An Animated Introduction to the Discrete Wavelet Transform

Revised Lecture Notes New Delhi December 2001

Arne Jensen

Aalborg University

An Animated Introduction to the Discrete Wavelet Transform - p.1/98

Reference

This is a tutorial introduction to the discrete wavelet transform. It is based on the book

A. Jensen and A. la Cour-Harbo: *Ripples in Mathematics The Discrete Wavelet Transform* Springer-Verlag 2001.

Book cover



A signal with 8 samples:

56, 40, 8, 24, 48, 48, 40, 16

We compute a transform as shown here:

56	40	8	24	48	48	40	16
48	16	48	28	8	-8	0	12
32	38	16	10	8	-8	0	12
35	-3	16	10	8	-8	0	12

To interpretation





56	40	8	24	48	48	40	16
48	16			8	-8		

56	40	8	24	48	48	40	16
48	16	48		8	-8	0	

56	40	8	24	48	48	40	16
48	16	48	28	8	-8	0	12

56	40	8	24	48	48	40	16
48	16	48	28	8	-8	0	12
				8	-8	0	12

56	40	8	24	48	48	40	16
48	16	48	28	8	-8	0	12
32		16		8	-8	0	12

56	40	8	24	48	48	40	16
48	16	48	28	8	-8	0	12
32	38	16	10	8	-8	0	12

56	40	8	24	48	48	40	16
48	16	48	28	8	-8	0	12
32	38	16	10	8	-8	0	12
		16	10	8	-8	0	12

56	40	8	24	48	48	40	16
48	16	48	28	8	-8	0	12
32	38	16	10	8	-8	0	12
35	-3	16	10	8	-8	0	12



32	38						
35	-3	16	10	8	-8	0	12

32	38	16	10	8	-8	0	12
35	-3	16	10	8	-8	0	12

48	16	48	28				
32	38	16	10	8	-8	0	12
35	-3	16	10	8	-8	0	12

48	16	48	28	8	-8	0	12
32	38	16	10	8	-8	0	12
35	-3	16	10	8	-8	0	12

56	40	8	24	48	48	40	16
48	16	48	28	8	-8	0	12
32	38	16	10	8	-8	0	12
35	-3	16	10	8	-8	0	12



35	35						
35	0	16	10	8	-8	0	12



51	19	45	25				
35	35	16	10	8	-8	0	12
35	0	16	10	8	-8	0	12

51	19	45	25	8	-8	0	12
35	35	16	10	8	-8	0	12
35	0	16	10	8	-8	0	12

59	43	11	27	45	45	37	13
51	19	45	25	8	-8	0	12
35	35	16	10	8	-8	0	12
35	0	16	10	8	-8	0	12

We now replace samples in the transformed signal below 9 by zero (thresholding) and then repeat the reconstruction procedure. The final result is:

51	51	19	19	45	45	37	13
51	19	45	25	0	0	0	12
35	35	16	10	0	0	0	12
35	0	16	10	0	0	0	12

Here is now a graphical representation of the results. Full line original signal, and dashed line for thresholding, left hand side 4, right hand side 9.



We now look at the transform in the first example. The direct transform $(a, b) \rightarrow (d, s)$ is given by

$$s = \frac{a+b}{2},$$
$$d = a - s.$$

and the inverse $(d, s) \rightarrow (a, b)$ by

$$a = s + d;,$$
$$b = s - d.$$

They can be realized as in-place transforms in two steps. The direct transform as

> First step: $a, b \rightarrow a, \frac{1}{2}(a+b)$ Second step: $a, s \rightarrow a-s, s.$

and the inverse transform as

First step: $d, s \rightarrow d+s, s$ Second step: $a, s \rightarrow a, 2s-a$.

Notation: Finite sequence of numbers (samples of a signal) of length 2^j is denoted by $s_j = \{s_j[1], s_j[2], \ldots, s_j[2^j]\}$. Basic idea in lifting is given in this figure:



An alternative to the first example is difference and mean computation, in that order:

$$a, b \to \delta, \mu$$

where

$$\delta = b - a$$
$$\mu = \frac{a+b}{2} = a + \frac{\delta}{2}$$

Predict: In the difference-mean case:

$$d_{j-1}[n] = s_j[2n+1] - s_j[2n].$$

In general:

$$\mathbf{d}_{j-1} = \mathsf{odd}_{j-1} - P(\mathsf{even}_{j-1}).$$

Update: In the difference-mean case:

$$s_{j-1}[n] = s_j[2n] + d_{j-1}[n]/2.$$

In general:

$$\mathbf{s}_{j-1} = \mathsf{even}_{j-1} + U(\mathbf{d}_{j-1}).$$
The transform $s_j \rightarrow s_{j-1}, d_{j-1}$ is called one step lifting. In the the first example we repeatedly applied the transform to the s-components, ending with s_0 of length 1. Two step discrete wavelet transform:



The difference and mean computations in the in place form:

$s_3[0]$	$s_{3}[1]$	$s_{3}[2]$	$s_{3}[3]$	$s_{3}[4]$	$s_{3}[5]$	$s_{3}[6]$	$s_{3}[7]$	
$s_3[0]$	$d_2[0]$	$s_{3}[2]$	$d_2[1]$	$s_{3}[4]$	$d_2[2]$	$s_{3}[6]$	$d_2[3]$	P
$s_2[0]$	$d_2[0]$	$s_2[1]$	$d_2[1]$	$s_2[2]$	$d_2[2]$	$s_2[3]$	$d_2[3]$	U
$s_2[0]$	$d_2[0]$	$d_1[0]$	$d_2[1]$	$s_2[2]$	$d_2[2]$	$d_1[1]$	$d_2[3]$	P
$s_1[0]$	$d_2[0]$	$d_1[0]$	$d_2[1]$	$s_1[1]$	$d_2[2]$	$d_1[1]$	$d_2[3]$	U
$s_1[0]$	$d_2[0]$	$d_1[0]$	$d_2[1]$	$d_0[0]$	$d_2[2]$	$d_1[1]$	$d_2[3]$	P
$s_0[0]$	$d_2[0]$	$d_1[0]$	$d_2[1]$	$d_0[0]$	$d_2[2]$	$d_1[1]$	$d_2[3]$	U

$s_3[0]$	$s_{3}[1]$	$s_{3}[2]$	$s_{3}[3]$	$s_{3}[4]$	$s_{3}[5]$	$s_{3}[6]$	$s_{3}[7]$







$s_3[0]$	$s_3[1]$	$s_{3}[2]$	$s_{3}[3]$	$s_{3}[4]$	$s_{3}[5]$	$s_{3}[6]$	$s_{3}[7]$	
$s_3[0]$	$d_2[0]$	$s_{3}[2]$	$d_2[1]$	$s_{3}[4]$	$d_2[2]$	$s_{3}[6]$	$d_2[3]$	P
$s_2[0]$	$d_2[0]$	$s_2[1]$	$d_2[1]$	$s_2[2]$	$d_2[2]$	$s_2[3]$	$d_2[3]$	U
$s_2[0]$	$d_2[0]$	$d_1[0]$	$d_2[1]$	$s_2[2]$	$d_2[2]$	$d_1[1]$	$d_2[3]$	P
$s_1[0]$	$d_{2}[0]$	$d_1[0]$	$d_2[1]$	$s_1[1]$	$d_2[2]$	$d_1[1]$	$d_2[3]$	U

$s_3[0]$	$s_{3}[1]$	$s_{3}[2]$	$s_{3}[3]$	$s_{3}[4]$	$s_{3}[5]$	$s_{3}[6]$	$s_{3}[7]$	
$s_3[0]$	$d_2[0]$	$s_{3}[2]$	$d_2[1]$	$s_{3}[4]$	$d_2[2]$	$s_{3}[6]$	$d_2[3]$	P
$s_2[0]$	$d_2[0]$	$s_2[1]$	$d_2[1]$	$s_2[2]$	$d_2[2]$	$s_2[3]$	$d_2[3]$	U
$s_2[0]$	$d_2[0]$	$d_1[0]$	$d_2[1]$	$s_2[2]$	$d_2[2]$	$d_1[1]$	$d_2[3]$	P
$s_1[0]$	$d_{2}[0]$	$d_1[0]$	$d_2[1]$	$s_1[1]$	$d_2[2]$	$d_1[1]$	$d_2[3]$	U
$s_1[0]$	$d_2[0]$	$d_1[0]$	$d_{2}[1]$	→ $d_0[0]$	$d_2[2]$	$d_1[1]$	$d_2[3]$	P

$s_3[0]$	$s_{3}[1]$	$s_{3}[2]$	$s_{3}[3]$	$s_{3}[4]$	$s_{3}[5]$	$s_{3}[6]$	$s_{3}[7]$	
$s_3[0]$	$d_2[0]$	$s_{3}[2]$	$d_2[1]$	$s_{3}[4]$	$d_2[2]$	$s_{3}[6]$	$d_2[3]$	P
$s_2[0]$	$d_2[0]$	$s_2[1]$	$d_2[1]$	$s_2[2]$	$d_2[2]$	$s_2[3]$	$d_2[3]$	U
$s_2[0]$	$d_2[0]$	$d_1[0]$	$d_2[1]$	$s_2[2]$	$d_2[2]$	$d_1[1]$	$d_2[3]$	P
$s_1[0]$	$d_2[0]$	$d_1[0]$	$d_2[1]$	$s_1[1]$	$d_2[2]$	$d_1[1]$	$d_2[3]$	U
$s_1[0]$	$d_2[0]$	$d_1[0]$	$d_{2}[1]$	$-d_0[0]$	$d_2[2]$	$d_1[1]$	$d_2[3]$	P
$s_0[0]$	$d_{2}[0]$	$d_{1}[0]$	$d_2[1]$	$d_0[0]$	$d_2[2]$	$d_1[1]$	$d_2[3]$	U

In place transform with pattern of computed values:

$s_3[0]$	$s_{3}[1]$	$s_{3}[2]$	$s_{3}[3]$	$s_{3}[4]$	$s_{3}[5]$	$s_{3}[6]$	$s_{3}[7]$	
$s_3[0]$	$d_{2}[0]$	$s_{3}[2]$	$d_{2}[1]$	$s_{3}[4]$	$d_{2}[2]$	$s_{3}[6]$	$d_{2}[3]$	P
$s_2[0]$	$d_2[0]$	$s_2[1]$	$d_{2}[1]$	$s_2[2]$	$d_2[2]$	$s_2[3]$	$d_2[3]$	U
$s_2[0]$	$d_2[0]$	$d_1[0]$	$d_2[1]$	$s_2[2]$	$d_2[2]$	$d_1[1]$	$d_2[3]$	P
$s_1[0]$	$d_2[0]$	$d_1[0]$	$d_2[1]$	$s_1[1]$	$d_2[2]$	$d_1[1]$	$d_2[3]$	U
$s_1[0]$	$d_2[0]$	$d_1[0]$	$d_2[1]$	$d_0[0]$	$d_2[2]$	$d_1[1]$	$d_2[3]$	P
$s_0[0]$	$d_2[0]$	$d_1[0]$	$d_2[1]$	$d_0[0]$	$d_2[2]$	$d_1[1]$	$d_2[3]$	U

A second example of lifting: Base prediction on assumption that signal is linear, ie $s_j[n] = \alpha n + \beta$. Prediction of $s_j[2n+1]$ is then $\frac{1}{2}(s_j[2n] + s_j[2n+2])$, and we need to save only $d_{j-1}[n] = s_j[2n+1] - \frac{1}{2}(s_j[2n] + s_j[2n+2])$.



The update step: Keep mean of $s_j[n]$ sequence equal to mean of $s_{j-1}[n]$ sequence. Final result is

$$d_{j-1}[n] = s_j[2n+1] - \frac{1}{2}(s_j[2n] + s_j[2n+2]),$$

$$s_{j-1}[n] = s_j[2n] + \frac{1}{4}(d_{j-1}[n-1] + d_{j-1}[n]).$$

Inverse transform:

$$s_j[2n] = s_{j-1}[n] - \frac{1}{4}(d_{j-1}[n-1] + d_{j-1}[n]),$$

$$s_j[2n+1] = d_{j-1}[n] + \frac{1}{2}(s_j[2n] + s_j[2n+2]).$$

Summary of one step lifting and inverse lifting:



Generalized lifting 1

One can generalize the lifting step by allowing several pairs of predictions and updates.



Generalized lifting 2

An example, Daubechies 4

$$s_{j-1}^{(1)}[n] = s_j[2n] + \sqrt{3}s_j[2n+1]$$

$$d_{j-1}^{(1)}[n] = s_j[2n+1] - \frac{1}{4}\sqrt{3}s_{j-1}^{(1)}[n] - \frac{1}{4}(\sqrt{3}-2)s_{j-1}^{(1)}[n-1]$$

$$s_{j-1}^{(2)}[n] = s_{j-1}^{(1)}[n] - d_{j-1}^{(1)}[n+1]$$

$$s_{j-1}[n] = \frac{\sqrt{3}-1}{\sqrt{2}}s_{j-1}^{(2)}[n]$$

$$d_{j-1}[n] = \frac{\sqrt{3}+1}{\sqrt{2}}d_{j-1}^{(1)}[n]$$

Generalized lifting 3

Last two steps are normalization steps, in order to preserve the energy in the transform, ie

$$\sum_{n} |s_j[n]|^2 = \sum_{n} |s_{j-1}[n]|^2 + \sum_{n} |d_{j-1}[n]|^2$$

now holds. Note that

$$\frac{\sqrt{3}-1}{\sqrt{2}} \cdot \frac{\sqrt{3}+1}{\sqrt{2}} = 1 \; .$$

Finally we can introduce the Discrete Wavelet Transform (DWT). Block diagrams are used for our lifting and inverse lifting based one step transforms:



A DWT over four scales

A DWT over four scales



A DWT over four scales



The inverse DWT over four scales

A DWT over four scales



The inverse DWT over four scales



A family of transforms (Cohen, Daubechies, Faveau)

$$\begin{split} d_{j-1}^{(1)}[n] &= s_j[2n+1] - \frac{1}{2}(s_j[2n] + s_j[2n+2]) \\ \texttt{CDF(2,2)} \qquad s_{j-1}^{(1)}[n] &= s_j[2n] + \frac{1}{4}(d_{j-1}[n-1] + d_{j-1}[n]) \\ \texttt{CDF(2,4)} \qquad s_{j-1}^{(1)}[n] &= s_j[2n] - \frac{1}{64}(3d_{j-1}[n-2] - 19d_{j-1}[n-1] \\ &\quad -19d_{j-1}[n] + 3d_{j-1}[n+1]) \\ \texttt{CDF(2,6)} \qquad s_{j-1}^{(1)}[n] &= s_j[2n] - \frac{1}{512}(-5d_{j-1}[n-3] + 39d_{j-1}[n-2] \\ &\quad -162d_{j-1}[n-1] - 162d_{j-1}[n] \\ &\quad + 39d_{j-1}[n+1] - 5d_{j-1}[n+2]) \\ d_{j-1}[n] &= \frac{1}{\sqrt{2}}d_{j-1}^{(1)}[n] \\ &\quad s_{j-1}[n] = \sqrt{2}s_{j-1}^{(1)}[n] \end{split}$$

Now some examples on synthetic signals: The first problem is how to visualize the action of the wavelet transform. We start with a simple signal and perform a three-scale Haar transform.



The coefficients separately. Note vertical range in plots.



Multiresolution representation of the DWT of a signal: Transform a signal $W_a^{(3)}$: $s_9 \rightarrow s_6, d_6, d_7, d_8$. Replace all entries but one in the transform by zeroes, and do the inverse transform. Schematically

$$W_{a}^{(3)} \colon \mathbf{s}_{9} \longrightarrow \underbrace{\mathbf{s}_{6}, \mathbf{d}_{6}, \mathbf{d}_{7}, \mathbf{d}_{8}}_{\bigvee}$$

$$W_{s}^{(3)} \colon \underbrace{\mathbf{0}_{6}, \mathbf{d}_{6}, \mathbf{0}_{7}, \mathbf{0}_{8}}_{\bigvee} \longrightarrow \mathbf{s}_{9}^{\prime}$$

Multiresolution representation of sine signal, three scales, Haar transform.



Singularity detection. Singularities can be localized in time using DWT. A sine plus a spike located at position 200:



We do some denoising examples. First based on the Haar transform. Here is the sine plus spike, and its multiresolution representation:



The idea in denoising is to keep the largest coefficients. On the left hand side we kept the 15% largest coefficients, and on the right hand side the 10% largest coefficients.



To get better performance one must use better wavelets. Same example, with CDF(2,2) (linear prediction) on the left, Daubechies 4 on the right. Largest 10% coefficients retained.



Same example with Daubechies transforms of length 8 and 12.



In the last example we show how to separate slow and fast variations in a signal. The function $\log(2 + \sin(3\pi\sqrt{t}))$, $0 \le r \le 1$, sampled 1024 times, and spikes added:



Multiresolution analysis, 6 scales, CDF(2,2):



Slow variation removed: Reconstruction based on d-components.



Interpretation 1

We recall the first example. We now apply the inversion procedure to the signals [1, 0, 0, 0, 0, 0, 0], [0, 1, 0, 0, 0, 0, 0], and [0, 0, 1, 0, 0, 0, 0].

1	1	1	1	1	1	1	1
1	1	1	1	0	0	0	0
1	1	0	0	0	0	0	0
1	0	0	0	0	0	0	0
1	1	1	1	-1	-1	-1	-1
---	----	----	----	----	----	----	----
1	1	-1	-1	0	0	0	0
1	-1	0	0	0	0	0	0
0	1	0	0	0	0	0	0

1	1	-1	-1	0	0	0	0
1	-1	0	0	0	0	0	0
0	0	1	0	0	0	0	0
0	0	1	0	0	0	0	0

Linear algebra interpretation as a matrix:



We do the same for the direct transform. Here is one example computation:

1	0	0	0	0	0	0	0
$\frac{1}{2}$	0	0	0	$\frac{1}{2}$	0	0	0
$\frac{1}{4}$	0	$\frac{1}{4}$	0	$\frac{1}{2}$	0	0	0
$\frac{1}{8}$	$\frac{1}{8}$	$\frac{1}{4}$	0	$\frac{1}{2}$	0	0	0

The result in matrix form for direct transform:

$$\mathbf{W}_{a}^{(3)} = \begin{bmatrix} \frac{1}{8} & \frac{1}{$$

Here is a graphical representation of the contents of $\mathbf{W}_{a}^{(3)}$:



It is one of the nontrivial results in wavelet theory that there always are either 2 or 4 waveforms behind each DWT. These waveforms get scaled and translated. By reconstructing from signals with zeroes except a single 1, one can find these waveforms. Here is an example using the inverse of the Daubechies 4 transform. We take the inverse transform of a signal with a one at place 6, and take lengths 8, 32, 128, 512, and 2048. The result is shown on the next slide.











Another example: inverse CDF(2,2), signal length 64, 1 at positions 40, 50, and 60.



Example using direct CDF(2,2):



We now present a generalization of the DWT to the Wavelet Packet Transform. Block diagram representation of one step DWT:



Note that we now put the average s components on the top, and the difference d components on the bottom, in this one step representation.



Our first example, full decomposition:

Our first example, full decomposition: Recall example

56	40	8	24	48	48	40	16
48	16	48	28	8	-8	0	12
32	38	16	10	8	-8	0	12
35	-3	16	10	8	-8	0	12

-8 -6-3-3

Our first example, full decomposition:

Our first example, full decomposition:





Possible representations of the signal:



48	16	48	28	8	-8	0	12

32	38	16	10	8	-8	0	12	

35	-3	16	10	8	-8	0	12			





WPT complexity 1

The number of possible representations of a signal grows very fast with the number of decomposition steps. We have:

Number of levels	Minimum signal length	Number of bases
1	1	1
2	2	2
3	4	5
4	8	26
5	16	677
6	32	458330
7	64	210066388901
8	128	44127887745906175987802

WPT complexity 2



The number of possible decompositions of a signal using j levels is denoted by A_j . We have $A_{j+1} = 1 + A_j^2$. We have the estimate $2^{2^{j-1}} < A_j < 2^{2^j}$. Example j = 10: $2^{2^9} \approx 10^{154}$ and $2^{2^{10}} \approx 10^{308}$.

Solution to complexity problem is the best basis algorithm. This is a very flexible algorithm, based on a cost function. A cost function is denoted by \mathcal{K} . It maps a finite length signal a to a number $\mathcal{K}(a)$. [a b] denotes the concatenation of two signals a and b. We require two properties:

Solution to complexity problem is the best basis algorithm. This is a very flexible algorithm, based on a cost function. A cost function is denoted by \mathcal{K} . It maps a finite length signal a to a number $\mathcal{K}(a)$. [a b] denotes the concatenation of two signals a and b. We require two properties:

Solution to complexity problem is the best basis algorithm. This is a very flexible algorithm, based on a cost function. A cost function is denoted by \mathcal{K} . It maps a finite length signal a to a number $\mathcal{K}(a)$. [a b] denotes the concatenation of two signals a and b. We require two properties:

An example: $\mathcal{K}(\mathbf{a}) =$ number of nonzero entries in \mathbf{a} .

$$5 = \mathcal{K}([1, 0, -1, 22, 0, 0, 2, -7])$$

= $\mathcal{K}([1, 0, -1, 22]) + \mathcal{K}([0, 0, 2, -7]) = 3 + 2$

Cost functions Threshold $\mathcal{K}_{thres}(\mathbf{a})$ equals number of elements in \mathbf{a} with absolute value greater than the threshold $\boldsymbol{\varepsilon}$. Example:

$$\varepsilon = 2.0:$$
 $\mathcal{K}_{\text{thres}}([1, 2, 30, -1, -4]) = 2$
 $\varepsilon = 1.0:$ $\mathcal{K}_{\text{thres}}([1, 2, 30, -1, -4]) = 3$
 $\varepsilon = 0.5:$ $\mathcal{K}_{\text{thres}}([1, 2, 30, -1, -4]) = 5$

Problem: Look out for rescaling hidden in transforms.

Cost functions ℓ^p -norm Notation: $\mathbf{a} = \{a[n]\}, 0 (useful values are$ <math>0) $<math>\mathcal{K}_{\ell^p}(\mathbf{a}) = \sum |a[n]|^p$.

Note that for p = 2 this is the energy in the signal.

Shannon entropy

$$\mathcal{K}_{\text{Shannon}}(\mathbf{a}) = \sum_{n} |a[n]|^2 \log(|a[n]|^2)$$

The best basis algorithm through the first example. Do a full decomposition. Result is:

56	40	8	24	48	48	40	16
48	16	48	28	8	-8	0	12
32	38	16	10	0	6	8	-6
35	-3	13	3	3	-3	1	7

Cost function: Number of entries with absolute value > 1. Compute cost of each vector in full decomposition:



Cost values are computed, and components are marked with cost values.












Compare cost of a pair of elements with the one just above. In case of lower or equal cost, move up. Adjust marking, if necessary.



Compare cost of a pair of elements with the one just above. In case of lower or equal cost, move up. Adjust marking, if necessary.

















Some things to note:

The best basis is not unique.

Some things to note:

- The best basis is not unique.
- A best basis with all components at the same level is called a best level basis.

Some things to note:

- The best basis is not unique.
- A best basis with all components at the same level is called a best level basis.
- With J levels the search algorithm is of order O(J log J). The full decomposition and the costs have to be computed only once.

Some things to note:

- The best basis is not unique.
- A best basis with all components at the same level is called a best level basis.
- With J levels the search algorithm is of order O(J log J). The full decomposition and the costs have to be computed only once.
- The size of the tree to be searched is independent of the length of the signal.

Discrete signal with finite energy

$$\mathbf{x} = \{x[n]\}_{n \in \mathbf{Z}}, \qquad \sum_{n \in \mathbf{Z}} |x[n]|^2 < \infty$$

Frequency contents ($j = \sqrt{-1}$):

$$X(\omega) = \sum_{n} x[n]e^{-jn\omega},$$

or with period T, ie n corresponds to sampling time nT,

$$X_T(\omega) = \sum x[n]e^{-jnT\omega}.$$

For a real signal $\overline{X_T}(\omega) = X_T(-\omega)$. Frequency contents in any interval $[k\pi/T, (k+1)\pi/T]$.



Discrete signal x[0], x[1], x[2], x[3], frequency interval $[0, \pi/T]$.



Same signal downsampled by 2, frequency interval $[0, \pi/2T]$.



FT of signal Filter response







DWT low pass



Product of FT and filters

IFT and $2\downarrow$

DWT high pass

One step DWT, eight samples. Energy distribution.

	$ d_2[0] ^2$	$ d_2[1] ^2$	$ d_2[2] ^2$	$ d_2[3] ^2$
$\frac{\pi}{2}$				
0	$ s_2[0] ^2$	$ s_2[1] ^2$	$ s_2[2] ^2$	$ s_2[3] ^2$
C) 1 2	2 3 4	4 5 6	5 7 8

Two step DWT, eight samples. Energy distribution.

π		I			
π	$ d_2[0] ^2$	$ d_2[1] ^2$		$ d_2[2] ^2$	$ d_2[3] ^2$
2	$ d_1 $	$0] ^2$		$ d_1 $	$[1] ^2$
0	$ s_1 $	$0] ^2$		$ s_1 $	$[1] ^2$
0) 1 2	2 3	4	5	6 7

Three step DWT, eight samples. Energy distribution.

π	ſ	1		
<u>π</u>	$ d_2[0] ^2$	$ d_2[1] ^2$	$ d_2[2] ^2$	$ d_2[3] ^2$
2	$ d_1[$	$0] ^2$	$ d_1 $	$[1] ^2$
		$ d_0 $	$[0] ^2$	
0		$ s_0 $	$[0] ^2$	
() 1 2	2 3	4 5	6 7 8

The first example, again:

(1)	56	40	8	24	48	48	40	16
(2)	48	16	48	28	8	-8	0	12
(3)	32	38	16	10	8	-8	0	12
(4)	35	-3	16	10	8	-8	0	12



More examples:



Explanation for previous example:



Frequency contents in WP decomposition, ideal filters:





Solution: Swap order in every other application of the DWT:



Significance of ordering, linear chirp. Left filter bank order, right natural frequency order.



Three frequencies, DWT and best level, J = 6.



A complicated signal, length 1024: Sum of

$$x[n] = \begin{cases} 25 & \text{if } n = 300 \ , \\ 1 & \text{if } 500 \le n \le 700 \ , \\ 15 & \text{if } n = 900 \ , \\ 0 & \text{otherwise} \ . \end{cases}$$

and

$$\sin(\omega_0 t) + \sin(2\omega_0 t) + \sin(3\omega_0 t) ,$$

with $\omega_0 = 405.5419$.

The signal



Time-frequency plane, Daubechies 4, DWT and best level, J = 6.



The Fourier transform 1

Review of the Fourier transform. There are at least four variants:

Acronym	Time	Frequency
CTCFFT	Continuous	Continuous
DTCFFT	Discrete	Continuous
CTDFFT	Continuous	Discrete
DTDFFT	Discrete	Discrete

The Fourier transform 2

CTCFFT $x(t) \longleftrightarrow \hat{x}(\omega)$

$$\hat{x}(\omega) = \int_{-\infty}^{\infty} x(t)e^{-j\omega t}dt \qquad x(t) = \frac{1}{2\pi} \int_{-\infty}^{\infty} \hat{x}(t)e^{j\omega t}d\omega$$

$$\int_{-\infty}^{\infty} |x(t)|^2 dt == \frac{1}{2\pi} \int_{-\infty}^{\infty} |x(\omega)|^2 d\omega$$

x(t) real-valued:

$$\overline{\hat{x}(\omega)} = \hat{x}(-\omega)$$

The Fourier transform 3

DTCFFT
$$x[n] \longleftrightarrow X(\omega)$$

$$X(\omega) = \sum_{n \in \mathbf{Z}} x[n] e^{-jn\omega} \qquad x[n] = \frac{1}{2\pi} \int_0^{2\pi} X(\omega) e^{in\omega} d\omega$$

$$\sum_{n \in \mathbf{Z}} |x[n]|^2 = \frac{1}{2\pi} \int_0^{2\pi} |X(\omega)|^2 d\omega$$

x[n] real-valued:

$$\overline{X(\omega)} = X(-\omega)$$

CTDFFT Interchange role of time and frequency above.
The Fourier transform 4

DTDFFT $\mathbf{x} \longleftrightarrow \hat{\mathbf{x}}$ Orthogonal basis for $\mathbf{C}^N \{\mathbf{e}_k\}_{k=0,...,N-1}$ given by

$$e_k[n] = e^{j2\pi nk/N}, \quad k, n = 0, \dots, N-1$$

$$\hat{x}[k] = \sum_{n=0}^{N-1} x[n] e^{-j2\pi nk/N} \qquad x[n] = \frac{1}{N} \sum_{k=0}^{N-1} \hat{x}[k] e^{j2\pi nk/N}$$

$$\sum_{n=0}^{N-1} |x[n]|^2 = \frac{1}{N} \sum_{k=0}^{N-1} |\hat{x}[k]|^2$$

The Fourier transform 5

 $\mathbf{x} \in \mathbf{C}^N$ realvalued. Then

$$\overline{\hat{x}}[k] = \sum_{n=0}^{N-1} x[n] e^{j2\pi nk/N} = \sum_{n=0}^{N-1} x[n] e^{-j2\pi n(N-k)/N} = \hat{x}[N-k]$$

Comparing DTDF with DTCF we see that $\hat{\mathbf{x}}$ is obtained by sampling $X(\omega)$ at the frequencies $0, 2\pi/N, \ldots, 2\pi(N-1)/N$, ie

$$\hat{x}[k] = X(2\pi k/N)$$

Sampling 1

A continuous signal x(t) is sampled at times nT, $n \in \mathbb{Z}$. Fourier series with this time unit:

$$X_T(\omega) = \sum_n x[n]e^{-jnT\omega}$$

Relation to the CTCFFT:

$$X_T(\omega) = \frac{1}{T} \sum_{k \in \mathbf{Z}} \hat{x} \left(\omega - \frac{2k\pi}{T} \right)$$

Sampling 2

Illustration of aliasing effect (undersampling):



Short Time Fourier Transform 1

The Short Time Fourier Transform (STFT) is based on DTCFFT and a window function:

$$X_{\text{STFT}}(k,\omega) = \sum_{n \in \mathbf{Z}} w[n-k]x[n]e^{-jnT\omega}$$

Let x be a signal of length N. Usual choice of k is for N even is k = mN/2, $m \in \mathbb{Z}$, and for N odd k = m(N-1)/2, $m \in \mathbb{Z}$.

The window function w gives a localization in time. Example is Hanning window:

$$w[n] = \sin^2(\pi(n-1)/N), \quad n = 1, \dots, N$$

Short Time Fourier Transform 2

Examples with N = 16: Rectangular, triangular, Hanning and Gaussian windows.



Short Time Fourier Transform 3

The spectrogram is obtained by plotting

$$\frac{1}{2\pi} |X_{\rm STFT}(k, 2\pi n/N)|^2$$

for values of k determined by the length of the window, and for n = 0, ..., N - 1. Visualized in the time-frequency plane by using cells of a size determined by the length of the window in the frequency direction and by the length of the signal and the overlap in the time direction.

An example

We compute the spectrogram of the signal used above. On the left hand side we use a Hanning window of length 256, on the right hand side the length is 64.



In the final example we compare the methods on a complicated signal. We perform two wavelet and two Fourier based analyses of the signal. The first two are STFT based, with a long and a short window. forcing us to identify either slow or fast oscillations, but not both. The wavelet based analysis shows first the result of a level basis analysis. The final one uses the best basis algorithm with the Shannon entropy. This clearly gives a superior result.

Spectrogram, 1024 point FFT, windows 64, overlap 16.



Spectrogram, 1024 point FFT, windows 512, overlap 400.



Wavelet packet level basis, symlet 12.



Wavelet packet, best basis, Shannon entropy, symlet 12.



Final remarks

I have introduced you to the discrete wavelet transform and its generalization, the wavelet packet transform. I have also reviewed some results from Fourier analysis, and shown you a comparative study on two signals. If you want to learn more, start by reading the book mentioned in the introduction, and then start experimenting with the transforms, both on synthetic signals, and on real world signals.

Thank your for your attention!

Technical afterword

One question often asked, after someone has seen this presentation, is how it is produced. It is produced using $extsf{E}T_{E}X$ with the document class prosper. It can be found at

http://prosper.sourceforge.net
It is of course in the public domain.