Convolution

- 1. Assume x(n) = u(n)-u(n-10) and h(n)= an u(n), with a=0.5.
- a) Find the theoretical closed-form expression for y(n) = x(n)*h(n)
- b) Plot h(n)
- c) Plot x(n) and y(n).

answer)
$$y[n] = \sum_{k=-\infty}^{k=+\infty} x[k]h[n-k]$$
 or $y[n] = \sum_{k=-\infty}^{k=+\infty} h[k]x[n-k]$

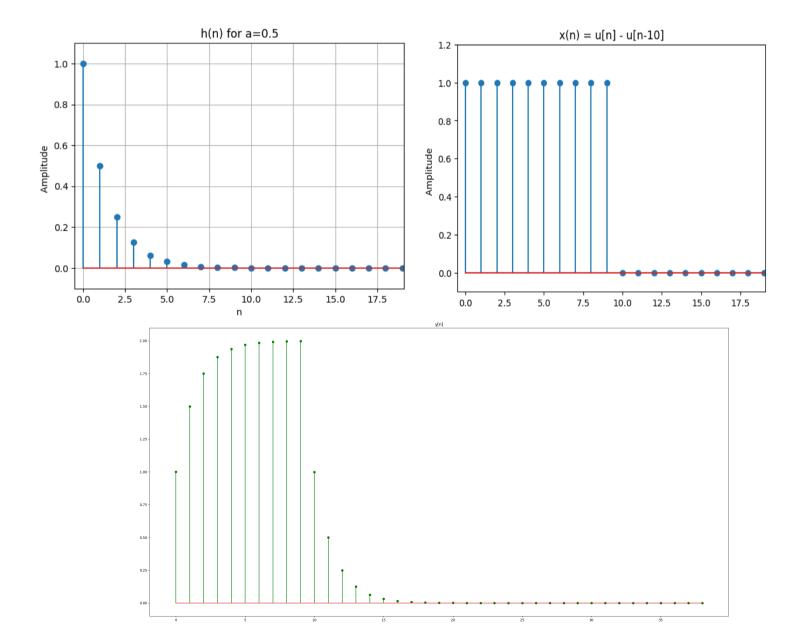
based on eq1 : $y[n] = \sum_{k=-\infty}^{k=+\infty} \infty^k u[k]x[n-k] = \sum_{k=0}^{k=n} \infty^k$ (Partial overlap is starting from 0 to n=N-1=9)

for $n \le N - 1$ (n ≤ 9), the sequence is finite

$$y[n] = \sum_{k=0}^{k=n} \alpha^k = \frac{(\alpha)^0 - (\alpha)^{n+1}}{1 - \alpha} = \frac{1 - \alpha^{n+1}}{1 - \alpha} \to 0 \le n \le 9$$

for $n \ge 10$

$$y[n] = \sum_{k=9}^{k=n} \alpha^k = \frac{(\alpha)^{n-9} - (\alpha)^{n+1}}{1-\alpha} = \alpha^{n-11} \left(\frac{1-\alpha^{10}}{1-\alpha}\right) \rightarrow n \ge 10$$



Difference Equation

- 2. Let y(n) 0.4y(n-1) = x(n) and assume y(n) = 0 when n < 0.
 - a) Find the impulse response.
 - b) Using $x(n) = \delta(n) + \sin(0.08\pi n) u(n)$, find and plot y(n). Hint: use recursive (iterative) computation to solve the difference equation instead of the impulse response.
 - c) What can you expect when n is close to zero?
 - d) What can you expect when n is very large?

a)

$$x[n] = \delta[n]$$
 and $y[n] = h[n]$

rewrite the problem equation:

$$h[n] - 0.4h[n-1] = \delta[n] \rightarrow h[n] = \delta[n] + 0.4h[n-1]$$
 we know $h[-1] = h[-2] = ... = 0$

$$n = 0$$
 $h[0] = \delta[0] + 0.4 h[0-1] \approx 1$

$$n = 1$$
 $h[1] = \delta[1] + 0.4 h[0] \approx 0.4$

$$n = 2$$
 $h[2] = \delta[2] + 0.4 h[1] \approx 0.16$
0 0.4

b)
$$y[n] = 0.4 y[n-1] + \delta[n] + \sin[0.8\pi n] u[n]$$

n=0
$$y[0] = 0.4 y[-1] + \delta[0] + \sin[0.8\pi 0] u[0] \approx 1$$

n=1
$$y[1] = 0.4 y[0] + \delta[1] + \sin[0.8\pi] u[1] \approx 0.64$$

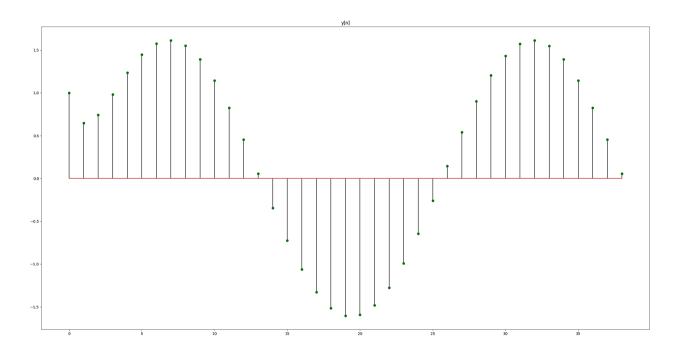
0.4 0 0.24 1

n=2
$$y[1] = 0.4 y[0] + \delta[1] + \sin[0.8\pi] u[1] \approx 0.736$$

0.64 0 0.48 1

$$y[n] = \{ ..., 1, 0.64, 0.736, ... \}$$

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n: 1, y[n] \Rightarrow 0.6487
                               n: 2, y[n] \Rightarrow 0.7412
                                                                n: 3, y[n] => 0.9810
n: 4, y[n] \Rightarrow 1.2367
                               n: 5, y[n] \Rightarrow 1.4458
                                                                n: 6, y[n] => 1.5763
                               n: 8, y[n] \Rightarrow 1.5500
n: 7, y[n] \Rightarrow 1.6128
                                                                n: 9, y[n] \Rightarrow 1.3905
n: 10, y[n] \Rightarrow 1.1440 n: 11, y[n] \Rightarrow 0.8257
                                                                n: 12, y[n] \Rightarrow 0.4556
n: 13, y[n] \Rightarrow 0.0569
                               n: 14, y[n] \Rightarrow -0.3454 n: 15, y[n] \Rightarrow -0.7259
n: 16, y[n] \Rightarrow -1.0609 \text{ n: } 17, y[n] \Rightarrow -1.3292 \text{ n: } 18, y[n] \Rightarrow -1.5140
n: 19, y[n] \Rightarrow -1.6036 \text{ n}: 20, y[n] \Rightarrow -1.5925 \text{ n}: 21, y[n] \Rightarrow -1.4813
n: 22, y[n] \Rightarrow -1.2771 \text{ n: } 23, y[n] \Rightarrow -0.9926 \text{ n: } 24, y[n] \Rightarrow -0.6457
n: 25, y[n] \Rightarrow -0.2583 n: 26, y[n] \Rightarrow 0.1454
                                                                n: 27, y[n] => 0.5399
                                                                n: 30, y[n]
n: 28, y[n] => 0.9005
                               n: 29, y[n] \Rightarrow 1.2045
n: 31, y[n] \Rightarrow 1.5712
                               n: 32, y[n] \Rightarrow 1.6108
                                                                n: 33, y[n] \Rightarrow 1.5491
n: 34, y[n] \Rightarrow 1.3902
                               n: 35, y[n] => 1.1439
                                                                n: 36, y[n] => 0.8257
n: 37, y[n] => 0.4556
                               n: 38, y[n] => 0.0569
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• c, d) If n is near zero, we expect to see behaviour like a step function, and when n is going infinity, we expect to see a periodic behaviour of the y[n].

3. The ideal lowpass filter in Example 2.18 is not implementable. A simple approximation can be achieved by truncating the impulse response. Assume a noncausal L-tap (L is odd) FIR filter is obtained by such a truncation. To make this filter causal, one can shift the impulse response to the right by (L-1)/2 samples, giving

$$h_{FIR}(n) = \frac{\sin[\omega_C(n - \frac{L-1}{2})]}{\pi(n - \frac{L-1}{2})}$$
, $0 \le n \le L-1$.

- a) Assuming ω_C =0.2 π , compute and plot the amplitude and phase responses of this causal FIR filter for L=19 and L=101.
- b) What are the advantages and disadvantages of increasing L?
- c) Plot the amplitude and phase responses of $(-1)^n h_{FIR}(n)$ and compare with the results in a).

Answer:

a)

Truncation of the impulse response equals multiplying h[n] or shifted by a rectangular window

$$\begin{array}{ccc} \mathsf{DTFT} \\ h(n) & \leftrightarrow & H(w) \end{array}$$

Frequency response:
$$H(w) = H_r(w)e^{j\theta(w)}$$
Real part

Symmetric impulse response with M, which is an odd value and defined by the following amplitude and phase response formulas:

$$H_r(w) = h\left(\frac{M-1}{2}\right) + \sum_{n=0}^{\frac{M-3}{2}} 2h(n)\cos(w\left[n - \frac{m-1}{2}\right])$$

$$\sphericalangle H_r\left(w\right) = \begin{cases} -w\left(\frac{M-1}{2}\right) & ; H_r\left(w\right) > 0\\ -w\left(\frac{M-1}{2}\right) + \pi & ; H_r\left(w\right) < 0 \end{cases}$$

Now, we can initial constants by given values in the aforementioned previous page formulas:

$$M = 19$$
, $w_c = 0.2\pi$

$$H_r(w) = h(9) + \sum_{n=0}^{8} 2h(n) \cos(\frac{\pi}{5}[n-9])$$

⇓

amplitude:

$$H_r(w) = \frac{\sin\left[\frac{\pi}{5}\times(9-9)\right]}{\pi(9-9)} + \sum_{n=0}^{8} 2\frac{\sin\left[\frac{\pi}{5}\times(n-9)\right]}{\pi(n-9)}\cos\left(\frac{\pi}{5}\left[n-9\right]\right)$$

$$0$$

$$H_r(w) = \frac{1}{\pi}\sum_{n=0}^{8} 2\frac{\sin\left[\frac{\pi}{5}\times(n-9)\right]}{(n-9)}\cos\left(\frac{\pi}{5}\left[n-9\right]\right) \implies (\text{Based on } \sin 2\theta = 2\sin\theta\cos\theta)$$

$$= \frac{1}{\pi}\sum_{n=0}^{8} \frac{\sin 2\left[\frac{\pi}{5}\times(n-9)\right]}{(n-9)} = 0 + \frac{1}{\pi}\sum_{n=1}^{8} \frac{\sin\left[\frac{2\pi}{5}\times(n-9)\right]}{(n-9)} \implies (\text{Based on } \sum_{k=1}^{n}k = \frac{1}{2}n(n+1))$$

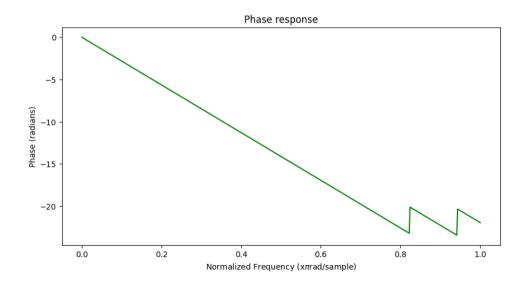
$$n=1 \rightarrow \frac{1}{\pi}\frac{\sin\left[-\frac{2\pi}{5}\times(8)\right]}{-8} = 0.00676 \quad |n=2 \rightarrow \frac{1}{\pi}\frac{\sin\left[-\frac{2\pi}{5}\times(7)\right]}{-7} = 0.00695$$

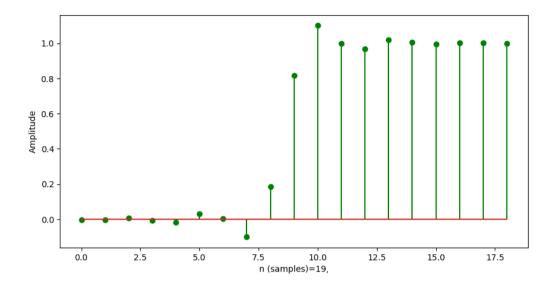
$$n=3 \rightarrow \frac{1}{\pi}\frac{\sin\left[-\frac{2\pi}{5}\times(6)\right]}{-6} = 0.00696, \dots n=8 \rightarrow \frac{1}{\pi}\frac{\sin\left[-\frac{2\pi}{5}\times(1)\right]}{-1} = 0.02193$$

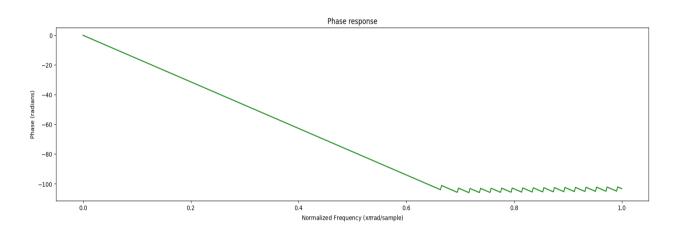
phase:

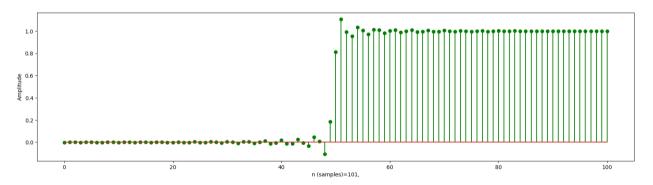
$$\sphericalangle H_r\left(w\right) = \begin{cases} \frac{-9\pi}{5} & ; H_r\left(w\right) > 0\\ \frac{-9\pi}{5} + \pi & ; H_r\left(w\right) < 0 \end{cases}$$

We should precisely repeat this process for M=101 with just different window sizes. In the following corresponding plots are showing.









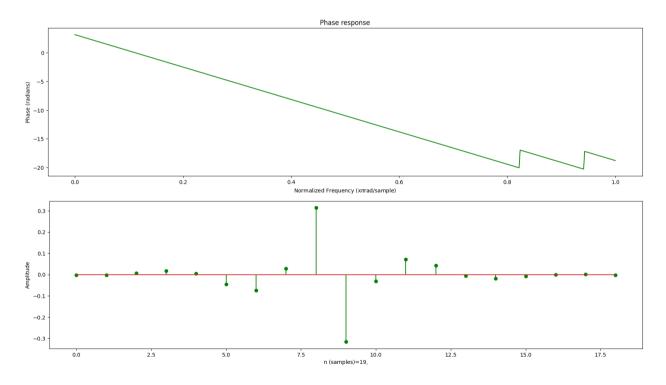
Advantages:

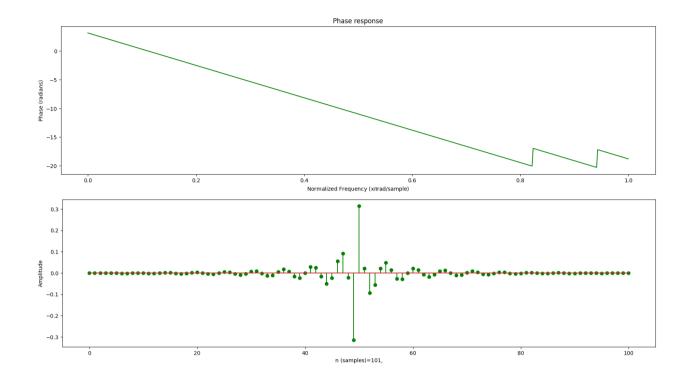
- This method doesn't need huge computational recourses, so it's the fastest technique in designing an FIR filter, but it's not the best.
- We receive a linear-phase frequency corresponding to delay. Therefore, we expect a smoother transition band and ripples than the ideal filter.
- o Picking good coefficients improves filter performance

Disadvantages:

- O The system has a delay of $\frac{L-1}{2}$ samples. For example, at n=0, the system will not react until the value $\frac{L-1}{2}$. So, in some real-time applications, it may cause a problem.
- Other drawbacks include: inefficiency unequal perturbation in ripples difficulty in choosing cut-off frequency.
- Selecting a window needs a trade-off between lobe width and peak side lobe amplitude and width of the transition band.
- o As L increases, the main lobe window decreases

c)





- As a comparison, more delay ends up having more zero response magnitude and very narrow band-pass windowing (range), but there is no effect on phase response.