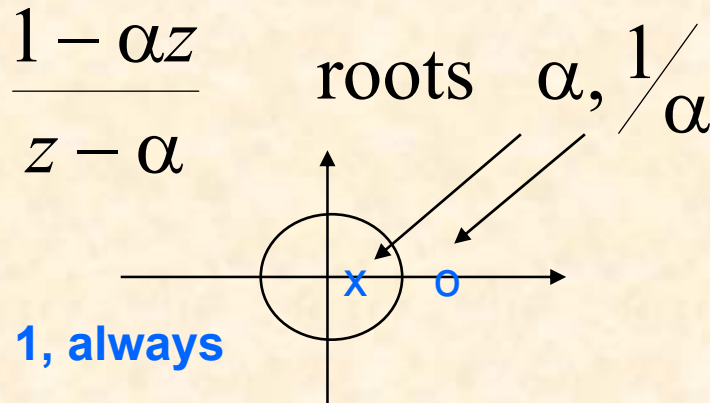
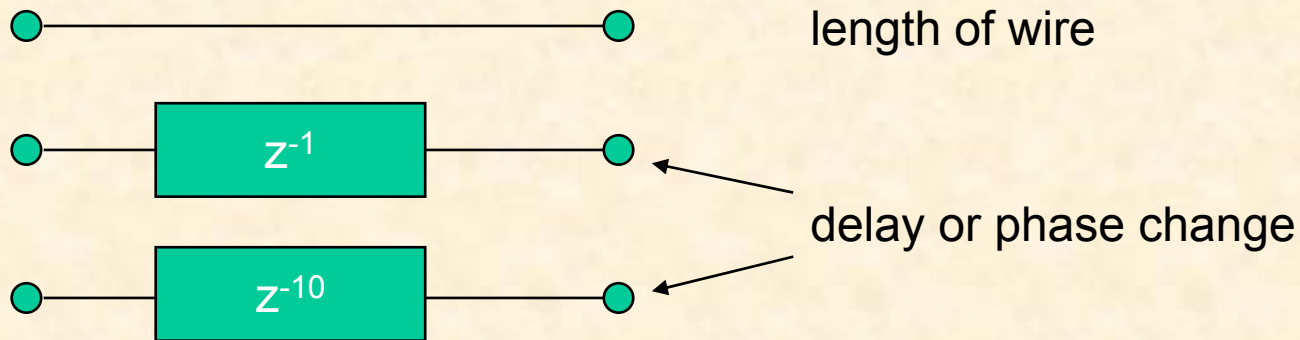
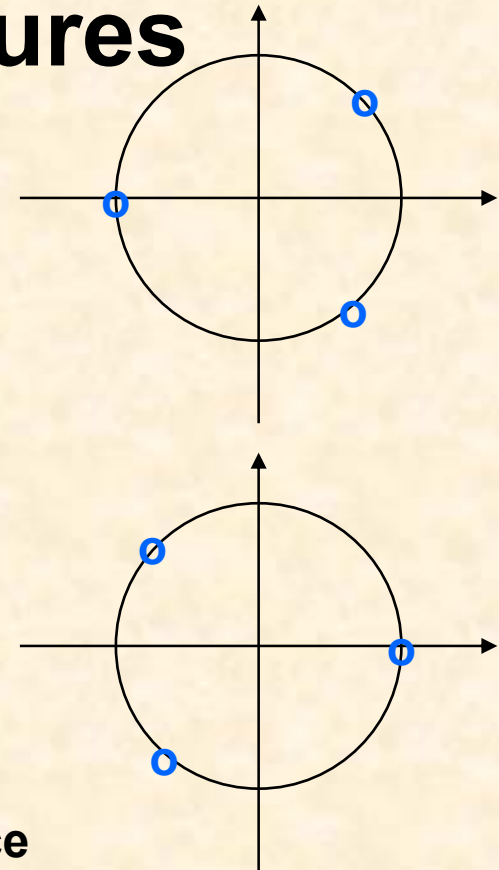
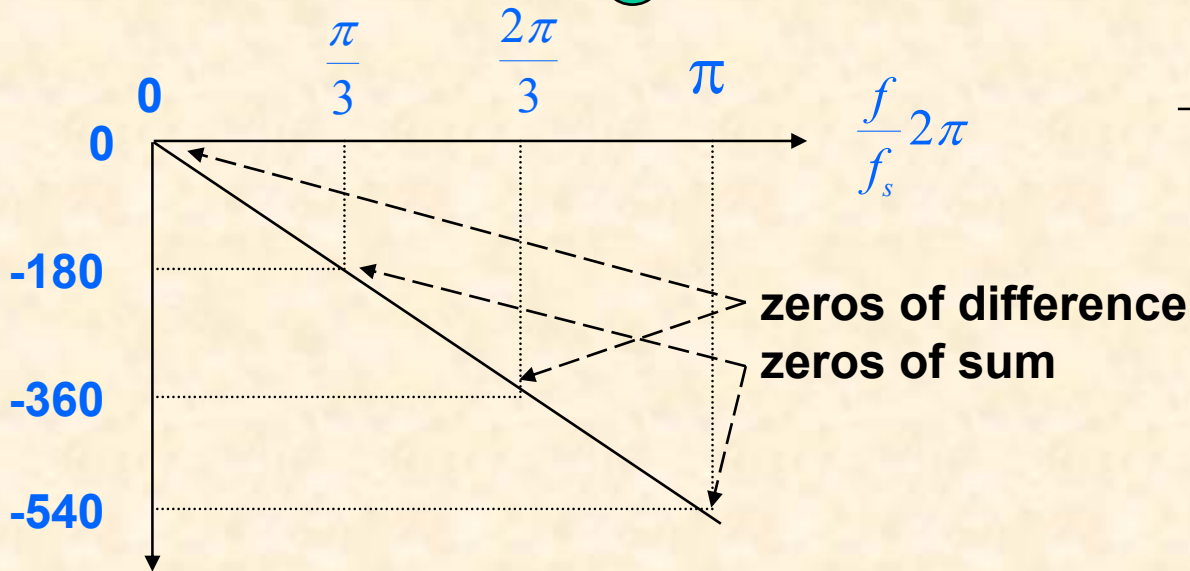
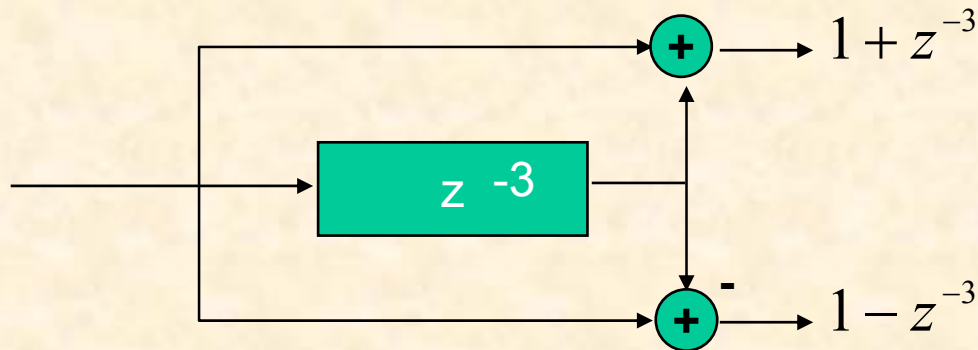


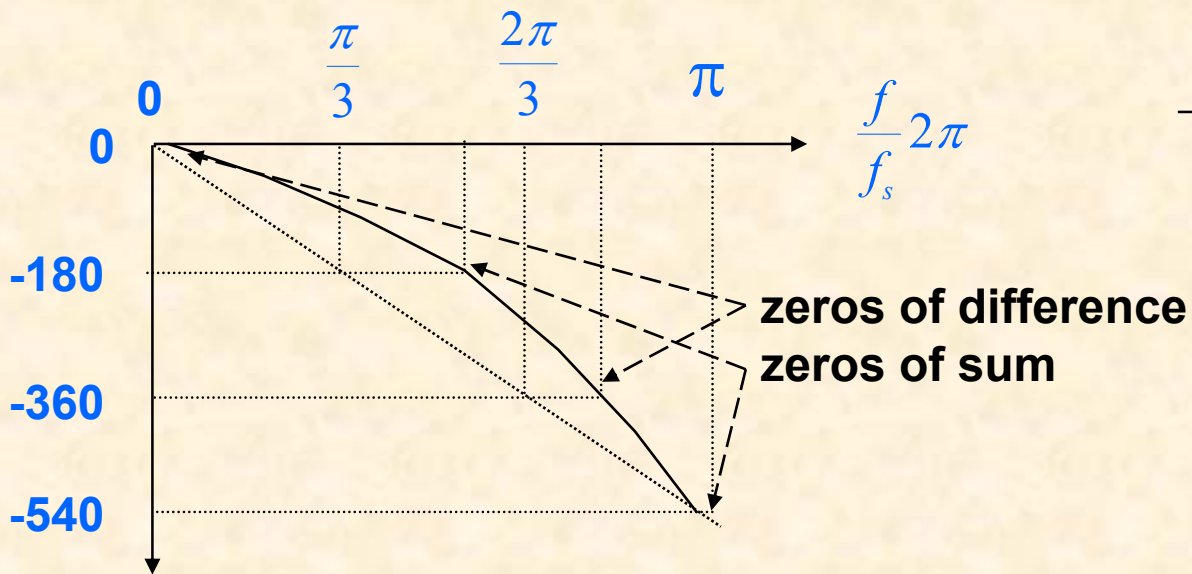
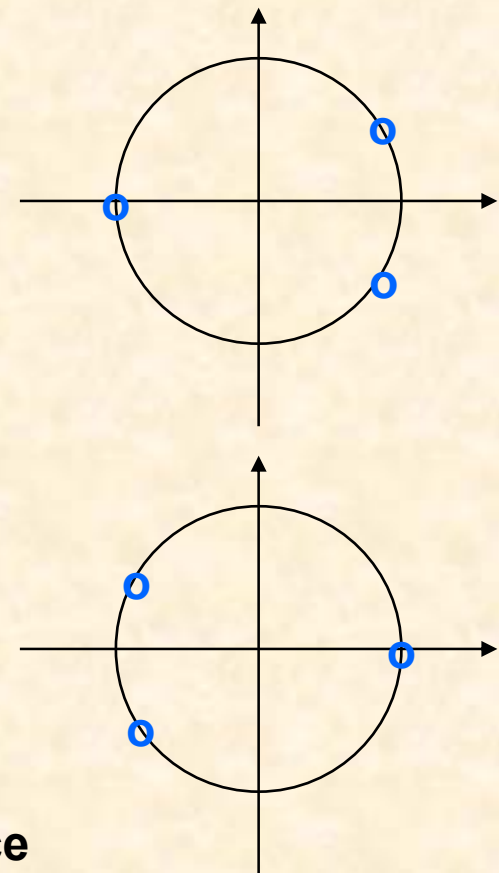
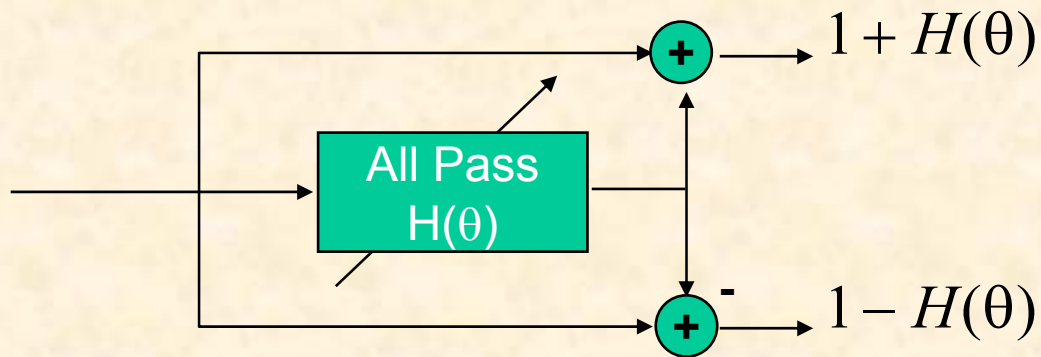
# All Pass Filter Structures



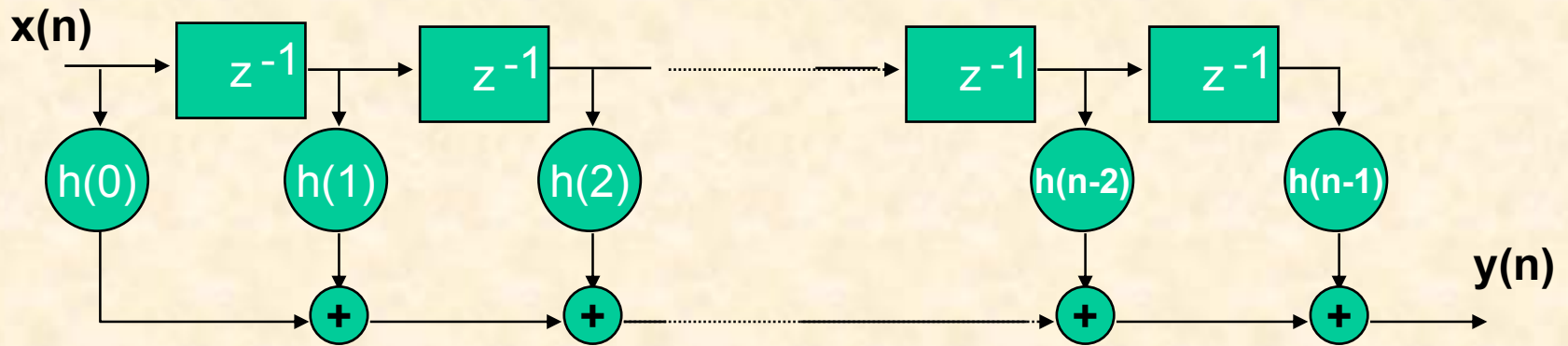
Filter works because of destructive interference due to delays ( $z^{-1}$ )  
 Why not design filter with no coefficients, only delays?

# All Pass Filter Structures





# FIR Filter Structure



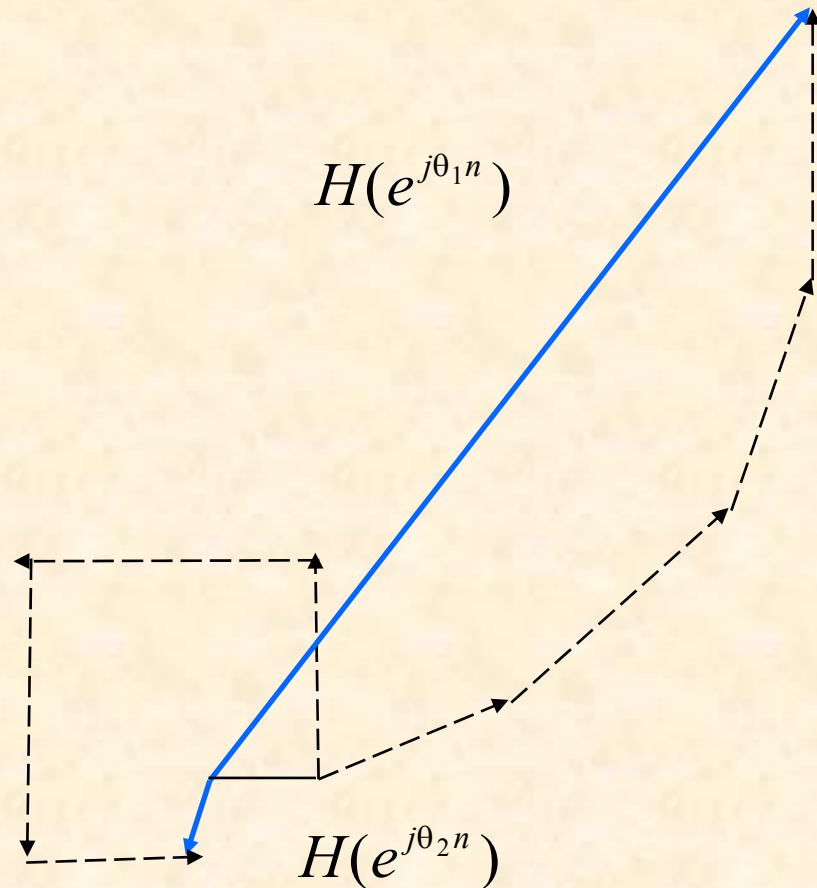
$$y(n) = \sum_{k=0}^{N-1} x(n-k)h(k)$$

$$Y(z) = X(z)H(z)$$

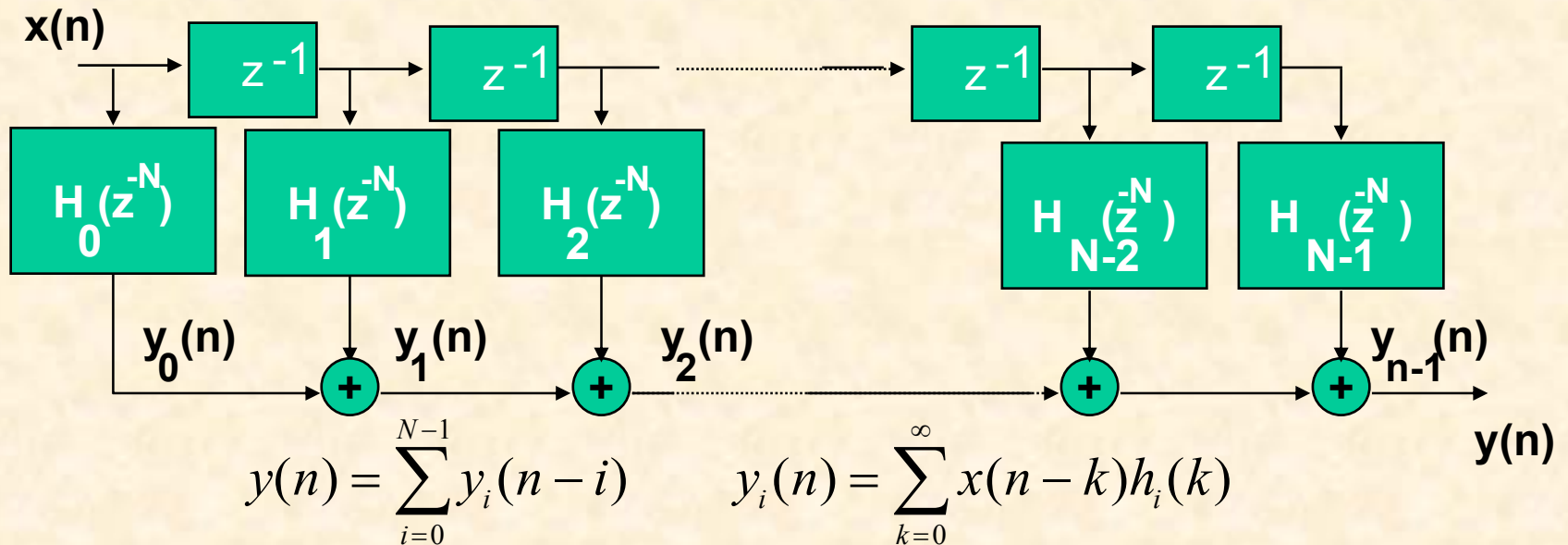
$$H(z) = \sum_{n=0}^{N-1} h(k)z^{-n}$$

$$H(\theta_k) = \sum_{n=0}^{N-1} h(n)e^{j\theta_k n}$$

Phasors are of unequal length and the phase shift between phasors is constant



# All Pass Polyphase Structure

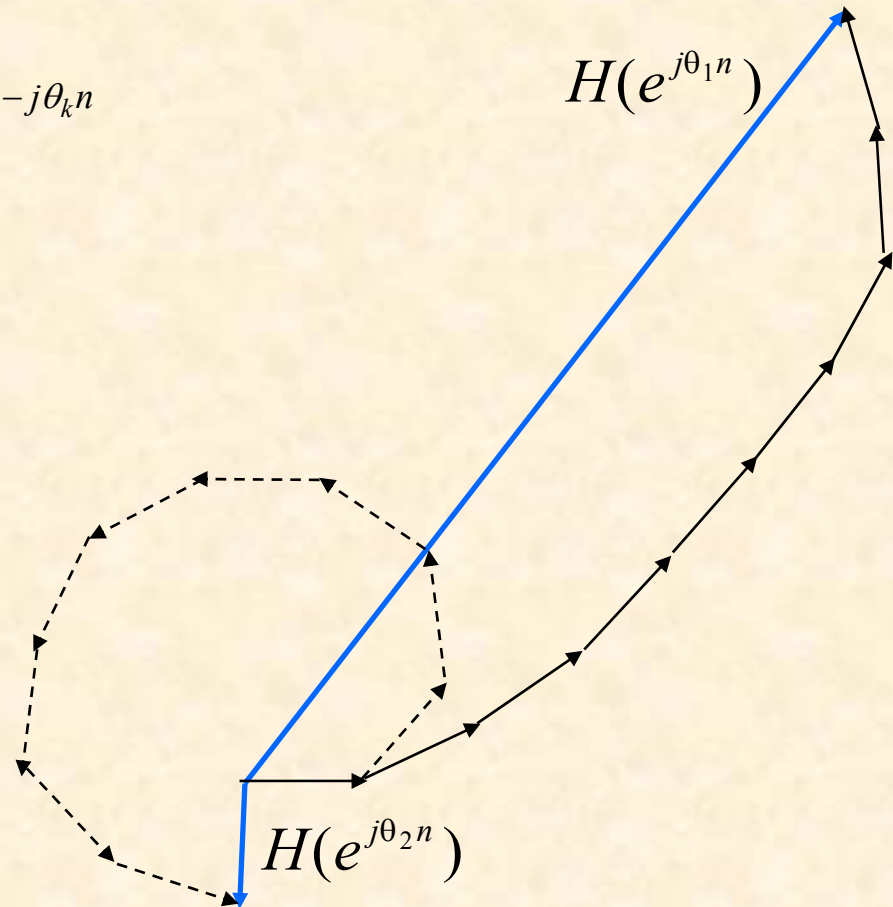


$$Y(z) = X(z) \sum_{i=0}^{N-1} Y_i(z)$$

$$Y_i(z) = \sum_{n=0}^{\infty} h_i(n)z^{-n}$$

$$H(\theta_k) = \sum_{n=0}^{N-1} H_n (e^{-jN\theta_k}) e^{-j\theta_k n}$$

**Phasors are of equal length but the phase shift between phasors is variable**



# All Pass Structures

Makes for very efficient filters

Bilinear Transform is an all-pass network

$$z = \frac{1 + s}{1 - s} = \frac{re^{j\theta_1}}{re^{j\theta_2}} = e^{j(\theta_1 - \theta_2)}$$

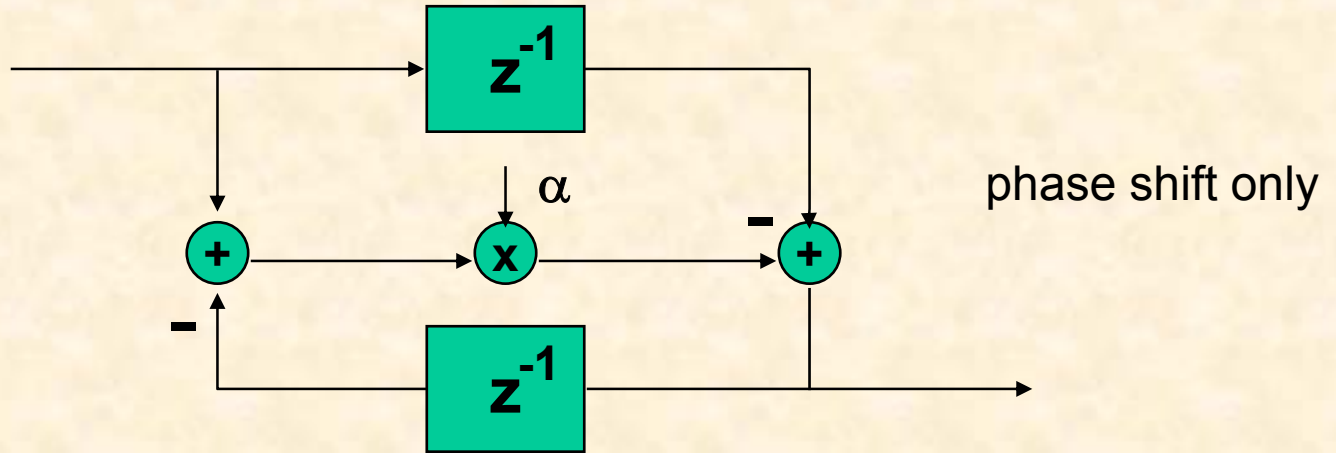
Used to convert analog filter spec into IIR filter spec

In the z-plane we can use the substitution

$$z = \frac{1 - \alpha z}{z - \alpha}$$

All-Pass Transform  
(harris)

# All Pass Structures

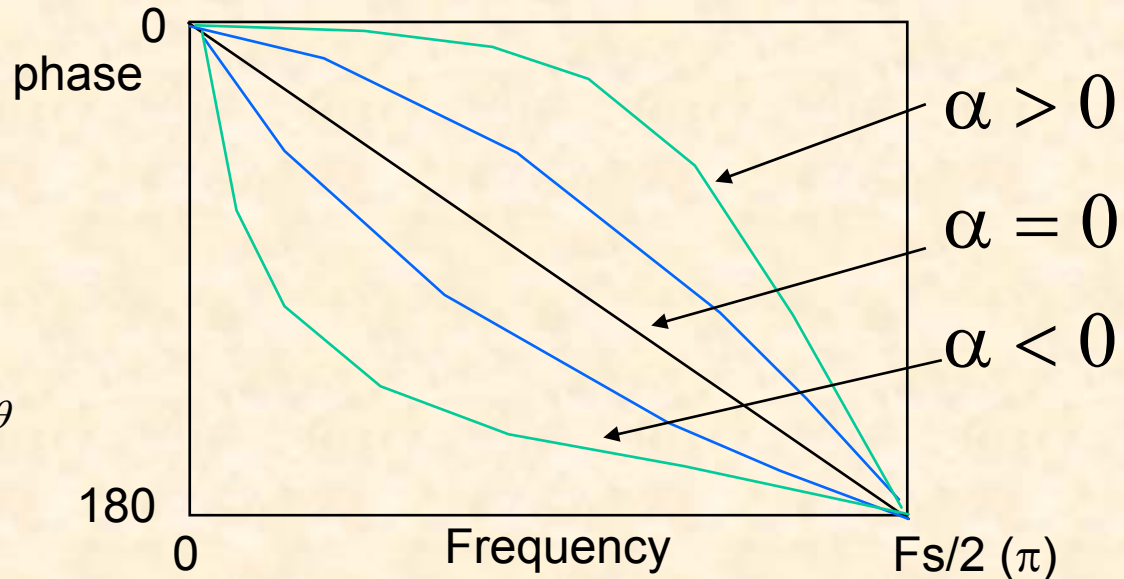


$$T(z) = \frac{1 - \alpha z}{z - \alpha}$$

$$\angle T(e^{j0}) = 0$$

$$\angle T(e^{j\pi}) = \pi$$

$$|T(z)| = 1 \quad \forall z = e^{j\theta}$$



# Why is it All Pass?

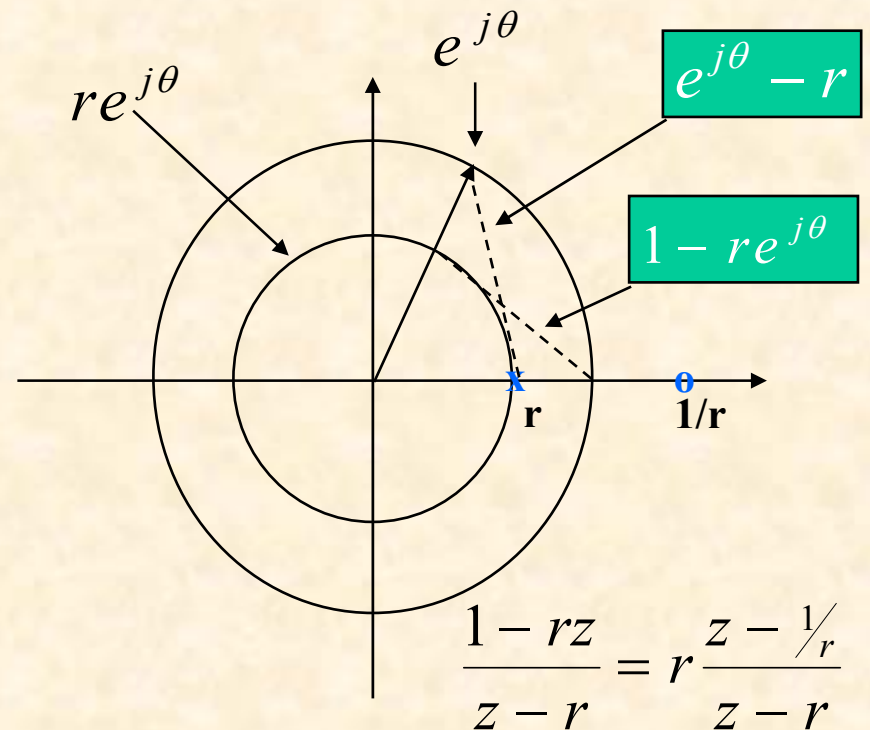
$$T(z) = \frac{1 - \alpha z^N}{z^N - \alpha}$$

for the case of  $N = 1$

$$T(z) = \frac{1 - \alpha z}{z - \alpha}$$

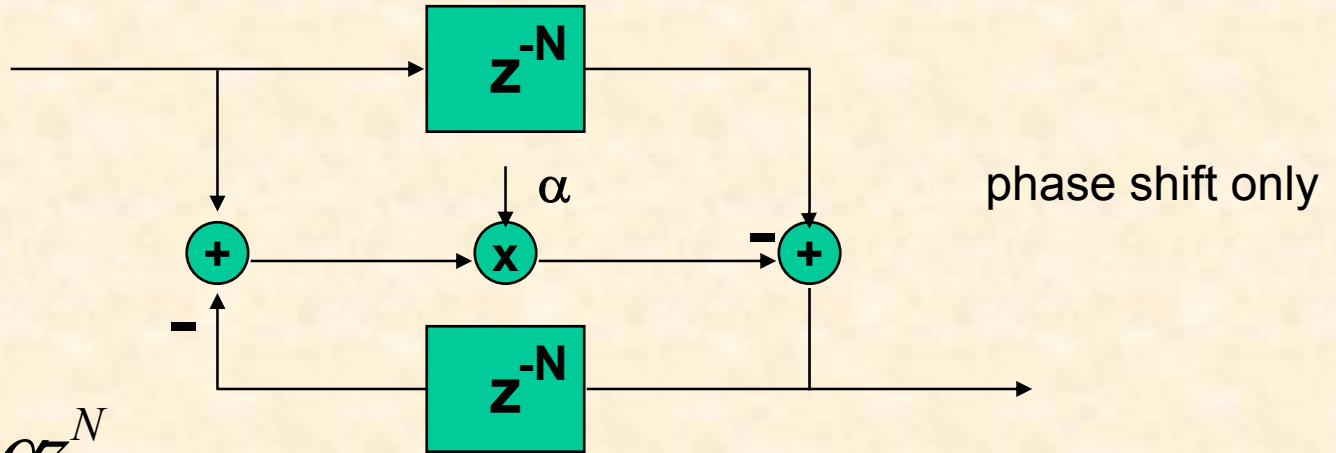
Pole at  $\alpha$ , Zero at  $1/\alpha$

Let  $r = \alpha$ , and  $z = e^{j\theta}$



# All Pass Structures

Note  $z \rightarrow z^2$   
replicates  
spectrum

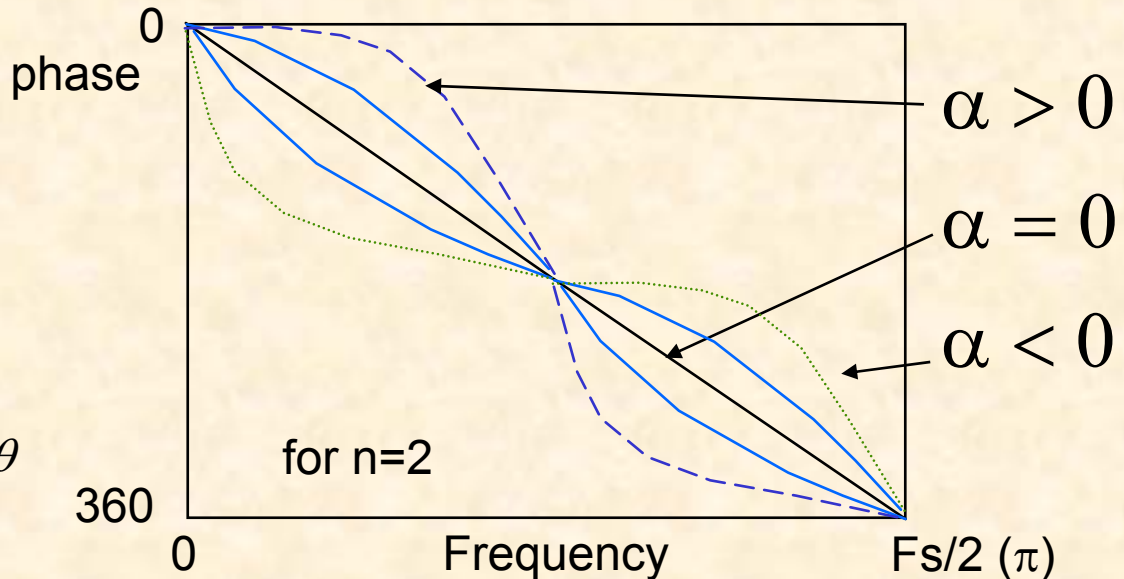


$$T(z) = \frac{1 - \alpha z^N}{z^N - \alpha}$$

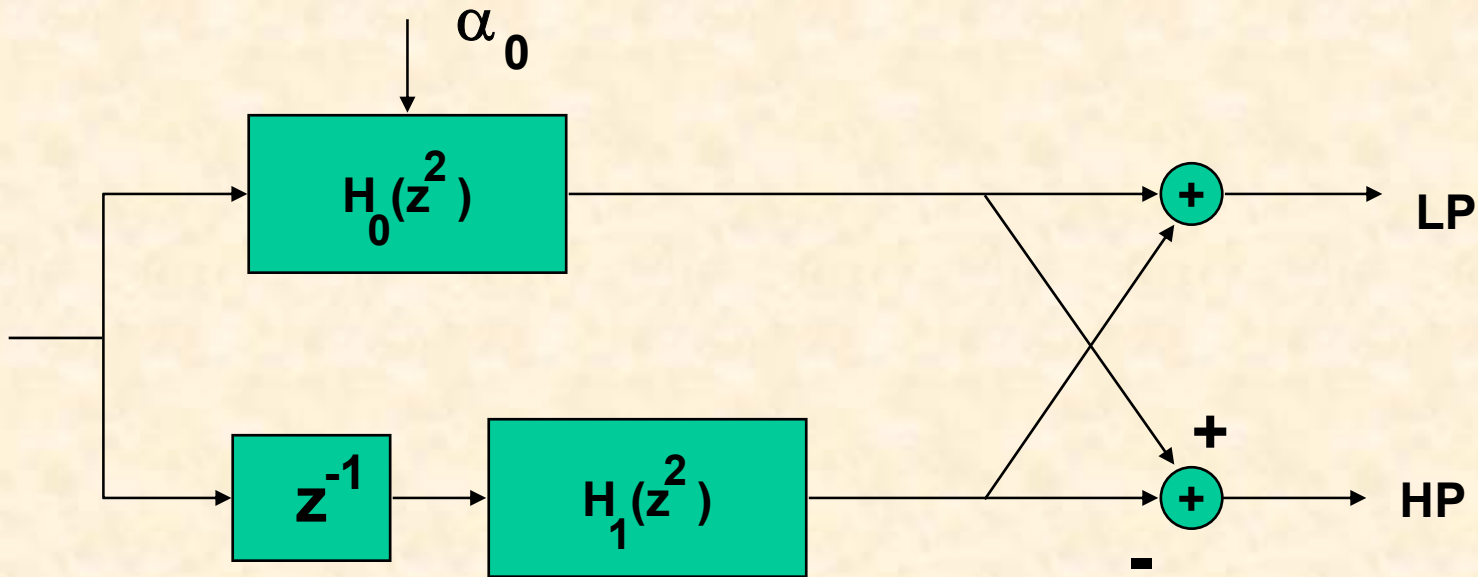
$$\angle T(e^{j0}) = 0$$

$$\angle T(e^{j\pi}) = n\pi$$

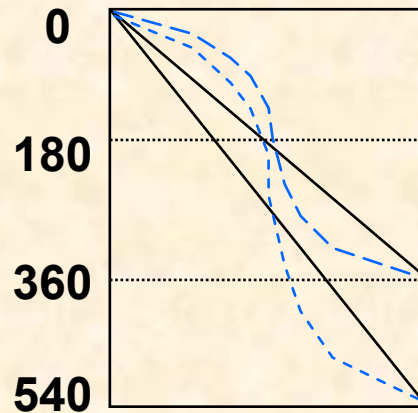
$$|T(z)| = 1 \forall z = e^{j\theta}$$



# Filter Structure



Very efficient filter  
two multipliers yield  
5 poles and 5 zeros



Adjust  
parameters  
to match  
phase in the  
pass band

# Demonstration

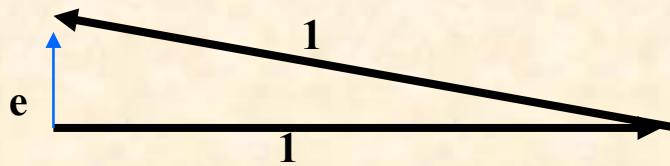
- Fildes (Max) program (research code)
- type a
- responds with
  - $n > 2$  (number of paths)
  - $k_0 > 2$  (stages on top path)
  - $k_1 > 2$  (stages on bottom path)
  - $w_0 > 0.3$  (passband  $> 0.25$ )
- 4 mult per input point yields 9 poles and 9 zeros
- Traditional IIR requires 18 multiplies
- 70 dB stopband attenuation,  $10^{-6}$  dB passband ripple  
(sometimes called microripple filters)

Also try  
h, z, p, w, x, u

# Why Are They Called Microripple?

## Stop Band

Note: (nominal passband gain = 2)



60dB attenuation

$$\delta_s = 10^{\frac{-60}{20}} = 10^{-3}$$

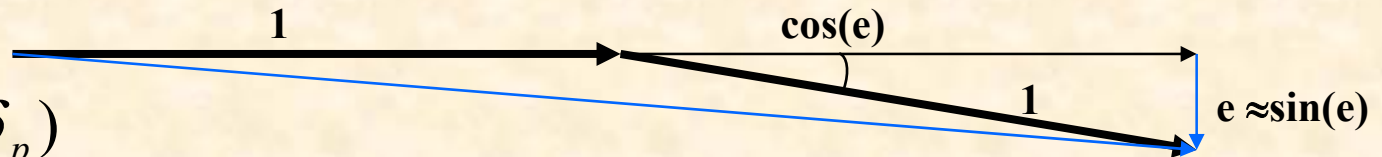
$$e = \delta_s / 2 = 5 \times 10^{-4}$$

Passband delta is

$$\delta_p = (1 - \cos(e)) / 2$$

$$\approx \left( 1 - \left( 1 - \frac{e^2}{2} \right) \right) / 2 = \frac{e^2}{4} = 6.25 \times 10^{-8}$$

## Pass Band



$$20 \log_{10}(1 + \delta_p)$$

$$= 20 \log_{10}(1 + 6.25 \times 10^{-8})$$

$$= 5.4 * 10^{-7}$$

**That is, 5.4x10<sup>-7</sup> dB!**