

## II) Continuous wavelets transform

### Definition (wavelet)

A wavelet is a function  $\psi \in L^*(\mathbb{R})$  s.t.:

- (i)  $\|\psi\|_2 = 1$
- (ii) has 0-average
- (iii)  $\psi$  is centred around 0
- (iv)  $\psi$  satisfies the following admissibility condition

$$0 < c_\psi = \int_0^{+\infty} \frac{|\hat{\psi}(v)|^2 dv}{v} < +\infty$$

$\Rightarrow$  We define for  $(v, s) \in \mathbb{R} \times \mathbb{R}_+^*$ , the family of functions  $\{\psi_{s,v} = \frac{1}{\sqrt{s}} \psi\left(\frac{\cdot - v}{s}\right)\}$  obtained by translating (by  $v$ ) and dilating (by  $s$ ) the function  $\psi$

Definition (continuous wavelet transform)

The continuous wavelet transform (CWT) of a signal  $f \in L^2(\mathbb{R})$  at time  $v \in \mathbb{R}$

and scale  $s > 0$

is defined as  $Wf(v, s) = \langle f, \psi_{v,s} \rangle$  (\*)

$$= \int_{\mathbb{R}} f(t) \frac{1}{\sqrt{s}} \psi^* \left( \frac{t-v}{s} \right) dt$$

Remark

(i)  $Wf$  can be interpreted as the result of a linear filtering operation

Introducing  $\tilde{\psi}_s = \frac{1}{\sqrt{s}} \psi^* \left( \frac{\cdot}{s} \right)$

then (\*) can be rewritten as

$$Wf(v, s) = \int_{\mathbb{R}} f(t) \tilde{\psi}_s(v-t) dt = f * \tilde{\psi}_s(v)$$

Since  $\hat{\Psi}(0) = \int_{\mathbb{R}} \Psi(f) df = 0$  (by definition)

and

$$\mathcal{F}[\tilde{\Psi}_s](\omega) = \sqrt{D} \hat{\Psi}^*/(\omega)$$

$\tilde{\Psi}_s$  can be interpreted as (the impulse response of)  
a band-pass filter

- 2 types (i) analytic wavelets (complex-valued)  
aimed at defining / capturing "instantaneous  
frequencies"  $\rightarrow$  not covered here
- (ii) real wavelets to detect sharp  
transitions, singularities  
and characterize textures  
(local regularity)

## 2) Real wavelets ( $\Psi(t) \in \mathbb{R}$ )

Example

continuous function per morlet

- (i) Haar wavelets (appropriate for the description of piece-wise constant signals  $\rightarrow$  sparse representation)
- (ii) Mescion hot  $\rightarrow$  see slides  
 $\Rightarrow$  band-pass filter, capturing details/texture information from an input signal

Remark

The decay of wavelet coefficient as  $D \rightarrow 0$  characterizes the regularity of a signal around the time-instant  $a$

## 2.1 Inversion on energy conservation

Theorem (Eiduron (1964), Grossman, Morlet (1984))

Let  $\psi \in L^2(\mathbb{R})$  be real valued, then

for any  $f \in L^2(\mathbb{R})$ ,

$$f \in \mathbb{R}, f(t) = \frac{1}{C_\psi} \int_0^{+\infty} \int_{\mathbb{R}} W_f(u, s) \frac{1}{\sqrt{s}} \psi(t-u) du ds$$

inversion formula

and

$$\|f\|_2^2 = \frac{1}{C_\psi} \int_0^{+\infty} \int_{\mathbb{R}} |W_f(u, s)|^2 du ds$$

energy conservation

↳ do not need to retain the expressions

## 2.2. Scoring function

When  $Wf(n, s)$  is known only for  $s < s_0$ ,  
the one also needs to know  $Wf(n, s)$  for  $s > s_0$   
to apply the reconstruction formula.

- This is obtained by introducing the so-called Scoring function  $\phi$  that is an exaggeration of waves at scales larger than  $s_0$ . The scoring function is such that :-

$$|\hat{\phi}(z)|^2 = \int_1^{+\infty} |\hat{\psi}(s)|^2 \frac{ds}{s} = \int_{\gamma}^{+\infty} |\hat{\psi}(\xi)|^2 \frac{d\xi}{\xi}$$

and its phase can be defined arbitrarily  $\tilde{\xi}$

- To guarantee that  $\hat{\phi}(0)$  is finite, the admissibility condition ensures that  $\lim_{|z| \rightarrow 0} |\hat{\phi}(z)|^2 = C_\phi < +\infty$

- The scoring function can be associated with the impulse response of a low-pass filter, complemented the band-pass filter associated with  $\psi$ .

Summary:  $\psi$  (mother wavelet)  $\rightarrow$  low-pass filter  
 $\phi$  (scaling function)  $\rightarrow$  low-pass "father wavelet" filter

$$\tilde{\psi}_D = \frac{1}{\sqrt{D}} \cdot \phi^* \left( -\frac{\cdot}{D} \right) \text{ a low-frequency}$$

approximation of  $f \in L^2(\mathbb{R})$  at a scale  $D$ , around  $u$  is:

$$Wf(u, D) = \left\langle f, \frac{1}{\sqrt{D}} \phi \left( \frac{\cdot - u}{D} \right) \right\rangle = f^* \tilde{\psi}_D(u)$$

Rewriting the reconstruction formula leads to

$$f(t) = \frac{1}{C_\psi} \int_0^{D_0} Wf(\cdot, D) * \psi_D(u) \frac{du}{D^2} + \frac{1}{C_\psi D_0} [f(\cdot, D_0) * \phi_{D_0}(t)]$$

low-frequency approximation

$$\left\{ \begin{array}{l} \psi_s = \frac{1}{\sqrt{s}} \psi\left(\frac{\cdot}{s}\right) \\ \phi_s = \frac{1}{\sqrt{s}} \phi\left(\frac{\cdot}{s}\right) \end{array} \right.$$

## Summary of representation covered

### Time-frequency representation

- Short-time Fourier transform (STFT)

$$Sf(v, \xi) = \langle f, g_{v, \xi} \rangle = \int_{\mathbb{R}} f(t) g(t-v) e^{-2\pi v \xi} dt$$

- Objectives

- time-frequency analysis

- fixed-size tessellation of the time-frequency plane (Heisenberg boxes)

$$g^*_{v, \xi}(t)$$

### Time-scale representation

- Continuous wavelet transform (CWT)

$$Wf(v, s) = \langle f, \Psi_{v, s} \rangle = \int_{\mathbb{R}} f(t) \frac{1}{\sqrt{s}} \Psi^* \left( \frac{t-v}{s} \right) ds$$

- Objectives

(i) real wavelets

- detect sharp transition band regularity

• Wavelet  $\Psi$   $\rightarrow$  band-pass filter  
 $\rightarrow$  captures details

• Scaling function  $\phi$   $\rightarrow$  low-pass filter  
 $\rightarrow$  smooth approximation at scale  $s$ ,

(ii) analytic wavelets

- characterize the temporal evolution of the frequency content of a signal using scale-adaptive Heisenberg boxes

## IV Orthogonal wavelet bases

### 1) Practical interest and objectives

Objectives obtain a representation of a signal composed of a limited number of non zero coefficients (sparse representation)

#### a) Approximation

Let  $f \in L^2(\mathbb{R})$  and  $\{e_m\}_{m \in \mathbb{N}}$  an orthonormal basis of  $L^2(\mathbb{R})$ .

• Linear approximation error

$$E_L(N) = \|f - f_N\|_2^2 = \sum_{m \geq N} |\langle f, e_m \rangle|^2 \text{ with}$$

$$f_N = \sum_{m=0}^{N-1} \langle f, e_m \rangle e_m$$

• Non linear approximation error

Let  $I_N$  denote the subspace of indices corresponding to the  $N$  largest value of  $|\langle f, e_m \rangle|$

$$\text{Then: } \varepsilon_{NL}(N) = \left\| \left\| 1 - \sum_{n \in I_N} \langle f, e_n \rangle e_n \right\|_2 \right\|^2$$

$\underbrace{\phantom{\sum_{n \in I_N}}}_{f_N}$

$$= \sum_{m \notin I_N} |\langle f, e_m \rangle|^2$$

$\rightarrow$  equivalent to using a fine approximation around features of interest (singularities for the wavelet transform)

## b) Denoising

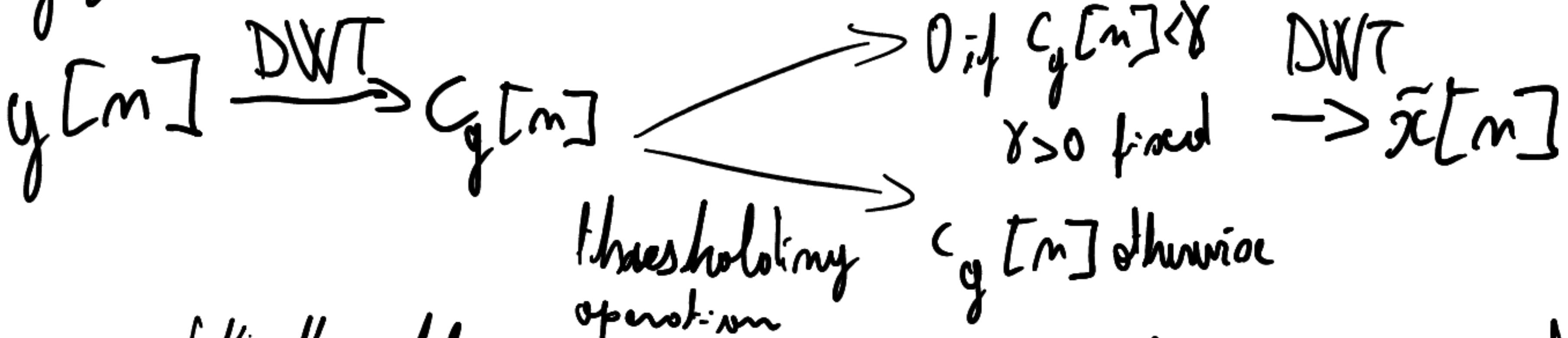
In general, the noise affects all the coefficients of a signal in the original domain

→ a representation adapted to the content of a signal concentrates the energy of the signal onto a few coefficients, corresponding to those with the highest magnitude

Example Donoho & Johnstone (1994)

have shown that a simple thresholding operation of the wavelet coeffs is efficient for denoising

$$y[n] = x[n] + w[n]$$



→ a difficult problem in the original can be addressed with a basic procedure

c) Compression: Keep only  $N$  most significant decomposition coefficient

JPEG  $\leftarrow$  DCT (discrete cosine transform)