

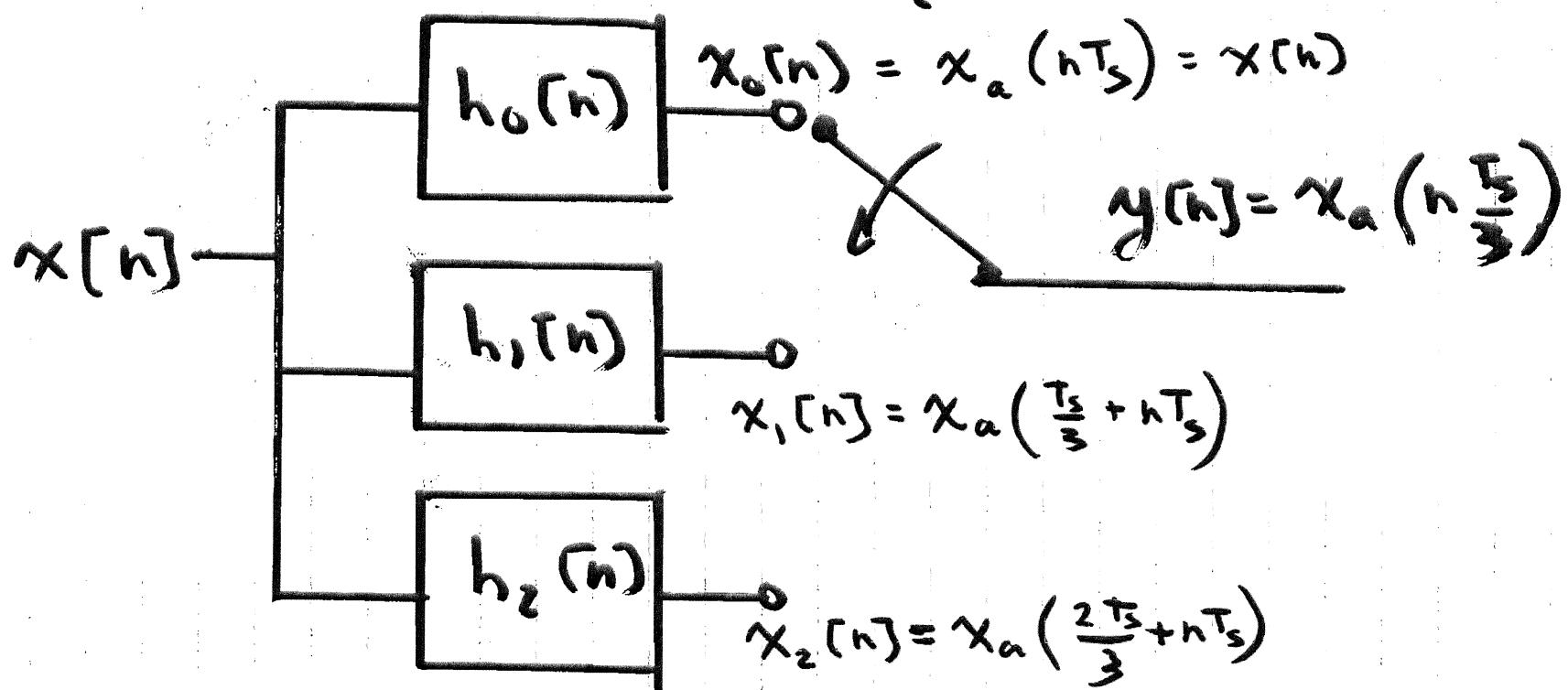
Recall: Single Signal 3x upsampling

$$h[n] = \frac{3 \sin\left(\frac{\pi}{3}n\right)}{\pi n}$$

$$h_0[n] = h[ln]$$

$$h_1[n] = h[l(n+1)]$$

$$h_2[n] = h[l(n+2)]$$

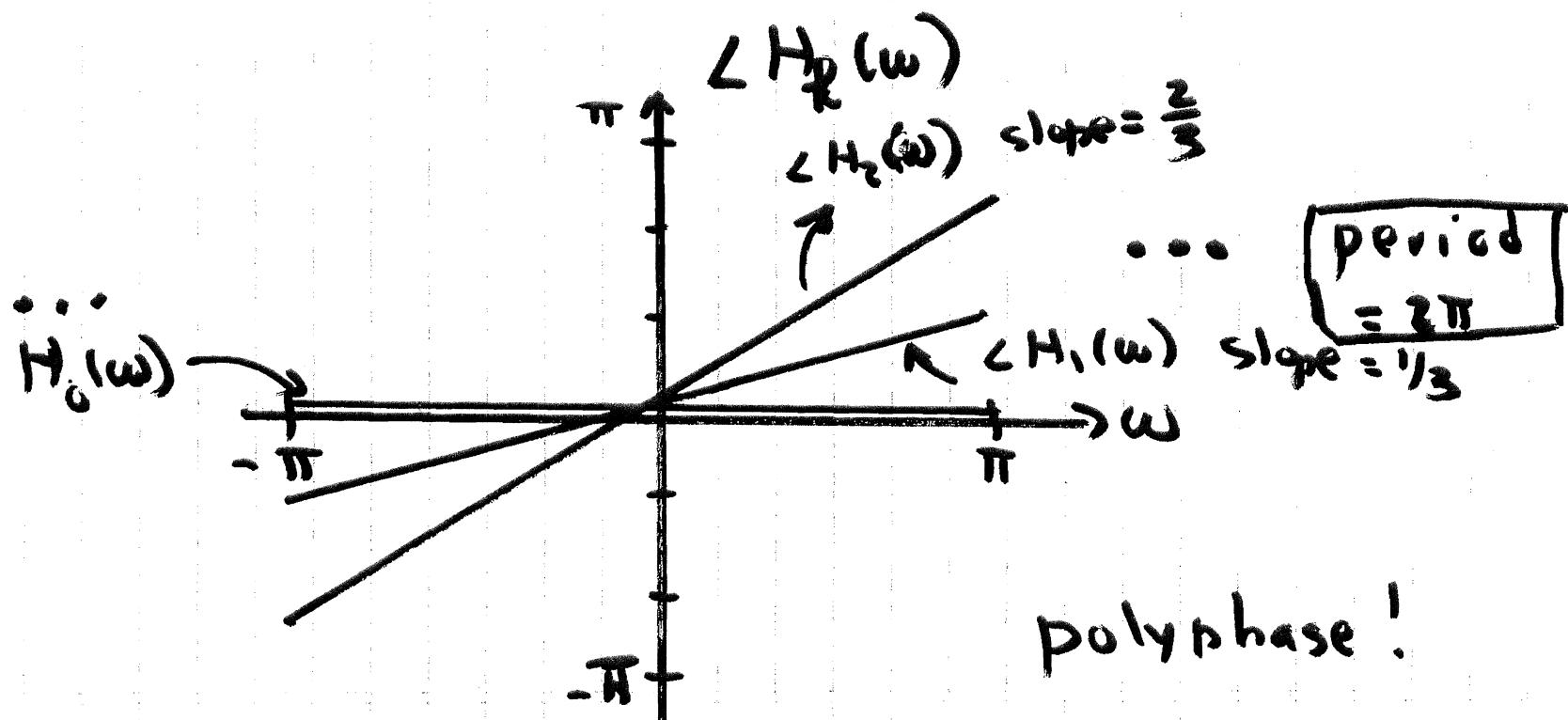


Recall: Single Signal \Rightarrow 3x upsampling

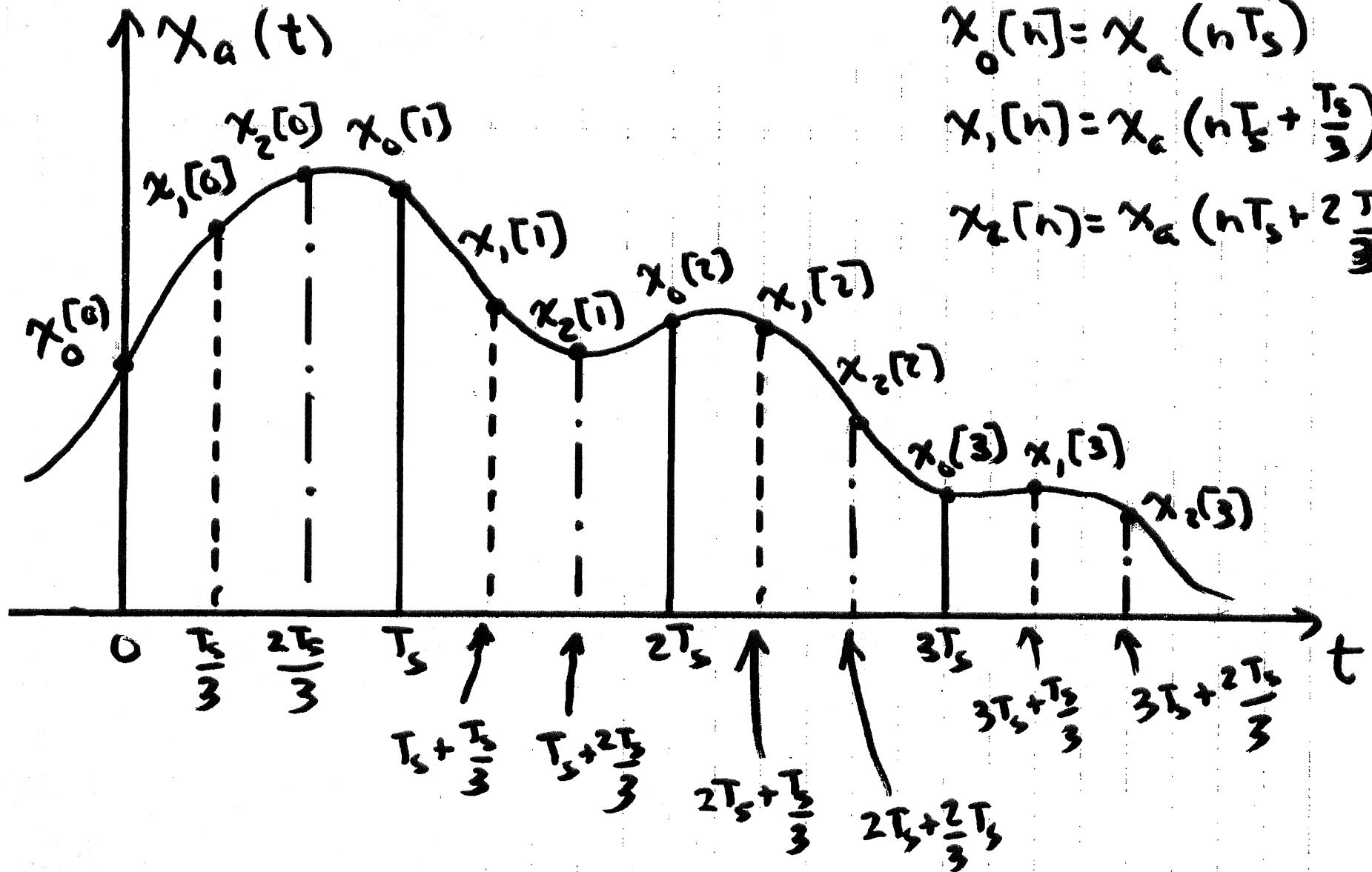
for $|\omega| < \pi$: $H_0(\omega) = 1$

$$H_1(\omega) = e^{j\frac{1}{3}\omega} \Rightarrow \angle H_1(\omega) = \frac{\omega}{3}$$

$$H_2(\omega) = e^{j\frac{2}{3}\omega} \Rightarrow \angle H_2(\omega) = \frac{2}{3}\omega$$



Single Signal: Upsampling 3x



$$x_0[n] = x_a(nT_s)$$

$$x_1[n] = x_a\left(nT_s + \frac{T_s}{3}\right)$$

$$x_2[n] = x_a\left(nT_s + 2\frac{T_s}{3}\right)$$

Straightforward Modulation After Digital Upsampling

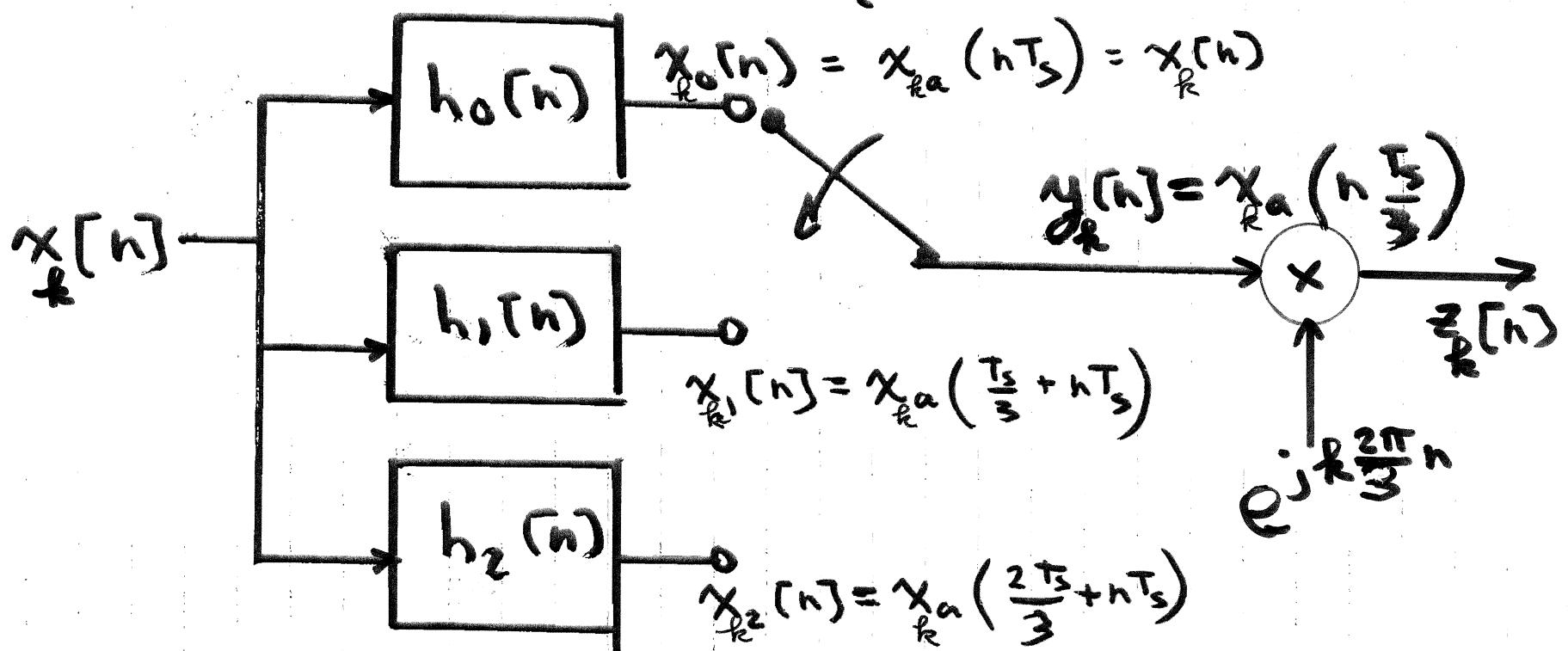
$$h[n] = 3 \frac{\sin\left(\frac{\pi}{3}n\right)}{\pi n}$$

$$h_0[n] = h[n]$$

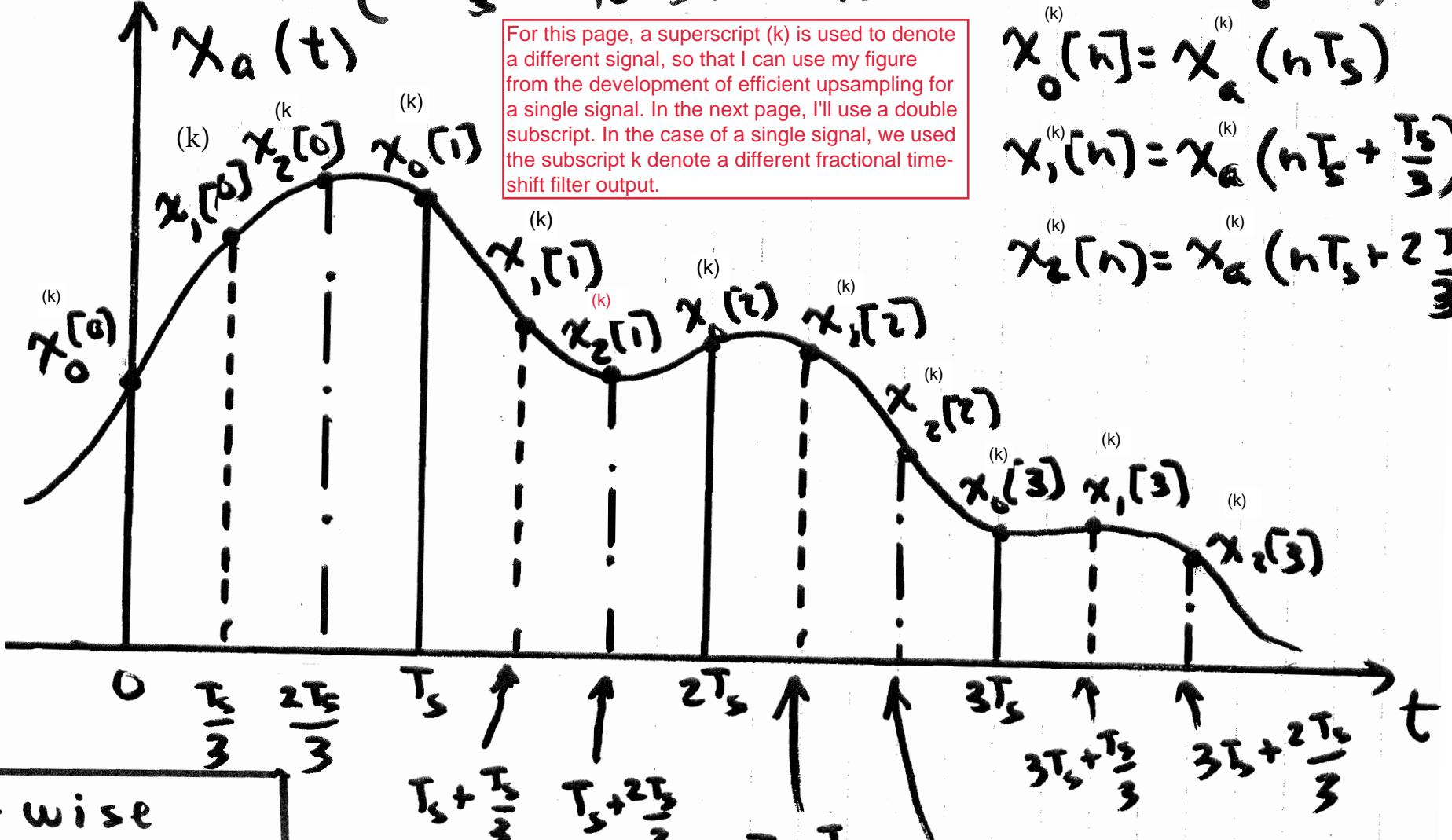
$$h_1[n] = h[n+1]$$

$$h_2[n] = h[n+2]$$

$L=3$



After 3x upsampling, pt-wise multiply by sinewave $e^{j\frac{2\pi}{3}n}$ to shift to different frequency band



$$\begin{aligned} x_0^{(k)} &= x_a^{(k)}(nT_s) \\ x_1^{(k)} &= x_a^{(k)}\left(nT_s + \frac{T_s}{3}\right) \\ x_2^{(k)} &= x_a^{(k)}\left(nT_s + 2\frac{T_s}{3}\right) \end{aligned}$$

pt-wise
multiply
with sinewave

$$.* \left\{ 1 \ \gamma_k \ \gamma_k^2 \ 1 \ \gamma_k \ \gamma_k^2 \dots \right\}$$

where: $\gamma_k = e^{j\frac{k}{3}}$, $k=0, 1, 2$

period of each
sinewave = 3

Post Up Sampling Modulation Exploiting Sinewave Period 3

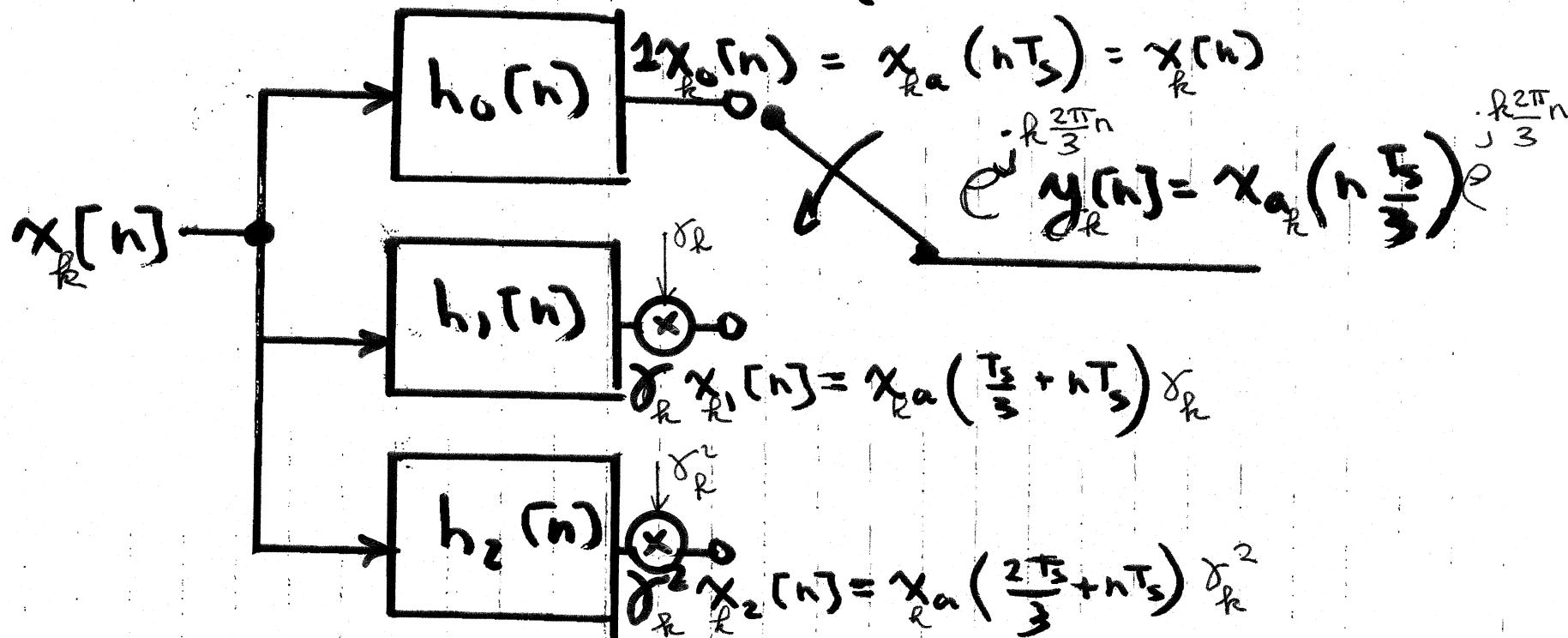
$$h[n] = 3 \frac{\sin\left(\frac{\pi}{3}n\right)}{\pi n}$$

$$h_0[n] = h[n]$$

$$h_1[n] = h[Ln+1]$$

$$h_2[n] = h[Ln+2]$$

$L=3$



$$\text{where: } \delta_k = e^{jk\frac{2\pi}{3}}$$

$$k=0, 1, 2$$

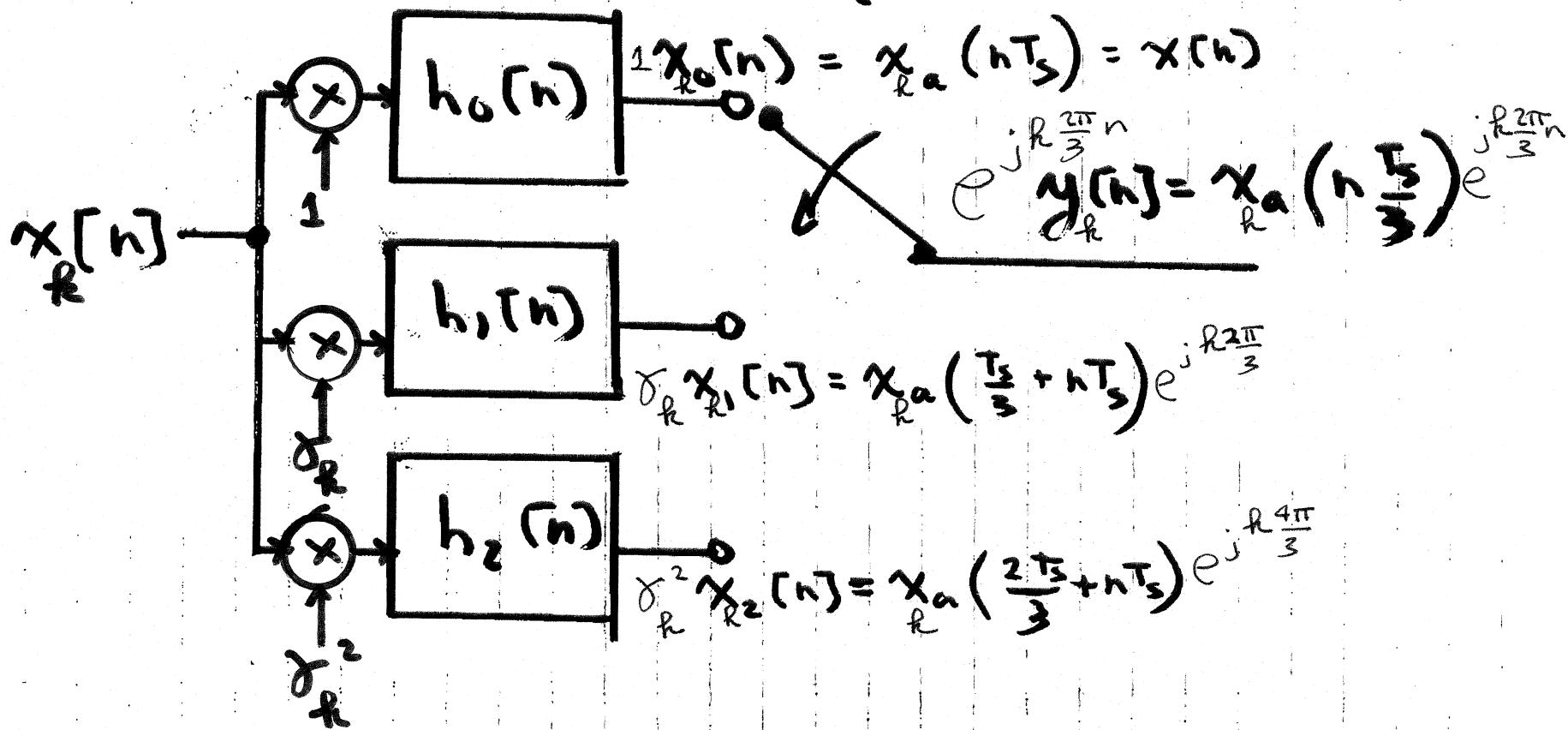
NEXT: Exploit linearity to Move Scaling Before Filter

$$h[n] = 3 \frac{\sin\left(\frac{\pi}{3}n\right)}{\pi n}$$

$$h_0[n] = h[0]$$

$$h_1[n] = h[1]$$

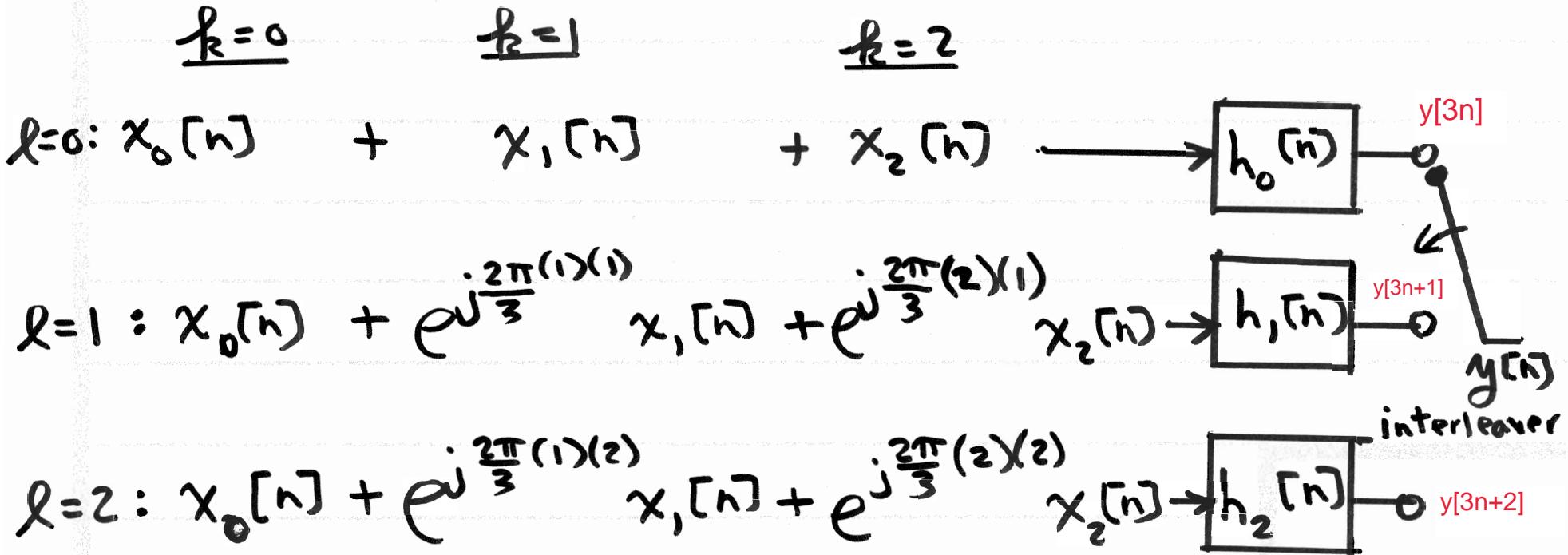
$$h_2[n] = h[2]$$



where: $\delta_R = e^{j\frac{2\pi}{3}}$ $k=0,1,2$

- And recall: we ultimately sum everything together since the signals are in different frequency bands

- THUS: Below, $x_0[n]$, $x_1[n]$ and $x_2[n]$ represent 3 DIFFERENT signals, each sampled above Nyquist (but presumably not much above the Nyquist Rate.) The subscript k also represents a different subband since each signal is ultimately upsampled and placed in a different subband.

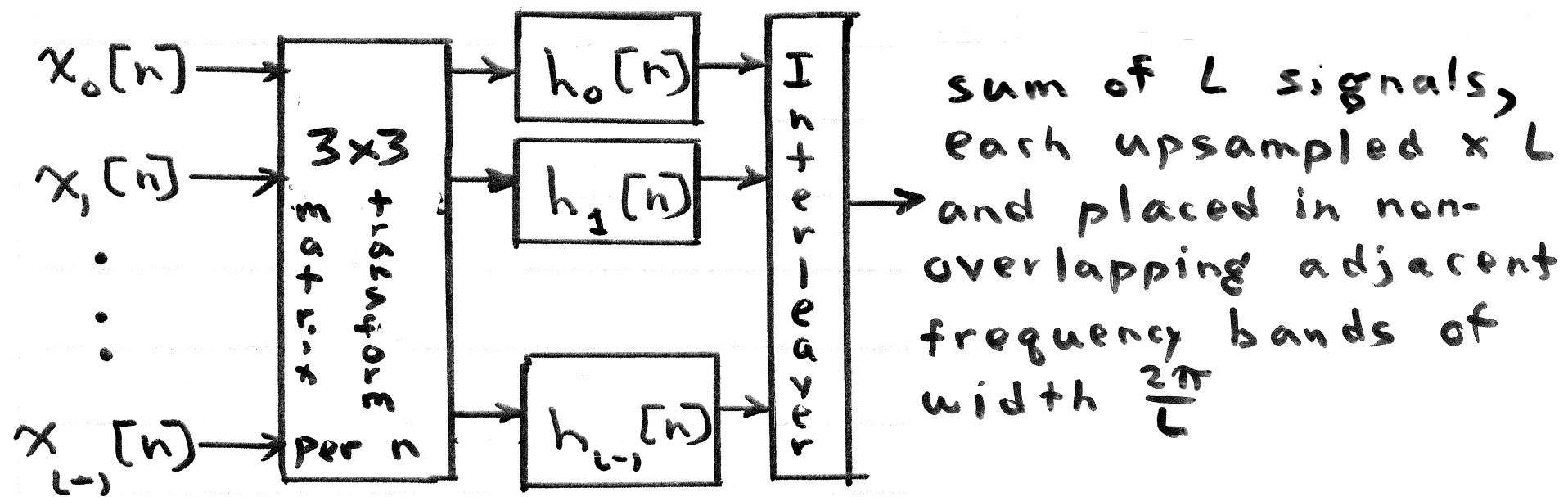


• again: k = signal index = subband index, $k = 0, 1, 2$

ℓ = polyphase component index, $\ell = 0, 1, 2$

=> see matlab example Multiplex3Sigs.m

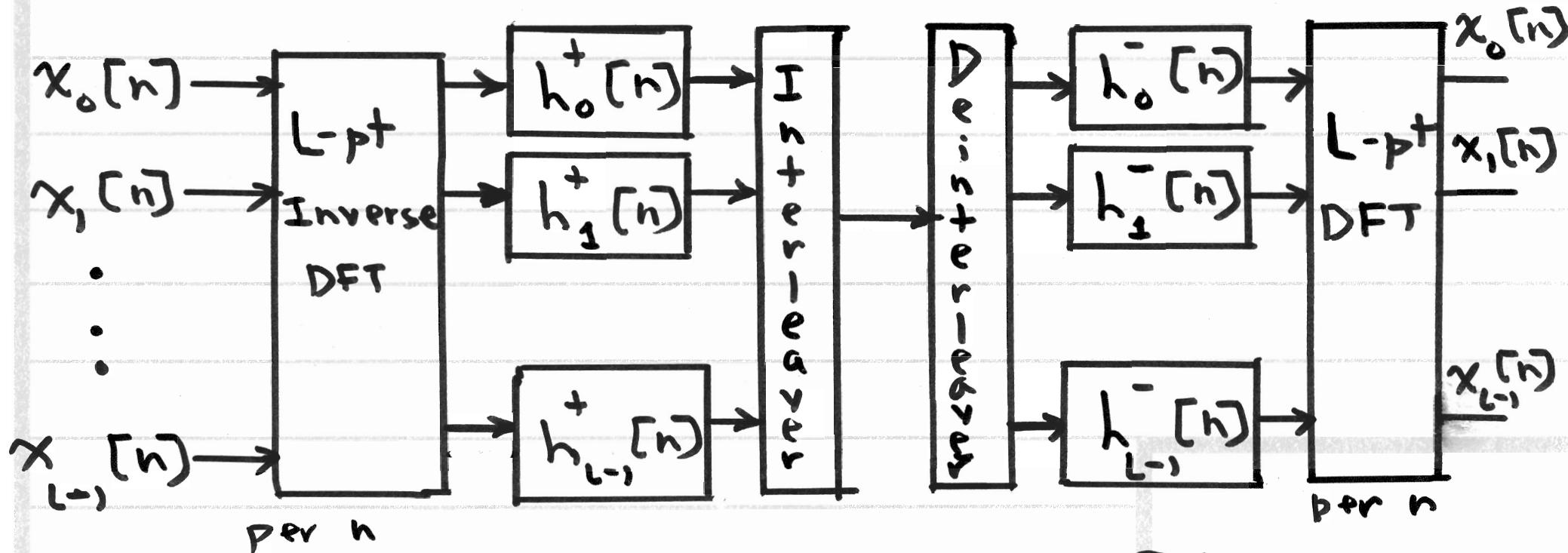
- The scalars on the polyphase components of the LP filter turn out to comprise 3×3 matrix



- In ideal case where $h_{LP}(n) = L \frac{\sin(\frac{\pi}{L}n)}{\pi n}$

$$H_L(\omega) = e^{j \frac{\omega}{L} \omega} \quad \text{for } -\pi < \omega < \pi$$

- The scalars on the polyphase components of the LP filter turn out to comprise either an L-pt DFT or an L-pt Inverse DFT

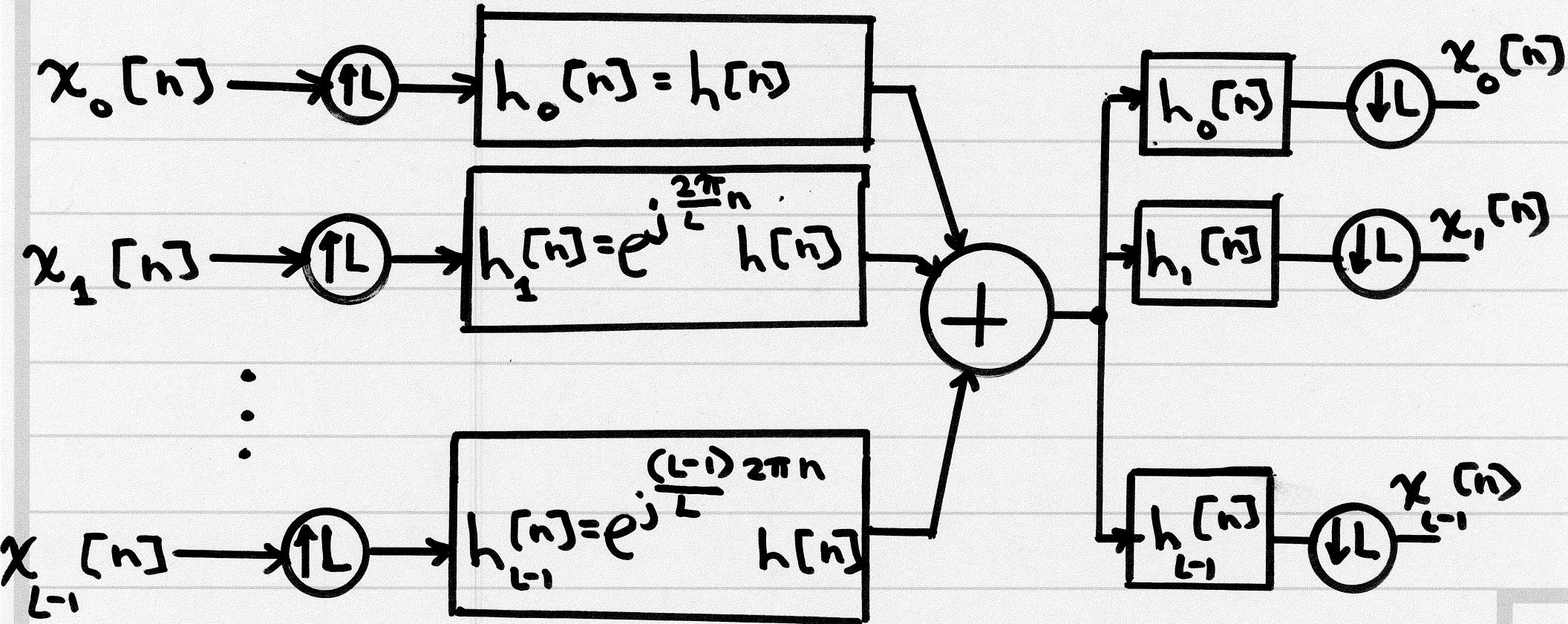


In ideal case where $h_{LP}[n] = L \frac{\sin(\frac{\pi}{L}n)}{\pi n}$

$$H_L^-(\omega) = e^{-j\frac{\omega}{L}\omega} \text{ for } -\pi < \omega < \pi$$

See Multiplex 3 Sigs Alt. m

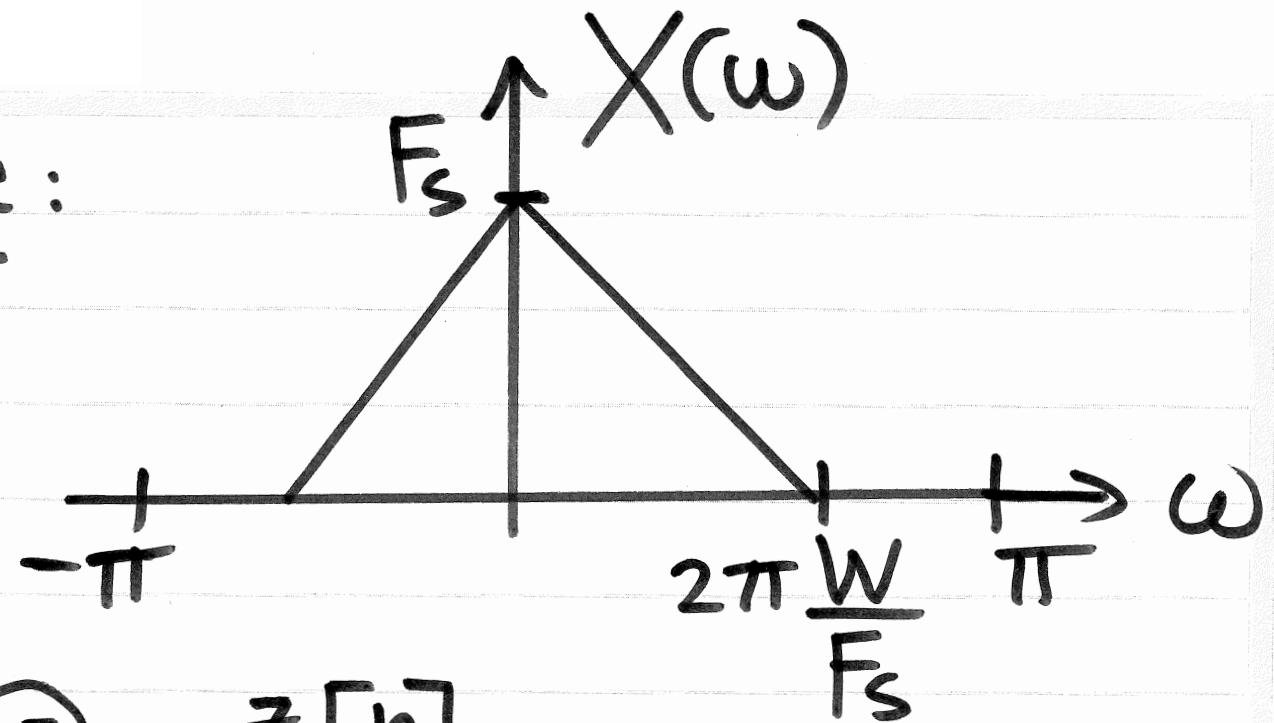
Previous page is efficient implementation of :



Recall:

L4

Example:



$$x[n] \rightarrow \textcircled{15} \rightarrow z[n]$$

Diagram illustrating the mapping of the discrete-time signal $x[n]$ to the discrete-time Fourier transform $z[n]$ via the circled value 15.

