

15MT73

Seventh Semester B.E. Degree Examination, Aug./Sept.2020 Signal Process

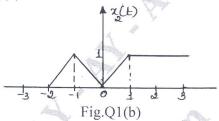
Time: 3 hrs.

Max. Marks: 80

Note: Answer any FIVE full questions, choosing ONE full question from each module.

Module-1

- a. Define Signal. List the types of signals with an example for each. (08 Marks)
 - b. Evaluate even and odd component of the signal (i) $x_1(t) = \cos t + \sin t + \sin t \cdot \cos t$ (ii) $x_2(t)$, is given in Fig.Q1(b). (08 Marks)



OR

- Check whether the given signal is periodic or not? If periodic find the fundamental period $x(t) = \cos(t + \pi/4).$ (04 Marks)
 - For the signal $x(n) = \{0.5, 1, 2, 4, 8\}$

Sketch $y_1(n) = x(n-3)$ and $y_2(n) = x(-n+4)$ (04 Marks) Two signal x(t) and g(t) are shown in Fig.Q2(c). Express the signal x(t) in terms of g(t).

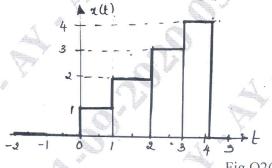




Fig.Q2(c)

(08 Marks)

Module-2

Derive an expression for convolution sum.

(08 Marks)

Determine the convolution of $x_1(n) = \{1, 2, 3\}$ and $x_2(n) = \{1, 2, 3, 4\}$

(08 Marks)

OR

State and prove commutative and distributive property for convolution integral. Evaluate the convolution of $x_1(t) = e^{-2t} u(t)$ and $x_2(t) = u(t+2)$

(08 Marks)

(08 Marks)

- Find the 4-point discrete fourier transform of $x(n) = \{1, 2, 3, 1\}$. (04 Marks) 5
 - A long sequence x(n) is filtered through a filter with a impulse response h(n) to yield the output y(n). If $x(n) = \{1, 2, 3, 3, 2, 1, -1, -2, -3, 5, 6, -1, 5, 0, 2, 1\}$ and $h(n) = \{1, 2\}$ Compute y(n) using overlap add technique assuming block length as 7. (12 Marks)

a. Compute the N-point DFT of $x(n) = a^n$ for $0 \le n \le N-1$. (04 Marks) b. Find the 8-point DFT of the sequence $x(n) = \{1, 1, 1, 1, 0, 0, 0, 0, 0\}$ using DIT-FFT radix-2 (12 Marks) algorithm. Draw the signal flow graph.

- a. Let $H(s) = \frac{1}{s^2 + s + 1}$ represent the transfer function of a lowpass filter with a passband of 1 rad/sec. Use frequency transformation to find the transfer function of the following analog filters.
 - (i) A lowpass filter with a passband of 10 rad/sec.
 - (ii) A high pass filter with a cutoff frequency of 1 rad/sec. (08 Marks)
 - b. Derive an expression for the order of Butterworth low pass filter. (08 Marks)

- Transform the analog filter $H_a(s) = \frac{(s+1)}{s^2 + 5s + 6}$ into H(z) using impulse invariant (08 Marks) transformation with T = 0.1 sec. (08 Marks)
 - Compare Butterworth and Chebyshev filters.

A lowpass filter is to be designed with the following desired frequency response. 9

$$H_{d}(e^{jw}) = H_{d}(w) = \begin{cases} e^{-j2w} & |w| < \frac{\pi}{4} \\ 0, & \frac{\pi}{4} < |w| < \pi \end{cases}$$

Determine the filter coefficients $h_d(n)$ h(n) if w(n) is a rectangular window defined as

$$W_{R}(n) = \begin{cases} 1, & 0 \le n \le 4 \\ 0, & \text{otherwise} \end{cases}$$
 (10 Marks)

b. Draw the block diagram of direct form-I and direct form-II for system function

$$H(z) = \frac{8 - 4z^{-1} + 11z^{-2} - 2z^{-3}}{1 - \frac{5}{4}z^{-1} + \frac{3}{4}z^{-2} - \frac{1}{8}z^{-3}}$$
(06 Marks)

OR

- (08 Marks) a. Compare IIR filter and FIR filters.
 - Obtain a cascade realization for a system described by