

Quiz Number: 3	Course Title: Digital Signal and Image Processing	Course Code: ELEC 421	Date: Thursday, November 21, 2024	Duration of Quiz: 15 minutes	Number of Questions: 5	Instructor: Siamak Najarian, Ph.D., P.Eng.	University: UBC	Department: Electrical and Computer Engineering
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Please carefully read the following instructions and guidelines:

1. *This quiz is closed books/notes.*
2. *You will not get a negative mark for choosing the incorrect answer.*
3. *Each question carries 1 mark.*

Best of Luck!

Question 1:

In digital signal processing, a bandlimited signal with a maximum frequency ω_B can be perfectly reconstructed from its samples if the sampling frequency ω_S satisfies which of the following conditions?

- A. $\omega_S < 2\omega_B$
 - B. $\omega_S = 2\omega_B$
 - C. $\omega_S \geq 2\omega_B$
 - D. $\omega_S > 2\omega_B$
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Question 2:

To mitigate the aliasing effect in signal sampling, what technique can be used to prevent high-frequency components from distorting the sampled signal?

- A. Apply a highpass filter to remove low-frequency components before sampling, ensuring the signal conforms to the Nyquist Rate (NR).
 - B. Apply a bandpass filter to ensure the signal remains within a specific frequency range.
 - C. Apply a lowpass filter to the signal before sampling to remove high-frequency components, ensuring the signal conforms to the Nyquist Rate (NR).
 - D. Apply a frequency modulation technique to shift the signal frequencies before sampling.
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Question 3:

In the context of zero-order hold reconstruction, which two functions should be convolved with each other to generate the reconstructed signal?

- A. The original continuous signal and a unit impulse function
- B. The discrete-time delta functions and a rectangular pulse function
- C. The sampled signal and a triangular pulse function
- D. The discrete-time delta functions and a Gaussian pulse function

Question 4:

In digital signal processing, one key difference between IIR and FIR low-pass filters is related to the filter's impulse response and phase characteristics. Which of the following statements is true about IIR low-pass filters?

- A. IIR filters have a finite impulse response and typically exhibit linear phase characteristics.
- B. IIR filters have an infinite impulse response and can achieve a steeper roll-off with fewer coefficients than FIR filters.
- C. IIR filters require more coefficients than FIR filters to achieve the same roll-off.
- D. IIR filters are always stable and have a linear phase response.

Question 5:

Which of the following is true regarding a Finite Impulse Response (FIR) low-pass filter in Digital Signal Processing (DSP)?

- A. FIR filters have a finite impulse response, are always stable, have a linear phase response, and require more coefficients for similar performance compared to IIR filters.
 - B. FIR filters always have a non-linear phase response, which distorts the order of frequencies in the signal.
 - C. FIR filters are always unstable, but they are easier to design for specific phase requirements.
 - D. FIR filters can achieve a steeper roll-off for a given filter order compared to IIR filters, but they are less computationally efficient.
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End of Questions

Answer Sheet:

Question Number	A	B	C	D
1				
2				
3				
4				
5				