Quiz, 15 questions

1 point

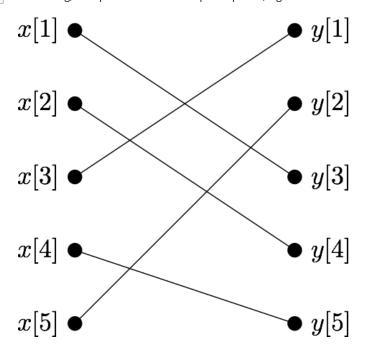
1.

(Difficulty: *) Among the choices below, select all the **linear** systems. (Please note that some of the choices use functions rather than discrete-time signals; the concept of linearity is identical in both cases).

AM radio modulation, i.e. multiply a signal x[n] by a cosine at the carrier frequency :

$$y[n] = x[n]\cos(2\pi\omega_c n)$$

Scrambling, i.e. a permutation to the input sequence, e.g.:



Clipping, i.e. enforce a maximum signal amplitude M,e.g.:

$$y[n] = \left\{ egin{array}{ll} x[n] & , \ x[n] \leq M \ M & , \ ext{otherwise} \end{array}
ight.$$

The DTFT, i.e. transform a sequence \mathbf{x} into $\mathrm{DTFT}\{\mathbf{x}\}$.

Time-stretch, i.e. $y(t)=x(\alpha t)$, e.g. if you play an old LP-45 vinyl disc at 33 rpm, the time-stretch coefficient would be $\alpha=33/45$.

Second derivative, i.e.

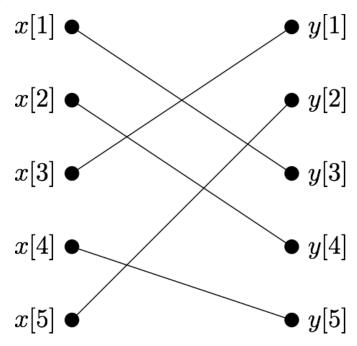
$$y(t)=rac{d^2}{dt^2}x(t)$$

Envelope detection (via squaring), i.e. $y[n] = |x[n]|^2 * h[n]$, where h[n] is the impulse response of a *lowpass* filter such as the moving average filter.

Quiz, 15 questions

(Difficulty: \star) Among the choices below, select all the **time-invariant** systems. (Please note that some of the choices use functions rather than discrete-time signals; the concept of time invariance is identical in both cases).

Scrambling : apply a permutation to the input sequence, e.g. :



Clipping, i.e. enforce a maximum signal amplitude M,e.g.:

$$y[n] = \left\{ egin{array}{ll} x[n] & , \ x[n] \leq M \ M & , \ {
m otherwise} \end{array}
ight.$$

AM radio modulation, i.e. multiply a signal x[n] by a cosine at the carrier frequency :

 $y[n] = x[n]\cos(2\pi\omega_c n)$

- Envelope detection (via squaring), i.e. $y[n] = |x[n]|^2 * h[n]$, where h[n] is the impulse response of a *lowpass filter* such as the moving average filter.
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 $y(t)=rac{d^2}{dt^2}x(t)$

The DTFT, i.e. transform a sequence ${f x}$ into ${
m DTFT}\{{f x}\}.$

1 point

(Difficulty: *) The impulse response of a room can be recorded by producing a sharp noise (impulsive sound source) in a silent room, thereby capturing the scattering of the sound produced by the walls.

\leftarrow

Homework for Module 4 Part 1

The impulse response h[n] of <u>Lausanne Cathedral</u> was measured by *Dokmanic et al.* by recording the sound of balloons being popped (hear it!).

The balloon is popped at time n=0 and after a number of samples N, the reverberations die out, i.e. h[n]=0 for n<0 or n>N.

The acoustic of this large space can then be artificially recreated by convolving any audio recording with the impulse response, e.g. this <u>cello recording</u> becomes <u>this</u>.

What are the properties of h[n] ? (tick all the correct answers)

BIBO stable

IIR

Causal

1 point

4

(Difficulty: ★) Let

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

$$x[n] = \left\{ egin{array}{ll} 1 & , \ n \geq 0, \ 0 & , \ \mathrm{else}. \end{array}
ight. ext{ }$$

$$y[n] = x[n] * h[n].$$

Compute y[-1], y[0], y[1], y[2] and write the result as space-separated values. E.g.: If you find y[-1]=-2, y[0]=-1, y[1]=0, y[2]=1, you should enter

1 -2 -1 0 1

Enter answer here

1 point

(Difficulty: **) Consider the filter $h[n] = \delta[n] - \delta[n-1]$,

$\leftarrow \quad \begin{array}{ll} \text{Homework} & \text{for}_n \text{Module, 4 Part 1} \\ & \text{the signal } x[n] = \left\{ \begin{array}{ll} 0 & \text{, else.} \end{array} \right. \end{array}$

and the output y[n] = x[n] * h[n].

Compute y[-1], y[0], y[1], y[2] and write the result as space-separated values. E.g.: If you find y[-1]=-2, y[0]=-1, y[1]=0, y[2]=1, you should enter

1 -2 -1 0 1

Enter answer here

1 point

6

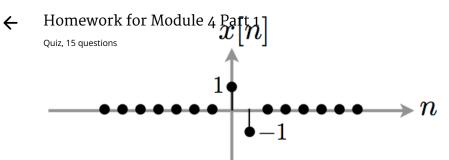
(Difficulty: $\star \star \star$) Which of the following filters are BIBO-stable?

Assume $N \in \mathbb{N}$ and $0 < \omega_c < \pi$.

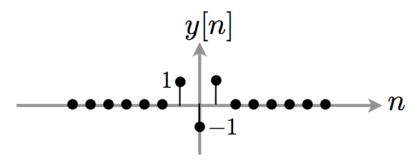
- The moving average: $h[n] = rac{\delta[n] + \delta[n-1]}{2}$.
- The filter $h[n] = \sum_{k=0}^{N-1} \delta[n-k] \sin\left(2\pi rac{k}{N}
 ight)$.
- The following smoothing filter: $h[n] = \sum_{k=0}^{\infty} \frac{1}{k+1} \frac{\delta[n-k]}{\delta[n-k]}$.
- The ideal low pass filter with a cutoff frequency ω_c : $H(e^{j\omega})=egin{cases}1&&|\omega|\leq\omega_c\\0&& ext{otherwise}\end{cases}$

1 point

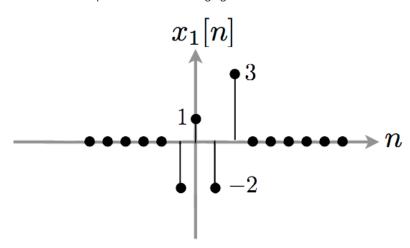
(Difficulty: **) Consider an LTI system ${\cal H}$. When the input to ${\cal H}$ is the following signal



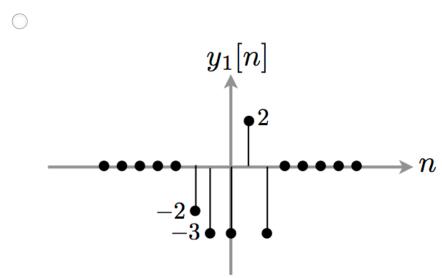
then the output is

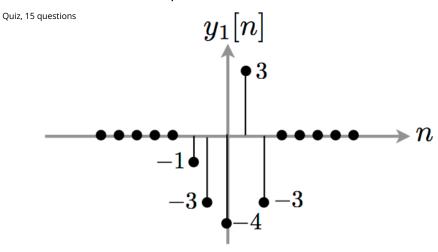


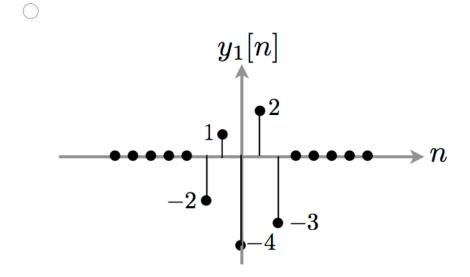
Assume now the input to ${\cal H}$ is the following signal



Which one of the following signals is the system's output?





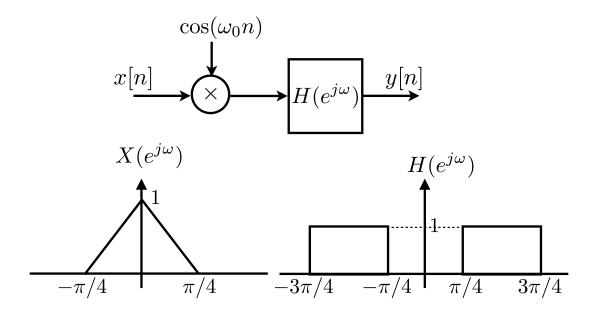


1 point

(Difficulty: \star) Consider the system shown below, consisting of a cosine modulator at frequency ω_0 followed by an ideal bandpass filter h[n] whose frequency response is also shown in the figure; assume that the input to the system is the significant part.1

 \leftarrow

Quiz, 15 questions



Determine the value of $\omega_0 \in [0,2\pi]$ that maximizes the energy of the output y[n] when the input is x[n].

Remember that $\boldsymbol{\pi}$ must be entered in the answer box as pi.

Preview

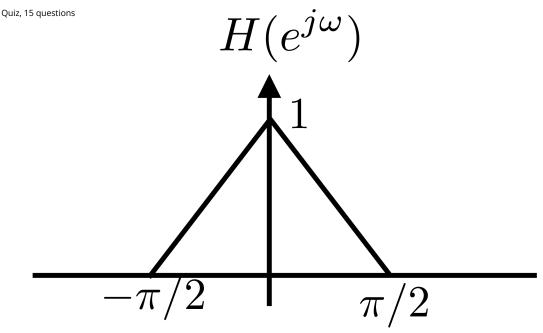
Enter math expression here

1 point

 \leftarrow

(Difficulty: ★) Consider a lowpass filter with the following frequency response.

Homework for Module 4 Part 1



What is the output y[n] when the input to this filter is $x[n] = \cos(\frac{\pi}{5}n) + \sin(\frac{\pi}{4}n) + 0.5\cos(\frac{3\pi}{4}n)$?

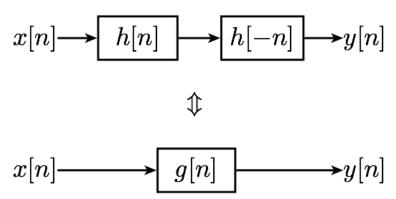
Preview

Enter math expression here

1 point

10.

(Difficulty: \star) Consider a filter with real-valued impulse response h[n]. The filter is cascaded with another filter whose impulse response is h'[n] = h[-n], i.e. whose impulse response is the time-reversed version of h[n]:



The cascade system can be seen as a single filter with impulse response g[n].

What is the phase of $G(ej\omega)$?

Preview

Quiz, 15 questions

1 point

11.

(Difficulty: \star) Let $x[n] = \cos(\frac{\pi}{2}n)$ and $h[n] = \frac{1}{5}\mathrm{sinc}(\frac{n}{5})$. Compute the convolution y[n] = x[n] * h[n], and write the value of y[5].

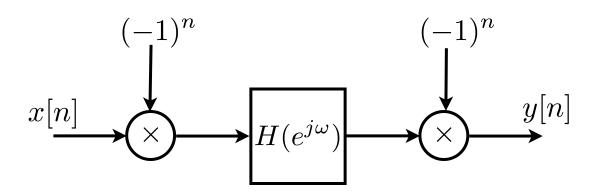
Hint: First find the convolution result in the frequency domain.

Enter answer here

1 point

12.

(Difficulty: **) Consider the system below, where $H(e^{j\omega})$ is an ideal lowpass filter with cutoff frequency $\omega_c=\pi/4$:



Consider two input signals to the system:

- $x_1[n]$ is bandlimited to $[-\pi/4,\pi/4]$
- $x_2[n]$ is band-limited to $[-\pi, -3\pi/4] \cup [3\pi/4, \pi]$.

Which of the following statements is correct?

- Both $x_1[n]$ and $x_2[n]$ are eliminated by the system.
- $x_2[n]$ is not modified by the system while $x_1[n]$ is eliminated.
- $x_1[n]$ is not modified by the system while $x_2[n]$ is eliminated.
- Both $x_1[n]$ and $x_2[n]$ are not modified by the system.

1 point

(Difficulty: \star) x[n] and y[n] are two square-summable signals in $\ell_2(\mathbb{Z})$; $X(e^{j\omega})$ and $Y(e^{j\omega})$ are their corresponding DTETs.

← Homework for Module 4 Part 1

We want to compute the value.

$$\sum_{n=-\infty}^{\infty} x[n]y^*[n].$$

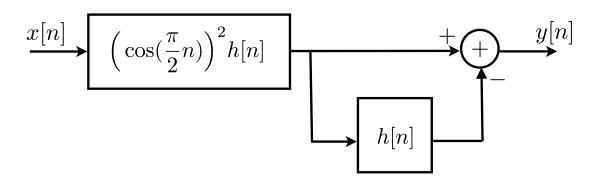
in terms of $X(e^{j\omega})$ and $Y(e^{j\omega})$. Select the correct expression among the choices below.

- $\int rac{1}{2\pi} \int_{-\pi}^{\pi} X(e^{j\omega}) Y(e^{-j\omega}) d\omega$
- $\frac{1}{2\pi}X(e^{j\omega})Y^*(e^{j\omega})$
- $X(e^{j\omega})Y(e^{-j\omega})$
- $\frac{1}{2\pi}X(e^{j\omega})Y(e^{-j\omega})$
- $X(e^{j\omega}) * Y(e^{-j\omega})$
- $\frac{1}{2\pi}\int_{-\pi}^{\pi}X(e^{j\omega})Y^*(e^{j\omega})d\omega$

1 point

14.

(Difficulty: $\star\star\star$) h[n] is the impulse response of an ideal lowpass filter with cutoff frequency $\omega_c<\frac{\pi}{2}$. Select the correct description for the system represented in the following figure?



Hint: Use the trigonometric identity $\cos(x)^2 = \frac{1}{2}(1 + \cos(2x))$

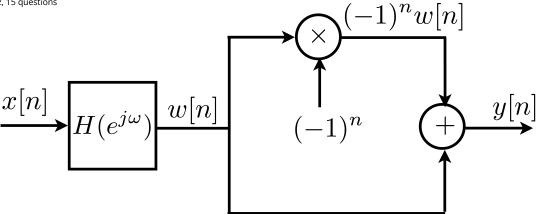
- A highpass filter with gain 1 and pass band $[\omega_c,\pi-\omega_c]$.
- A lowpass filter with gain 1 and cutoff frequency $\omega_c/2$.
- A highpass filter with gain $\frac{1}{4}$ and cutoff frequency ω_c .
- A lowpass filter with gain 1 and cutoff frequency ω_c .
- A highpass filter with gain $\frac{1}{2}$ and cutoff frequency $\pi \omega_c$.
- A lowpass filter with gain $\frac{1}{2}$ and cutoff frequency $2\omega_c$.

1 point

(Difficulty: $\star\star\star$) Consider the following system, where $H(e^{j\omega})$ is a half-band filter, i.e. an ideal lowpass with cutoff

frequency $\omega_c=\pi/2$: Homework for Module 4 Part 1 \leftarrow

Quiz, 15 questions



Assume the input to the system is $x[n] = \delta[n]$. Compute

$$\sum_{n=-\infty}^{\infty}y[n]$$

• Hint: Perform the derivations in the frequency domain.

Enter answer here

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