



**27<sup>th</sup> EUSIPCO 2019**  
European Signal  
Processing Conference

A Coruña, Spain, September 2-6, 2019

Second-order time-reassigned synchrosqueezing:  
Application to Draupner wave analysis

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September 5, 2019



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## Purpose of this work

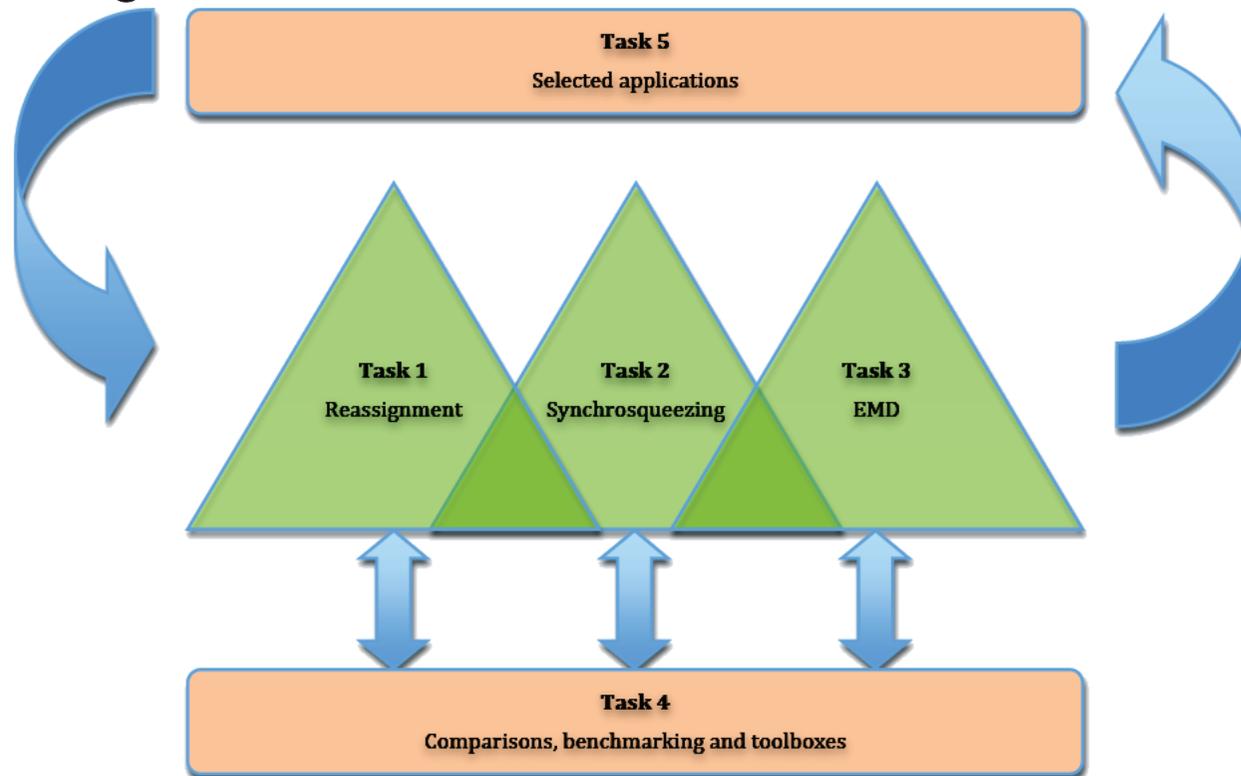
### Goals

- Computing efficient representations for handling non-stationary multicomponent signals
- Dealing with impulsive-like and/or strongly modulated signals
- Reversible representation allowing an extraction of the elementary signal components
- Meaningful **information extraction** from observation

⇒ **Proposed approach:** Time-frequency analysis combined with reassignment-based post-processing methods (i.e. synchrosqueezing)

Context of our research:  
the ASTRES project 2014-2017 funded by the French ANR

**ASTRES:** Analysis, Synthesis and Transformation by Reassignment, EMD and Synchrosqueezing.



Offers a new toolbox for non-stationary multicomponent signal processing:  
[https://github.com/dfourer/ASTRES\\_toolbox](https://github.com/dfourer/ASTRES_toolbox)

D. Fourer, J. Harmouche, J. Schmitt, T. Oberlin, S. Meignen, F. Auger and P. Flandrin. The ASTRES Toolbox for Mode Extraction of Non-Stationary Multicomponent Signals. Proc. EUSIPCO 2017, Aug. 2017. Kos Island, Greece.

## The new French ANR ASCETE project: 2019-2022

**ASCETE:** Analysis and Separation of Complex signal: Exploiting the Time-frequency structure



Project holder: Sylvain Meignen (LJK, Grenoble)

- Extends the previous methods with stochastic models
- Combines signal processing methods with machine learning
- New applications to audio, biomedicine, astrophysics, etc.

# Plan

- 1 Introduction
- 2 Time-reassigned synchrosqueezing
  - Rationale
  - Second-order time-reassigned synchrosqueezing
- 3 Application to Draupner wave analysis
  - Data description
  - Application to Draupner wave analysis
- 4 Conclusion

## STFT definitions

- We define the STFT of a signal  $x$  as a function of time  $t$  and frequency  $\omega$  computed using a differentiable analysis window  $h$  as:

$$F_x^h(t, \omega) = \int_{\mathbb{R}} x(\tau) h(t - \tau)^* e^{-j\omega\tau} d\tau \quad (1)$$

with  $j^2 = -1$  the imaginary unit and  $z^*$  the complex conjugate of  $z$ .

- A time-frequency representation is provided by the spectrogram defined as:  $|F_x^h(t, \omega)|^2$ .
- Signal reconstruction formula:

$$x(t) = \frac{1}{2\pi h(0)^*} \int_{\mathbb{R}} F_x^h(t, \omega) e^{j\omega t} d\omega \quad (2)$$

when  $h(0) \neq 0$

## The reassignment method [Kodera *et al.* 1978] [Auger, Flandrin, 1995]

### Principle

Improves the readability of a time-frequency representation (TFR):

Reassignment moves the signal energy according to:  $(t, \omega) \mapsto (\hat{t}_x(t, \omega), \hat{\omega}_x(t, \omega))$ , where  $\hat{t}_x(t, \omega)$  is a group-delay estimator and  $\hat{\omega}_x(t, \omega)$  is an instantaneous frequency estimator.

Both time-frequency reassignment operators can be computed as follows (STFT case):

$$\hat{t}_x(t, \omega) = \text{Re}(\tilde{t}_x(t, \omega)), \text{ with } \tilde{t}_x(t, \omega) = t - \frac{F_x^{\mathcal{T}}h(t, \omega)}{F_x^h(t, \omega)} \quad (3)$$

$$\hat{\omega}_x(t, \omega) = \text{Im}(\tilde{\omega}_x(t, \omega)), \text{ with } \tilde{\omega}_x(t, \omega) = j\omega + \frac{F_x^{\mathcal{D}}h(t, \omega)}{F_x^h(t, \omega)} \quad (4)$$

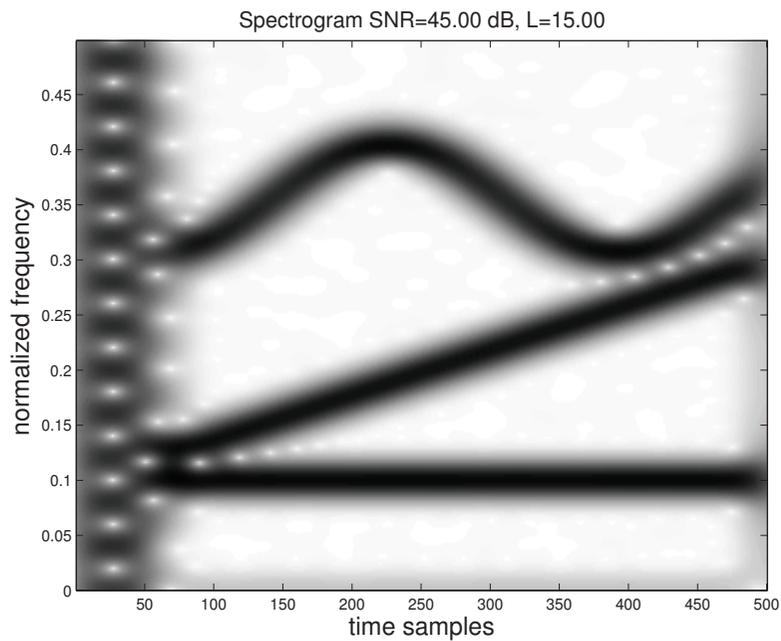
where  $\mathcal{T}h(t) = th(t)$  and  $\mathcal{D}h(t) = \frac{dh}{dt}(t)$  are modified versions of the analysis window  $h$ . The reassigned spectrogram is computed by:

$$\text{RF}_x^h(t, \omega) = \iint_{\mathbb{R}^2} |F_x^h(\tau, \Omega)|^2 \delta(t - \hat{t}_x(\tau, \Omega)) \delta(\omega - \hat{\omega}_x(\tau, \Omega)) d\tau d\Omega. \quad (5)$$

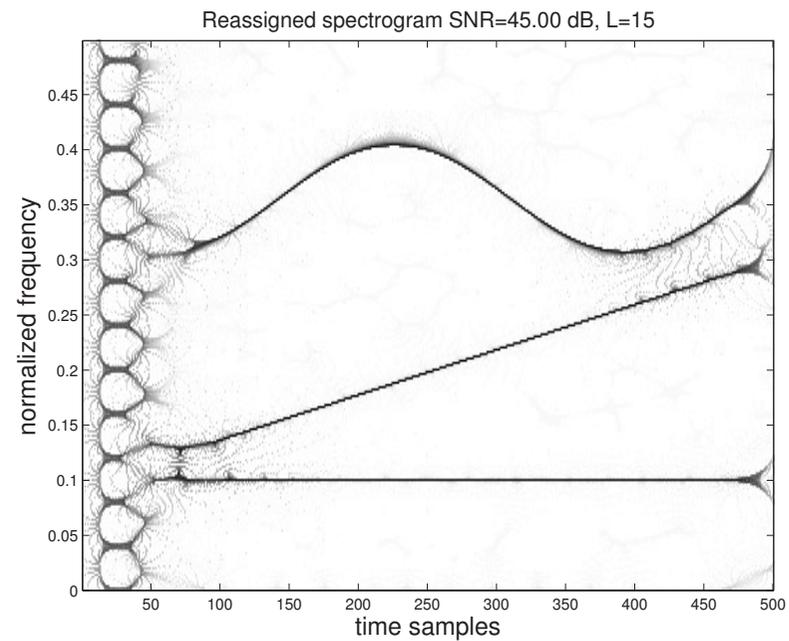
The resulting reassigned spectrogram  $\text{RF}_x(t, \omega)$  is a sharpened but non-reversible TFR due to the loss of the phase information.

# The reassigned spectrogram

$$R_x(t, \omega) = \iint_{\mathbb{R}^2} |F_x^h(\tau, \Omega)|^2 \delta(t - \hat{t}(\tau, \Omega)) \delta(\omega - \hat{\omega}(\tau, \Omega)) d\tau \frac{d\Omega}{2\pi} \quad (6)$$



(a)  $|F_x^h(t, \omega)|^2$



(b)  $R_x^h(t, \omega)$

# Synchrosqueezing [Daubechies 1996, 2011] [Thakur 2011]

## Principle

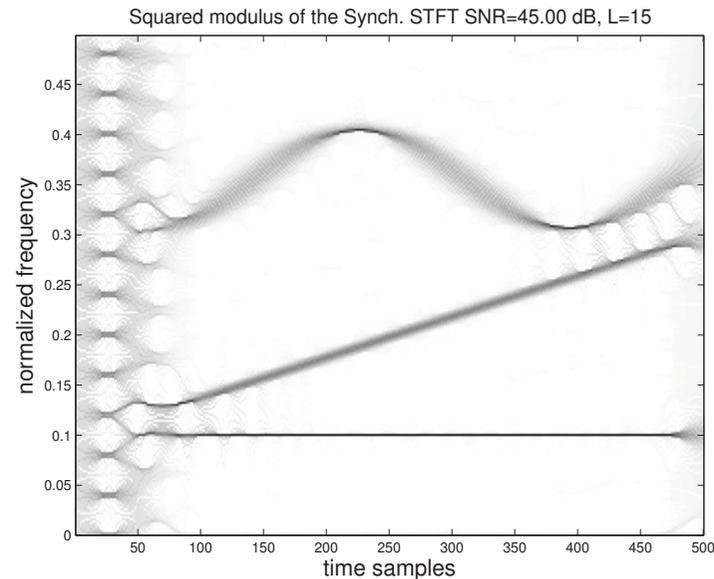
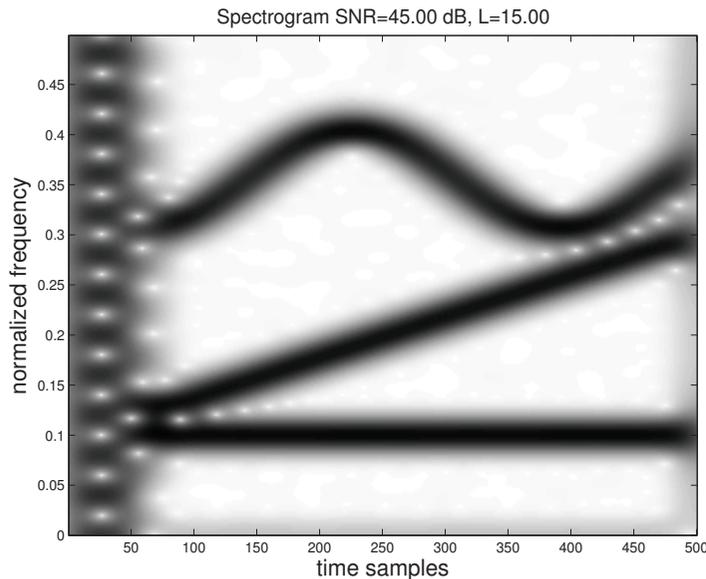
A variant of the reassignment method to compute sharpen and reversible TFRs by reassigned the transform instead of its energy to preserve the phase information.

Computation of the synchrosqueezed STFT:

$$S_x(t, \omega) = \frac{1}{h(0)} \int_{\mathbb{R}} F_x^h(t, \Omega) \delta(\omega - \hat{\omega}(t, \Omega)) \frac{d\Omega}{2\pi} \quad (7)$$

Its signal reconstruction formula:

$$\hat{x}(t) = \int_{\text{supp}_{\Omega}(x)} S_x(t, \Omega) d\Omega \quad (8)$$



## Mode extraction

### Principle

Apply a time-frequency mask over a previously computed reversible TFR.  
⇒ Performs better if the components are sparse and well disentangled.

[Brevdo, 2011] method finds the best frequency curve  $\Omega(t)$  to maximize the energy with a smooth constraint through a total variation penalization term:

$$\hat{\Omega} = \operatorname{argmax}_{\Omega} \int_{\mathbb{R}} |S_x(t, \Omega(t))|^2 dt - \lambda \int_{\mathbb{R}} \left| \frac{d\Omega}{dt}(t) \right|^2 dt, \quad (9)$$

where  $\lambda$  controls the importance of the smoothness of  $\Omega$ .

Notice: When the ridges of several components have to be estimated, this method can be iterated after subtracting the energy located at the previously estimated ridge.

## STFT properties

The STFT marginalization over time of  $F_x^h(t, \omega)$  leads to:

$$\int_{\mathbb{R}} F_x^h(t, \omega) dt = \iint_{\mathbb{R}^2} h(t - \tau)^* x(\tau) e^{-j\omega\tau} dt d\tau \quad (10)$$

$$= \iint_{\mathbb{R}^2} h(u)^* x(\tau) e^{-j\omega\tau} du d\tau \quad (11)$$

$$= \int_{\mathbb{R}} h(u)^* du \int_{\mathbb{R}} x(\tau) e^{-j\omega\tau} d\tau \quad (12)$$

$$= F_h(0)^* F_x(\omega) \quad (13)$$

with  $F_x(\omega) = \int_{\mathbb{R}} x(t) e^{-j\omega t} dt$  the Fourier transform of signal  $x$ .  
 Hence, the signal can be recovered by:

$$x(t) = \frac{1}{2\pi F_h(0)^*} \iint_{\mathbb{R}^2} F_x(\tau, \omega) e^{j\omega t} d\tau d\omega \quad (14)$$

## Time-reassigned synchrosqueezed STFT [He 2019]

### Principle

Synchrosqueezes the original transform along the time axis instead of the frequency axis exploiting the STFT properties when marginalized over time.

The time-reassigned synchrosqueezed STFT can be defined as:

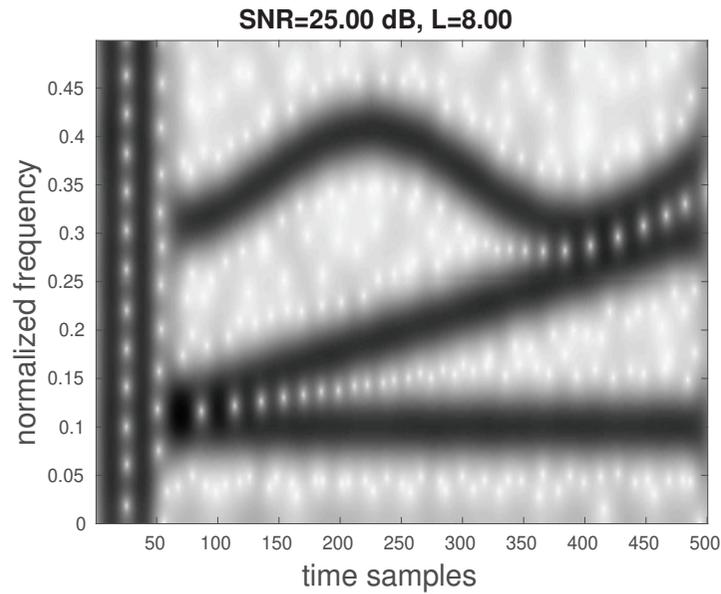
$$T_x^h(t, \omega) = \int_{\mathbb{R}} F_x^h(\tau, \omega) \delta(t - \hat{t}_x(\tau, \omega)) d\tau \quad (15)$$

where  $\hat{t}_x(t, \omega)$  corresponds to a group-delay estimator related to the time reassignment operator given by Eq. (3).

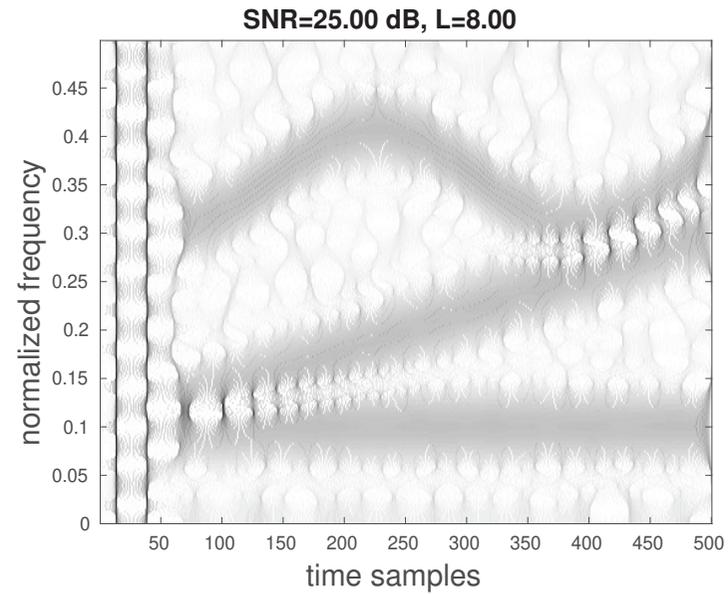
The original signal can thus be reconstructed using the following exact formula:

$$x(t) = \frac{1}{2\pi F_h(0)^*} \iint_{\mathbb{R}^2} T_x^h(\tau, \omega) e^{j\omega t} d\tau d\omega \quad (16)$$

## Numerical results



(e)  $|F_x^h(t, \omega)|^2$



(f)  $|T_x^h(t, \omega)|^2$

- Perfect localization of the 2 impulses
- Poor localization of the sinusoidal components

## Signal model

$$x(t) = e^{\lambda_x(t) + j\phi_x(t)} \quad (17)$$

$$\text{with } \lambda_x(t) = l_x + \mu_x t + \nu_x \frac{t^2}{2} \quad (18)$$

$$\text{and } \phi_x(t) = \varphi_x + \omega_x t + \alpha_x \frac{t^2}{2} \quad (19)$$

where  $\lambda_x(t)$  and  $\phi_x(t)$  respectively stand for the log-amplitude and phase and with  $q_x = \nu_x + j\alpha_x$  and  $p_x = \mu_x + j\omega_x$ .

For such a signal, it can be shown that [Fourer et al., 2017]:

$$\omega_x = \text{Im}(\tilde{\omega}_x(t, \omega) - q_x \tilde{t}_x(t, \omega)) = \hat{\omega}_x(t, \omega) - \text{Im}(q_x \tilde{t}_x(t, \omega)) \quad (20)$$

## Enhanced group-delay estimation

The new proposed second-order horizontal synchrosqueezing consists in moving  $F_x^h(t, \omega)$  from the point  $(t, \omega)$  to the point  $(t_x^{(2)}, \omega)$  located on the instantaneous frequency curve, *i.e.* such that  $\dot{\phi}(t_x^{(2)}) = \frac{d\phi_x}{dt}(t_x^{(2)}) = \omega_x + \alpha_x t_x^{(2)} = \omega$ .

This leads to:

$$t_x^{(2)} = \frac{\omega - \omega_x}{\alpha_x} = \hat{t}_x(t, \omega) + \frac{\omega - \hat{\omega}_x(t, \omega)}{\alpha_x} + \frac{\nu_x}{\alpha_x} \text{Im}(\tilde{t}_x(t, \omega)) \quad (21)$$

which can be estimated by:

$$\hat{t}_x^{(2)}(t, \omega) = \begin{cases} \frac{\omega - \hat{\omega}_x(t, \omega) + \text{Im}(\hat{q}_x(t, \omega) \tilde{t}_x(t, \omega))}{\hat{\alpha}_x(t, \omega)} & \text{if } \hat{\alpha}_x(t, \omega) \neq 0 \\ \hat{t}_x(t, \omega) & \text{otherwise} \end{cases} \quad (22)$$

where  $\hat{q}_x(t, \omega) = \hat{\nu}_x(t, \omega) + j\hat{\alpha}_x(t, \omega)$  is an unbiased estimator of  $q_x$ .

## Implementation considerations

In [Fourer et al.,2017] and [Fourer et al., 2018] we introduced two families of unbiased estimators called  $(tn)$  and  $(\omega n)$  involving  $n$ -order derivatives ( $n \geq 2$ ) with respect to time (resp. to frequency) which enable to compute Eqs. (22).

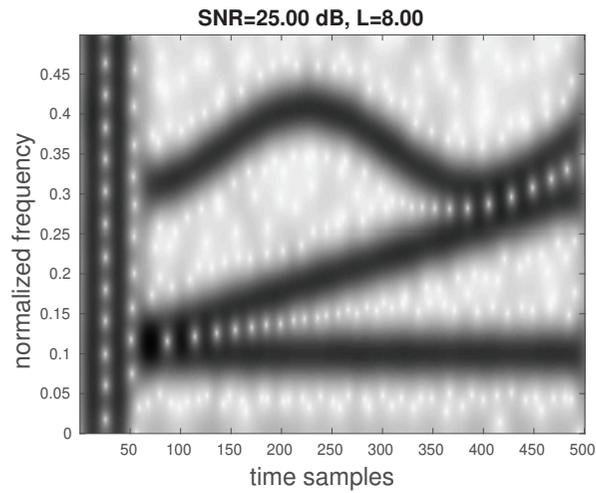
$$\hat{q}_x^{(tn)}(t, \omega) = \frac{F_x^{\mathcal{D}^n} h F_x^h - F_x^{\mathcal{D}^{n-1}} h F_x^{\mathcal{D}h}}{F_x^{\mathcal{T}h} F_x^{\mathcal{D}^{n-1}} h - F_x^{\mathcal{T} \mathcal{D}^{n-1}} h F_x^h} \quad (23)$$

$$\hat{q}_x^{(\omega n)}(t, \omega) = \frac{(F_x^{\mathcal{T}^{n-1}} \mathcal{D}h + (n-1) F_x^{\mathcal{T}^{n-2}} h) F_x^h - F_x^{\mathcal{T}^{n-1}} h F_x^{\mathcal{D}h}}{F_x^{\mathcal{T}^{n-1}} h F_x^{\mathcal{T}h} - F_x^{\mathcal{T}^n} h F_x^h} \quad (24)$$

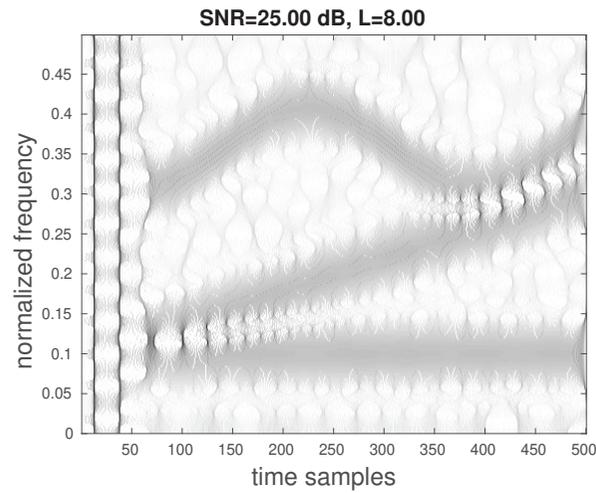
with  $\mathcal{D}^n h(t) = \frac{d^n h}{dt^n}(t)$  and  $\mathcal{T}^n h(t) = t^n h(t)$ .

- discrete-time reformulations of our previously described expressions combined with the rectangle approximation method. Thus  $F_x^h[k, m] \approx F_x^h\left(\frac{k}{F_s}, 2\pi \frac{m F_s}{M}\right)$ , where  $F_s$  denotes the sampling frequency,  $k \in \mathbb{Z}$  is the time sample index and  $m \in \mathcal{M}$  is the discrete frequency bin.
- The number of frequency bins  $M$  is chosen as an even number such as  $\mathcal{M} = [-M/2 + 1; M/2]$
- Our implementation uses a Gaussian window expressed as  $h(t) = \frac{1}{\sqrt{2\pi T}} e^{-\frac{t^2}{2T^2}}$  where  $T$  is the time-spread of the window which can be related to  $L = TF_s$ .

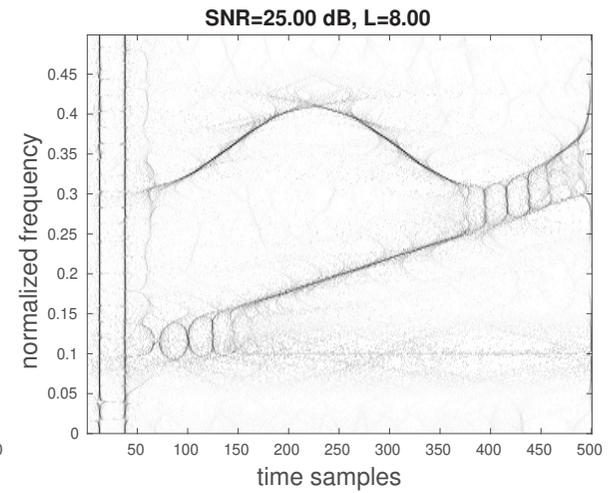
# Comparative numerical results 1/2



(g) spectrogram

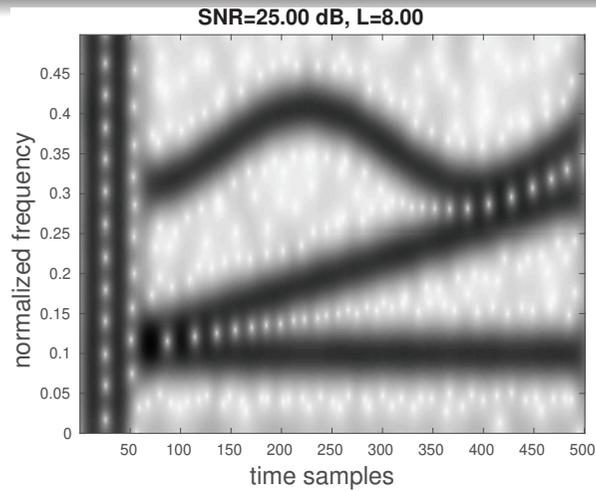


(h) time-reassigned chrosqueezing

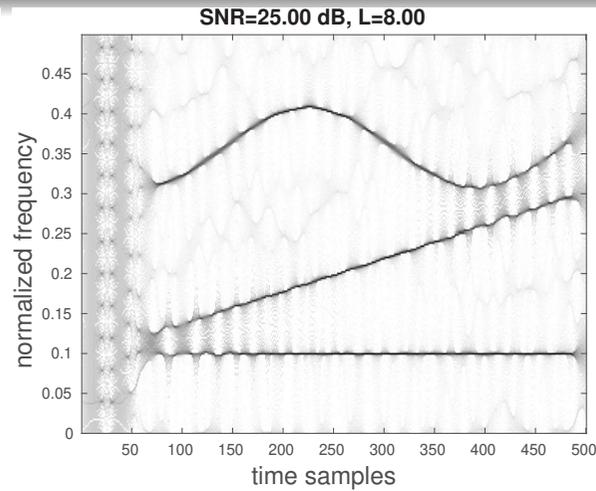


(i) second-order horizontal synchrosqueezing

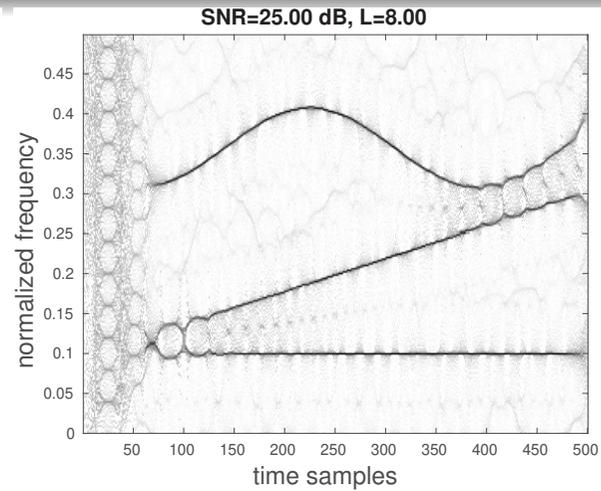
# Comparative numerical results 2/2



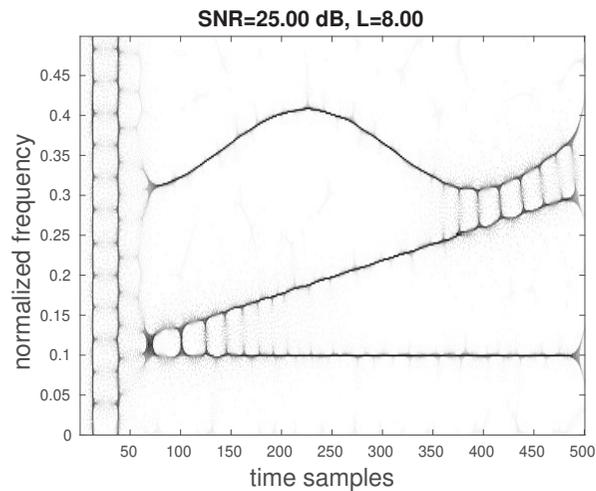
(j) spectrogram



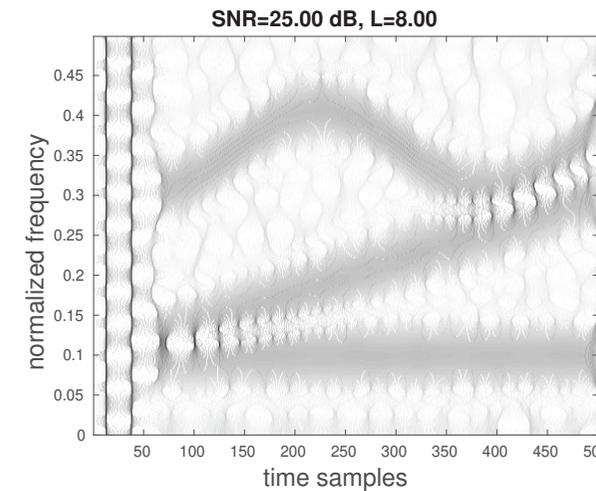
(k) synchrosqueezing



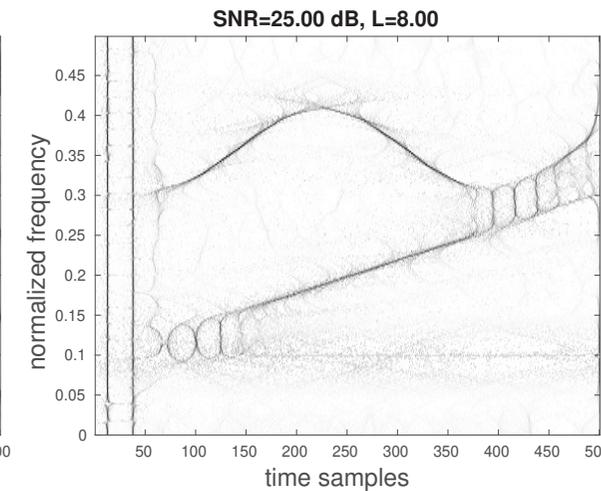
(l) second-order vertical synchrosqueezing



(m) reassignment



(n) time-reassigned synchrosqueezing



(o) second-order horizontal synchrosqueezing

## Signal reconstruction

Whole signal reconstruction expressed in terms of Reconstruction quality Factor (RQF):

$$\text{RQF} = 10 \log_{10} \left( \frac{\sum_n |x[n]|^2}{\sum_n |x[n] - \hat{x}[n]|^2} \right) \quad (25)$$

Method	RQF (dB)
STFT	269.27
reassignment	N/A
classical synchrosqueezing	35.89
second-order vertical synchrosqueezing	23.80
time-reassigned synchrosqueezing	116.67
second-order time-reassigned synchrosqueezing	116.67

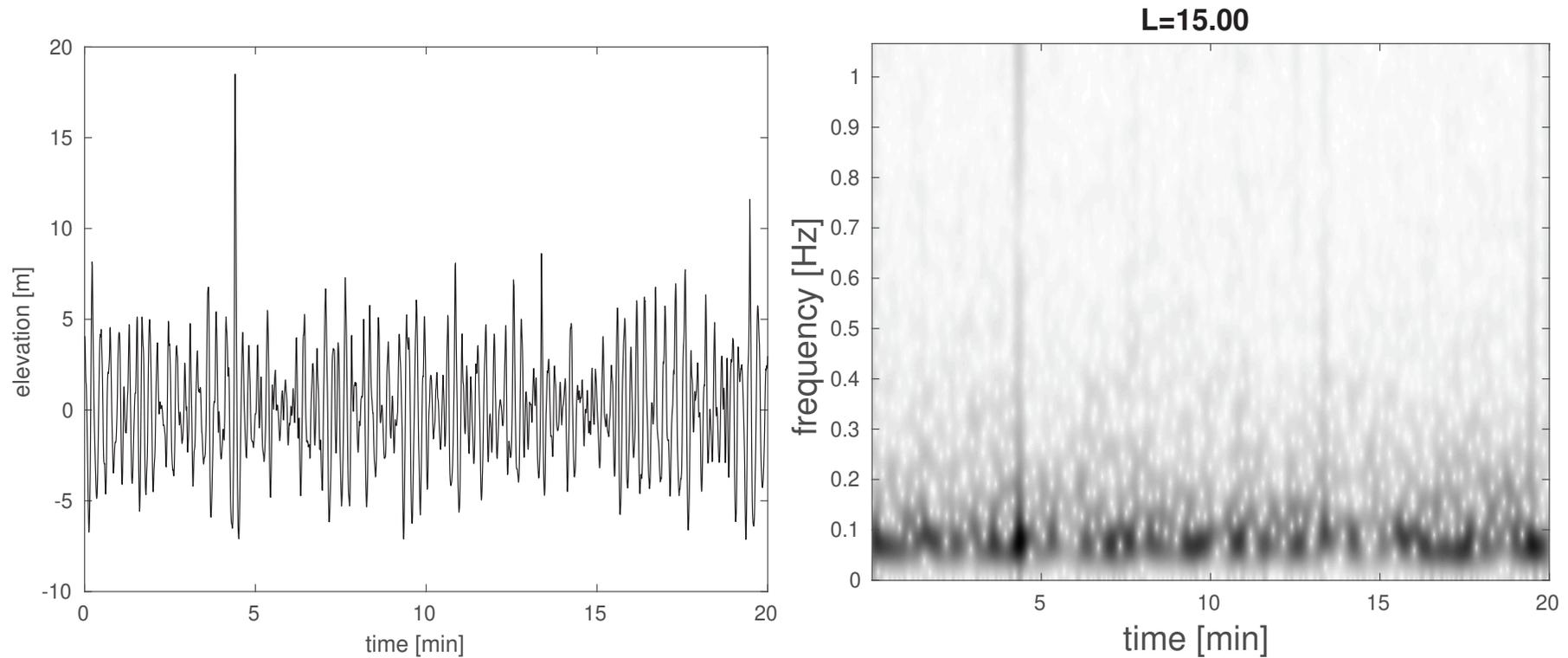
## Draupner wave recording

We consider a possible freak wave event measured in the North Sea on the Draupner Platform the 1st of January 1995.



## Draupner wave signal [Haver, 2004]

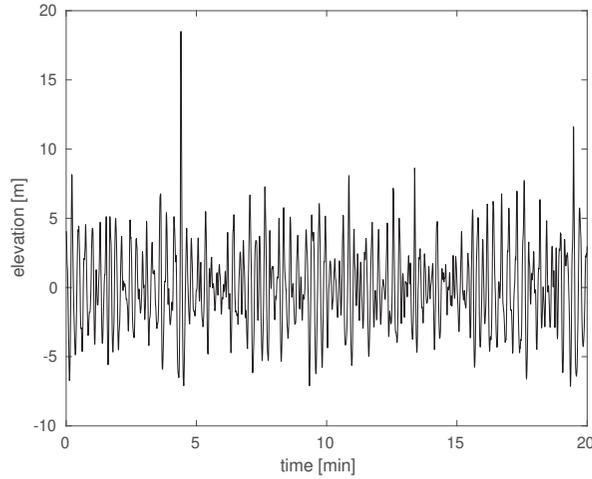
- The signal corresponds to the sea surface elevation deduced from the measures provided by a wave sensors consisting of a down-looking laser.
- The sampling frequency of this signal is  $F_s = 2.13$  Hz and the duration is 20 minutes.



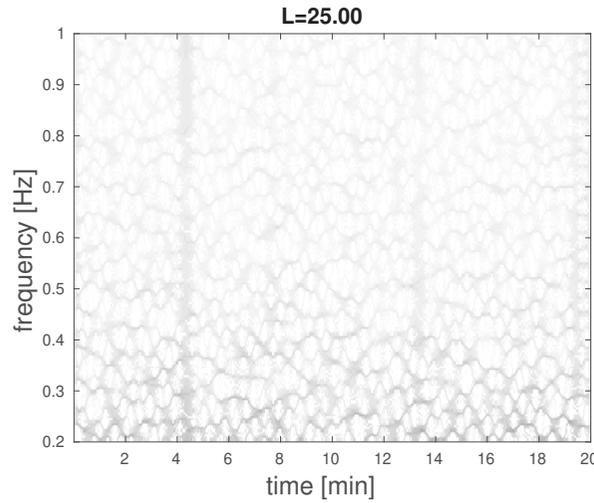
(p)  $x(t)$

(q)  $|F_x(t, \omega)|^2$

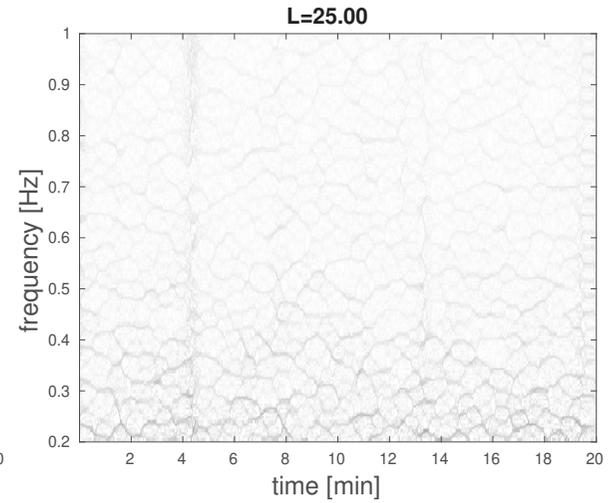
# Time-frequency representations



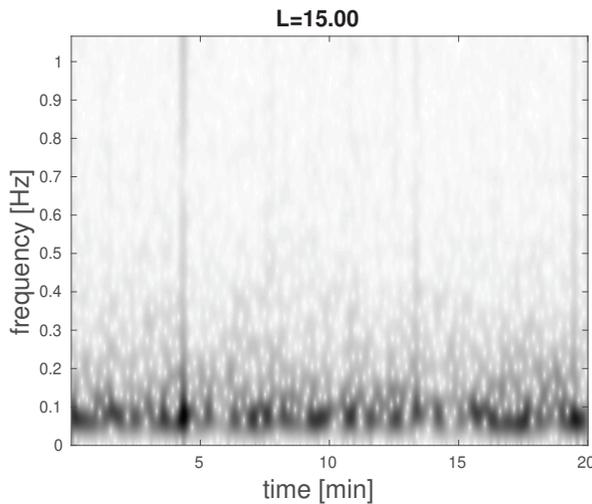
(r) signal



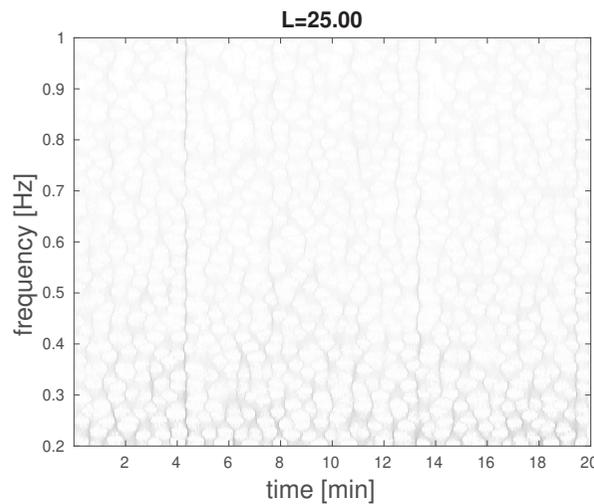
(s) synchrosqueezing



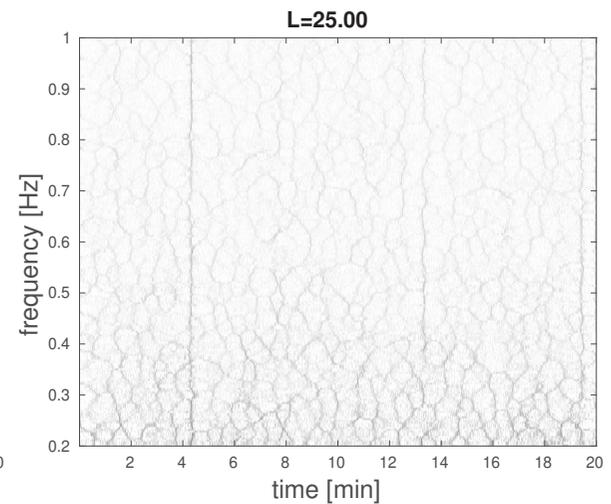
(t) second-order vertical synchrosqueezing



(u) spectrogram



(v) time-reassigned synchrosqueezing



(w) second-order horizontal synchrosqueezing

## Impulses detection

### Proposed Saliency function

Defined as the root mean square of the marginal over frequency band  $\Omega = [0.4; 1]$  Hz of the signal energy contained in the considered time-frequency representation.

$$G(t) = \left( \int_{\Omega} |TFR_x^h(t, \omega)|^2 d\omega \right)^{\frac{1}{2}}. \quad (26)$$

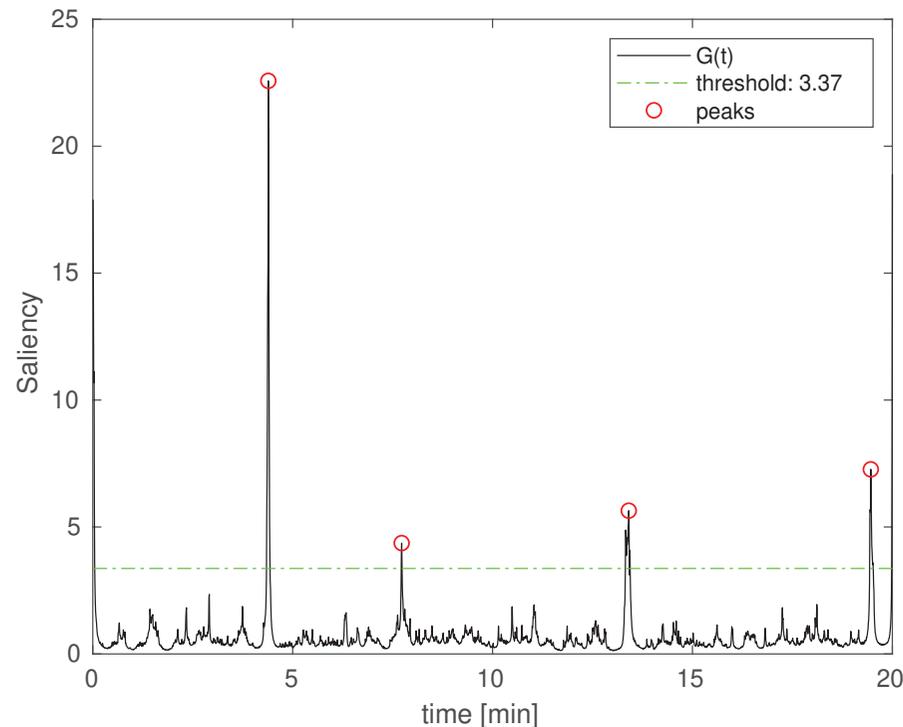


Figure : Result provided using the second-order horizontal synchrosqueezing.

## Impulses disentangling

A binary masked version of the transform  $S_x^h(t, \omega)$  can thus be computed using  $G(t)$  as:

$$\hat{S}(t, \omega) = \begin{cases} S_x^h(t, \omega) & \text{if } G(t) > \Gamma \\ 0 & \text{otherwise} \end{cases} . \quad (27)$$

where  $\Gamma$  is a defined threshold.

Our numerical computation uses  $\Gamma = 3.37$  which corresponds to 5 times the mean value of  $G(t)$ .

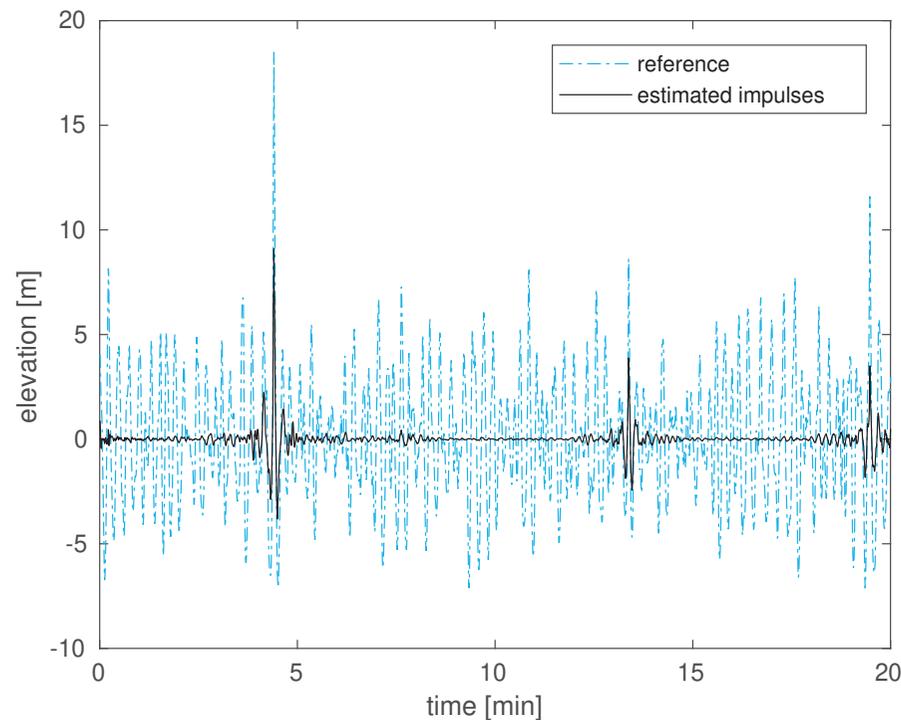


Figure : Draupner signal and recovered waveform of its impulsive components.

## Contributions summary

- A novel horizontal synchrosqueezing method based on an enhanced group-delay estimator.
- Efficient computation from the STFT using specific analysis windows.
- New applicative results based on the Draupner wave signal.

Matlab code freely available on IEEE Code Ocean at: <http://fourer.fr/hsst>