These are sample MCQs to indicate pattern, may or may not appear in examination
University of Mumbai
Online Examination 2020
Program: TE Electronics and Telecommunication Engineering
Curriculum Scheme: Revised 2016
Examination: Third Year Semester V
Course Code: ECC 504 and Course Name: Discrete Time Signal Processing
Time: 1 hour
Max. Marks: 50

Note to the students:- All the Questions are compulsory and carry equal marks .

| Q1. | If DFT of $\mathrm{x}(\mathrm{n})$ is $\mathrm{X}(\mathrm{K})$ then DFT of $\mathrm{x}(\mathrm{n}-\mathrm{N} / 2)$ as per half period shift property of DFT is |
| :--- | :--- |
| Option A: | $\mathrm{X}(\mathrm{K})$ |
| Option B: | $-\mathrm{X}(\mathrm{K})$ |
| Option C: | $(-1)^{\wedge} \mathrm{k} . \mathrm{X}(\mathrm{K})$ |
| Option D: | $2 \mathrm{X}(\mathrm{K})$ |
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| Q2. | The circular convolution of two sequences in time domain is equivalent to |
| Option A: | Multiplication of DFTs of two sequences |
| Option B: | Summation of DFTs of two sequences |
| Option C: | Difference of DFTs of two sequences |
| Option D: | Square of multiplication of DFTs of two sequences |
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| Q3. | If sequence is imaginary and odd the DFT is |
| Option A: | Real and Even |
| Option B: | Imaginary and odd |
| Option C: | Imaginary and even |
| Option D: | Real and Odd |
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| Q4. | If $\mathrm{X}(\mathrm{k})$ is the N -point DFT of a sequence $\mathrm{x}(\mathrm{n})$, then what is the DFT of $\mathrm{x} *(\mathrm{n})$ ? ( note that <br> $\mathrm{X}^{*}(\mathrm{k})$ is complex conjugate of $\left.\mathrm{X}(\mathrm{K})\right)$ <br> Option A: <br> $\mathrm{X}(\mathrm{N}-\mathrm{k})$ <br> Option B: <br> $\mathrm{X}^{*}(\mathrm{k})$ <br> Option C: <br> $\mathrm{X}^{*}(\mathrm{~N}-\mathrm{k})$ <br> Option D: <br> $\mathrm{X}(\mathrm{k})$ <br> Q5. <br> Option A: <br> Option B: <br> Option C: <br> In DIT FFT algorithm input is arranged in <br> Reverse order <br> Bit reversed order |


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| :--- | :--- |
| Q6. | The total number of complex additions required to compute N point DFT by radix-2 FFT <br> is? (Note the base of all Log is 2 ) |
| Option A: | $(\mathrm{N} / 2) \log (\mathrm{N})$ |
| Option B: | $(\mathrm{N} / 2) \log \left(\mathrm{N}^{*} 2\right)$ |
| Option C: | N log(N) |
| Option D: | N |
|  |  |
| Q7. | What is the relation between analog frequency $(\Omega)$ and digital frequency $(\omega)$ in impulse <br> invariant method? |
| Option A: | $\Omega=\omega \mathrm{T}$ |
| Option B: | $\Omega=\omega / \mathrm{T}$ |
| Option C: | $\Omega=\mathrm{T} / \omega$ |
| Option D: | $\Omega=\omega$ |
|  |  |
| Q8. | The methods used for designing IIR filters is |
| Option A: | Impulse Invarience Method |
| Option B: | Window method |
| Option C: | Kaiser Window |
| Option D: | Frequency Sampling Method |
|  |  |
| Q9. | Bilinear Transformation make use of.............. |
| Option A: | one to one mapping from s-domain to z-domain |
| Option B: | Sampling the impulse response of an equivalent analog filter |
| Option C: | Taking backward difference for the derivative |
| Option D: | Approximation of Derivatives |
|  |  |
| Q10. | Which of the following method is not used for designing IIR Filter |
| Option A: | Impulse Invariant Method |
| Option B: | Bilinear Transformation |
| Option C: | Approximation of Derivatives |
| Option D: | Window method |
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| Q11. | In IlR digital filter the present output depends on |
| Option A: | Present input samples and past Inputs samples only |
| Option B: | Present input samples and past output samples only |
| Option C: | Present input samples only |
| Option D: | Present Input sample, Past input samples and output samples |
|  |  |
| Q12. | If N is Unquantised number and Nt is number obtained by truncation of Error is, Et = |
| Option A: | Nt-N |
| Option B: | Et-Nt |
| Option C: | N-Et |
| Option D: | N -Nt |
|  |  |
| Q13. | What is NTF |
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| Option A: | Negative Transfer Function |
| :--- | :--- |
| Option B: | Noise Truncation Function |
| Option C: | Negative Truncation Function |
| Option D: | Noise Transfer Function |
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| Q14. | Overflow limit cycle in output are oscillation due to which of following |
| Option A: | Overflow of Division |
| Option B: | Overflow of Subtraction |
| Option C: | Overflow of Multiplication |
| Option D: | Overflow of Addition |
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| Q15. | Which type of architecture uses different storage space for program code and the data? |
| Option A: | Von Neumann architecture |
| Option B: | Harvard architecture |
| Option C: | Fragmented architecture |
| Option D: | Split cell architecture |
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| Q16. | In DAGs, which register/s provide/s increment or step size for index register especially <br> during the register move? |
| Option A: | Index Register |
| Option B: | Length \& Base Register |
| Option C: | Modify Register |
| Option D: | Only Base Register |
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| Q17. | In TMS 320 C6x processor architecture, which functional unit is adopted for transferring <br> the data from register to and from control register? |
| Option A: | L2 |
| Option B: | M2 |
| Option C: | S2 |
| Option D: | D2 |
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| Q18. | Which units are generally involved in Multiply and Accumulate (MAC)? |
| Option A: | Only Subtractor |
| Option B: | Multiplier |
| Option C: | Accumulator |
| Option D: | Both B \& C |
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| Q19. | Which of the following block is not required in digital processing of a RADAR signal? |
| Option A: | Typical DTMF frequencies range approximately from. |
| Option B: | 700Hz to 1700 Hz |
| Option B: | A/D converter |
| Option C: | DSP converter |
| Option D: | Both A \& B |


| Option C: | 700 Hz to 1200 Hz |
| :--- | :--- |
| Option D: | 1200 Hz to 1700 Hz |
|  |  |
| Q21. | In DTMF 770 Hz \& 1633Hz Frequency combination related to |
| Option A: | A |
| Option B: | B |
| Option C: | C |
| Option D: | D |
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| Q22. | In FIR filter design using Window method, which among the following parameters is/are <br> separately controlled using Kaiser window? |
| Option A: | Order of filter (M) |
| Option B: | Shape of Window function |
| Option C: | Order of the filter (M) as well as Shape of window function |
| Option D: | Gain in passband and attenuation in stop band of the filter |
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| Q23. | Which among the following represent/s the characteristic/s of an ideal filter? |
| Option A: | Infinite gain in passband |
| Option B: | Zero attenuation in stop band |
| Option C: | Constant gain in passband and zero gain in stopband |
| Option D: | Passes all frequencies at the input to output without attenuation |
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| Q24. | The principle of Frequency Sampling method of fIR filter design is : |
| Option A: | DTFT H(w) is used to find impulse response h(n) of the filter |
| Option B: | DFT samples H(k) are used to find impulse response h(n) of the filter |
| Option C: | System function H(z) is used to find impulse response h(n) of the filter |
| Option D: | Truncation of Fourier series coefficients is used to find impulse response $\mathrm{h}(\mathrm{n})$ of the <br> filter |
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| Q25. | Impulse response of linear phase FIR filter is h(n) =\{ 1,2,3,2,1\}. This filter is of: |
| Option A: | Type-I |
| Option B: | Type-II |
| Option C: | Type-III |
| Option D: | Type-IV |

