ELL 319: Digital Signal Processing (Fall 2015)

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Minor Exam # 2

Time allowed: 60 mins

Justify your answer. 1. Let x[n] be a real-valued minimum phase sequence, then is $y[n] = (-1)^n x[n]$ a minimum phase sequence?

(2) X = [u] x 00 (Z-) X = (Z)/

simupse ready muminim as [n]x sinil (2/2)X = [4] h (2

(5) / 6 ords a ci o5-= 9 10 muth (5) x po and a oi os fe all poles and zeroes of X(2) are inside unit with

of 20 is a pole of X(2) And 2=0-26 is a pole of Y(2)

. for ocond win or 15) / (= 1) /2-1 00 E (2) < 1 = (-: x (5) in min - phone)

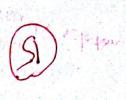
(25 points) $H(e^{10}) = 1$ at n = 0. rejects completely a frequency component at $\omega_0 = 2\pi/3$ and its frequency response is normalized so that 2. Determine the coefficients of a linear-phase FIR filter $y[n] = b_0x[n] + b_1x[n-1] + b_2x[n-2]$ such that it

Y(2) = 60 X(2) + 6,2 x(2) + 6,2 x (2)

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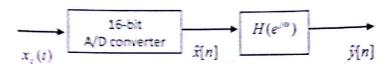
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5/1145 = w too

5/m/2 = m

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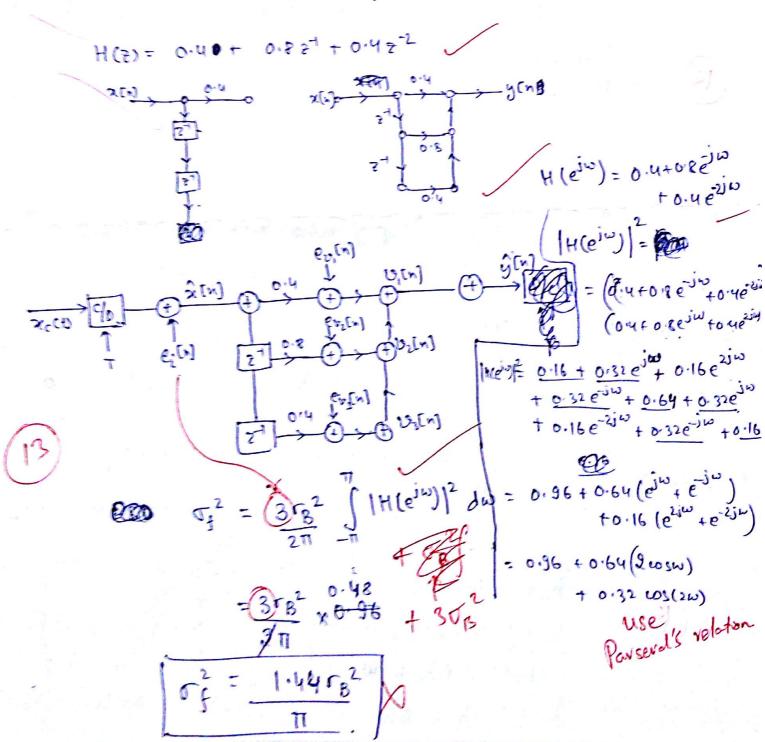
Consider the system shown below:



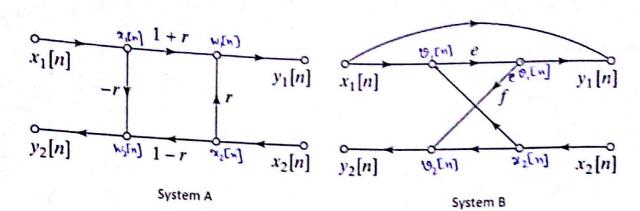
(25 points)

Impulse response of the digital filter is: $h[n] = 0.4\delta[n] + 0.8\delta[n-1] + 0.4\delta[n-2]$. Assume that the filter is implemented with 16-bit fixed point arithmetic (using two's complement) and the products are rounded to 16-bits before being accumulated to produce the output. Use the linear noise model to analyse this system.

- (a) Determine the maximum magnitude of $\hat{x}[n]$ such that no overflow can possibly occur in the implementing the digital filter.
- (b) Draw a detailed linear noise model for the complete system (including the A/D converter).
- (c) Determine the total noise power at the output σ_f^2 . (You can express the result in terms of σ_B^2)



4. Consider the two systems A and B shown below, each of which is a two-input, two-output system:

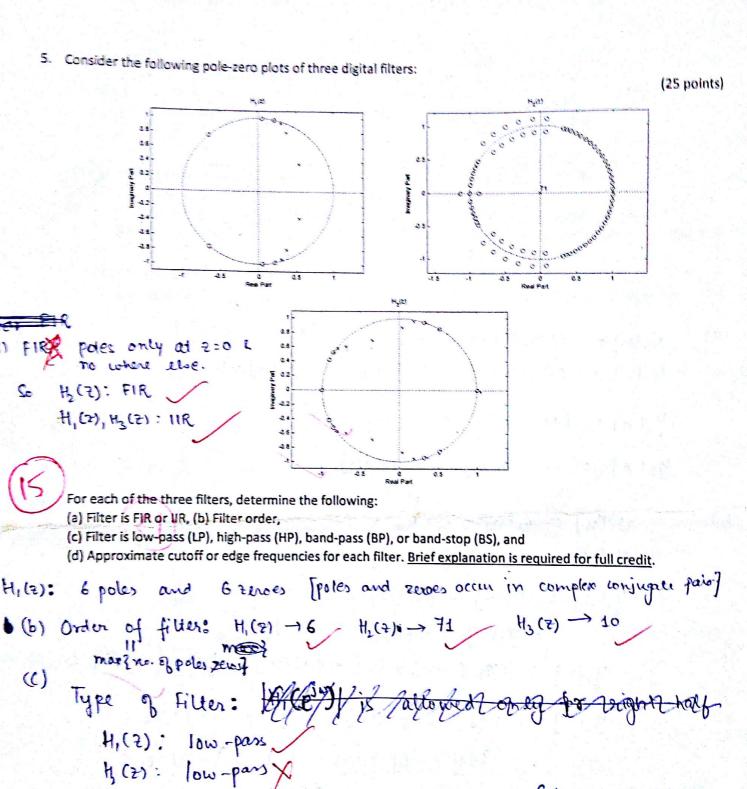


(a) Determine the difference equation for System A.

(20 points)

- (b) Determine the values of e and f for System B in terms of r such that the two systems are equivalent.
- (c) Which implementation structure might be preferred? Give justification for your answer.

(a)
$$w_1[n] = (1+r)x_1[n] + rx_2[n]$$
 $w_1[n] = y_1[n]$
 $w_1[n] = (1+r)x_1[n] - rx_1[n]$
 $y_2[n] = (1+r)x_1[n] + rx_2[n]$
 $y_2[n] = -rx_1[n] + (1+r)x_2[n]$
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(1) High -pass (does not allow freq - p. zero at 2=1)