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filters $G_0(z)$ and $G_1(z)$ so that the structure of figure below is a perfect reconstruction filter bank.



(b) Let the analysis filters $H_0(z)$ and $H_1(z)$ of the above structure be the power complementary FIR filters of order *N* each. Show that this structure becomes a perfect reconstruction filter bank if the synthesis filters $G_0(z)$ and $G_1(z)$ are chosen as

$$G_0(z) = z - NH_0(z^{-1})$$
 and $G_1(z) = z - NH_1(z^{-1})$.
6 + 6 = 12

- 7 (a) What do you mean by adaptive filters? Explain with basic adaptive filter structure.
 - (b) Derive weight update expression of an adaptive filter using LMS algorithm.
 - (c) Explain how adaptive filters are used in system identifications.

3 + 5 + 4 = 12

Group - E

- 8. (a) What is the need for spectral estimation? How can the energy density spectrum be determined?
 - (b) A discrete-time signal x(n) is given as under: $x(n) = \cos(2\pi f_1 n) + \cos(2\pi f_2 n), n = 0, 1, 2, ..., 7.$ Determine the power spectrum for the data sequence length L = 8,16 for different values of f_1 and f_2 , where $f_2 = f_1 + \Delta f$ and Δf is small deviation from f_1 or simply frequency separation. (2 + 2) + 8 = 12
- 9. (a) What are AR and ARMA processes?

(b) A 1st order auto regressive (AR) process is given by $x(n) = Ax(n-1) + \omega(n)$, where *A* is a constant and $\omega(n)$ is a white noise process of zero mean and variance σ^2 . Find the mean and autocorrelation function of the process $x\{x(n)\}$.

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$$(2+2)+8=12$$

ADVANCED DIGITAL SIGNALS AND SYSTEMS (AEIE 5101)

Time Allotted : 3 hrs

Full Marks: 70

Figures out of the right margin indicate full marks.

Candidates are required to answer Group A and <u>any 5 (five)</u> from Group B to E, taking <u>at least one</u> from each group.

Candidates are required to give answer in their own words as far as practicable.

Group – A (Multiple Choice Type Questions)

1. Choose the correct alternative for the following:

 $10 \times 1 = 10$

(i) Condition for aliasing problem is: (a) $f_s < f_m$ (b) $f_s < 2f_m$ (c) $f_s = f_m$ (d) all of these.

(ii) Which of the following system is causal?
(a)
$$y(n) = x(-n) - x(2n-1)$$
 (b) $y(n) = x^2(n) - x(n+1)$
(c) $h(n) = n \left(\frac{1}{2}\right)^n u(n)$ (d) $h(n) = n \left(\frac{1}{2}\right)^n u(n+1)$.

(iii) Which of the following substitution is done in Bilinear transformations? (a) $2 \begin{bmatrix} 1+z^{-1} \end{bmatrix}$ (b) $2 \begin{bmatrix} z^{-1}+1 \end{bmatrix}$

(d)
$$s = \frac{1}{T} \begin{bmatrix} \frac{1+2}{1-z^{-1}} \end{bmatrix}$$

(c) $s = \frac{2}{T} \begin{bmatrix} \frac{1-z^{-1}}{1+z^{-1}} \end{bmatrix}$
(d) None of the above.

(iv) A continuous-time signal $x_a(t)$ is known to be uniquely recoverable from its samples $x_a(nT_s)$ when $T_s = 1$ ms. What is the highest frequency in $x_a(f)$? (a) 5000 Hz (b) 1000 Hz (c) 500 Hz (d) none of these.

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(v)

If
$$DFT \{x(n)\} = X(k)$$
, then $DFT \{x(n+m)_N\}$
(a) $X(k)e^{-j\frac{2\pi km}{N}}$ (b) $X(k)e^{-j\frac{2\pi k}{mN}}$ (c) $X(k)e^{j\frac{2\pi km}{N}}$ (d) $X(k)e^{j\frac{2\pi k}{mN}}$.

- For a system y(n) = x(n-3), the impulse response of the system (vi) and the inverse system will be, (a) $h(n) = \delta(n+3)$ and x(n) = y(n-3) respectively (b) $h(n) = \delta(3n)$ and $x(n) = y\left(\frac{n}{3}\right)$ respectively (c) $h(n) = \delta(n-3)$ and x(n) = y(n+3) respectively (d) $h(n) = \delta(n+3)$ and x(n) = y(3n) respectively.
- Which of the following is the difference equation of the FIR filter of (vii) length *M* , input x(n) and output y(n)?

(a)
$$y(n) = \sum_{k=0}^{M+1} b_k x(n+k)$$

(b) $y(n) = \sum_{k=0}^{M+1} b_k x(n-k)$
(c) $y(n) = \sum_{k=0}^{M-1} b_k x(n-k)$
(d) None of the above.

- Which operation is implemented by keeping every M -th sample of (viii) x(n) and removing M-1 in between samples to generate y(n)?
 - (a) Up-sampling (b) Down-sampling (c) Both (a) and (b) (d) None of the above.
- If $X_1(k)$ and $X_2(k)$ are the N-point DFTs of $x_1(n)$ and $x_2(n)$ (ix) respectively, then what is the *M* -point DFT of $x(n) = ax_1(n) + bx_2(n)$? (a) $X_1(ak) + X_2(bk)$ (b) $aX_1(k) + bX_2(k)$ (c) $e^{ak}X_1(k) + e^{bk}X_2(k)$ (d) None of the mentioned.
- (x) E[X - E(X)] is equal to: (a) E(X)(b) Var(X)(d) E(X) - X. (c) 0

Group - B

2. (a) Determine whether the following system is time-invariant or time variant: v(n) = x(-n)

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- (b) Represent the sequence $x(n) = \{2, 4, -1, -2\}$ as sum of shifted unit impulses.
- Determine whether the following signal is energy or power signal: (c) $x(n) = \left(\frac{3}{8}\right)^n u(n)$

5 + 2 + 5 = 12

- 3. Test the stability of the system whose impulse response (a) $h(n) = \left(\frac{1}{2}\right)^n u(n).$
 - Find the circular convolution of two sequences $x_1(n) = \{1, 2, 3\}$ and (b) $x_2(n) = \{1, -1, -2\}$ using DFT-IDFT method.

$$4 + 8 = 12$$

Group - C

- Design a lowpass Butterworth IIR digital filter using bilinear 4. (a) transformation technique with T = 1, from its analog filter that satisfies the following specifications: Passband cutoff: $\Omega_p = 0.2\pi$; Passband ripple: $R_p = 7dB$ Stopband cutoff: $\Omega_s = 0.3\pi$; Stopband ripple: $A_s = 16 dB$.
 - Transform the analog filter $H_a(s) = \frac{s+1}{s^2 + 5s + 6}$ into a digital filter (b) H(z) using the impulse invariance technique in which T = 0.1.

8 + 4 = 12

2 + 10 = 12

- State the main disadvantage of direct form realization of digital filters. 5. (a)
 - Obtain the direct form and cascade form realization for the transfer (b) function of an FIR system given by

$$H(z) = \left(1 - \frac{1}{4}z^{-1} + \frac{3}{8}z^{-2}\right) \left(1 - \frac{1}{8}z^{-1} - \frac{1}{2}z^{-2}\right).$$

Group - D

Consider the analysis- synthesis filter bank shown in figure below. 6. (a) Develop the input output relation of this structure in the z-domain. Let $H_0(z) = \frac{(1+z^{-1})}{2}$ and $H_1(z) = \frac{(1-z^{-1})}{2}$. Determine the synthesis

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