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**GANDHI INSTITUTE OF ENGINEERING AND TECHNOLOGY UNIVERSITY, ODISHA, GUNUPUR
(GIET UNIVERSITY)**



**Ph.D. (Second Semester-Summer) Examinations, May - 2025
23SPPEEC2011 - Advanced Digital Signal Processing
(ECE)**

Time: 3 hrs

Maximum: 70 Marks

The figures in the right hand margin indicate marks.

Answer ANY FIVE Questions.**(14 x 5 = 70 Marks) Marks**

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| 1.a. | Explain the decimation-in-time (DIT) FFT algorithm with a signal flow graph for an 8-point FFT. Compare its computational complexity with the DFT. | 8 |
| b. | Design a linear phase FIR filter using the Hamming window for a given cutoff frequency. Discuss the trade-offs of using Hamming vs. Kaiser windows. | 6 |
| 2. | Explain the polyphase decomposition of an FIR filter and derive its efficient structure for decimation. How does this reduce computational load? Discuss an application in software-defined radios (SDR). | 14 |
| 3.a. | Derive the Wiener-Hopf equations for an optimum FIR Wiener filter. How does it minimize mean square error? | 7 |
| b. | Compare AR lattice and ARMA lattice-ladder filters in terms of structure and prediction error minimization. | 7 |
| 4. | Explain the LMS algorithm with its stability conditions. Compare its convergence behavior with the RLS algorithm in the context of channel equalization. | 14 |
| 5.a. | Describe the Bartlett and Welch methods for non-parametric spectrum estimation. How do they reduce variance compared to the periodogram? | 7 |
| b. | Explain the Yule-Walker method for AR model-based spectrum estimation. Derive the normal equations. | 7 |
| 6.a. | Discuss the role of multirate DSP in radar systems (e.g., pulse compression). How does polyphase filtering improve efficiency? | 7 |
| b. | Explain the design of a Hilbert transformer using an FIR filter and its application in single-sideband (SSB) modulation. | 7 |
| 7. | Describe the MUSIC algorithm for spectrum estimation. How does eigenanalysis help in resolving closely spaced frequencies? Provide a mathematical formulation. | 14 |
| 8.a. | Explain how adaptive noise cancellation is achieved using LMS filters in speech processing. | 7 |
| b. | Discuss the application of wavelets in image compression (e.g., JPEG 2000). Why are wavelets superior to DCT? | 7 |

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