

The filter representation of wavelets

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An $N \times N$ -matrix T , with N even, is called an MRA-matrix if the columns are translates of the first two columns in alternating order, in the same way as the columns of a circulant Toeplitz matrix.

We denote by H_0 the (unique) filter with the same first row as H , and by H_1 the (unique) filter with the same second row as H . H_0 and H_1 are also called the *DWT filter components*.

Let \mathbf{c}_m be the coordinates in ϕ_m , and let H_0, H_1 be defined as above. Any stage in a DWT can be implemented in terms of filters as follows:

- Compute $H_0 \mathbf{c}_m$. The even-indexed entries in the result are the coordinates \mathbf{c}_{m-1} in ϕ_{m-1} .
- Compute $H_1 \mathbf{c}_m$. The odd-indexed entries in the result are the coordinates \mathbf{w}_{m-1} in ψ_{m-1} .

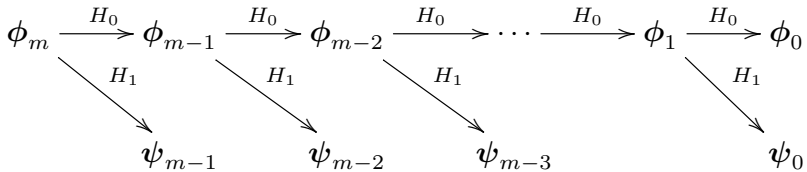


Figure: Detailed illustration of a wavelet transform.

We denote by G_0 the (unique) filter with the same first column as G , and by G_1 the (unique) filter with the same second column as G . G_0 and G_1 are also called the *IDWT filter components*.

Let G_0, G_1 be defined as above. Any stage in an IDWT can be implemented in terms of filters as follows:

$$\mathbf{c}_m = G_0 \begin{pmatrix} c_{m-1,0} \\ 0 \\ c_{m-1,1} \\ 0 \\ \dots \\ c_{m-1,2^{m-1}N-1} \\ 0 \end{pmatrix} + G_1 \begin{pmatrix} 0 \\ w_{m-1,0} \\ 0 \\ w_{m-1,1} \\ \dots \\ 0 \\ w_{m-1,2^{m-1}N-1} \end{pmatrix}.$$

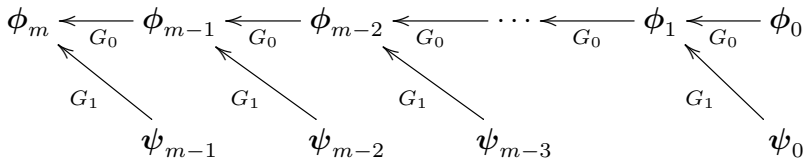


Figure: Detailed illustration of an IDWT.

The DWT can be computed with the help of two filters H_0, H_1 , as explained in Theorem 6.3. Any linear transformation computed from two filters H_0, H_1 in this way is called a *forward filter bank transform*. The IDWT can be computed with the help of two filters G_0, G_1 as explained in Theorem 6.5. Any linear transformation computed from two filters G_0, G_1 in this way is called a *reverse filter bank transform*.

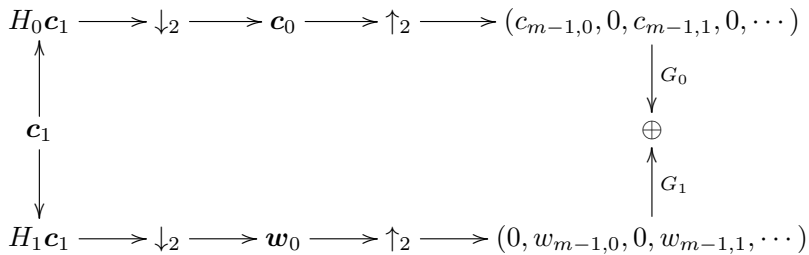


Figure: Detailed illustration of DWT and IDWT.

Dual filter bank transforms, Definition 6.7

Assume that H_0, H_1 are the filters of a forward filter bank transform, and that G_0, G_1 are the filters of a reverse filter bank transform. By the *dual transforms* we mean the forward filter bank transform with filters $(G_0)^T, (G_1)^T$, and the reverse filter bank transform with filters $(H_0)^T, (H_1)^T$.

Fact about the `dual`-parameter in the DWT:

- If the `dual` parameter is false, the DWT is computed as the forward filter bank transform with filters H_0, H_1 , and the IDWT is computed as the reverse filter bank transform with filters G_0, G_1 .
- If the `dual` parameter is true, the DWT is computed as the forward filter bank transform with filters $(G_0)^T, (G_1)^T$, and the IDWT is computed as the reverse filter bank transform with filters $(H_0)^T, (H_1)^T$.

Frequency responses

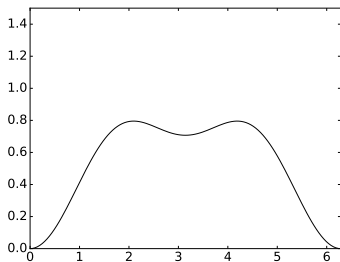
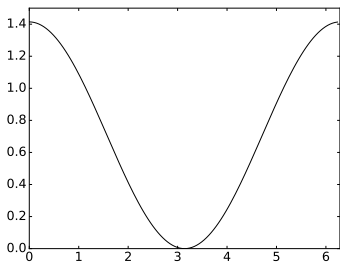
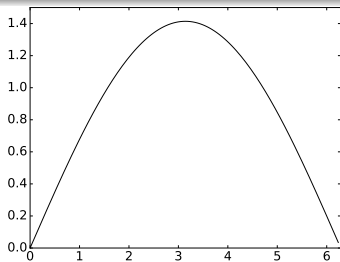
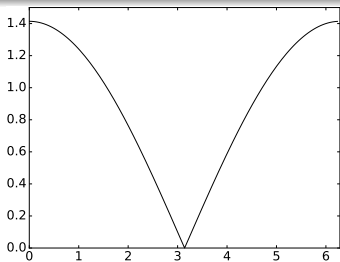


Figure: The frequency responses $\lambda_{G_0}(\omega)$ and $\lambda_{G_1}(\omega)$ for the Haar wavelet (top), and for the alternative piecewise linear wavelet (bottom).

Other frequency responses

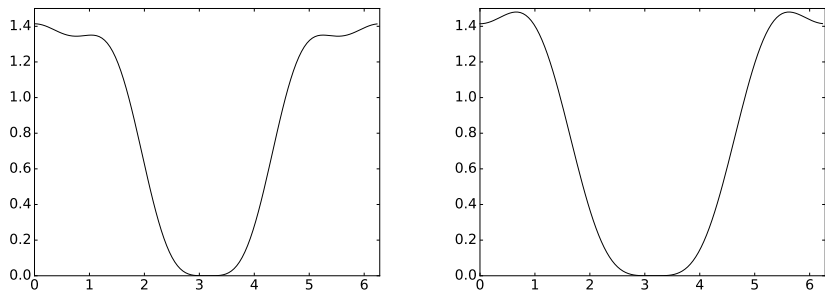


Figure: The frequency responses $\lambda_{H_0}(\omega)$ (left) and $\lambda_{G_0}(\omega)$ (right) for the CDF 9/7 wavelet.

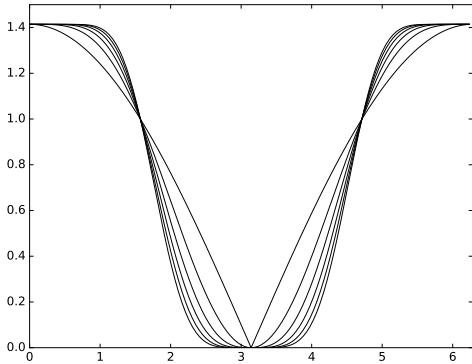


Figure: The magnitudes $|\lambda_{G_0}(\omega)| = |\lambda_{H_0}(\omega)|$ for the first orthonormal wavelets.

Forward filter bank transform, Definition 6.21

Let H_0, H_1, \dots, H_{M-1} be $N \times N$ -filters. A *forward filter bank transform* H produces output $\mathbf{z} \in \mathbb{R}^N$ from the input $\mathbf{x} \in \mathbb{R}^N$ in the following way:

- $z_{iM} = (H_0 \mathbf{x})_{iM}$ for any i so that $0 \leq iM < N$.
- $z_{iM+1} = (H_1 \mathbf{x})_{iM+1}$ for any i so that $0 \leq iM + 1 < N$.
- ...
- $z_{iM+(M-1)} = (H_{M-1} \mathbf{x})_{iM+(M-1)}$ for any i so that $0 \leq iM + (M - 1) < N$.

In other words, the output of a forward filter bank transform is computed by applying filters H_0, H_1, \dots, H_{M-1} to the input, and by downsampling and assembling these so that we obtain the same number of output samples as input samples (also in this more general setting this is called *critical sampling*). H_0, H_1, \dots, H_{M-1} are also called *analysis filter components*, the output of filter H_i is called *channel i channel*, and M is called the number of channels. The output samples z_{iM+k} are also called the *subband samples* of channel k .

Let G_0, G_1, \dots, G_{M-1} be $N \times N$ -filters. An *reverse filter bank transform* G produces $\mathbf{x} \in \mathbb{R}^N$ from $\mathbf{z} \in \mathbb{R}^N$ in the following way:

- Define $\mathbf{z}_k \in \mathbb{R}^N$ as the vector where $(\mathbf{z}_k)_{iM+k} = z_{iM+k}$ for all i so that $0 \leq iM+k < N$, and $(\mathbf{z}_k)_s = 0$ for all other s .

$$\mathbf{x} = G_0 \mathbf{z}_0 + G_1 \mathbf{z}_1 + \dots + G_{M-1} \mathbf{z}_{M-1}.$$

G_0, G_1, \dots, G_{M-1} are also called *synthesis filter components*.

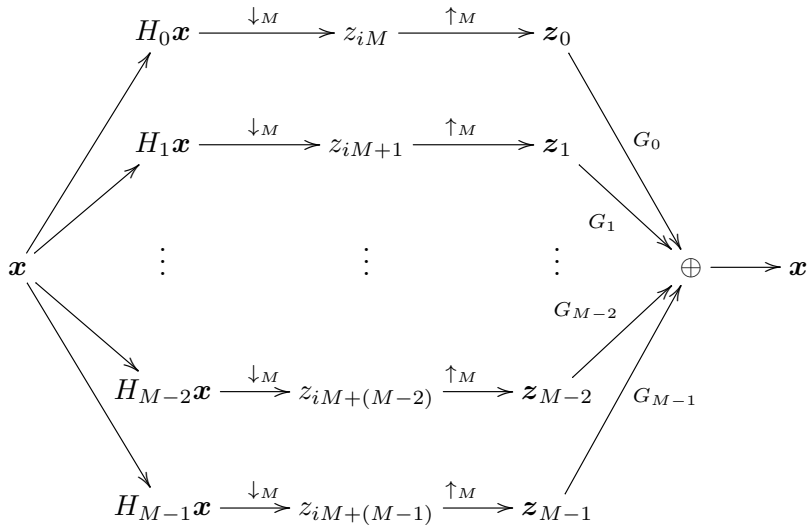


Figure: Illustration of forward and reverse filter bank transforms.