Reg. No.



MANIPAL INSTITUTE OF TECHNOLOGY (A constituent Institution of MAHE, Manipal)

VII SEMESTER B.TECH (ELECTRICAL & ELECTRONICS ENGINEERING) **END SEMESTER EXAMINATIONS, NOVEMBER 2018**

SUBJECT: ADVANCED DIGITAL SIGNAL PROCESSING [ELE 4012]

REVISED CREDIT SYSTEM

lours	Date: 27, NOVEMBER 2018	Max. Marks: 50

Time: 3 H **Instructions to Candidates:**

Answer ALL the questions.

Missing data may be suitably assumed.

Consider the multi-rate structure with input transform $X(e^{j\omega})$ is shown in Fig. Q 1A. Sketch 1A. the following (i) $V_1(e^{j\omega})$ (ii) $V_2(e^{j\omega})$ (iii) $V_3(e^{j\omega})$ (iv) $S(e^{j\omega})$; and (v) $Y(e^{j\omega})$. (04)

1B. The transform of input signal x[n] is depicted in Fig. Q 1B. The signal x[n] is modulated by signal $s[n] = \cos(\omega_n)$ and then up-sampled by factor 2. Sketch the following

(i)
$$X_M(e^{j\omega})$$
 (ii) $Y(e^{j\omega})$ for the modulated frequency $\omega_0 = \frac{\pi}{4}$ and $\omega_0 = \frac{\pi}{2}$ (03)

- A multi-rate discrete-time system with input and output shown in Fig. Q1C. Simply the system **1C.** to where down-sampler with input x[n] is followed by one period sample delay. In case of simplification mention the reasoning.
- 2A. Design an efficient two stages decimator with two suitable pair of decimation factors for the following specifications:

Input sampling frequency: 8 kHz; signal is decimated by factor M = 50; the highest frequency of interest after decimation: 75Hz; It is specified that the filter has overall passband ripple

 $\delta_p = 0.01$ and a stopband ripple $\delta_s = 10^{-4}$. Justify the answer with computational complexities.

- Considering the general form of poly-phase representation of interpolator type-1, develop a 2B. computationally efficient realization of a factor of 3 interpolator employing a length of 12 linear phase FIR low pass filter. Use the symmetry of the impulse response. (04)
- **2C** Consider a co-sinusoidal random variable so that

Y = g [X] = Cos(X). X is random variable uniformly distributed in the interval $(-\pi, \pi)$. Find the expected value of random variable Y. (02)

(03)

(04)

3A. Consider the random process $X(t) = A\cos(\omega_c t + \theta)$ where θ is the random variable with uniform distribution in the interval is $(-\pi, \pi)$ and A & ω_c are constants.

Determine (i) Is the process wide sense stationary (WSS)? (ii) If random process X(t) isWSS, find its Power spectral density (PSD) and plot power spectrum.(05)

3B. A continuous random variable *X* has probability density function given by

$$F_{X}(x) = \begin{cases} \left(k - \frac{x}{4}\right); & 1 \le x \le 3\\ 0 & ; & otherwise \end{cases}$$

Find

(i) the value of k (ii) $P(X \le 2.5)$ (iii) value of expectation E(X) for value k obtained in (i) (03)

3C. Prove that if the two events *A* and B are not disjoint then probability of their union event is given by

$$P[A \cup B] = P(A) + P(B) - P(A \cap B)$$
(02)

4A. A linear system is described by difference equation

$$y[n] = \frac{4}{5}y[n-1] + x[n] + x[n-1]$$
, where signal $x[n]$ is wide sense stationary (WSS) random

process with zero mean and autocorrelation $R_{\chi\chi}(m) = (0.5)^{|m|}$

- (a) Determine the power density spectrum of the output y(n)
- (b) Determine the autocorrelation $R_{_{VV}}(m)$ of the output
- (c) Determine the variance σ_{v}^{2}
- **4B.** In the application of random process the input-output relation through a linear system is described as shown in fig. Q4B. If the random process X(t) be the input to LTI system is wide

sense stationary (WSS), then show that (i) output random process Y(t) is also WSS. (ii)

$$S_{YY}(F) = S_{XX}(F) |H(F)|^2$$
 (04)

- **4C.** If the sample sequence of a random process has N=2000 samples. Determine (i) the frequency resolution of the Bartlett, Welch (for 50% overlap), and Blackman-Tukey methods for a quality fact Q = 20 (ii) the record lengths (M) for the Bartlett, Welch (for 50% overlap), and Blackman-Tukey methods.
- 5A. Discuss the properties of Wavelet Transform?
- **5B.** Determine the 2D DWT Haar decomposition of 2D pixel values:

7	3	25	9
35	1	9	3
11	15	9	27
7	33	5	11

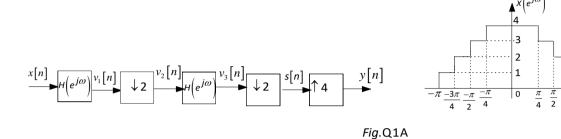
Also reconstruct the pixel values from the decomposed pixel values with threshold value of 5. (04)

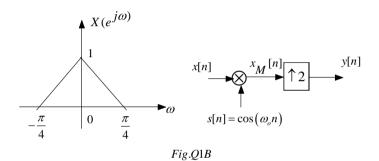
(04)

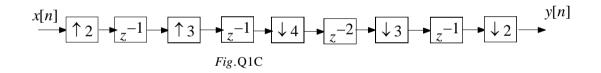
(02)

(02)

- **5C.** Perform system modelling task assuming the unknown system is two coefficient FIR filter.
 - i. Setup the LMS algorithm to implement the adaptive filter, assuming that initial weights are: $w(0) = w(1) = 0.5 \& \mu = 0.6$
 - ii. Perform adaptive filtering to obtain error signal e[n] for n = 0, 1, 2, 3. Given : x(0) = 1, x(1) = -1, x(2) = 1, x(3) = -1;d(0) = 0.5, d(1) = 0.5, d(2) = -0.5, d(3) = 0.5







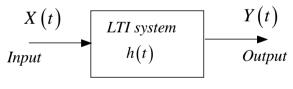


Fig.Q4B

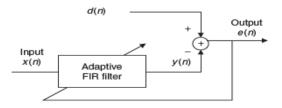


Fig.Q5

(04)

ω

3π

4

π