<b>ROLL</b>	NO.	

Code: AE77/AC77/AE121 Subject: DIGITAL SIGNAL PROCESSING

## **AMIETE - ET/CS (Current & New Scheme)**

**Time: 3 Hours** 

**June 2018** 

Max. Marks: 100

PLEASE WRITE YOUR ROLL NO. AT THE SPACE PROVIDED ON EACH PAGE IMMEDIATELY AFTER RECEIVING THE QUESTION PAPER.

**NOTE:** There are 9 Questions in all.

- Question 1 is compulsory and carries 20 marks. Answer to Q.1 must be written in the space provided for it in the answer book supplied and nowhere else.
- The answer sheet for the Q.1 will be collected by the invigilator after 45 minutes of the commencement of the examination.
- Out of the remaining EIGHT Questions answer any FIVE Questions. Each question carries 16 marks.
- Any required data not explicitly given, may be suitably assumed and stated.

#### Q.1 Choose the correct or the best alternative in the following:

 $(2\times10)$ 

a. Determine the convolution y(n) of two sequences  $x(n) = \{1, 2, 1, 2\}$  and  $h(n) = \{1, 0, 1, 0\}$ 

(A) 
$$y(n) = \{1,2,2,4,1,2,0\}$$

**(B)** 
$$y(n) = \{1,2,2,4,2,1,0\}$$

(C) 
$$y(n) = \{0,2,1,4,2,2,1\}$$

**(D)** 
$$y(n) = \{1,2,1,2,1,0,1\}$$

- As the window becomes shorter, the ability to resolve closely spaced sinusoids in the spectral estimate
  - (A) increases

(B) decreases

(C) does not change

- (D) increases then decreases
- c. The minimum sampling frequency for  $x_a(t)$  is real with  $X_a(f)$  non-zero only for 9 KHz < |f| <12 KHz is
  - (A) 4.5 KHz

**(B)** 12 KHz

(C) 9 KHz

- **(D)** 6 KHz
- d. If x[n + N] = x[n] then X[k + N] is
  - (A) 1

**(B)** 0

(C) X [k]

- (D) X [N]
- e. LTI behavior of a system depends on
  - (A) Input signal must be bandlimited
  - (B) System must be Linear and Time-Invariant
  - (C) Sampling Rate must be high enough so that any aliased components are removed
  - (D) All of these
- f. If the Fourier series coefficients of a signal are periodic then the signal must be

1

- (A) discrete-time, periodic
- (B) continuous-time, non-periodic
- (C) discrete-time, non-periodic
- (D) continuous-time, periodic

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- g. Time dependent Fourier Transform can be analyzed using
  - (A) Overlap add method
- (B) Overlap save method
- (C) Both (A) & (B)
- (D) None of these
- h. Z transform of  $\delta$  (n) is
  - (A)  $Z^{-n}$

**(B)** 1

(C) 1/z

- **(D)** 1/(1-z)
- Canonical form of structure is
  - (A) Direct Form I

- (B) Direct Form II
- (C) Both (A) & (B)
- (D) None of these
- j. In high-speed filtering applications
  - (A) parallel realization is preferred (B) cascaded realization is preferred.
  - (C) linear realization is preferred
- (D) None of these

#### Answer any FIVE Questions out of EIGHT Questions. Each question carries 16 marks.

- a. The continuous time signal  $x_a(t) = \sin(20\pi t) + \cos(40\pi t)$  is sampled with a sampling  $\mathbf{Q2}$ period T to obtain the discrete time signal  $x[n] = \sin(\frac{\pi n}{5}) + \cos(\frac{2\pi n}{5})$ . (i) Determine a choice for T consistent with this information. (ii) Is your choice for T in part (i) unique? If so, explain why? If not, specify another choice of T consistent with the information given.
  - b. With the help of complete calculations show that SNR increases by approximately 6 dB with each bit added to word length. **(8)**
- a. The Difference equation of a causal discrete time LTI system is given as Q.3  $y[n] = -\frac{1}{2}y[n-1] + x[n]$ 
  - (i) Find the frequency response  $H(e^{j\omega})$  for the system.
  - (ii) Find the output response of this system to the input  $x(n) = (1/2)^n u(n)$ (8)
  - b. (i) What are Inverse systems?

**(2)** 

- (ii) Explain minimum phase systems and discuss their unique fundamental properties. **(6)**
- **Q.4** a. By what factor do coefficient multipliers reduce/increase in Direct Form structures for linear phase FIR systems? Draw the structures. **(8)** 
  - b. Discuss the factors that influcence the choice of structure of realization of a LTI (8)system.

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- Q.5 a. Determine the magnitude response of  $y(n) = \frac{1}{2} [x(n) + x(n-2)]$ . (8)
  - b. Discuss the Parks- McClellan algorithm for type-I low pass filter. (8)
- **Q.6** a. Find out the DFT for a finite duration sequence x[n] with period N=5 (8)
  - b. Discuss overlap add method for performing linear convolution of large length signal.

    (8)
- Q.7 a. Consider a sequence  $x(n) = \{1,2,3,4\}$  its DFT is given by  $x(k) = \{10,-2+j2,-2,-2-j2\}$ . The sampling rate is 10 Hz. (8)
  - (i) Determine the sampling period, time index and sampling time instant for a discrete time sample x(3) in time domain.
  - (ii) Determine the frequency resolution, frequency bin number and frequency for each of the DFT coefficients X(1) and X(3) in frequency domain.
  - b. Explain Goertzel Algorithm approach to compute DFT. (8)
- Q.8 a. Discuss the effect of windowing on Fourier analysis of sinusoidal signals. (8)
  - b. Explain properties of periodogram. (8)
- Q.9 a. Explain Hilbert Transform relationships mentioning how it eliminates the constraints on the Fourier Transform.(8)
  - b. Give the Real and Imaginary-part sufficiency of the Fourier Transform for causal sequence. (8)