

Chap 12.4

Recall that downsampling/decimation : reduces sampling frequency
decreases number of sample points

- often used to reduce memory requirement

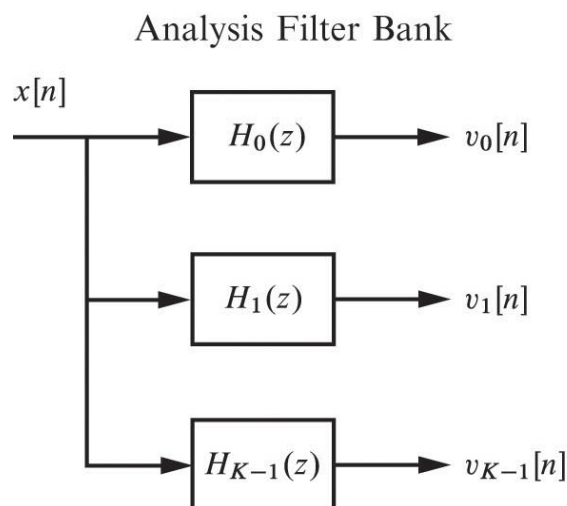
Recall that upsampling/interpolation : increase sampling frequency
increase number of sample points

- used to "fill" in missing values

Another important application of decimation and interpolation is filter bank

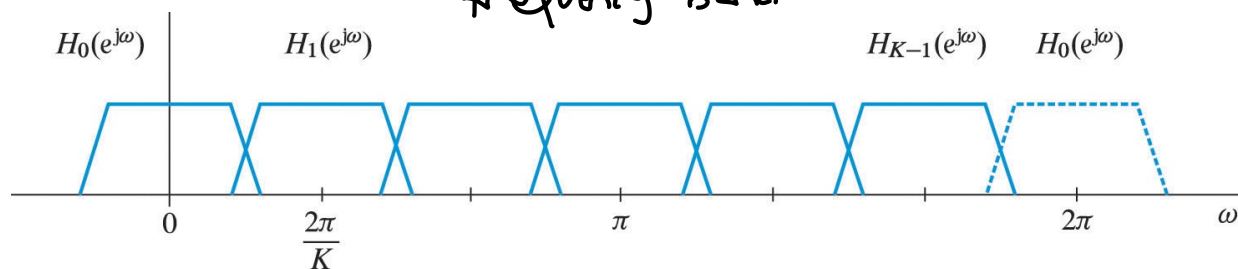
Filter bank is a collection of filters with a common input or a common output

There are two types of filter bank :
analysis filter and synthesis filter bank



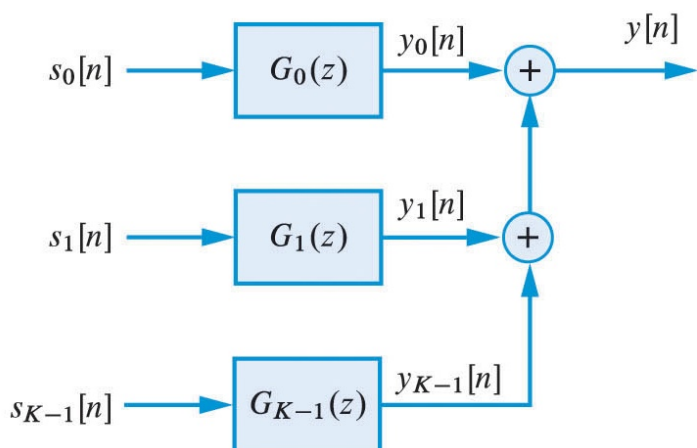
Analysis filter splits a signal $x[n]$ into K signals $v_k[n]$ known as sub-band signals using analysis filter $H_k(z)$

Example analysis filter that splits $x[n]$ into different frequency band



- think equalizer in the music player

Synthesis Filter Bank



Synthesis filter bank consists of K synthesis filter $G_k(z)$, which combines K signals $s_k[n]$ into a signal $y[n]$

But there is a big problem with this design. If $x[n]$ has 200 points in the time domain. Then

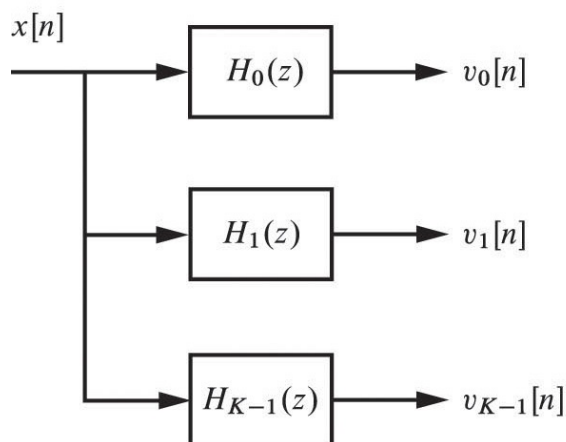
$v_0[n]$ has 200 points in the time domain

$v_1[n]$ has 200 points in the time domain

\vdots

we have $(200)K$ points to store

Analysis Filter Bank



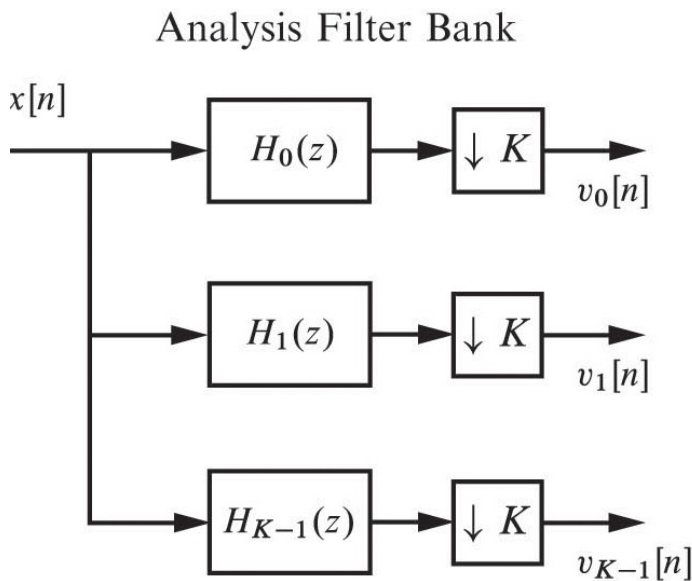
Practical analysis filter bank

$x[n]$ has 200 points

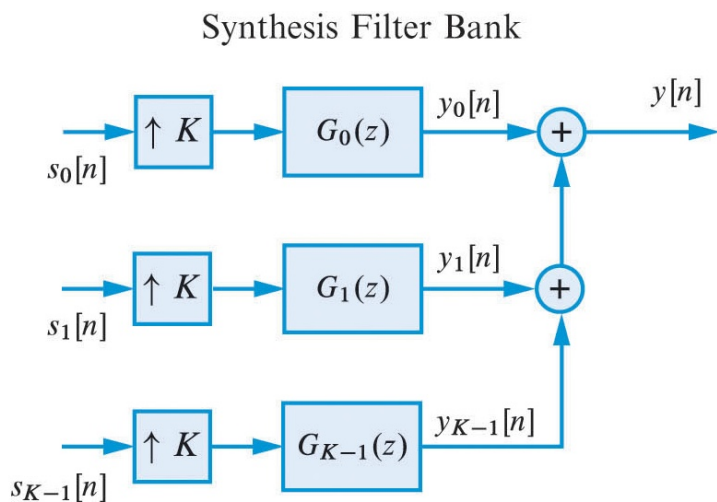
$V_0[n]$ has $\frac{200}{K}$ points

$V_1[n]$ has $\frac{200}{K}$ points
 \vdots

we have $(\frac{200}{K})K = 200$
 points to store

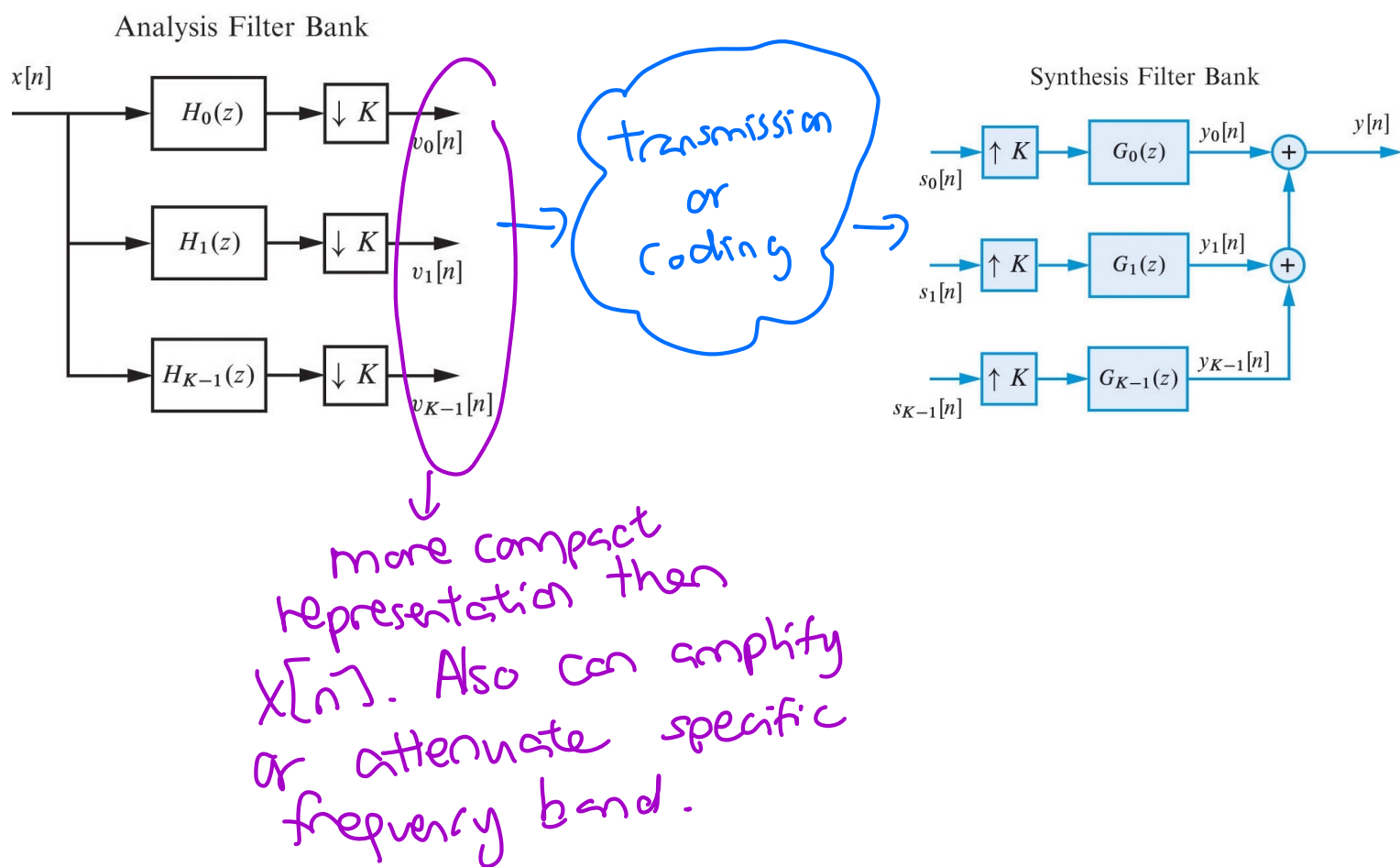


Analysis filter is followed by a synthesis filter.
 So practical synthesis filter have an interpolation step before the synthesis filter



If synthesis filter "undo" the work of analysis filter, why do we use filter bank?

For efficient transmission of information



If filter banks are used for transmission, then we want $y[n] \approx x[n]$. We can handle $y[n] = Gx[n - n_d]$

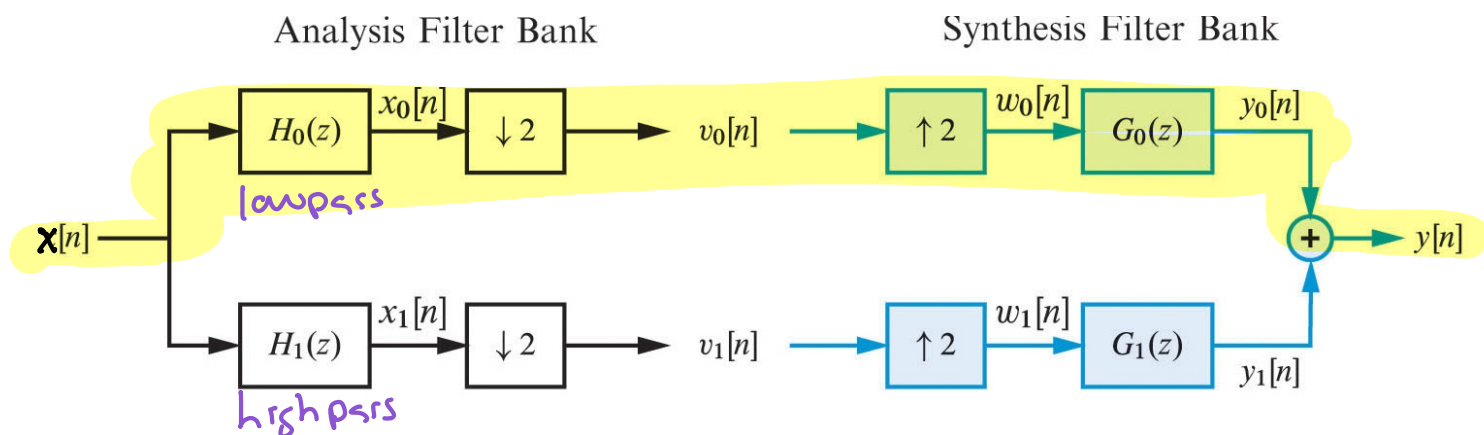
This will restrict the design of $H_k(z)$ and $G_k(z)$

In the frequency domain perfect (or distortionless) reconstruction means

$Y(z) = G z^{-n_d} X(z) = G e^{-j\omega n_d} X(z)$

↳ linear phase response

Two-channel filter bank is the simplest filter bank and will let us see what conditions we need on $H_k(z)$ and $G_k(z)$ to ensure perfect reconstruction.



We don't want to have to deal with convolution in the time domain

Consider the 0th channel. Usually this is the lower frequency channel

analysis filter {
$$X_0(z) = X(z)H_0(z)$$

$$V_0(z) = \frac{1}{2}X_0(z^{\frac{1}{2}}) + \frac{1}{2}X_0(-z^{\frac{1}{2}})$$
 this is from decimation

synthesis filter {
$$W_0(z) = V_0(z^2)$$
 this is from upsampling

$$Y_0(z) = V_0(z^2)G_0(z)$$

$$= \frac{1}{2}(H_0(z)X(z) + H_0(-z)X(-z))G_0(z)$$

We can consider the other band the same way and get

$$Y_1(z) = \frac{1}{2} (H_1(z)X(z) + H_1(-z)X(-z))G_1(z)$$

The output of the synthesis filter

$$Y(z) = Y_0(z) + Y_1(z)$$

we shuffle terms

$$= \frac{1}{2} (T(z)X(z) + A(z)X(-z))$$

$$= \frac{1}{2} \left[(H_0(z)G_0(z) + H_1(z)G_1(z))X(z) + (H_0(-z)G_0(z) + H_1(-z)G_1(z))X(-z) \right]$$

Recall we want distortion $Y(z) = Gz^{-n_d}X(z)$

This means we want

$$H_0(z)G_0(z) + H_1(z)G_1(z) = Gz^{-n_d}$$

$$H_0(-z)G_0(z) + H_1(-z)G_1(z) = 0$$

We can write this as a system of linear equations

$$\begin{bmatrix} H_0(z) & H_1(z) \\ H_0(-z) & H_1(-z) \end{bmatrix} \begin{bmatrix} G_0(z) \\ G_1(z) \end{bmatrix} = \begin{bmatrix} G z^{-n_d} \\ 0 \end{bmatrix}$$

(recall this looks like $Ax=b$)

This is only solvable if the determinant of the matrix, $\Delta_m(z) \neq 0$

$$\Delta_m(z) = H_0(z)H_1(-z) - H_0(-z)H_1(z) \neq 0$$

If we can solve the system of equations, we have

$$G_0(z) = \frac{z^{-n_d} H_1(-z)}{\Delta_m(z)}$$

$$G_1(z) = -\frac{z^{-n_d} H_0(-z)}{\Delta_m(z)}$$

To guarantee perfect reconstruction, we need

$$z^{n_d} H_0(z) G_0(z) + z^{n_d} H_1(z) G_1(z) = G = 2$$

Let $R(z) \equiv z^{n_d} H_0(z) G_0(z)$, we will call this the product filter

Designing analysis and synthesis filters
reduces to first designing $R(z)$ and
then finding $H_0(z), H_1(z), \dots, G_0(z), G_1(z)$