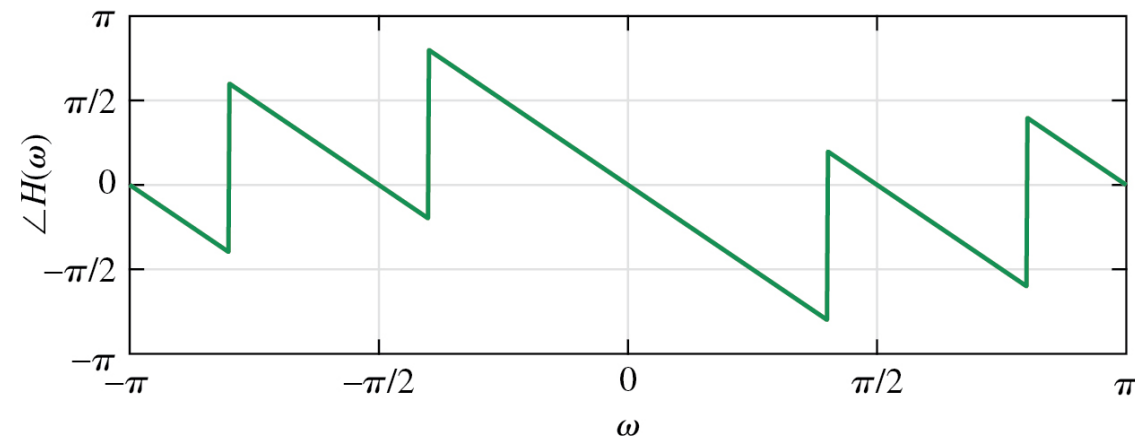
$$H(\omega) = \frac{1}{M+1} \sum_{k=0}^M e^{-j\omega k} = \frac{1}{M+1} \frac{1 - e^{-j\omega(M+1)}}{1 - e^{-j\omega}}$$

$$H(\omega) = \frac{1}{M+1} \frac{\sin(\omega(M+1/2))}{\sin(\omega/2)} e^{-j\omega/2}$$

Frequency-Domain Analysis of LTI Systems



This is a low-pass filter



M=4

Frequency-Domain Analysis of LTI Systems

- Example with Infinite impulse response

$$y(n) = ay(n-1) + bx(n)$$

We have found the impulse response for this system a few times:

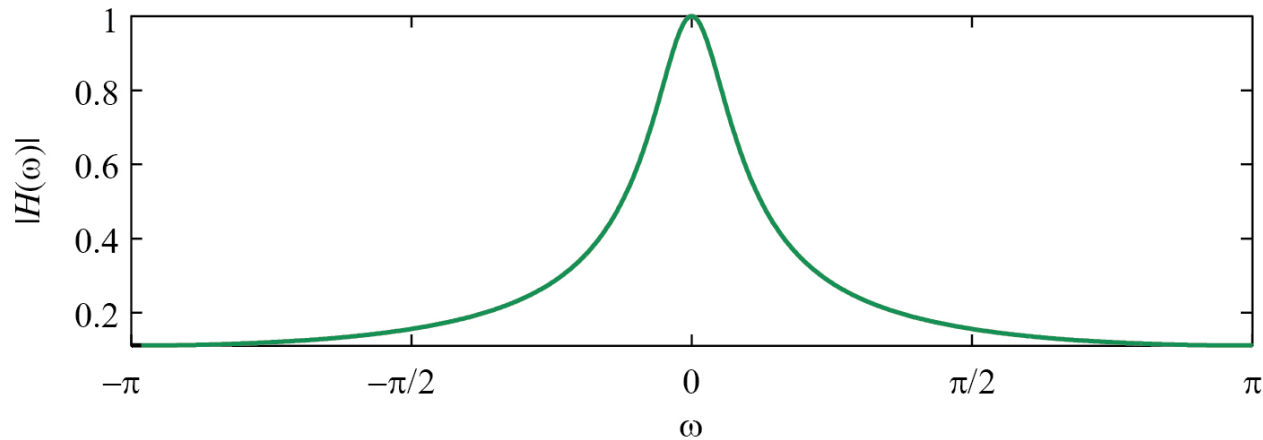
$$H(z) = \frac{b}{1 - az^{-1}}$$

$$H(\omega) = \frac{b}{1 - ae^{-j\omega}}$$

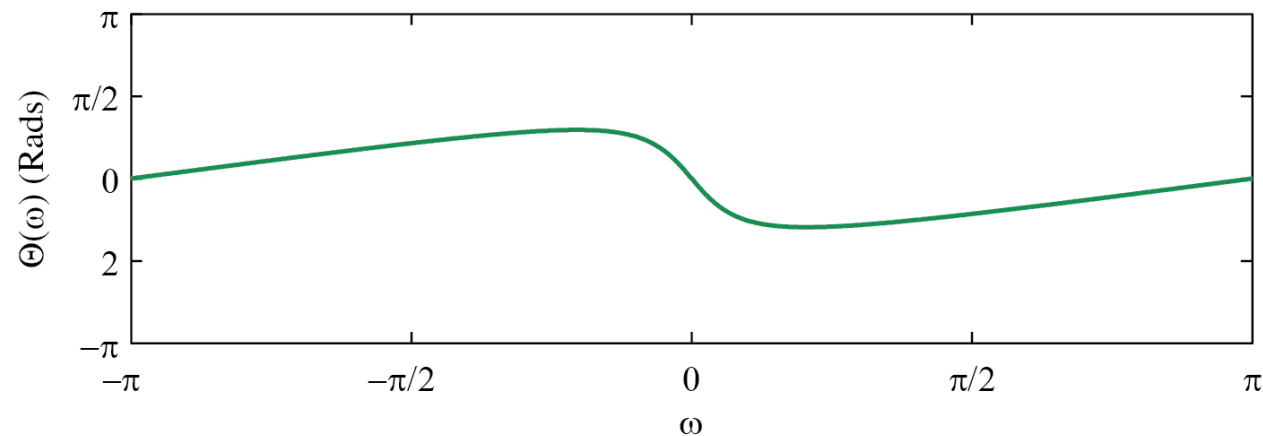
$$|H(\omega)| = \frac{1 - a}{\sqrt{1 - 2a \cos \omega + a^2}}$$

$$\phi = -\tan^{-1} \left(\frac{a \sin \omega}{1 - a \cos \omega} \right)$$

Frequency-Domain Analysis of LTI Systems



This is a also low-pass filter



$a=0.8$

Frequency-Domain Analysis of LTI Systems

- Transient and steady-state response of system
 - Example

$$y(n) = ay(n-1) + x(n), \quad y(-1)$$

$$y(0) = ay(-1) + x(0)$$

$$y(1) = a[ay(-1) + x(0)] + x(1)$$

$$y(2) = a[a[ay(-1) + x(0)] + x(1)] + x(2)$$

\vdots

$$y(n) = a^{n+1}y(-1) + \sum_{k=0}^n a^k x(n-k)$$

Frequency-Domain Analysis of LTI Systems

- Transient and steady-state response of system
 - If the input is a complex exponential: $x(n) = Ae^{j\omega n}$

$$y(n) = a^{n+1}y(-1) + \sum_{k=0}^n a^k Ae^{j\omega(n-k)} = a^{n+1}y(-1) + A \left[\sum_{k=0}^n a^k e^{-j\omega k} \right] e^{j\omega n}$$

$$y(n) = a^{n+1}y(-1) + A \left[\sum_{k=0}^n \left(ae^{-j\omega} \right)^k \right] e^{j\omega n}$$

Using sum of finite geometric series

$$y(n) = a^{n+1}y(-1) + A \left[\frac{1 - \left(ae^{-j\omega} \right)^{n+1}}{1 - ae^{-j\omega}} \right] e^{j\omega n} = a^{n+1}y(-1) - \frac{Aa^{n+1}e^{-j\omega(n+1)}}{1 - ae^{-j\omega}} e^{j\omega n} + \frac{A}{1 - ae^{-j\omega}} e^{j\omega n}$$

These die off as n increases

Steady-state

Frequency-Domain Analysis of LTI Systems

- Steady-state for periodic input
 - Discrete Fourier Series of input:

$$x(n) = \sum_{k=0}^{N-1} c_k e^{j2\pi k/N}$$

Output of each harmonic gets modified by: $H\left(\frac{2\pi k}{N}\right)$

$$y(n) = \sum_{k=0}^{N-1} c_k H\left(\frac{2\pi k}{N}\right) e^{j2\pi k/N}$$

so also periodic with modified Fourier Series coefficients

Frequency-Domain Analysis of LTI Systems

- Steady-state for aperiodic input
 - Use convolution to find output:

$$Y(\omega) = H(\omega)X(\omega)$$

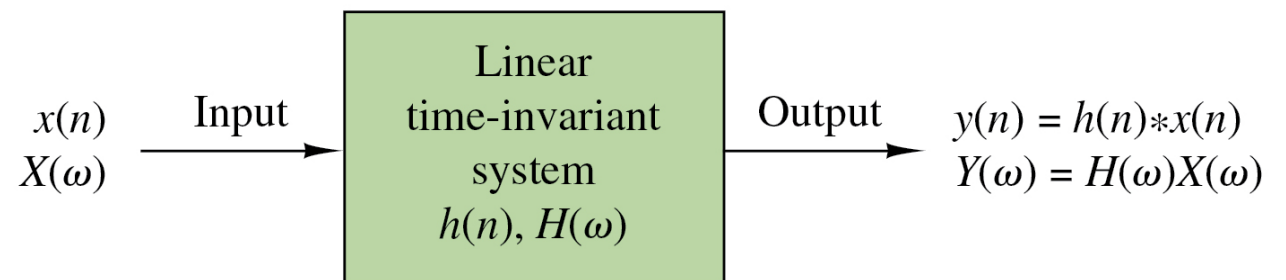
$$|Y(\omega)| = |H(\omega)| |X(\omega)|$$

$$\angle Y(\omega) = \angle H(\omega) + \angle X(\omega)$$

Energy Density:

$$|Y(\omega)|^2 = |H(\omega)|^2 |X(\omega)|^2$$

$$S_{yy} = |H(\omega)|^2 S_{xx}$$



Time for some applications rather than theory

- Two tasks presented which will be subject of the next lab
 1. Phone tones
 2. Speech recognition

Time for some applications rather than theory

- Phone tones
 - Can you figure out what numbers are being “dialed” by the tones they produce?
 - Tools to use:
 - Fourier transform to see if you can identify discrete frequency that can be associated with numbers on the “dial”
 - Segment tones for numbers “dialed” and try to map to numbers



Time for some applications rather than theory

- I have a set of *.mp3 files for numbers being dialed

(show some code)

```
phone_number = 'phone_number_2.mp3';
[phn,fs] = audioread(phone_number);
phone_part1 = split(phone_number, '.');
phone_call = split(phone_part1 {1}, '_');
callnum = phone_call {3};
phn1 = phn(:,1);
Normalize amplitudes to have maximum value of 1
phn1 = phn1/max(abs(phn1));
tm = (1/fs)*[1:length(phn1)];
figure(1)
plot(tm,phn1);
xlabel('Time (sec)')
ylabel('Magnitude')
title(['Phone Number ',callnum])
```

```
segment = phn1(startseg(k):endseg(k));
tmseg = (1/fs)*[1:length(segment)];
figure
plot(tmseg,segment)
nsamp = length(segment);
fnyquist = fs/2;
x_mag = abs(fft(segment))/nsamp;
bins = [0:nsamp-1];
freq_hz = bins*fs/nsamp;
% Plot only positive frequencies
n_2 = ceil(nsamp/2);
figure()
plot(freq_hz(1:n_2), x_mag(1:n_2))
xlabel('Frequency (Hz)')
ylabel('Magnitude');
title('Single-sided Magnitude spectrum (Hertz)');
axis([0,2000,0,0.2])
```


Time for some applications rather than theory

- Speech recognition
 - There are some very sophisticated methods of speech recognition, which actually seem to work some time.
 - We won't be using these.
 - Dataset with single words from google.
 - We will just try to distinguish two words, like “yes” and “no”
 - This will involve obtaining attributes in the frequency domain and using them in a classifier
 - A very neat tool in Matlab called classificationLearner
 - A good start for features is finding the power in some set of frequency bands
 - Need to normalize by total power