

PROBLEM #1 (15 Points) PART1- Conceptual Questions

Indicate whether the following statements are **true or false**. If true, give a brief explanation. If false, give a simple counter-example or a clear reason.

1) All periodic continuous-time signals remain periodic after sampling.

2) An all-pass filter can be FIR.

3) Increasing the window length can lower stopband ripples.

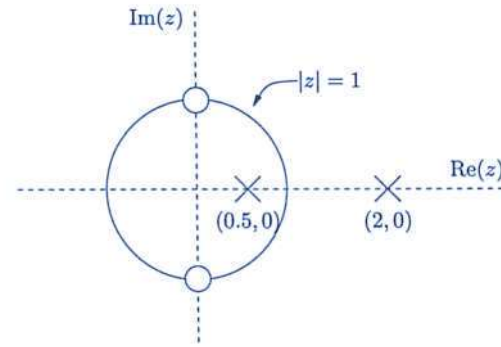
4) Bilinear transformation of analog IIR filter to digital IIR filter can preserve linear phase.

5) A modulator $y(n) = (-1)^n x(n)$ is a linear and time-invariant system.

PROBLEM #1 (30 Points) PART2- Filter Analysis

The pole-zero plot for $H(z)$ for a DT-LTI system is shown below.

- 1) Write the expression for the transfer function that is associated with the shown pole-zero plot. You can use K for a scaling constant.
- 2) Can the system be stable? If yes, what kind of a signal would $h(n)$ be *right sided, left sided, double sided*?
- 3) Can this system be both stable and causal? Explain your answer?
- 4) Write an expression for $H(\omega)$, and *magnitude and phase of $H(\omega)$* .
Use $K = 1$
- 5) Plot magnitude of $H(\omega)$ (not bode plot), i.e. $|H(\omega)|$ *versus frequency ω* .
- 6) What kind of a filter can this be and why?



Use the specifications of the following ideal filter to design an FIR Filter, answer the following questions:

$$H_d(\omega) = \begin{cases} 0 & |\omega| < \pi/4 \\ 1 & \pi/4 \leq |\omega| \leq \pi/3 \\ 0 & \pi/3 \leq |\omega| \leq \pi \end{cases}$$

And using,

$$W_v(n) = \begin{cases} 0.5 + 0.5 \cos\left(\frac{\pi n}{9}\right) & -9 \leq n \leq 9 \\ 0 & \text{otherwise} \end{cases}$$

- 1) Sketch the frequency response for the desired filter.
- 2) Determine the impulse response this FIR filter which approximates this frequency response.
- 3) Compute by hand, show all your work, the first 4 coefficients of the impulse response $h(n)$ coefficients.
- 4) Use MATLAB to plot both $h(n)$, and the frequency response magnitude and phase of the filter you designed.
- 5) Comment on the expected performance of the filter you designed.

PROBLEM #3 Filters and frequency response

Consider the following filter $H(z)$

$$H(\omega) = \frac{0.75(1 + z^{-2})}{1 + 0.5 z^{-2}}$$

- 1) Give the expression for its frequency response
- 2) Provide a plot of the magnitude of the frequency response (i.e $|H(\omega)|$ versus frequency ω).
- 3) Determine the exact form of the steady state output (do not use z-transform) when the input is

$$h(n) = 2u(n) + 3 \sin\left(\frac{\pi n}{2}\right) + 4 \cos\left(\frac{\pi n}{2}\right) u(n) + 5 \cos(\pi n) u(n)$$

- 4) Based on the filter's frequency response, explain the relation between $y(n)$ and $x(n)$ (that is why some sinusoids from $x(n)$ passed while others were filtered out).