## Program: BE Electronics & Telecommunication Engineering

## Curriculum Scheme: Revised 2016

Examination: Third Year Semester V

## Course Code: ECC504 and Course Name: Discrete Time Signal Processing

Time: 1 hour

Max. Marks: 50

Note to the students:- All the Questions are compulsory and carry equal marks . Q1. What is the value of complex addition required to compute DFT for N=8? Option A: 56 Option B: 64 Option C: 32 Option D: 16 For a decimation-in-frequency FFT algorithm, which of the following is true? Q2. Option A: Both input and output are in order Option B: Both input and output are shuffled Option C: Input is shuffled and output is in order Option D: Input is in order and output is shuffled Q3. If  $x_1(n)$  and  $x_2(n)$  are two real valued sequences of length N, and let x(n) be a complex valued sequence defined as  $x(n)=x_1(n)+jx_2(n), 0 \le n \le N-1$ , then what is the value of  $x_2(n)$ ?  $x_2(n) = \frac{x(n) + x^*(n)}{2}$ Option A:  $x_2(n) = \frac{x(n) - x^*(n)}{2}$ Option B:  $x_2(n) = \frac{x(n) - x^*(n)}{2j}$ Option C:  $x_2(n) = \frac{x(n) + x^*(n)}{2i}$ Option D: Q4. What is DFT of {1,1,1,1} Option A:  $X(K) = \{1, 0, 0, 0, \}$ Option B:  $X(K) = \{4, 4, 4, 4\}$ Option C:  $X(K) = \{4,0,0,0\}$ Option D:  $X(K) = \{4, 0, 1, 0\}$ 

Q5.	Aliasing occur in which of the following method?
Option A:	DFT & IDFT
Option B:	FFT & IFFT
Option D:	Overlap Save method
Option D:	Overlap Add method
Option D.	
Q6.	Which of the filter exhibit more transition band?
Option A:	Chebyshev-1
Option B:	Chebyshev-2
Option C:	Butterworth
Option D:	FIR Filter
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Q7.	Due to aliasing error the impulse invariant method is not suitable for design of
Option A:	Low pass filter
Option B:	Band pass filter
Option C:	High pass filter
Option D:	High pass and band pass filter
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Q8.	While designing IIR digital filter, Aliasing effect occurs due to which
	transformation method?
Option A:	Bilinear transformation Method
Option B:	Impulse Invariant Method
Option C:	Butterworth method
Option D:	Windowing Method
Q9.	Which of the following is true in the case of Butterworth filters?
Option A:	Smooth pass band
Option B:	Wide transition band
Option D:	Not so smooth stop band
Option D:	All of the mentioned
Option D.	All of the mentioned
Q10.	. If the conversion technique is to be effective, then the LHP of s-plane should be
Q10.	mapped
Option A:	Outside of unit circle
Option B:	On the Unit circle
Option C:	Inside unit circle
Option D:	Does not matter
Q11.	In Bilinear Transformation Method poles are transferred by using
Option A:	$s = \frac{2}{Ts} \left( \frac{Z-1}{Z+1} \right)$
Option B:	$s = \frac{Ts}{2} \left( \frac{Z-1}{Z+1} \right)$
Option C:	$s = \frac{2}{Ts} \left( \frac{Z+1}{Z-1} \right)$

Ontion D:	$T_{\rm S}$ (7 $\pm$ 1)
Option D:	$s = \frac{Ts}{2} \left( \frac{Z+1}{Z-1} \right)$
	2(Z-1)
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Q12.	Which of the following windows has a time domain sequence $h(n) = 1 - \frac{2\left n - \frac{M-1}{2}\right }{M-1}$ ?
Option A:	Bartlett window
Option B:	Blackman window
Option C:	Hanning window
Option D:	Hamming window
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Q13.	Which of the following is the difference equation of the FIR filter of length M, input $x(n)$ and output $y(n)$ ?
Option A:	M+1
	$y(n) = \sum_{k=0}^{N-1} b_k x(n+k)$
Option B:	<u>M+1</u>
1	$y(n) = \sum_{k=0}^{\infty} b_k x(n-k)$
Option C:	$y(n) = \sum_{k=0}^{M-1} b_k x(n-k)$
Option D:	$y(n) = \sum_{k=0}^{M-1} b_k x(n+k)$
014	A filter is said to be linear above filter if the above delay and shown delay are
Q14.	A filter is said to be linear phase filter if the phase delay and group delay are
Option A:	High
Option B:	Moderate
Option C:	Low
Option D:	Constant
Q15.	for design of multirate system, which should be the ideal choice
Option A:	IIR filter
Option B:	FIR filter
Option D:	Feedback filter
Option D:	Non linear filter
Option D.	
Q16.	Which of the following windows has a time domain sequence?
	$\mathbf{C}$

	$W(n) = 0.42 - 0.5COS\left(\frac{2\pi n}{M-1}\right) + 0.08COS\left(\frac{4\pi n}{M-1}\right)$
Option A:	Rectangular window
Option B:	Hanning window
Option C:	Triangular window
Option D:	Blackman window
Q17.	The effects caused due to finite word lengths are
Option A:	Coefficient quantization error
Option B:	Granular error
Option C:	Ringing effect
Option D:	Slope overflow error
Q18.	Whenever the quantization of filter coefficients is done,
Option A:	The degradation of frequency response takes place.
Option B:	The enhancement of frequency response takes place.
Option C:	The magnitude response remains unchanged.
Option D:	The input response remains unchanged.
Q19.	In the frequency response characteristics of FIR filter, the number of bits per coefficient should bein order to maintain the same error.
Option A:	Increased
Option B:	Constant
Option C:	Decreased
Option D:	Either decrease or increase
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Q20.	Quantization is a process.
Option A:	Non linear
Option B:	Reversible
Option C:	Non linear & Reversible
Option D:	linear
Q21.	What is function of MAC unit
Option A:	Multiply and Add data in one cycle
Option B:	Multiply and Add data in multiple cycle
Option C:	Multiply and Shift data in one cycle
Option D:	Multiply and Shift data in multiple cycle
Q22.	The function of Boot Loader is
Option A:	Speed the operation
Option B:	To access Internal Memory
Option C:	Decide the action after DSP reset.
Option D:	Helps for serial communication
Q23.	Advantage of DSP processor over simple processor is

Option A:	Accuracy is high
Option B:	Accuracy is less
Option C:	Stability is more
Option D:	Less dynamic range
Q24.	Function of DSP processor in ECG is to
Option A:	To modulate signal
Option B:	To detect signal
Option C:	Remove high frequency components
Option D:	Detect Target
Q25.	Function of DSP processor in RADAR is to
Option A:	To modulate signal
Option B:	To detect signal
Option C:	To amplify signal
Option D:	Detect Target