

GUJARAT TECHNOLOGICAL UNIVERSITY**M.E –IIst SEMESTER–EXAMINATION – JULY- 2012****Subject code: 1722601****Date: 06/07/2012****Subject Name: Advanced Digital Signal Processing and Applications****Time: 10:30 am – 13:00 pm****Total Marks: 70****Instructions:**

1. Attempt all questions.
2. Make suitable assumptions wherever necessary.
3. Figures to the right indicate full marks.

Q.1 (a) Demonstrate that wide sense stationary random process can be represented as the output of a causal and causally invertible linear system excited by a white noise process. **07**

(b) What do you understand by Backward Linear prediction. For a Backward Linear predictor find the expression for the system function $B_p(z)$. Show that $B_p(z)$ is the reverse polynomial of $A_p(z)$, the system function of Forward linear prediction. **07**

Q.2 (a) Explain and compare AR and ARMA random processes with mathematical expressions for system functions as well as lattice structures. **07**

(b) Given a quadratic MSE function for the Wiener filter: **07**

$$J = 40 - 20w + 10w^2$$

Find the optimal solution for w^* to determine $J_{\min}$ by

(i) achieving the minimum MSE $J_{\min}$ and

(ii) using the steepest descent method with an initial guess as $w_0 = 0$ and $\mu = 0.04$ by iterating three times.

OR

(b) Design a Wiener FIR filter of length of $M=2$ with input to the filter with sequence $x(n)=s(n)+w(n)$ for an AR(1) process satisfying the equation $s(n)=0.4s(n-1)+v(n)$ where $\{v(n)\}$ is white noise sequence with variance **07**

$\sigma_v^2 = 0.49$ and $\{w(n)\}$ is the white noise sequence with

$\sigma_w^2 = 1$. Determine (i) $\Gamma_{ss}(f)$ (ii) For filter coefficients $h(0)$ and $h(1)$ for $\gamma_{ss}(m) = (0.4)^m$.

Q.3 (a) Obtain estimate of the statistical mean and autocorrelation of the random process from a single realization $x(n)$. **07**

(b) An ARMA Process has an autocorrelation $\{\gamma_{xx}(m)\}$ whose Z transform is given as **07**

$$\Gamma_{xx}(z) = \frac{9\left(z - \frac{1}{3}\right)(z-3)}{\left(z - \frac{1}{2}\right)(z-2)} \quad \frac{1}{2} < |z| < 2$$

Determine (i) the filter $H(z)$ for generating $\{x(n)\}$ from a white noise input sequence (ii) a stable whitening filter for the sequence $\{x(n)\}$.

OR

Q.3 (a) Show that for the optimum IIR Wiener filter, **07**

$$(i) MMSE_{\infty} = \sigma_d^2 - \sum_{k=0}^{\infty} h_{opt}(k) h_{dx}^*(k) \quad (ii) Q(z) = \frac{1}{\sigma_i^2} [\Gamma_{di}(z)]_+$$

- (b) Determine the system function $H(z)$ for the FIR filter described by the lattice coefficients by the lattice coefficients $K_1=0.6$, $K_2=0.3$, $K_3=0.5$, $K_4=0.9$. Find $h(n)$. **07**

Q.4 (a) Explain the application of adaptive filter for Echo cancellation in data transmission over telephone lines using Symbol rate echo canceller. **07**

- (b) Define : (i) Stationary Random processes (ii) Ensemble Averages (iii) Prediction error (iv) Cross Power density spectrum (v) innovation process (vi) Adaptive filter (vii) Toeplitz matrix **07**

OR

Q.4 (a) What is Least mean-squared error criterion? Explain in detail LMS algorithm for FIR filters adaptive filter based on steepest descent method. **07**

- (b) Determine the reflection coefficients $|k_m|$ of the Lattice filter corresponding to an FIR filter described by the system function **07**

$$H(z) = A_3(z) = 1 + \frac{13}{24}z^{-1} + \frac{5}{8}z^{-2} + \frac{1}{3}z^{-3}$$

Q.5 (a) Obtain the relation between filter parameters and autocorrelation sequence and obtain Yule-Walker equation for AR process from the generalized relations for ARMA process. **07**

- (b) Explain the cases of linear estimation problems such as signal prediction and smoothing of Wiener filter subject to an input sequence $x(n)=s(n)+w(n)$ where $s(n)$ and $w(n)$ are uncorrelated random sequences, and output sequence is $y(n)$ with normal linear Wiener-Hopf equations for the minimization of the minimum mean square error for both. **07**

OR

Q.5 (a) Explain Polyphase decomposition process. Illustrate for FIR structure. **07**

- (b) For a single-stage decimator with the following specifications: **07**
 Sampling rate = 8 kHz, Decimation factor $M = 4$
 Input audio frequency range = 0–800 Hz,
 Passband ripple = 0.02 dB, Stopband attenuation = 46 dB
 a. draw the block diagram for the decimator;
 b. determine the window type, filter length, and cutoff frequency if the window method is used for the anti aliasing FIR filter design.
