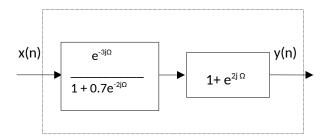
TEST 1 PART B

For the causal filter below



- a. Write the overall difference equation(show the equation clearly and define coefficients)
- b. Give the transfer function of the system (show the equation clearly)
- c. Compute with pencil and paper the impulse response in closed form
- d. Use MATLAB to determine steady state response due to x(n)=u(n)
- e. Write a MATLAB program to compute and plot the frequency response of the overall system. Give plots in dB and the program.

(20pts)

For the following causal filter:

(15 pts)

$$y(n)=x(n)-x(n-2)+1.2y(n-1)-0.36y(n-2)$$

- a) Determine the transfer function.
- b) Find the poles and zeros.
- c) Use the residue theorem for the inverse z-transform and calculate in <u>closed form</u> the transient and steady state response of the filter below due to the input x(n)=u(n).

Exercise 3 - This small programming exercise is on a simple linear prediction of speech and its utility in estimating the frequency response of the mouth cavity (vocal tract). You need to use the file *cleanspeech* and apply frame by frame processing. Identify voiced (vowel) frames in speech (those that have strong periodicity).

Apply the function **lpc** in MATLAB first on a voiced (vowel) frame. The lpc will provide to you coefficients that you can use to form an *all-pole* transfer function. The order of this transfer function should be 10 (hence you need 10 lpc coefficients). The frame size that you use should be N=256.

For a randomly chosen strongly voiced frame, Use MATLAB and Provide

- A time domain plot and an FFT spectral plot (in dB)
- A plot of the *all pole* filter magnitude response in dB
- Provide a pole zero plot for the all pole filter
- Provide the frequency in HZ of the first peak of the all pole filter (ignore peaks at DC)

Determine three 256-point frames. One perfectly periodic, one that looks random (but not silence), and one mixed (somewhat periodic but not perfectly periodic)

- Apply the lpc function for N=256 on all three frams. For each frame, Provide the time domain, the FFT (in dB) and the all-pole frequency response in dB.

To get full credit, label your graphs carefully and provide 1-2 sentences to explain each result.

Do your own work. Info in your text book on speech processing should be helpful. You can also use resources from MathWorks.