

# COMPREHENSIVE EXAMINATION

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Course No./Title : EEE G613/Advanced Digital Signal Processing

DATE: Dec. 17<sup>th</sup>, 2022. Test duration: 180 Mins. Max. marks: 40.

(Note: You may use standard results or formulae by stating them explicitly. Highlight final answers in boxes. Simplify your response to the extent possible. Use notation consistently. Provide key intermediate steps.)

**Q.1.** Consider the following filters:

$$\mathcal{H}_0(z) = 1 + z^{-1}, \mathcal{H}_1(z) = 1 - z^{-1}, \mathcal{F}_0(z) = 1 + z^{-1}, \mathcal{F}_1(z) = -1 + z^{-1},$$

where  $\mathcal{H}_0(z)$  and  $\mathcal{H}_1(z)$  are the analysis filters,  $\mathcal{F}_0(z)$  and  $\mathcal{F}_1(z)$  are the synthesis filters.

Can we construct two-channel QMF bank using the above set of filters? Yes/No? Justify your response. [1 point]

ii). Suppose the synthesis filters are selected such that  $\mathcal{F}_0(z) = \mathcal{H}_0(z)$  and  $\mathcal{F}_1(z) = -\mathcal{H}_1(z)$ . Comment on the nature of the matrix. Determine the Alias component matrix and its determinant. [3 points]

iii). Neatly sketch the QMF bank with analysis and synthesis filters. Provide all labels/notation in the block diagram. [2 points]

iv). Express the output  $\hat{X}(z)$  as a product of three matrices in which the middle matrix is the Alias component matrix. [1 point]

**Q.2.** Let  $\sigma_0^2, \sigma_1^2, \dots, \sigma_{N-1}^2$  denote the subband variances. Recall that variance signifies the power of a signal. Consider the coding gain in an ideal subband coding system with  $N$  subband channels (ideal filters). The subband coding gain (say  $\mathcal{G}_c$ ) is defined as

$$\mathcal{G}_c = \frac{\frac{1}{N} \sum_{m=0}^{N-1} \sigma_m^2}{\left(\prod_{m=0}^{N-1} \sigma_m^2\right)^{\frac{1}{N}}}.$$

Consider the following power spectrum of the input signal:

$$|X(e^{j\omega})|^2 = 1 - \frac{\omega}{\pi}, \quad |\omega| \leq \pi.$$

i). Let  $N = 2$ . Determine subband variances ( $\sigma_0^2, \sigma_1^2$ ) and their geometric mean (GM). Compute the subband gain in dB. Note: During the computation of variances, you need to account for positive and negative frequencies. [3 points]

ii). Consider the general scenario of  $N$  subband channels. Obtain the general expression for subband variance (say  $\sigma_k^2$ ) and subband coding gain (as a function of  $N$ ). Note: Final answer should be in the most simplified form and does *not* contain summation (' $\sum$ '). [4 points]

**Q.3.** i). Consider the discrete-time (DT) matched filter detector. Consider the following statement: "The signal output of the matched filter is maximized by sampling the output at  $n = (N - 1)$ ." Prove this statement analytically. The following equation is useful:

$$y[n] = \int_{-0.5}^{0.5} S^*(f)X(f) \exp\{j2\pi f(n - (N - 1))\} df.$$

ii). Consider the following signal detection:

$$s[n] = A_m \cos(2\pi f_0 n), \quad 0 < f_0 < 0.5, \quad n = 0, 1, \dots, (N - 1).$$

a). Determine the signal output of a discrete-time matched filter at time  $n = (N - 1)$ . Simplify to the extent possible. Hint: Try approximation (sum  $\rightarrow$  integral) for large  $N$ . Note: Your final answer should *not* contain summation (' $\sum$ ').

b). Assume that the signal is delayed by  $n_0 > 0$  samples so that we receive  $s[n - n_0]$ . Applying the same matched filter (in part (a)), determine the signal output as a function of  $N, A_m, f_0, n_0$  at  $n = (N - 1)$ . Comment on the final result (i.e., What is the impact of  $n_0$ ?) Note: Your final answer should *not* contain summation (' $\sum$ '). [2 + 2 + 3 points]

**[P.T.O]**

**Q.4.** Consider the following real-valued measurements described by

$$Y_1 = \beta + W_1, \quad Y_2 = \beta + W_2,$$

where  $\beta$  is a non-random parameter,  $W_1, W_2$ , are random, statistically independent unbiased measurement errors having mean zero. The objective is to design a data processing algorithm that combines the two measurements  $Y_1$  and  $Y_2$  to produce mean square error (MSE)-optimal estimate of  $\beta$ . The errors  $W_1, W_2$  have variances  $\sigma_1^2$  and  $\sigma_2^2$ , respectively.

Define  $\hat{\beta} = k_1 Y_1 + k_2 Y_2$ . Determine the optimal estimate  $\hat{\beta}^*$  in terms of the error variances and measurements.

*Note:* Let  $\beta_e = \hat{\beta} - \beta$ .  $\mathbf{E}[\beta_e] = 0$ . Find  $k_1$  and  $k_2$  (in terms of noise variances) that minimize  $\mathbf{E}[|\beta_e|^2]$ .

Determine the minimum mean square estimation error (MMSE). [4 + 1.5 points]

ii). Refer to Fig. 1. Identify and name the signal processing tasks. Note that (a), (b), and (c) illustrate three types of estimation problems; we desire an estimate at time  $t$ .

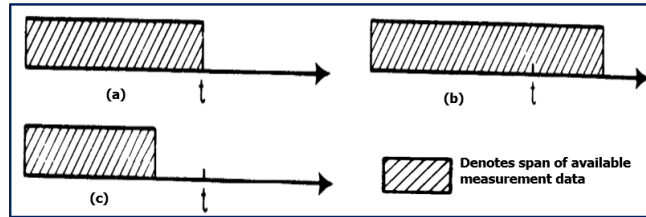


Fig. 1. Pertaining to Q.4.(ii).

**Q.5.** i). Consider the following adaptive filter model:

Let  $\mathbf{x}[n]$  denote the input vector. Let  $\mathbf{h}[n]$  characterize the unknown system (e.g., channel) and  $\hat{\mathbf{h}}[n]$  its estimate.

Let  $\mathbf{y}[n] = \mathbf{h}^T[n]\mathbf{x}[n]$  and  $\mathbf{d}[n] = \mathbf{y}[n] + \mathbf{I}[n]$ , where  $\mathbf{I}[n]$  denotes the interference  $\neq 0$ . Further, consider the following definitions:

$$\mathbf{e}[n] = \mathbf{d}[n] - \hat{\mathbf{y}}[n], \quad \delta[n] = \hat{\mathbf{h}}[n] - \mathbf{h}[n], \quad \mathbf{r}[n] = \hat{\mathbf{y}}[n] - \mathbf{y}[n], \quad \Delta_e[n] = \left| \hat{\mathbf{h}}[n] - \mathbf{h}[n] \right|^2.$$

Consider the following update:

$$\hat{\mathbf{h}}[n+1] = \hat{\mathbf{h}}[n] + \frac{\mu \mathbf{e}[n] \mathbf{x}[n]}{\mathbf{x}^T[n] \mathbf{x}[n]},$$

where  $\mu$  denotes learning rate.

Derive an expression for optimal learning rate  $\mu^*$  in terms of  $\sigma_r^2, \sigma_e^2$ . *Note:*  $\mathbf{E}[|\mathbf{r}[n]|^2] \triangleq \sigma_r^2$  and  $\mathbf{E}[|\mathbf{e}[n]|^2] \triangleq \sigma_e^2$ . *Hint:* Optimal learning rate minimizes  $\mathbf{E}[\Delta_e[n+1]]$ . [4 points]

ii). Consider the following model:  $r[n] \sim \mathcal{CN}(0, 1)$ . Further,  $e[n] = 9 \sin(n\omega_0 + \Psi)$  is a wide-sense stationary random process, where  $\Psi \sim \mathcal{U}[-\pi, \pi]$ . Using the formula obtained in (i), compute  $\mu^*$ . [3 points]

**Q.6. Valid or Invalid (5 points)** *Note: Write down Valid/Invalid. Justify your response.*

- 1) Consider the mathematical formulation of compressed sensing (CS): Let  $\mathbf{s}$  denote a column vector of weighting coefficients. Let  $\mathbf{y}$  denote the vector of measurements. Further, let  $\Phi$  be a stable measurement matrix, and  $\Psi$  is the basis matrix comprising orthonormal basis vectors. The following problem is *concave* in  $\mathbf{s}$ :

$$\min_{\mathbf{s}} \|\Theta \mathbf{s} - \mathbf{y}\|_2 + \lambda \|\mathbf{s}\|,$$

where  $\Theta = \Phi \mathbf{s}$ ,  $\lambda$  is some real positive constant that weighs importance of sparsity.

- 2) Recall the optimal filter (Wiener) weights  $\mathbf{w}^*$  given by  $\mathbf{w}^* = \mathcal{R}_x^{-1} \mathbf{r}_{dx}$ , where  $\mathcal{R}_x$  is the autocorrelation of  $x[n]$  and  $\mathbf{r}_{dx}$  is the crosscorrelation between  $d[n]$  and  $x[n]$ . Consider the quadratic cost  $\mathcal{C}_q(\mathbf{w}) = \sigma_d^2 - 2\mathbf{w}^T \mathbf{r}_{dx} + \mathbf{w}^T \mathcal{R}_x \mathbf{w}$ . For  $\sigma_d^2 = 0.54$ ,  $\mathbf{r}_{dx} = [\frac{1}{\sqrt{8}} \quad -\frac{1}{\sqrt{8}}]^T$  and  $\mathbf{w}^* = [\frac{1}{\sqrt{2}} \quad -\frac{1}{\sqrt{2}}]^T$ , we have  $\mathcal{C}_q(\mathbf{w}^*) = 0.04$ .
- 3) Consider a finite energy pulse  $s(t)$ . Its magnitude spectrum is given by  $|S(f)| = \frac{1}{2\sqrt{\pi}} \exp(-4\pi^2 f^2)$ . The mean square bandwidth = 0.25. *Note:*  $\int_{-\infty}^{\infty} e^{-\pi f^2} df = 1$ .
- 4) Consider a channel impulse response at  $t_n$  denoted by  $h(t_n) = h_{cn} + j h_{sn}$ , is a zero mean, complex Gaussian random process with variance  $\sigma^2(t_n)$ . The real and imaginary components  $h_{cn}$  and  $h_{sn}$  are uncorrelated Gaussian random processes each with variance  $\sigma^2(t_n)$ .
- 5) Let  $\mathbf{B}$  be a stable measurement matrix. The standard CS problem is to reconstruct a  $s$ -sparse vector  $\mathbf{x} \in \mathbb{R}^N$  from the measurements  $\mathbf{y} = \mathbf{B}\mathbf{x} \in \mathbb{R}^M$ , where  $M > N$ .

□ END OF QUESTION PAPER □