

## Lecture 13: Conjugate Quadrature Filter Bank

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### 1 Introduction

We continue in this lecture to build upon the particular class of filter bank which we have introduced in the previous lecture called a Conjugate Quadrature Filter (CQF) bank.

### 2 Conjugate Quadrature Filter bank

For the perfect reconstruction system we must first do away aliasing. The alias cancellation equation for the two band filter bank is given by

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

$$\frac{G_1(Z)}{G_0(Z)} = -\frac{H_0(-Z)}{H_1(-Z)}$$

Equating the numerator and denominator we get the relation between  $G_0(Z)$ ,  $H_1(-Z)$ ,  $G_1(Z)$  and  $H_0(-Z)$  as

$$\begin{aligned} G_1(Z) &= -H_0(-Z) \\ G_0(Z) &= H_1(-Z) \end{aligned}$$

The relation between the analysis HPF (high pass filter) and analysis LPF (low pass filter) called a conjugate quadrature relationship, is given by

$$H_1(Z) = z^{-D} H_0(-Z^{-1})$$

Here  $z^{-D}$  term is used to introduce causality. Putting  $Z = e^{j\omega}$  in the above equation we get the frequency response equation as

$$\begin{aligned} H_1(Z) &= z^{-D} H_0(-Z^{-1})|_{Z=e^{j\omega}} \\ H_1(e^{j\omega}) &= e^{-j\omega D} H_0(-e^{-j\omega}) \end{aligned}$$

The magnitude response is given by

$$\begin{aligned} |H_1(e^{j\omega})| &= |e^{-j\omega D} H_0(-e^{-j\omega})| \\ |H_1(e^{j\omega})| &= |e^{-j\omega D}| |H_0(-e^{-j\omega})| \\ |H_1(e^{j\omega})| &= |H_0(-e^{-j\omega})| \end{aligned}$$

$H_0(Z)$  is a Low pass filter with a real impulse response (real coefficients), therefore

$$H_0(e^{-j\omega}) = \overline{H_0(e^{j\omega})}$$

The magnitude response of LPF  $H_0(Z)$  is symmetric along the magnitude axis and phase response is anti-symmetric along the frequency axis  $\omega$ .

$$H_0(-e^{-j\omega}) = H_0(e^{-j(\omega \pm \pi)})$$

**NOTE:** LPF with cutoff frequency  $\frac{\pi}{2}$  (With shift by  $\pi$  on  $\omega$ )  $\rightleftharpoons$  HPF with cutoff frequency  $\frac{\pi}{2}$

We have shown,

$$H_1(Z) = z^{-D} H_0(-Z^{-1})$$

For the perfect reconstruction the equation must satisfy,

$$\begin{aligned} G_0(Z)H_0(Z) + G_1(Z)H_1(Z) &= C_0 z^{-D} \\ H_1(-Z)H_0(Z) - H_0(-Z)H_1(Z) &= C_0 z^{-D} \\ (-1)^{-D} z^{-D} H_0(Z^{-1})H_0(Z) - H_0(-Z)z^{-D} H_0(-Z^{-1}) &= C_0 z^{-D} \end{aligned}$$

We need the following for perfect reconstruction systems,

$$(-1)^{-D} H_0(Z^{-1})H_0(Z) - H_0(-Z)H_0(-Z^{-1}) = C_0$$

If we consider the Haar filter then the relationship between  $H_0(Z)$  and  $H_1(Z)$  is given by,

$$\begin{aligned} H_0(Z) &= 1 + z^{-1} \\ H_0(-Z^{-1}) &= 1 - z \end{aligned}$$

The above equation is non-causal so to make it causal by inserting delay, we get the below equation,

$$z^{-D} H_0(-Z^{-1}) = z^{-D}(1 - z)$$

Here  $z^{-D}$  retains causality.

If D is odd,

$$\begin{aligned} H_0(Z)H_0(Z^{-1}) + H_0(-Z)H_0(-Z^{-1}) &= -C_0 \\ H_0(Z)H_0(Z^{-1}) + H_0(-Z)H_0(-Z^{-1}) &= \text{Constant} \end{aligned}$$

Putting  $Z = e^{j\omega}$ , we get the above equation in the frequency domain as,

$$H_0(e^{j\omega})H_0(e^{-j\omega}) + H_0(-e^{j\omega})H_0(-e^{-j\omega}) = \text{Constant}$$

For real impulse response we have,

$$\begin{aligned} H_0(e^{-j\omega}) &= \overline{H_0(e^{j\omega})} \\ H_0(e^{j\omega})\overline{H_0(e^{j\omega})} + H_0(e^{j(\omega \pm \pi)})\overline{H_0(e^{j(\omega \pm \pi)})} &= \text{Constant} \\ |H_0(e^{j\omega})|^2 + |H_0(e^{j(\omega \pm \pi)})|^2 &= \text{Constant} \end{aligned}$$

Above equation is called the power complementary equation.  
 For perfect reconstruction system,

$$H_0(Z)H_0(Z^{-1}) + H_0(-Z)H_0(-Z^{-1}) = \text{Constant}$$

Lets assume  $\kappa_0(Z) = H_0(Z)H_0(Z^{-1})$

$$\kappa_0(Z) + \kappa_0(-Z) = \text{Constant}$$

We are going to choose even length of  $H_0(Z)$  ,i.e.  $D \rightarrow \text{Odd}$

$$h[n] : \begin{matrix} h_0 & h_1 & h_2 & \dots & h_D \\ \uparrow & & & & \uparrow \\ n & & & & D \end{matrix}$$

Similarly,  $H_0(Z^{-1})$  is given by,

$$h[n] : \begin{matrix} h_D & \dots & h_2 & h_1 & h_0 \\ \uparrow & & \uparrow & & \uparrow \\ n & & -D & & 0 \end{matrix}$$

Here  $H_0(Z)H_0(Z^{-1})$  corresponds to their convolution in time domain

$$\begin{matrix} (h_0 & h_1 & h_2 & \dots & h_D) \\ \uparrow & & & & \uparrow \\ 0 & & & & D \end{matrix} \otimes \begin{matrix} (h_D & \dots & h_2 & h_1 & h_0) \\ \uparrow & & \uparrow & & \uparrow \\ -D & & 0 & & 0 \end{matrix}$$

Let impulse response  $h[k]$  be as given below

$$h[k] : \begin{matrix} h_0 & h_1 & h_2 & \dots & h_D \\ \uparrow & & & & \uparrow \\ 0 & & & & D \end{matrix}$$

And impulse response  $g[k]$  is given below which is mirror image of  $h[k]$ , that means  $g[k] = h[-k]$

$$g[k] : \begin{matrix} h_D & \dots & h_2 & h_1 & h_0 \\ \uparrow & & \uparrow & & \uparrow \\ -D & & 0 & & 0 \end{matrix}$$

Similarly  $g[n - k]$  is shown below

$$g[n - k] : \begin{matrix} h_0 & h_1 & h_2 & \dots & h_D \\ \uparrow & & & & \uparrow \\ n & & & & n+D \end{matrix}$$

The convolution between  $h[k]$  and  $g[k]$  is given

$$\kappa_0[n] = \sum_{k=-\infty}^{k=+\infty} h[k]g[n - k]$$

Here  $h[k]$  is causal and filter length is  $(D + 1)$ .

The convolution at the sample  $n$  is  $y[n]$ .

Shown below is the multiplication of  $h[k]$  and  $g[k]$  (which is shifted by  $n$  samples)

$$\begin{array}{cccccccc}
 & h_0 & h_1 & h_2 & \dots & h_n & h_{n+1} & \dots & h_D \\
 \kappa_0[n]: & \uparrow & & & & & & & \\
 & \mathbf{0} & & & & h_0 & h_1 & h_2 & \\
 & & & & & \uparrow & & & \\
 & & & & & n & & & 
 \end{array}$$

In  $Z$ -domain  $\kappa_0(Z) = H_0(Z)H_0(Z^{-1})$ .

The  $m^{\text{th}}$  sample of the filter  $k_0[m]$  is  $\langle h[k], h[k \pm m] \rangle$

Let  $m = 2$  and filter length 4 ( $D = 3$ )

$$\begin{array}{cccccccc}
 & h_0 & h_1 & h_2 & h_3 & & & \\
 \kappa_0[2]: & & & & & & & \\
 & \uparrow & & & & h_0 & h_1 & h_2 & h_3 \\
 & \mathbf{0} & & & & \uparrow & & & \\
 & & & & & \mathbf{2} & & & 
 \end{array}$$

$$k_0[2] = h_0h_2 + h_1h_3$$

If  $m = -2$  and filter length 4 ( $D = 3$ )

$$\begin{array}{cccccccc}
 & & & & & h_0 & h_1 & h_2 & h_3 \\
 \kappa_0[-2]: & & & & & & & & \\
 & & h_0 & h_1 & h_2 & h_3 & & & \\
 & & \uparrow & & & \uparrow & & & \\
 & & \mathbf{-2} & & & \mathbf{0} & & & 
 \end{array}$$

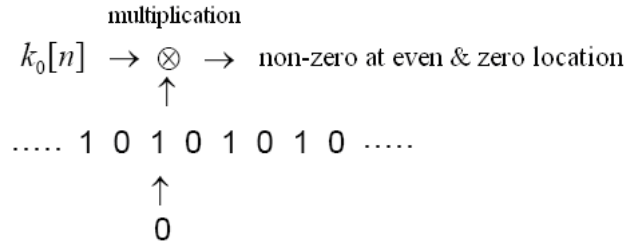
$$k_0[-2] = h_0h_2 + h_1h_3$$

That means the convolution between  $h[n]$  and  $h[-n]$  is symmetrical.

$$\begin{aligned}
 \kappa_0(Z) + \kappa_0(-Z) &= \text{Constant} \\
 \frac{1}{2}\{\kappa_0(Z) + \kappa_0(-Z)\} &= \text{Constant}
 \end{aligned}$$

From the above equation the summation  $\frac{1}{2}\{\kappa_0(Z) + \kappa_0(-Z)\}$  represents the nonzero sample value at even location and zero sample value at the odd location.

Let  $\kappa_0(Z)$  correspond to the sequence  $k_0[n]$ ,  $\frac{1}{2}\{\kappa_0(Z) + \kappa_0(-Z)\}$  impulse response is shown below.



But from the equation  $\frac{1}{2}\{\kappa_0(Z) + \kappa_0(-Z)\} = \text{Constant}$ , we want the non-zero sample value only at zero location and zero sample value for odd and even location.

So at the even location  $m = 2l$  and  $m \neq 0$  and ( $l \in \mathbb{Z}$ ), we want zero sample value.

Let Daubechies filter with length 4 ( $D = 3$ )

$$h_0[n] : \begin{array}{cccc}
 h_0 & h_1 & h_2 & \dots h_3 \\
 & \uparrow & & \uparrow \\
 & 0 & & 3
 \end{array}$$

In the Haar case,  $(1 - z^{-1})$  represents a High pass filter.

Here we consider the Daubechies filter with length 4 so two  $(1 - z^{-1})$  in the High pass filter which means  $(1 - z^{-1})^2$  factor in HPF.

Similarly, low pass filter has a factor  $(1 + z^{-1})^2$ .

A Daubechies low pass filter with length 4 is given by

$$H_0(Z) = h_0 + h_1z^{-1} + h_2z^{-2} + h_3z^{-3}$$

We can write this equation in the factor of  $(1 - z^{-1})^2$  *i.e.*

$$H_0(Z) = (1 + z^{-1})^2(1 + B_0z^{-1})$$

In the above equation, we need three zeros.

Two zeros are already chosen at unit circle which are  $-1, -1$  and one zero is selected based on value of  $B_0$ . This value can be obtained by comparing the above two equations.

Expanding the above two equations

$$\begin{aligned}
 H_0(Z) &= (1 + 2z^{-1} + z^{-2})(1 + B_0z^{-1}) \\
 H_0(Z) &= 1 + (2 + B_0)z^{-1} + (1 + 2B_0)z^{-2} + B_0z^{-3}
 \end{aligned}$$

The dot product of the impulse response of LPF with its even shifts must be zero. We will use this constraint to find the value of  $B_0$  in the next lecture.