

Introduction to Image Processing:

# Global Operations: Wavelet Transform

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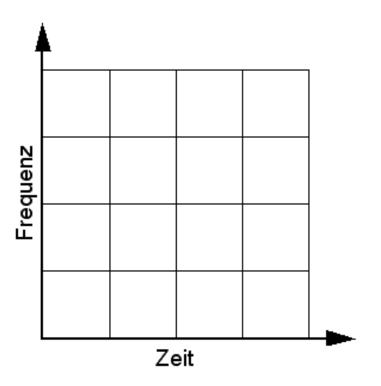


#### **Wavelet Transform**

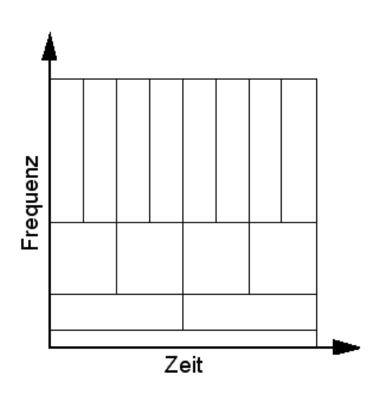
- The foundation of the Wavelet theory goes back to the 1930's.
- The today's wavelet transform was invented by the French physicists Jean Morlet and Alex Grossmann in the 1980's.
- A good introduction to wavelets is by Barbara Burke: "Wavelets: Die Mathematik der kleinen Wellen" [Burke97].
  - All images in this presentation are from this book.
- The motivation behind wavelets where some problems with the FT.

#### **Wavelet Transform vs. Windowowed Fourier Transform**

- The discrete Windowed Fourier transform (WFT) remains static.
- The wavelet transform analyses the signal in multiple resolutions:

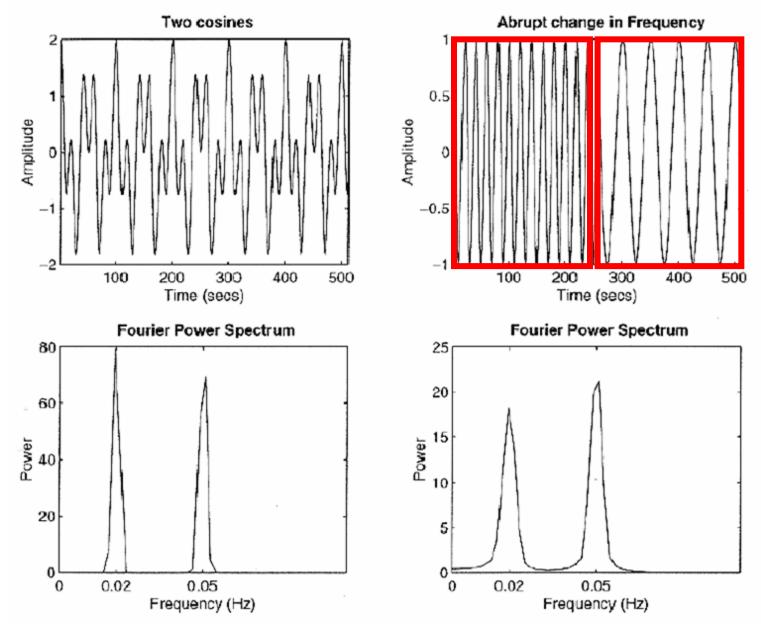


Static time-frequency resolution in FT



Multiple Resolution in WT

### **Wavelet Transform vs. Windowed Fourier Transform**

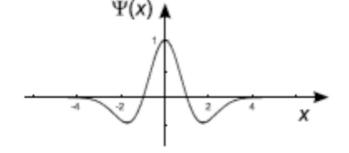


#### **Wavelet Transform: Base or Mother Wavelets**

- · Wavelets are functions that are used for sampling.
- By sampling we calculate the cover ration with the underlying signal section.
- Wavelets can be scaled and shifted.
- The unscaled and unshifted wavelet is called base or mother wavelet.

#### Basiswavelet:

Mexican hat



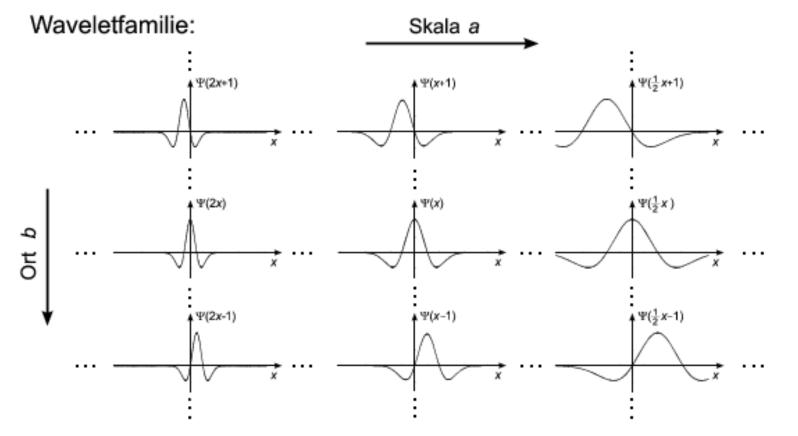
$$\Psi(x) = (1 - x^2) e^{-x^2/2}$$

#### **Wavelet Transform: Base Wavelet**

All versions of the base wavelet together are the wavelet family.

Basiswavelet:

Mexican hat  $\Psi(x) = (1 - x^2) e^{-x^2/2}$ 



#### **Wavelet Transform: Base Wavelet**

Base wavelets  $\Psi$  must fullfill the following conditions:

The area of the wavelet must be zero:

$$\int \psi(x) dx = 0$$

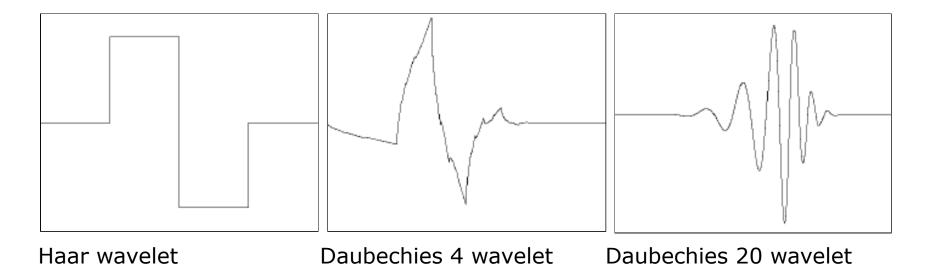
• A wavelet  $\Psi$  (a,b) of a wavelet family is defined as follows:

$$\psi_{a,b}(t) = \frac{1}{\sqrt{a}} \psi\left(\frac{t-b}{a}\right)$$

where a is the scale factor and b is the shift factor.

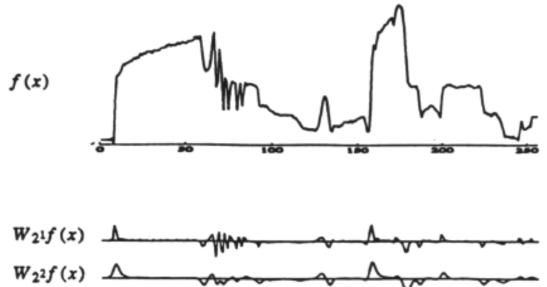
#### **Wavelet Transform: Base Wavelet**

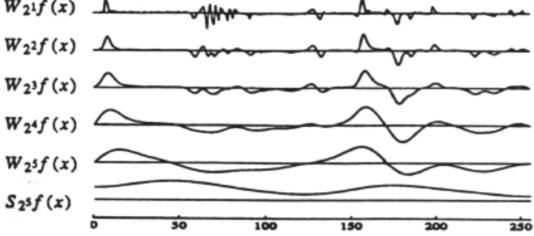
#### Some well known base wavelets:



#### **Wavelet Transform: Continuous WT**

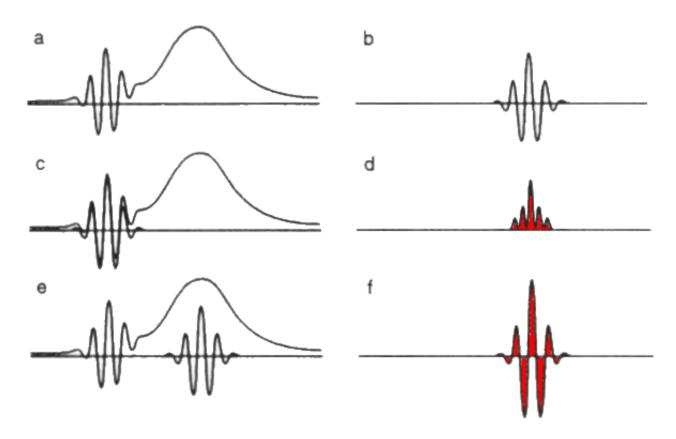
- By sampling the signal with scaled wavelets we get the wavelet transform.
- It corresponds to the cover ratio of the wavelets with the signal.
- The reconstruction is done as in the FT by summing up all wavelet coefficients.





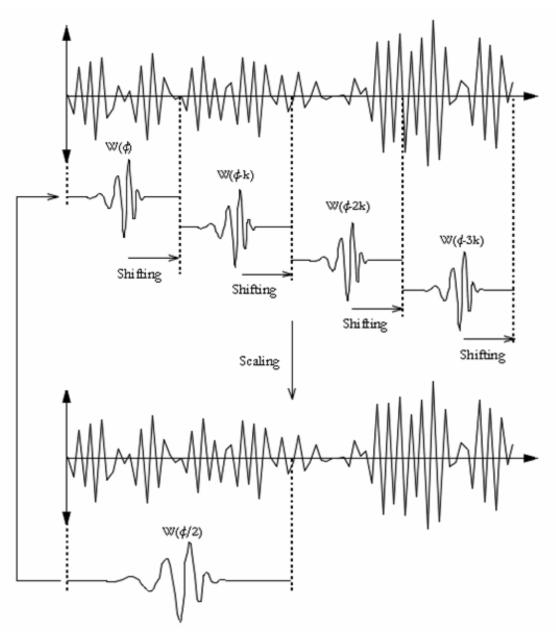
#### **Wavelet Transform: Continuous WT**

- How do we calculate the cover ratio of the wavelet with the signal?
- You multiply the wavelet function with the signal section.
- We calculate the area of the resulting function with integration.
- This value is called **Wavelet coefficient**.

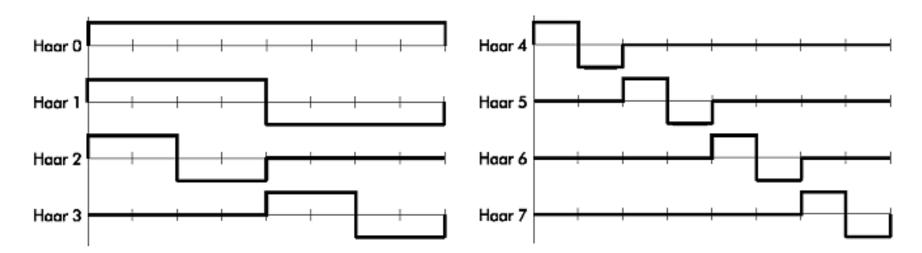


#### **Wavelet Transform: Discrete WT**

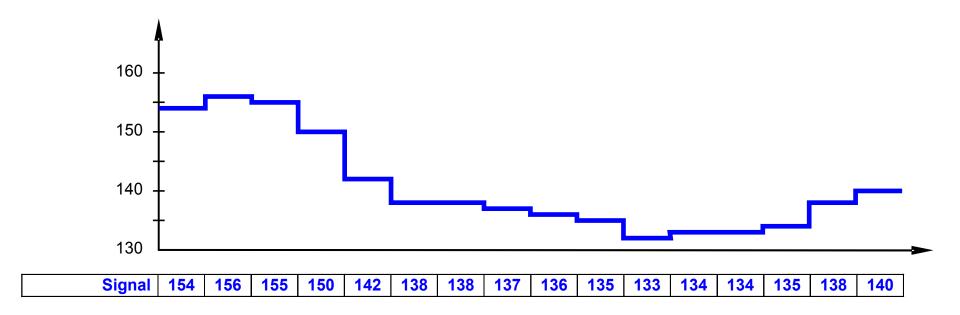
- The discrete WT (DWT)
   samples the signal in discrete
   step distances.
- The break through came in 1986 with the Fast Wavelet Transform (FWT) from Stéphane Mallat und Yves Meyer



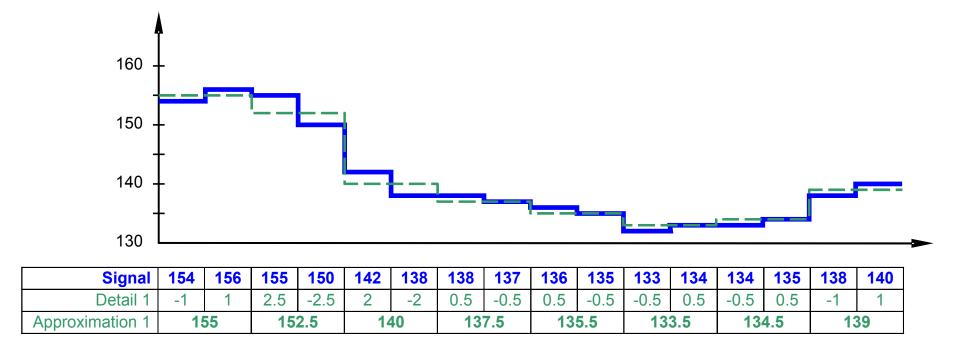
- The simplest Wavelet transform is the *Haar transform*.
- It is a simple and hard function.
- Through scaling and shifting we get the Haar Wavelet family:



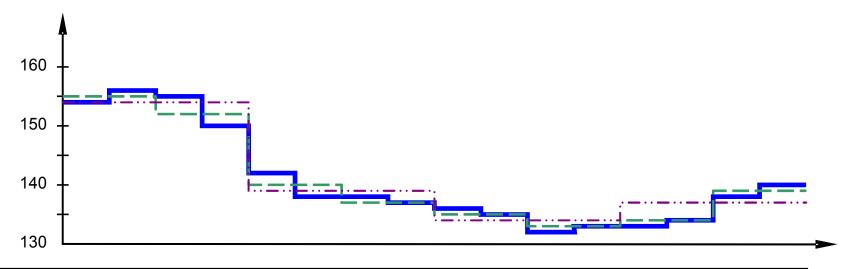
- We build the average of neigbouring cells (approximation).
- In the lines in between we write the difference (detail)



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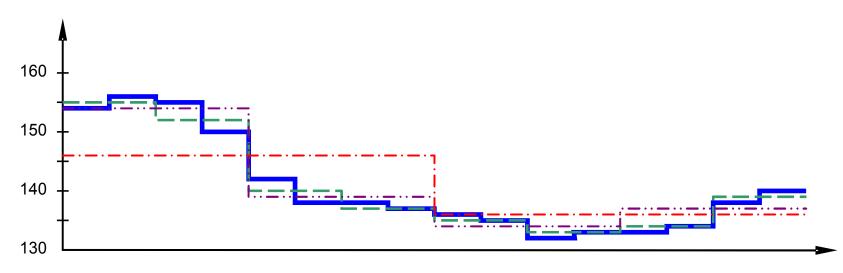


- We build the average of neigbouring cells (approximation).
- In the lines in between we write the difference (detail)



	Signal	154	156	155	150	142	138	138	137	136	135	133	134	134	135	138	140
	Detail 1	-1	1	2.5	-2.5	2	-2	0.5	-0.5	0.5	-0.5	-0.5	0.5	-0.5	0.5	-1	1
App	roximation 1	155		152.5		140		137.5		135.5		133.5		134.5		139	
Detail 2		1.	25	-1.25		1.25		-1.25		1		-1		-2.25		2.25	
Approximation 2			153	153.75			138.75			134.5				136.75			

- We build the average of neigbouring cells (approximation).
- In the lines in between we write the difference (detail)

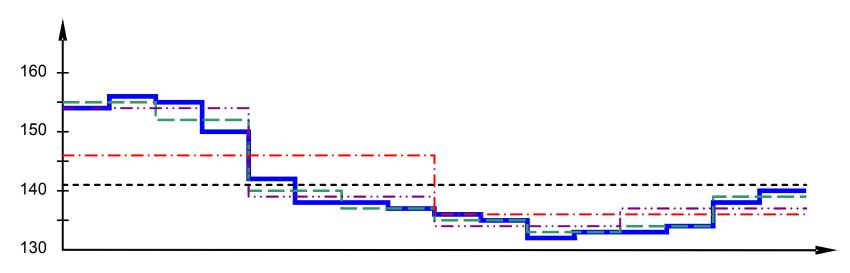


Signal	154	156	155	150	142	138	138	137	136	135	133	134	134	135	138	140	
Detail 1	-1	1	2.5	-2.5	2	-2	0.5	-0.5	0.5	-0.5	-0.5	0.5	-0.5	0.5	-1	1	
Approximation 1	155		152.5		140		137.5		135.5		133.5		134.5		139		
Detail 2	1.25		-1.25		1.3	1.25 -1.		25	1		-1		-2.25		2.25		
Approximation 2		153.75			138.75			134.5				136.75					
Detail 3	7.5			-7.5				-1.1	125		1.125						
Approximation 3		146.25								135.625							

• We 1 average value and 15 detail coefficiants:

```
140.9375, 5.3125, 7.5, -1.125, 1.25, 1,25, 1, -2.25, -1, 2.5, 2, 0.5, 0.5, -0.5, -0.5, -1
```

• Because the details are symmetric, we store only one detail value:



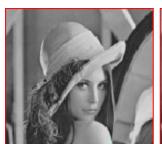
Signal	154	156	155	150	142	138	138	137	136	135	133	134	134	135	138	140	
Detail 1	-1	1	2.5	-2.5	2	-2	0.5	-0.5	0.5	-0.5	-0.5	0.5	-0.5	0.5	-1	1	
Approximation 1	1 155		152.5		140		137.5		135.5		133.5		134.5		139		
Detail 2	1.25		-1.	1.25 1.25		-1.	1.25 1		1	-1		-2.25		2.25			
Approximation 2		153	.75		138.75			134.5				136.75					
Detail 3		7	.5		-7.5				-1.1	125		1.125					
Approximation 3		146.25								135.625							
Detail 4	5.3125								-5.3125								
Approximation 4	140.9375																

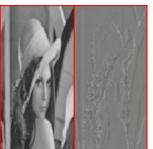
- The forward transform is called analysis.
- The approximation and details can also be built with filter masks:
- This is a 1D convolution (2 multiplies & 1 add):

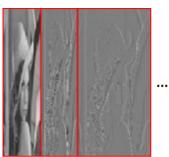
$$h_0 = \left[\frac{1}{2}, \frac{1}{2}\right]$$
 low pass filter (approximation)  $h_1 = \left[\frac{1}{2}, -\frac{1}{2}\right]$  high pass filter (detail)

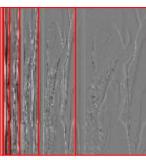
• The reconstruction called synthesis has the following filter masks:

$$f_0 = [1,1]$$
 left synthesis filter  $f_1 = [1,-1]$  right synthesis filter

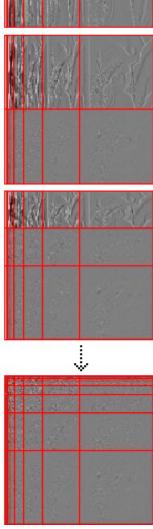








 If you apply the Haar transform first to all lines and then to all columns we get the standard decomposition of the 2D Haar transform.

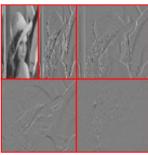


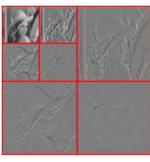


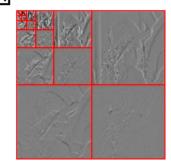


 If you apply the Haar transform alternately to all lines and columns we get the nonstandard decomposition of the 2D Haar transform.









- Image transformed with this hard Haar transform tend to get block artefacts and to be noisy.
- The normalized Haar transform is better.

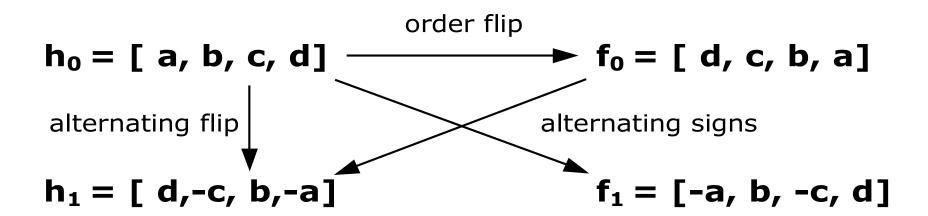
$$h_0 = \left[\frac{1}{\sqrt{2}}, \frac{1}{\sqrt{2}}\right] \text{ low pass filter (approximation)} \qquad h_1 = \left[\frac{1}{\sqrt{2}}, -\frac{1}{\sqrt{2}}\right] \quad \text{high pass filter (detail)}$$

The reconstruction has the following filter masks:

$$f_0 = \left[\frac{1}{\sqrt{2}}, \frac{1}{\sqrt{2}}\right]$$
 left synthesis filter  $f_1 = \left[\frac{1}{\sqrt{2}}, -\frac{1}{\sqrt{2}}\right]$  right synthesis filter

#### **Wavelet Transform: Filter bank**

- Other wavelet transforms have wider filter masks.
- We always need a low pass, a high pass and two synthesis filters.
- This set of filters is also called a filter bank.
- There is a specific relation between the filters.
- We therefore only need to define the low pass filter:



#### **Wavelet Transform: Filter bank**

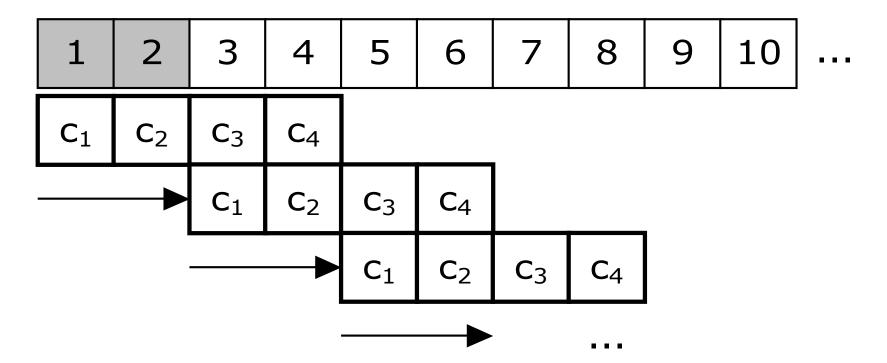
#### Normalized Haar (filter length 2)

```
h0 = [ 0.70710678118655, 0.70710678118655]
h1 = [ 0.70710678118655, -0.70710678118655]
f0 = [ 0.70710678118655, 0.70710678118655]
f1 = [ 0.70710678118655, 0.70710678118655]
```

#### • Daubechies 4: (filter length 8)

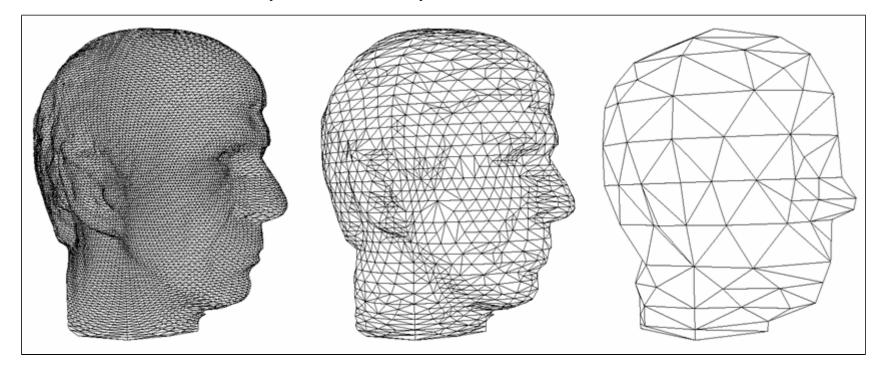
#### **Wavelet Transform: Filter bank**

- No matter how wide the filters are, they are move allways 2 cells.
- For filters wider than 2 we get a border problem:



# **Wavelet Transform: Applications**

- Image compression
- Multi scalen resolution (level of detail)



Audio analysis

## **Wavelet Transform: Applications**

Reconstruct the following 1D Haar transform:

89.6875,

1.6875, -5.875, 3.25, -4.5, -0.25, 2.25, 0.75, -1, -4, -1, -0.5, 1.5, 1, 0.5, 0

• The first value is called DC. Why?