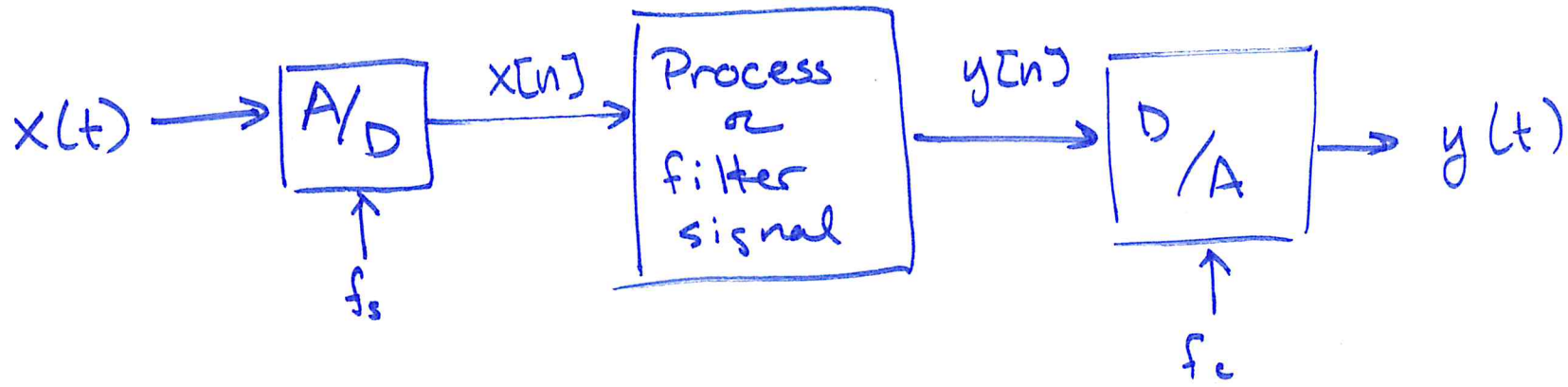


# Lecture 15: Digital Signal Processing Systems

①



Ex:

- $x(t)$  = music
- $x[n]$  = sampled music
- filter = dft + scaling (bass boost)
- $y[n]$  = samples w/ boosted bass
- $y(t)$  = music (played w/ soundsc)

Generally, a filter is a device or process for removing or modifying frequency components of a signal.

### Simple examples of filters

$$y[n] = x[n]^3$$

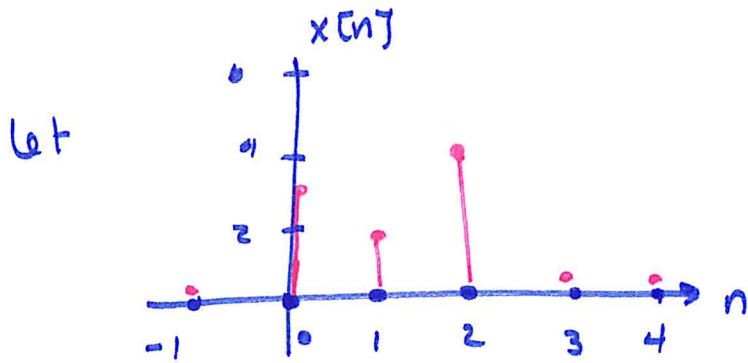
$$y[n] = \text{ifft}(\text{fft}(x[n]) \cdot * \text{weights}) \quad \leftarrow$$

$$y[n] = \max \{x[n], x[n-1], \dots, x[n-4]\}$$

$$y[n] = \frac{1}{2}x[n] + \frac{1}{3}x[n-1] + \frac{1}{5}x[n-2] \quad \leftarrow$$

Moving average filter :  $y[n] = \text{average of past + input samples}$ .

e.g.  $y[n] = \frac{1}{3} x[n] + \frac{1}{3} x[n-1] + \frac{1}{3} x[n-2]$



; what is  $y[n]$ ?

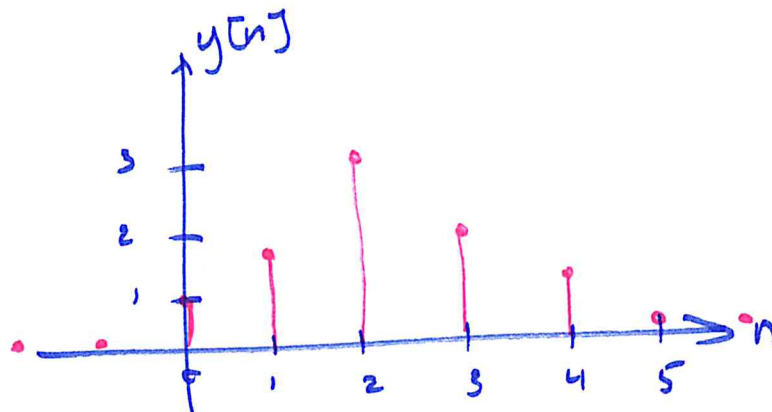
$$y[0] = \frac{1}{3} x[0] + \frac{1}{3} x[-1] + \frac{1}{3} x[-2] = \frac{1}{3} \cdot 3 = 1$$

$$y[1] = \frac{1}{3} x[1] + \frac{1}{3} x[0] + \frac{1}{3} x[-1] = \frac{1}{3} (2) + \frac{1}{3} (3) = \frac{2}{3}$$

$$y[2] = \frac{1}{3} x[2] + \frac{1}{3} x[1] + \frac{1}{3} x[0] = \frac{1}{3} \cdot 9 = 3$$

$$y[3] = \frac{1}{3} x[3] + \frac{1}{3} x[2] + \frac{1}{3} x[1] = \frac{1}{3} (4) = \frac{4}{3}$$

$$y[4] = \frac{1}{3} x[4] = \frac{1}{3}$$



# Finite Impulse Response (FIR) filters

FIR filters = weighted moving average filter

$$y[n] = \sum_{k=0}^M b_k x[n-k] = b_0 x[n] + b_1 x[n-1] + b_2 x[n-2] + \dots + b_M x[n-M]$$

e.g. Moving average filter from pg 3,

$$M = 2 \quad b_0 = b_1 = b_2 = 1/3$$

## Impulse Response:

impulse signal (delta function)  $S[n] = \begin{cases} 1 & \text{if } n=0 \\ 0 & \text{if } n \neq 0 \end{cases}$



impulse response of a filter is filter output if input is impulse signal



An FIR filter is entirely characterized by its response to an impulse. (5)

That is, if we know  $h[n]$ , then we can figure out how the filter will respond to any signal  $x[n]$ .

Ex: impulse response ~~to~~ of moving average filter?

$$\begin{aligned} y[n] &= \frac{1}{3} x[n] + \frac{1}{3} x[n-1] + \frac{1}{3} x[n-2] \\ &= \sum_{k=0}^2 \frac{1}{3} x[n-k] \Rightarrow \underline{b_k} = \begin{cases} \frac{1}{3} & \text{if } k=0, 1, 2 \\ 0 & \text{otherwise} \end{cases} \end{aligned}$$

to get impulse response, set  $x[n] = \delta[n]$

$$\underline{h[n]} = \frac{1}{3} \delta[n] + \frac{1}{3} \delta[n-1] + \frac{1}{3} \delta[n-2]$$

In general, impulse response of FIR filter is

$$h[n] = \sum_{k=0}^M b_k \delta[n-k] = \begin{cases} 0 & \text{if } n < 0 \\ b_0 & \text{if } n = 0 \\ b_1 & \text{if } n = 1 \\ b_2 & \text{if } n = 2 \\ \vdots & \\ b_M & \text{if } n = M \\ 0 & \text{if } n > M \end{cases}$$

only nonzero for  $k=n$

$$b_k = h[k] \text{ for } k = 0, 1, \dots, M$$

So we can calculate

$$y[n] = \sum_{k=0}^M h[k] x[n-k]$$

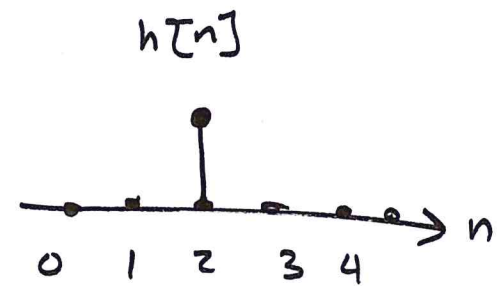
This type of sum is called a convolution of  $h$  and  $x$ .

We often write this as

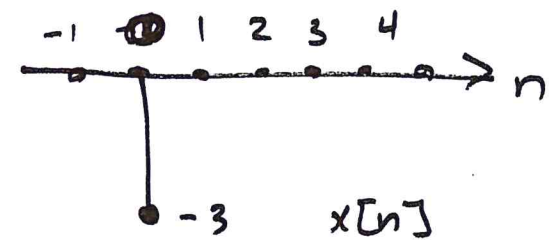
$$y = h * x$$



Ex: let  $h[n] = \begin{cases} 1 & \text{if } n=2 \\ 0 & \text{otherwise} \end{cases}$   
 "delay filter"



let  $x[n] = \begin{cases} -3 & \text{if } n=0 \\ 0 & \text{otherwise} \end{cases}$



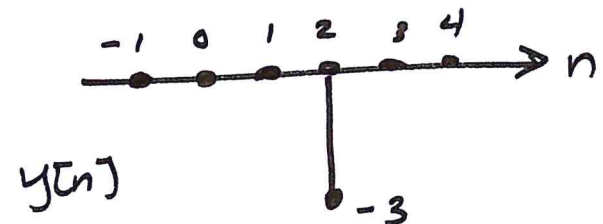
Compute  $y = h * x$ .

$y[0] = \sum_{k=-\infty}^{\infty} h[k] x[0-k] = h[0] \cdot x[0] = 0 \cdot -3 = 0$   
*only has value when  $k=0$*

$y[1] = \sum_{k=-\infty}^{\infty} h[k] x[1-k] = h[1] \cdot x[0] = 0$   
*only has value when  $k=1$*

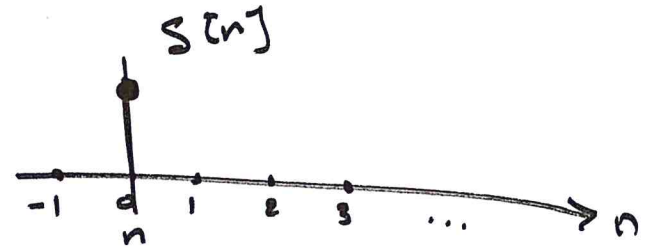
$y[2] = \sum_{k=-\infty}^{\infty} h[k] x[2-k] = h[2] \cdot x[0] = 1 \cdot -3 = -3$   
*value when  $k=2$*

$y[3] = \sum_{k=-\infty}^{\infty} h[k] x[3-k] = 0$   
*value when  $k$*

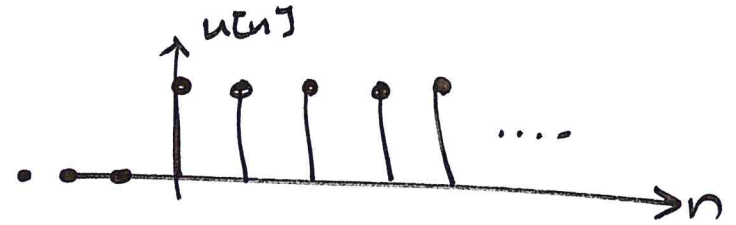


# Two important discrete-time signals

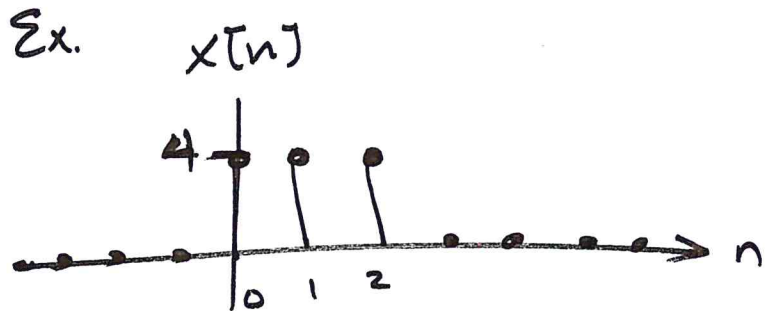
Impulse : 
$$s[n] = \begin{cases} 1 & \text{if } n=0 \\ 0 & \text{otherwise} \end{cases}$$



Step : 
$$u[n] = \begin{cases} 1 & \text{if } n \geq 0 \\ 0 & \text{otherwise} \end{cases}$$



Impulse + Step functions are building blocks we can use to express ~~any~~ any signal.



$$\begin{aligned} x[n] &= 4s[n] + 4s[n-1] + 4s[n-2] \\ &= 4u[n] - 4u[n-3] \end{aligned}$$