Instituto Superior Técnico / Tecnical University of Lisbon

Departament of Bioengineering

## Master on Biomedical Engineering

## **Digital Signal Processing in Bioengineering**

2nd Semester de 2012/2013

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Test  $2 \Box$  / Exam  $1 \Box$ 

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Name :

Number:

The duration of the test is 1h30m and of the exam is 3h. The score of each item in the test is 2 when right and -0.5 if wrong. In the exam the scores are half, 1 when right and -0.25 if wrong. Only one option can be selected in each question.

DON'T FORGET TO INDICATE THE QUESTIONS YOU ARE ANSWERING FOR THE SECOND TEST IN THE BOX LABELED WITH  $[T:\Box]$ . A maximum of 8 items are allowed in the test.

- 1.  $[\mathbf{T}:\Box]$  Let  $x(n) = [1, -1]^T$  and  $y = [j, -j]^T$ . What is the value of the inner product  $\langle x, y \rangle$ ?
  - □ a) 0
  - □ b) 2*j*
  - ■ c) -2*j*
  - $\Box$  d) None
- 2. [**T**: $\Box$ ] Let  $p = [p_1, p_2, ..., p_N]$  a complete, orthogonal and normed basis of a vector space S, were each element x can be obtained as the following linear combination,  $x = \sum_{i=1}^{N} c_i p_i$ . The norm of x in this space is
  - **a**) ||x|| = ||c||
  - $\Box$  b)  $||x|| \le ||c||$
  - $\Box$  c) ||x|| < ||c||
  - $\square$  d) None
- 3.  $[\mathbf{T}:\Box]$  Consider a flame intensity, x(t), captured by a photo diode (light sensor), x(t). If a 4096 length FFT is used to obtain the spectrum what should be the sampling rate used, form the list below, in the digitalization process to have a spectral resolution smaller (better) than 100Hz?

- $\square$  a) 100kHz
- □ b) 250kHz
- ■ c) 500kHz
- $\square$  d) None
- 4. [T: $\Box$ ] Consider the finite length sequence  $x(n) = \{1, 2, 3\}$  and the sequence  $y(n) = x((2-n)_5)$ . What is the value of y(3).
  - **a**) 0
  - 🗆 b) 1
  - □ c) 2
  - $\square$  d) None
- 5. [**T**: $\Box$ ] Consider the complex signal  $x(n) = \{1, 1 j, 0, 2 j, 3, -2 + j, 2j\}$ . What is the 8 length DFT value for  $k = 8, X_8(8)$ ?
  - ■ a) 5 + j
  - 🗆 b) 0
  - 🗆 c) 6
  - $\square$  d) None

6. [**T**: $\Box$ ] What is the period of the signal  $sin(0.01\pi n)$ ?

- **a**) 200
- $\square$  b) it is not periodic
- 🗆 c) 0.01
- $\square$  d) None
- 7. [T:□] The goal is to filter, in real time, an audio signal from a microphone with a 25 length impulse response FIR filter. The signal should be processed with a 500 sample length blocks and the convolution is performed by using a 512 length FFT algorithm. What is the number of overlapped samples of the input blocks?
  - □ a) 24.
  - **b**) 12.
  - □ c) 0.
  - $\square$  d) None

- 8. [**T**:□] With respect to the previous item how many processed samples should be added to the next output block if the corresponding invalid samples are set to zero?
  - **a**) 24.
  - 🗆 b) 12.
  - □ c) 0.
  - $\square$  d) None
- 9. [**T**: $\Box$ ] Let x(n) be a band limited signal, with a cut-off frequency of  $\omega_c = 3\pi/4$ . To change the sampling rate by a factor of R = 0.75,  $f_s(new) = Rf_s$ , where h is the impulse response of an ideal low-pass filter, what is the appropriated sequence of operations?
  - $\square$  a)  $T_{\uparrow 3}[T_{\downarrow 4}[h * x(n)]]$
  - $\square$  b)  $T_{\uparrow 3}[h * T_{\downarrow 4}[x(n)]]$
  - $\blacksquare$  c)  $T_{\downarrow 3}[h * T_{\uparrow 4}[x(n)]]$
  - $\square$  d) None
- 10. [**T**:□] What is the window that allows highest time frequency compression in the scope of time-frequency analysis?
  - **a**)  $e^{-t^2}/\sqrt{\pi}$ .
  - $\square$  b)  $e^{-t}/\sqrt{\pi}$ .
  - $\square$  c) sinc(t).
  - $\square$  d) None
- 11.  $[\mathbf{T}:\Box]$  Consider an infinite length discrete signal, x(n). For this signal there is/are
  - $\square$  a) a lower bound for scale resolution (coarser).
  - **b**) a upper bound for scale resolution (finer).
  - $\square$  c) no bounds for scale resolution.
  - $\square$  d) None

12. [**T**: $\Box$ ] What is the polyphase decomposition of  $H(z) = 1/(1 - 0.5z^{-1})$  for M=3?

- $\square$  a)  $1/(1 0.5z^{-1})$ .
- $\blacksquare$  b)  $(1 + 0.5z^{-1} + 0.25z^{-2})/(1 0.5^3z^{-3})$
- $\square$  c)  $z^{-2}/(1 0.5z^{-1})$
- $\Box$  d) None

- 13.  $[\mathbf{T}:\Box]$  A perfect reconstruction of a signal from the wavelet decomposition requires,
  - $\square$  a) the approximation and detail coefficients from all scales
  - ■ b) the detail coefficients from all scales and the approximation coefficients of the coarser scale
  - $\Box$  c) only the detail coefficients
  - $\square$  d) None
- 14.  $[\mathbf{T}:\Box]$  Consider a 1024 length discrete signal, x(n). How many decompositions scales is it possible to compute with a dyadic *discrete wavelet transform* (DWT)?
  - $\square$  a) 1024
  - 🗆 b) 102
  - **c**) 10
  - $\square$  d) None
- 15. [**T**: $\Box$ ] Consider the basis functions of the DFT,  $\phi_k(n) = e^{j\frac{2\pi}{N}kn}$ . What is the inner product  $\langle \phi_k(n), \phi_r(n) \rangle$ ? (where  $\delta(n)$  is the impulse and N is the length of the signals to analyse).
  - **a**)  $N\delta(k-r)$
  - 🗆 b) 0
  - □ c) N
  - $\square$  d) None
- 16.  $[\mathbf{T}:\Box]$  Consider the following prototype filter of a filter bank,

$$|H(\omega)| = \begin{cases} 1 - 4|\omega|/\pi & \text{if } |\omega| < \pi/4\\ 0 & \text{otherwise} \end{cases}$$
(1)

How many replicas of this filter are needed do maximally decompose a real discrete signal (since the signal and filter are real consider only the interval  $[0, \pi]$ )?

- 🗆 a) 6
- **b**) 5
- □ c) 4
- $\Box$  d) None

## Problems

Please present the details of your computations. For **test**, only one of these problem need to be solved.

1. **T**: $\Box$  Let x(n) be a N length real discrete signal and

$$y(n) = \begin{cases} x(n) & \text{if n is even} \\ 0 & \text{otherwise} \end{cases}$$
(2)

- (a) How are related the N length  $DFT_N$ s of x(n) and y(n)?
- (b) **T**: Let x(n) = [0, 1, 0, 2, 0, 3, 0, 4, 0, ....] be a 256-length discrete signal. What is the N length DFT of y(n),  $Y_{256}(k)$ ?
- 2. **T**:□Consider a *Linear Time Invariant* (LTI) filter described by the following difference equation

$$y(n) = x(n) + 0.5x(n-1) - 0.25y(n-1) + 0.5y(n-2)$$
(3)

- (a) Derive its transfer function, which is the Z-transform of its impulsive response, H(z) = TZ(h(n)).
- (b) Represent it graphically in the canonical *Direct Form II*, where the number of delay blocks is minimized.
- (c) Is this filter stable? Prove