

### 6.003: Signal Processing

#### Frequency Response and Filtering

- Discrete-Time Frequency Response
- Continuous-Time Frequency Response

October 14, 2021

#### Representing Systems with Difference/Differential Equations

Discrete-time systems that can be described by linear **difference** equations with constant coefficients are linear and time-invariant.

$$x[n] \longrightarrow \boxed{\text{LTI}} \longrightarrow y[n]$$

$$\sum_l c_l y[n-l] = \sum_m d_m x[n-m]$$

Continuous-time systems that can be described by linear **differential** equations with constant coefficients are linear and time-invariant.

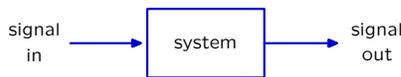
$$x(t) \longrightarrow \boxed{\text{LTI}} \longrightarrow y(t)$$

$$\sum_l c_l \frac{d^l y(t)}{dt^l} = \sum_m d_m \frac{d^m x(t)}{dt^m}$$

– natural and compact representations of many systems

#### Context: The System Abstraction

Describe a **system** (physical, mathematical, or computational) by the way it transforms an **input signal** into an **output signal**.



This abstraction is particularly powerful for **linear and time-invariant** systems, which are both **prevalent** and **mathematically tractable**.

Three important representations for LTI systems:

- **Difference/Differential Eq:** algebraic input/output **constraint** ✓
- **Convolution:** represent system by **unit-sample/impulse response** ✓
- **Filter:** represent a system by its frequency response

#### Unit-Sample Response and Impulse Response

Discrete-time systems that are linear and time-invariant can be completely specified by their response to a **unit-sample signal**.



$$\text{If } \delta[n] \rightarrow h[n], \text{ then } x[n] \rightarrow y[n] = (x * h)[n] = \sum_m x[m]h[n-m]$$

Continuous-time systems that are linear and time-invariant are completely specified by their response to a unit **impulse** function.



$$\text{If } \delta(t) \rightarrow h(t), \text{ then } x(t) \rightarrow y(t) = (x * h)(t) = \int x(\tau)h(t - \tau)d\tau$$

– an LTI system is completely characterized by a **single signal**

#### Unit-Sample Response

The unit-sample response is a **complete** description of a system.



This is a bit surprising since  $\delta[n]$  is such a **simple signal**.  
The unit-sample signal is the **shortest** possible non-trivial DT signal!

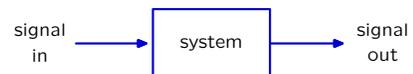


The response to this simple signal determines the response to **any** other input signal:

$$x[n] \rightarrow y[n] = (x * h)[n] = \sum_m x[m]h[n-m]$$

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- **Filter:** represent a system by its **frequency response**

**Frequency Response**

The **frequency response** is a third way to characterize a linear time-invariant system. This characterization is based on responses to **sinusoids**.



The idea is to characterize a system by the way  $A$  and  $\phi$  vary with  $\Omega$ .

Sinusoids differ from the unit-sample signal in important ways:

- **eternal** (longest possible signals) versus **transient** (shortest possible)
- comprises a **single** frequency versus a **sum** of all possible frequencies

**Frequency Response**

Using complex exponentials to characterize the frequency response.



Notice that the complex valued  $A$  can represent both amplitude and phase. We can find  $A$  using convolution.

$$y[n] = (x * h)[n] = \sum_{m=-\infty}^{\infty} x[n-m]h[m] = \sum_{m=-\infty}^{\infty} e^{j\Omega(n-m)}h[m]$$

$$= e^{j\Omega n} \sum_{m=-\infty}^{\infty} h[m]e^{-j\Omega m} = H(\Omega) e^{j\Omega n}$$

The response to a complex exponential is a complex exponential with the **same frequency** but possibly **different amplitude and phase**.

The map for how a system modifies the amplitude and phase of a complex exponential input is the **Fourier transform of the unit-sample response**.

**Frequency Response**

The frequency response is a **complete** characterization of an LTI system.

1. One can always find the frequency response of a system.



2. Scaling the input by a constant scales the output by the same constant.



3. Linearity implies that the response to a sum is the sum of the responses.

$$x[n] = \frac{1}{2\pi} \int_{-\pi}^{\pi} X(\Omega)e^{j\Omega n} d\Omega \rightarrow \text{LTI} \rightarrow y[n] = \frac{1}{2\pi} \int_{-\pi}^{\pi} X(\Omega)H(\Omega)e^{j\Omega n} d\Omega$$

4. The Fourier transform of the output is  $X(\Omega)H(\Omega)$ .



The transform of the output is  $H(\Omega)$  times the transform of the input.

**Frequency Response**

The frequency response is a **complete** description of a system.



This is a bit surprising since  $e^{j\Omega n}$  contains a **single** frequency.

→ can find the output of a system by breaking the input into its constituent frequencies and summing the responses to each frequency – one-at-a-time.



The frequency response can be used to find response to **any** input signal:

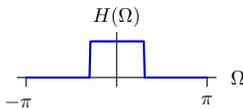
$$X(\Omega) \rightarrow Y(\Omega) = H(\Omega)X(\Omega)$$

**Frequency Response**

The frequency response can be an **insightful** description of a system.

Example:

A low-pass filter passes frequencies near 0 and rejects those near  $\pi$ .



Very natural way to describe audio components:

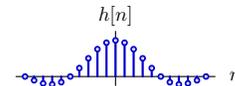
- microphones
- loudspeakers
- audio equalizers

**Unit-Sample Response and Frequency Response**

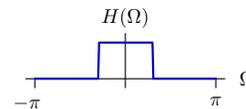
**Two complete** representations for linear, time-invariant systems.



**Unit-Sample Response:** responses across time for a unit-sample input.



**Frequency Response:** responses across frequencies for sinusoidal inputs.



The **frequency response** is Fourier transform of **unit-sample response**!

**Example**

Find the frequency response of a system described by the following:

$$y[n] - \alpha y[n-1] = x[n]$$

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**Method 1:**

Find the unit-sample response and take its Fourier transform.

$$h[n] - \alpha h[n-1] = \delta[n]$$

Solve the difference equation for  $h[n]$ .

$$h[n] = \delta[n] + \alpha h[n-1]$$

First order  $\rightarrow$  need one initial condition:  $h[-1] = 0$

$$h[0] = \delta[0] + \alpha h[-1] = 1$$

$$h[1] = \delta[1] + \alpha h[0] = \alpha$$

$$h[2] = \delta[2] + \alpha h[1] = \alpha^2$$

$$h[3] = \delta[3] + \alpha h[2] = \alpha^3$$

$$h[n] = \alpha^n u[n]$$

$$H(\Omega) = \sum_{n=-\infty}^{\infty} h[n]e^{-j\Omega n} = \sum_{n=0}^{\infty} \alpha^n e^{-j\Omega n} = \sum_{n=0}^{\infty} (\alpha e^{-j\Omega})^n = \frac{1}{1 - \alpha e^{-j\Omega}}$$

**Example**

Find the frequency response of a system described by the following:

$$y[n] - \alpha y[n-1] = x[n]$$

**Method 2:**

Find the response to  $e^{j\Omega n}$  directly.

$$x[n] = e^{j\Omega n}$$

Because the system is linear and time-invariant, the output will have the same frequency as the input, but possibly different amplitude and phase.

$$y[n] = H(\Omega)e^{j\Omega n}$$

$$y[n-1] = H(\Omega)e^{j\Omega(n-1)} = H(\Omega)e^{-j\Omega}e^{j\Omega n}$$

Substitute into the difference equation.

$$H(\Omega)e^{j\Omega n} - \alpha H(\Omega)e^{-j\Omega}e^{j\Omega n} = H(\Omega)(1 - \alpha e^{-j\Omega})e^{j\Omega n} = e^{j\Omega n}$$

Since  $e^{j\Omega n}$  is never 0, we can divide it out.

$$H(\Omega) = \frac{1}{1 - \alpha e^{-j\Omega}}$$

Same answer as method 1.

**Example**

Find the frequency response of a system described by the following:

$$y[n] - \alpha y[n-1] = x[n]$$

**Method 3:**

Take the Fourier transform of the difference equation.

$$Y(\Omega) - \alpha e^{-j\Omega}Y(\Omega) = X(\Omega)$$

Solve for  $Y(\Omega)$ .

$$Y(\Omega) = \frac{1}{1 - \alpha e^{-j\Omega}}X(\Omega)$$

Since  $Y(\Omega) = H(\Omega)X(\Omega)$ ,

$$H(\Omega) = \frac{1}{1 - \alpha e^{-j\Omega}}$$

Same answer as methods 1 and 2.

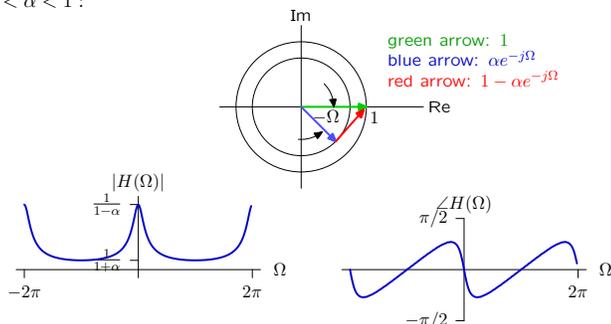
**Example**

Plot the frequency response.

$$H(\Omega) = \frac{1}{1 - \alpha e^{-j\Omega}}$$

Note that denominator is the difference of 2 complex numbers.

If  $0 < \alpha < 1$ :



Amplifies at low frequencies and attenuates high frequencies. Adds delay.

**The System Abstraction**

The system abstraction applies equally well for continuous-time signals.

**The System Abstraction**

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This abstraction is particularly powerful for **linear and time-invariant** systems, which are both **prevalent** and **mathematically tractable**.

Three important representations for LTI systems:

- **Differential Equation:** algebraic **constraint** on **derivatives** ✓
- **Convolution:** represent a system by its **impulse response** ✓
- **Filter:** represent a system by its **frequency response**

**Representing Systems with Difference/Differential Equations**

**Discrete-time** systems that can be described by linear **difference** equations with constant coefficients are linear and time-invariant.

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**Unit-Sample Response and Impulse Response**

**Discrete-time** systems that are linear and time-invariant can be completely specified by their response to a **unit-sample signal**.



If  $\delta[n] \rightarrow h[n]$ , then  $x[n] \rightarrow y[n] = (x * h)[n] = \sum_m x[m]h[n-m]$

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If  $\delta(t) \rightarrow h(t)$ , then  $x(t) \rightarrow y(t) = (x * h)(t) = \int x(\tau)h(t-\tau)d\tau$

– an LTI system is completely characterized by a single signal

**Impulse Response**

The impulse response is a **complete** description of a system.



This is a bit surprising since  $\delta(t)$  is zero almost everywhere. The impulse function is the **shortest** possible non-trivial CT signal!



The response to this signal determines the response to **any** other input.

$$x(t) \rightarrow y(t) = (x * h)(t) = \int x(\tau)h(t-\tau)d\tau$$

**Frequency Response**

The **frequency response** is a third way to characterize a linear time-invariant system. This characterization is based on responses to **sinusoids**.



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Using complex exponentials to characterize the frequency response.



Notice that the complex valued  $A$  can represent both amplitude and phase. We can find  $A$  using convolution.

$$y(t) = (x * h)(t) = \int_{-\infty}^{\infty} x(t-\tau)h(\tau)d\tau = \int_{-\infty}^{\infty} e^{j\omega(t-\tau)}h(\tau)d\tau$$

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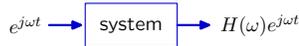
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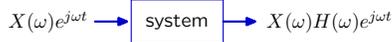
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$$\frac{1}{2\pi} \int_{-\infty}^{\infty} X(\omega)e^{j\omega t} d\omega \rightarrow \text{system} \rightarrow \frac{1}{2\pi} \int_{-\infty}^{\infty} X(\omega)H(\omega)e^{j\omega t} d\omega$$

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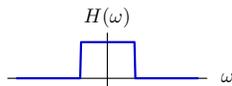
$$X(\omega) \rightarrow Y(\omega) = H(\omega)X(\omega)$$

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Very natural way to describe audio enhancements:

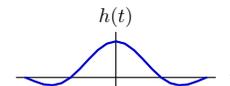
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**System Abstraction**

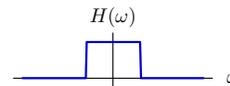
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**Example**

Find the frequency response of a system described by the following:

$$y(t) + \alpha \frac{dy(t)}{dt} = x(t)$$

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**Method 1:**

Find the response to  $e^{j\omega t}$  directly.

$$x(t) = e^{j\omega t}$$

Because the system is linear and time-invariant, the output will have the same frequency as the input, but possibly different amplitude and phase.

$$y(t) = H(\omega)e^{j\omega t}$$

$$\frac{dy(t)}{dt} = j\omega H(\omega)e^{j\omega t}$$

Substitute into the differential equation.

$$H(\omega)e^{j\omega t} + j\omega\alpha H(\omega)e^{j\omega t} = (1 + j\omega\alpha)H(\omega)e^{j\omega t} = e^{j\omega t}$$

Since  $e^{j\omega t}$  is never 0, we can divide it out.

$$H(\omega) = \frac{1}{1 + j\omega\alpha}$$

**Example**

Find the frequency response of a system described by the following:

$$y(t) + \alpha \frac{dy(t)}{dt} = x(t)$$

**Method 2:**

Take the Fourier transform of the differential equation.

$$Y(\omega) + j\omega\alpha Y(\omega) = X(\omega)$$

Solve for  $Y(\omega)$ .

$$Y(\omega) = \frac{1}{1 + j\omega\alpha} X(\omega)$$

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

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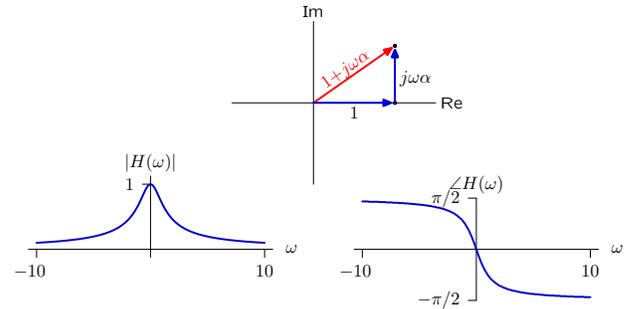
Same answer as method 1.

**Example**

Plot the frequency response.

$$H(\omega) = \frac{1}{1 + j\omega\alpha}$$

Note that denominator is sum of 2 complex numbers.



Amplifies low frequencies, attenuates high frequencies, adds phase delay.

**Check Yourself**

Find the frequency response of a rectangular box filter:

$$y(t) = \frac{1}{2} \int_{t-1}^{t+1} x(\tau) d\tau$$

(This CT filter is analogous to the three-point averager in DT.)

**Summary**

The Fourier transform of the response of a DT LTI system is the product of the Fourier transform of the input times the system's **frequency response**.

$$X(\Omega) \rightarrow \boxed{\text{LTI}} \rightarrow Y(\Omega)$$

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

The frequency response  $H(\Omega)$  is the Fourier transform of the unit-sample response  $h[n]$ .

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$$X(\omega) \rightarrow \boxed{\text{LTI}} \rightarrow Y(\omega)$$

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

The frequency response  $H(\omega)$  is the Fourier transform of the impulse response  $h(t)$ .