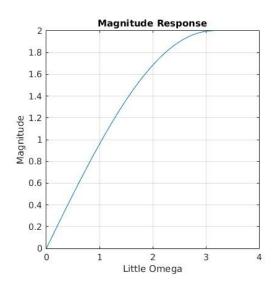
Lab 4 Report

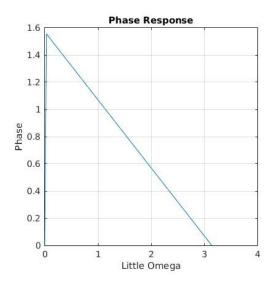
Wenqian Ye NetID: wenqian3

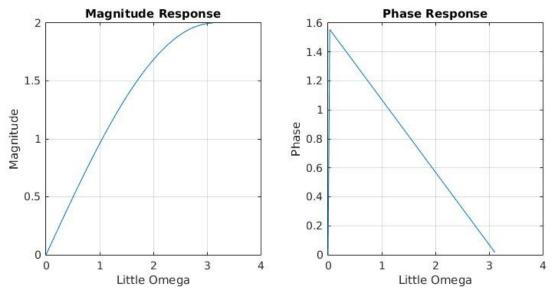
• Report item 0

1. [h,w] = freqz(b,a,n) returns the n-point frequency response vector, h, and the corresponding angular frequency vector, w, for the digital filter with numerator and denominator polynomial coefficients stored in b and a, respectively. It takes in 3 arguments b - numerator coefficients in the transfer function, a - denominator coefficient in the transfer function, n - the number of evaluation points that we desire between 0 and pi. The output is the frequency response.

2 . The first file approaches using z-transform to DFT, and the second file approaches using DTFT.

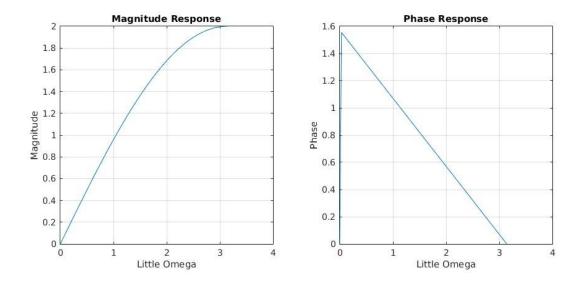






3. The two figure look the same. But for the w > 3.1102, the first figure miss this part. The reason is that for the first program, the ff only goes till 3.1102, so the plot is end when w = 3.1102.

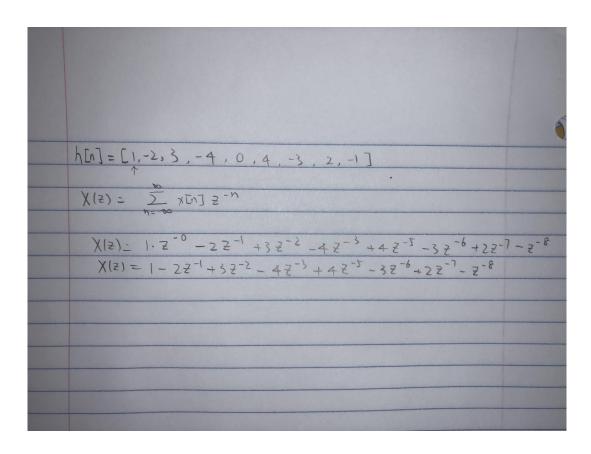
4.



5. P = angle(Z) returns the phase angles, in radians, for each element of complex array Z. The angles lie between $\pm \pi$.

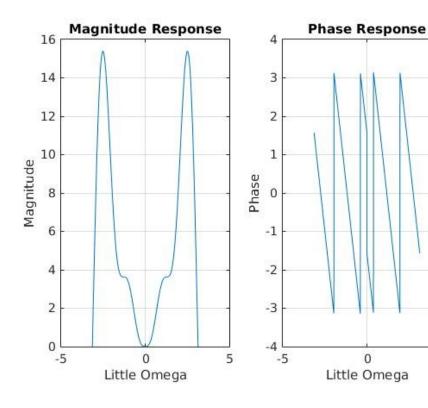
Y = phase(X) returns the phase angles of a signal, in radians, the phase lie between -infinite to +infinite.

• Report item 1 (1).

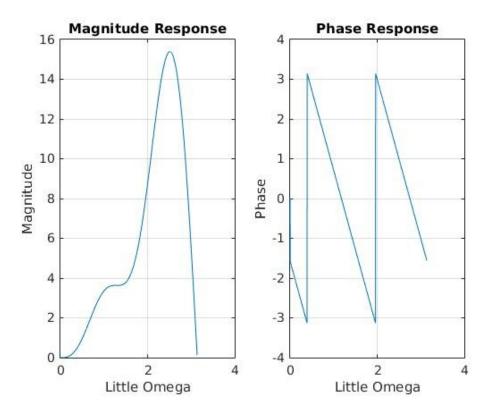


$$\begin{split} H(z) &= 1 - 2z - 1 + 3z - 2 - 4z - 3 + 4z - 4 - 3z - 6 + 2z - 7 - z - 8 \\ Hd(\omega) &= 1 - 2e^{(-j\omega)} + 3e^{(-2j\omega)} - 4e^{(-3j\omega)} + 4e^{(-4j\omega)} - 3e^{(-6j\omega)} + 2e^{(-7j\omega)} - e^{(-8j\omega)} \end{split}$$

(2).



(3).



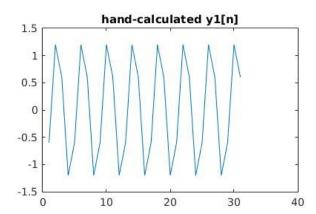
5

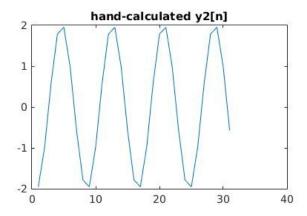
The responses are the same. The freq() are wrapped between -pi and pi, but looks like the same as (3).

• Report Item 2

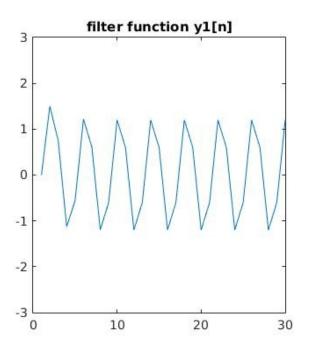
(1)

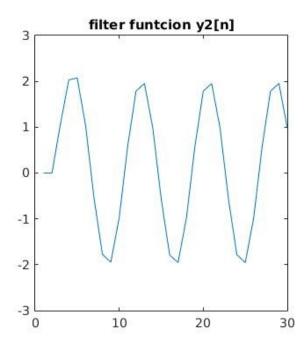
	8= 1	SEX
2, a)	y(n) = 0.5y[n-1] + 0.5x[n-1] H(Z) = 1 - 0.5z-1 H(W) =	0.5e-jw (-0.5e-jw
	y,[n] = Hd(w) 3 cog(xn/2 + LHd(w)) y_2[n] = Hd(w) 3 sin(xn/2 + LHd(w))	Hd(====================================
	$y_{1} = 3 \int_{0.2}^{0.2} \cos(\frac{\lambda}{2}n + 243, 43)^{0}$ $y_{2} = 3 \int_{0.2}^{0.2} \cos(\frac{\lambda}{2}n + \frac{3\lambda}{2})$	19 3
	1000000	



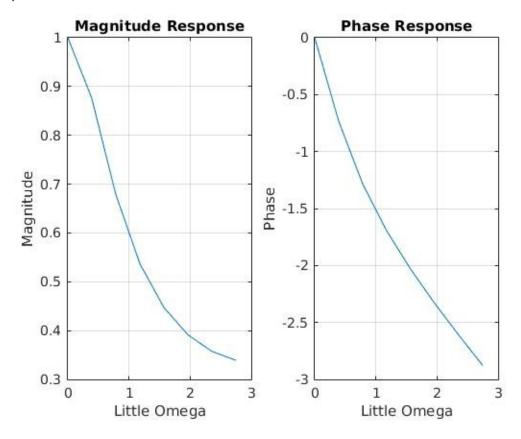


(3)



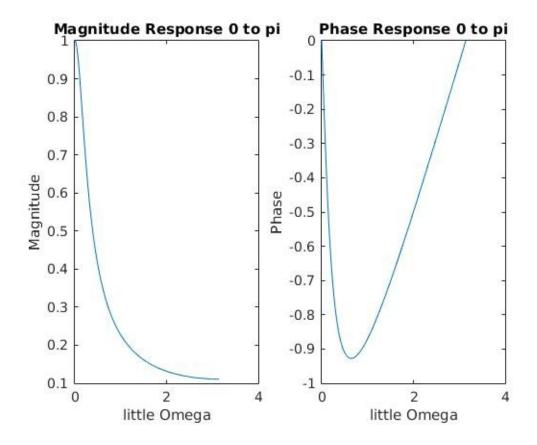


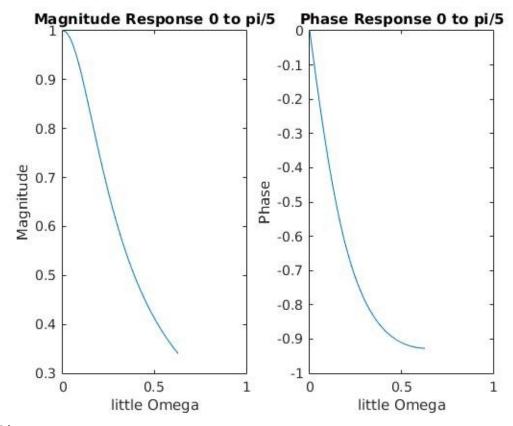
(4)



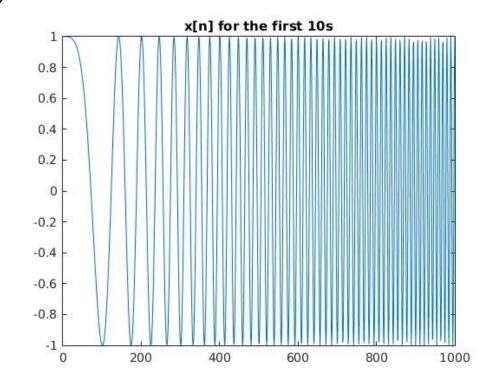
(5). (2) and (3) are different. They have the same shape but looks like both y1[n] and y2[n] computed from hand is shifted right relative to those from filter function.

• Report Item 3 (1)

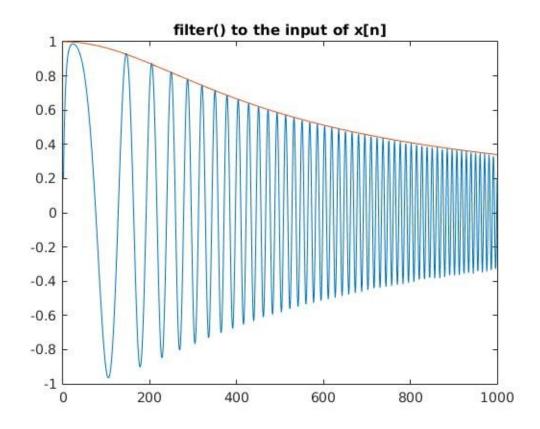






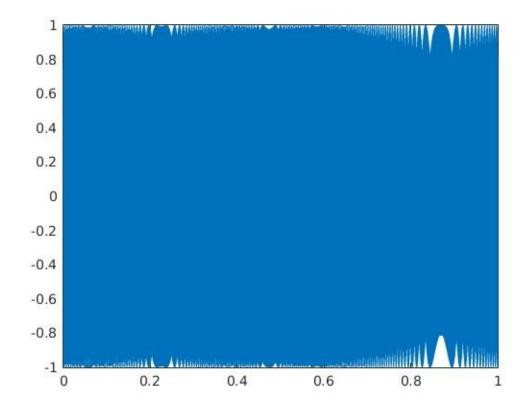


(3).



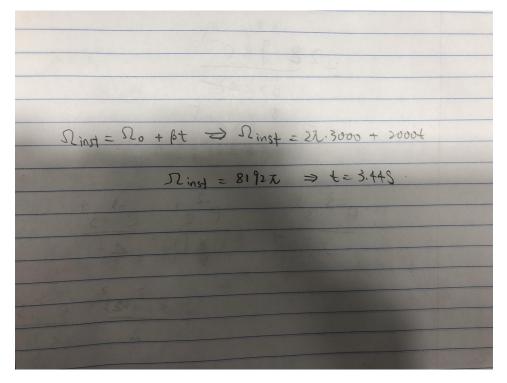
Each instantaneous frequency at time sample n is equal to the frequency at index n of the frequency response, making this valid.

• Report Item 4

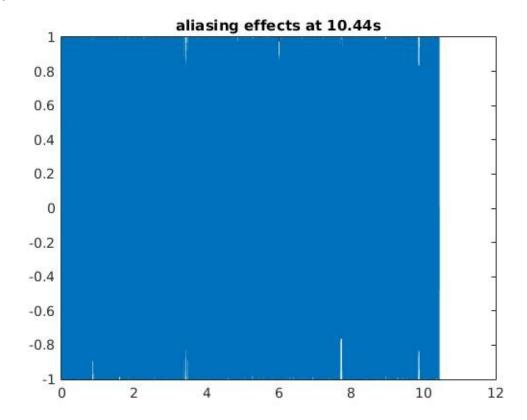


(2).I heard a high frequency increasing in pitch

(3).



(4).



I hear some decreasing frequencies.