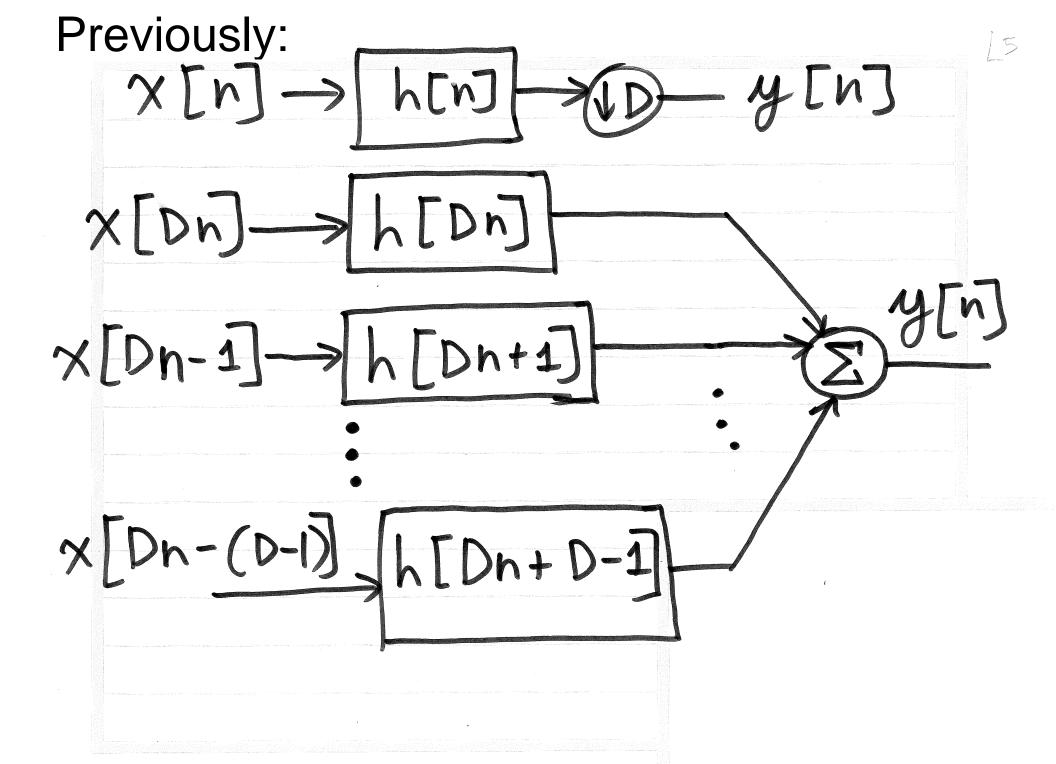
- · Addendum Notes on Efficient Upsampling and Efficient "Filtering Followed by Downsampling"
- . Efficient Upsampling was about how we read out the upsampled output values
- "Efficient Downsampling" was about how we counted through the summation variable in a summation
- · These same concepts will lead to the FFT: Fast Fourier Transform
- . We will develop different FFT schemes as a combination of how we read out the DFT values AND how we count thru the summation variable

· efficient implementation for for zero-inserts followed by filtering with h[n]



2

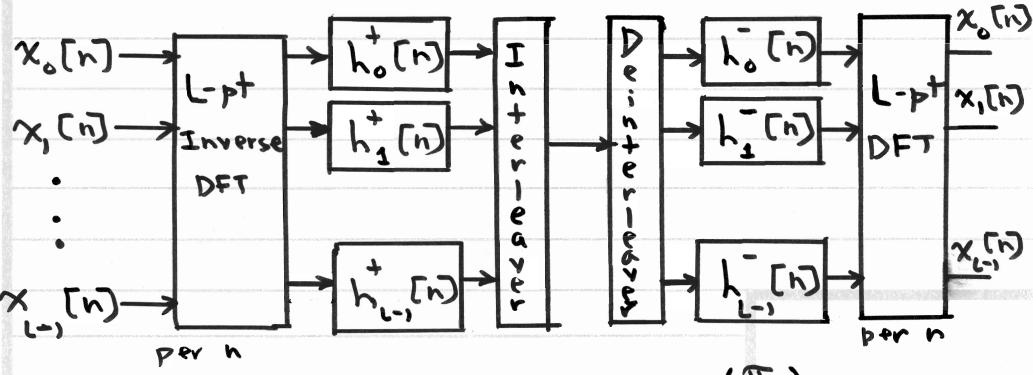
· Alternative Efficient Downsampling "Scheme

$$\frac{\chi(n)}{h[n]} \rightarrow \frac{\chi(n)}{h[n]} \rightarrow \frac{\chi(n)}{h[n]$$

$$= \sum_{k=0}^{\infty} \sum_{k'=-\infty}^{\infty} x \left[Dk' + k \right] \left[D(n-k') - k \right]$$

タ[n]= = ×[Dn+引*h。[n] · Leads to: 4° (4)=4[DV] -> L_[Dn-(D-1)] in our previous efficient implementation and - signs were reversed y[n]= = x[Dn-e] * h[Dn+e]

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 hep[n] = L sin(IN)

Previous page is efficient implementation of:

· Consider again where have a bank of filters each followed by decimation, with each filter related as:

and the decimation is by a factor of D=L

Form:
$$f_{R} = e^{jR} \left(L_{n-R} \right) \int_{L_{R}} \left[L_{n-R} \right]$$

had previously.