

## Contributions in the ASCETE project

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# Plan

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- 2 Objective 1 : New approaches for the study of MCSs
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## ANR ASCETE project



Goals : Combining deterministic and stochastic approaches to extend the proposed techniques to more complex signals

- Deterministic models combined with machine learning (e.g. deep neural networks)
- Difficult cases for mode recovery (e.g. overlapping components, noisy signals, etc.)
- Combining synchrosqueezing with Non-negative Matrix Factorization (NMF)
- Generalization to high dimension signals (images, tensors, etc.)
- New practical applications (perception, biomedicine, astronomy, etc.)

Project objectives :

- Objective 1 : New approaches for the study of MCSs with synchrosqueezing transforms
- Objective 2 : Improving signal representations using data-driven and machine learning approaches
- Objective 3 : Combining non negative matrix factorization and SST, Phase retrieval
- Objective 4 : Applications and Software developments

# Plan

- 1 Introduction
- 2 Objective 1 : New approaches for the study of MCSs
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## Second-order time-reassigned synchrosqueezing

- Partners : Nantes, Paris
- Involved Tasks : T1.1 : combining stochastic and deterministic approaches to improve SST

### Contributions

- A new second-order group-delay estimator for horizontal synchrosqueezing [Fourer, Auger 2019],
- Application to the S-transform and continuous wavelet transform [Fourer, Auger 2020]
- First analysis of the Draupner Wave signal analysis

### Motivation

- Computing enhanced TFR designed for impulsive signal and strongly modulated modes
- Generalization of the time-reassigned synchrosqueezed STFT first proposed by [He et al, 2019]

## Main Idea

### Definitions

STFT :

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

with  $j^2 = -1$

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

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

$$= \iint_{\mathbb{R}^2} h(u)^* x(\tau) e^{-j\omega\tau} du d\tau \quad (3)$$

$$= \int_{\mathbb{R}} h(u)^* du \int_{\mathbb{R}} x(\tau) e^{-j\omega\tau} d\tau \quad (4)$$

$$= F_h(0)^* F_x(\omega) \quad (5)$$

with  $F_x(\omega) = \int_{\mathbb{R}} x(t) e^{-j\omega t} dt$  the Fourier transform of signal  $x$ .

## Time-reassigned synchrosqueezing

### Horizontal synchrosqueezing

$$S_{\mathbf{x}}^h(t, \omega) = \int_{\mathbb{R}} F_{\mathbf{x}}^h(\tau, \omega) \delta(t - \hat{t}_{\mathbf{x}}^{(2)}(\tau, \omega)) d\tau \quad (6)$$

where  $\hat{t}_{\mathbf{x}}(t, \omega)$  corresponds to the time reassignment operator (group-delay).

Reconstruction :

$$x(t) = \frac{1}{2\pi F_h(0)^*} \iint_{\mathbb{R}^2} S_{\mathbf{x}}^h(\tau, \omega) e^{j\omega\tau} d\tau d\omega. \quad (7)$$

### Second-order group-delay estimator

$$\hat{t}_{\mathbf{x}}^{(2)}(\mathbf{t}, \omega) = \begin{cases} \frac{\omega - \hat{\omega}_{\mathbf{x}}(\mathbf{t}, \omega) + \text{Im}(\hat{q}_{\mathbf{x}}(\mathbf{t}, \omega) \tilde{x}_{\mathbf{x}}(\mathbf{t}, \omega))}{\hat{\alpha}_{\mathbf{x}}(\mathbf{t}, \omega)} & \text{if } \hat{\alpha}_{\mathbf{x}}(\mathbf{t}, \omega) \neq 0 \\ \hat{t}_{\mathbf{x}}(\mathbf{t}, \omega) & \text{otherwise} \end{cases} \quad (8)$$

with :

$$\hat{t}_{\mathbf{x}}(\mathbf{t}, \omega) = \text{Re}(\tilde{t}_{\mathbf{x}}(\mathbf{t}, \omega)), \text{ with } \tilde{t}_{\mathbf{x}}(\mathbf{t}, \omega) = t - \frac{F_{\mathbf{x}}^T h(\mathbf{t}, \omega)}{F_{\mathbf{x}}^h(\mathbf{t}, \omega)} \quad (9)$$

$$\hat{\omega}_{\mathbf{x}}(\mathbf{t}, \omega) = \text{Im}(\tilde{\omega}_{\mathbf{x}}(\mathbf{t}, \omega)), \text{ with } \tilde{\omega}_{\mathbf{x}}(\mathbf{t}, \omega) = j\omega + \frac{F_{\mathbf{x}}^D h(\mathbf{t}, \omega)}{F_{\mathbf{x}}^h(\mathbf{t}, \omega)} \quad (10)$$

$$\hat{\alpha}_{\mathbf{x}}(\mathbf{t}, \omega) = \text{Im}(\hat{q}_{\mathbf{x}}), \text{ with } \hat{q}_{\mathbf{x}}(\mathbf{t}, \omega) = \frac{F_{\mathbf{x}}^D h F_{\mathbf{x}}^h - F_{\mathbf{x}}^D h F_{\mathbf{x}}^D h}{F_{\mathbf{x}}^T h F_{\mathbf{x}}^D h - F_{\mathbf{x}}^T h F_{\mathbf{x}}^D h} \quad (11)$$

## Numerical results (comparison)

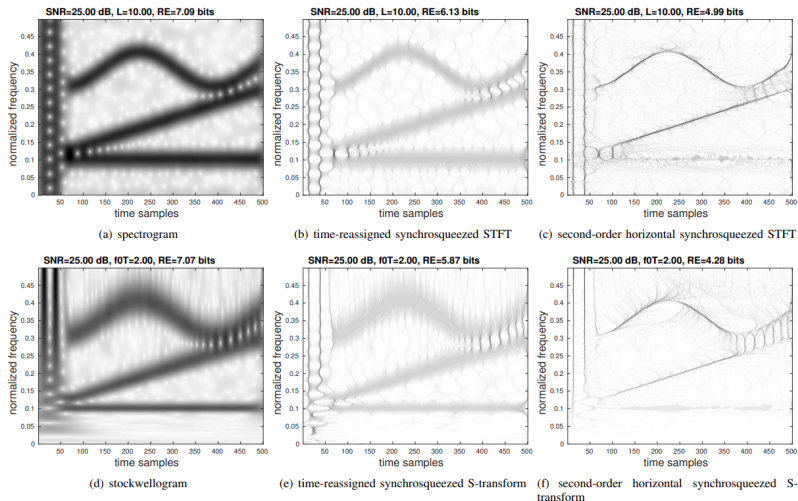


Fig. 1. Comparison of the resulting TFRs with Rényi Entropy (at order  $\alpha = 3$ ) of a synthetic multicomponent signal. The TFRs of the synchrosqueezing methods (b),(c), (e) and (f) correspond to their squared modulus.



## Numerical results on Draupner signal and codes

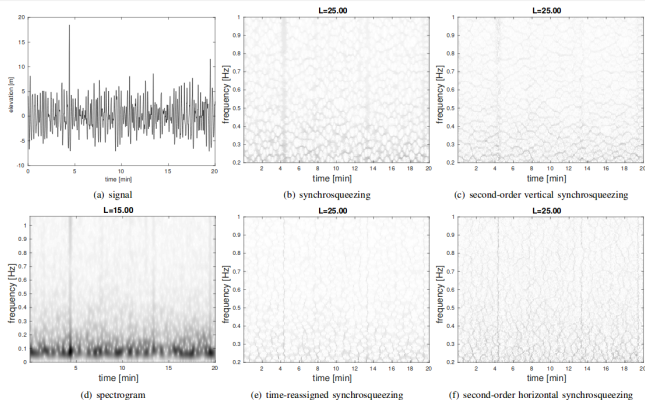


Fig. 2. Waveform (a) and TFRs of the Draupner wave signal. spectrogram (d), synchrosqueezing (b), second-order vertical synchrosqueezing (c), time-reassigned synchrosqueezing (e) and second-order horizontal synchrosqueezing (f).

### Matlab Codes

<https://fourer.fr/hsst>

<https://fourer.fr/sthsst>

## Possible extensions

- Robust to noise estimators using regularization
- High-order group-delay estimators (Pham, Meignen et al.)
- Theoretical analysis of time-reassigned synchrosqueezing wavelet transform [Li, Zhang, Auger et al. 2022]
- Self-matched extracting wavelet transform and signal reconstruction [Li, Auger, et al. 2022]
- etc.

## Pseudo-Bayesian approach for mode extraction

- Partners : Paris
- Involved Tasks : T1.1 + T1.2 : New ridge extraction technique

### Contributions

- A new noise-robust mode extraction method based on the STFT [Legros, Fourer, 2021],
- Extension using alpha-beta divergence and a new signal detector (submitted [Legros, Fourer, 2022])
- Application on both synthesized and real-world signals

### Motivation

- Robust mode extraction method usable to any time-frequency representation
- Robust Instantaneous Frequency estimator and Ridge detection method

## Main Idea

- Given a discrete-time spectrogram  $s_{n,m}$  ( $n$  the time index and  $m$  the frequency bin), model each spectrogram slice as :

$$s_{n,m} | \bar{m}_n \sim g(m - \bar{m}_n), \quad \text{with} \quad g(m) = \frac{2\sqrt{\pi}L}{M} e^{-\left(\frac{2\pi mL}{M}\right)^2} \quad (12)$$

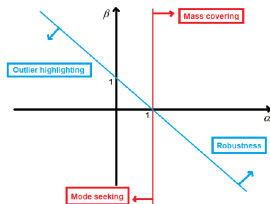
- Compute the joint likelihood (assuming independence between successive frames)

$$p(\mathbf{s}_n | \bar{\mathbf{m}}_n) = \prod_{m=0}^{M-1} p(s_{n,m} | \bar{m}_n). \quad (13)$$

- Instead of maximizing posterior given by  $p(\bar{\mathbf{m}}_n | \mathbf{s}_n) = \frac{p(\mathbf{s}_n | \bar{\mathbf{m}}_n)p(\bar{\mathbf{m}}_n)}{p(\mathbf{s}_n)}$  which is equivalent to minimize the KL divergence between  $p(\mathbf{s}_n)$  and  $p(\mathbf{s}_n | \bar{\mathbf{m}}_n)$ , we replace the KL-divergence by a robust  $\alpha - \beta$ -divergence leading to the following pseudo-posterior :
- Use the pseudo-posterior expressed in terms of alpha-beta cross-entropy :

$$p(\bar{\mathbf{m}}_n | \mathbf{s}_n) \propto e^{-M \text{CE}_{\mathbf{AB}}^{\alpha, \beta}(\bar{\mathbf{m}}_n)} p(\bar{\mathbf{m}}_n). \quad (14)$$

# Main Idea




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## ALGORITHM 1

- 1: Input: TFR  $S_0$ , GRW mean  $m_0$  and variance  $\sigma_0$ ,  $K$ ,  $g$ .
  - 2: **for**  $k = 1, \dots, K$  **do**
  - 3:   **for**  $n = 0, \dots, N - 1$  **do**
  - 4:     Compute  $p(m_n)$  by matching moments from Eq. (18).
  - 5:     Compute the pseudo-posterior  $p(m_n | s_n)$  from Eq. (16).
  - 6:     Perform MMSE estimation of  $\hat{m}_n$ .
  - 7:   **end for**
  - 8:   Repeat steps 4 to 6 iterating from  $n = N - 1, \dots, 0$
  - 9:   Update the TFR by subtracting the  $k$ th ridge (TFR support set to 0).
  - 10: **end for**
- 
-

## Numerical results

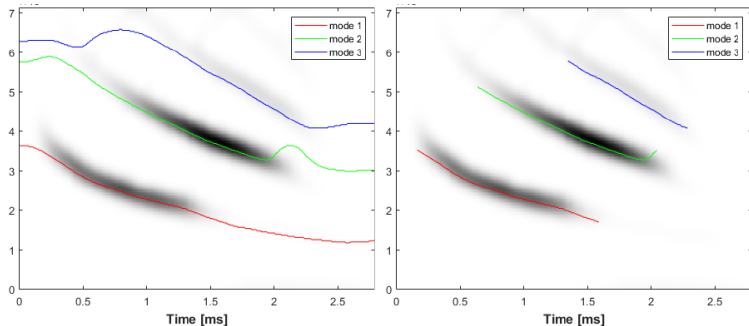


Fig. 13. Estimation of  $K = 3$  signal components of the real bat record signal using the proposed ABD method with  $\alpha = 0.4, \beta = 0.7$  with (right) and without (left) performing detection.

## Numerical results and codes

Table 1. RQF of each reconstructed components (averaged over 100 realizations of noise) for the different competing approaches for a SNR = 10dB. In second rows are displayed, for each case, the std of the estimators.

|                                  | C1                    | C2                    | C3                    | Average               |
|----------------------------------|-----------------------|-----------------------|-----------------------|-----------------------|
| Brevdo [5]                       | 15.86<br>±0.84        | 16.60<br>±0.84        | 6.52<br>±2.12         | 12.70<br>±1.4         |
| ABD, $\alpha = 0.4, \beta = 0.4$ | <b>17.12</b><br>±1.81 | 16.63<br>±0.81        | 11.04<br>±0.81        | <b>14.31</b><br>±0.63 |
| ABD, $\alpha = 0.2, \beta = 0.4$ | 17.07<br>±0.79        | 16.64<br>±0.78        | 10.89<br>±0.62        | 14.26<br>±0.74        |
| ABD, $\alpha = 0.4, \beta = 0.2$ | 16.93<br>±0.89        | <b>16.52</b><br>±0.76 | <b>10.90</b><br>±0.57 | 14.28<br>±0.75        |
| ABD, $\alpha = 0.2, \beta = 1.2$ | 14.96<br>±6.74        | 16.78<br>±0.80        | 9.33<br>±0.64         | 13.35<br>±3.93        |
| ABD, $\alpha = 0.7, \beta = 1.2$ | 11.62<br>±4.84        | 16.37<br>±0.72        | 9.51<br>±0.33         | 12.20<br>±8.03        |
| ABD, $\alpha = 0.2, \beta = 1.5$ | 5.52<br>±8.90         | 16.28<br>±2.43        | 8.64<br>±0.24         | 9.85<br>±2.83         |
| RD [24]                          | 16.22<br>±7.26        | 12.28<br>±7.14        | 5.05<br>±7.92         | 11.18<br>±7.45        |

### Matlab Codes

<https://fourer.fr/eusipco21>

<https://codeocean.com/capsule/8693890/tree/v1>

## EM approach for mode extraction and IF estimation

- Partners : Paris, Grenoble
- Involved Tasks : T1.1 + T1.2 : New ridge extraction technique

### Contributions

- A new noise-robust mode extraction method based on the STFT [Legros, Fourer, 2021]
- Application on synthesized signals

### Motivation

- Noise-robust mode extraction method
- Deal with overlapping components



## Main Idea

- Given a spectrogram  $s_{n,m}$ , assume a mixture model :

$$p(s_{n,m} | \mathbf{w}_n, \hat{\mathbf{m}}_n) = \sum_{k=1}^K w_n^k g(m - \hat{m}_n^k) + \frac{1}{M} \left( 1 - \sum_{k=1}^K w_n^k \right), \quad (15)$$

- Compute the joint-likelihood :

$$p(\mathbf{S} | \mathbf{W}, \hat{\mathcal{M}}) = \prod_n \prod_m p(s_{n,m} | \mathbf{w}_n, \hat{\mathbf{m}}_n). \quad (16)$$

- Consider two distinct prior :

- Total Variation :

$$p(\hat{\mathcal{M}} | \epsilon) \propto \exp \left[ -\epsilon \sum_{k=1}^K \|\hat{\mathbf{m}}_{k,:}^\top\|_{TV} \right], \quad (17)$$

- Laplacian :

$$p(\hat{\mathcal{M}} | \lambda) \propto \exp \left[ -\frac{\lambda}{2} \sum_{k=1}^K \|\mathbf{L}_a \hat{\mathbf{m}}_{k,:}^\top\|_2^2 \right], \quad (18)$$

## Main Idea

- Approximate joint posterior as :

$$p(\mathbf{W}, \hat{\mathcal{M}} | \mathbf{S}) \propto p(\mathbf{S} | \mathbf{W}, \hat{\mathcal{M}}) p(\hat{\mathcal{M}}) p(\mathbf{W}). \quad (19)$$

- (modified) Expectation-Maximization at iteration (i) :

$$\begin{aligned} \hat{Q}(\mathbf{W} | \mathbf{W}^{(i)}) &= \log p(\mathbf{S} | \mathbf{W}, \tilde{\mathcal{M}}) + \log [p(\hat{\mathcal{M}}) p(\mathbf{W})], \\ \mathbf{W}^{(i+1)} &= \underset{\mathbf{W}}{\operatorname{argmax}} \hat{Q}(\mathbf{W} | \mathbf{W}^{(i)}). \end{aligned} \quad (20)$$

## Numerical results

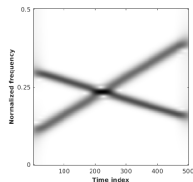
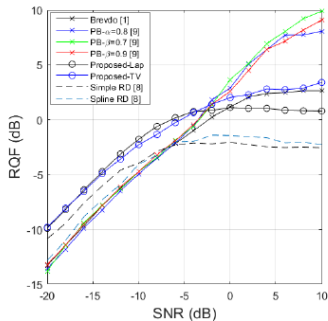
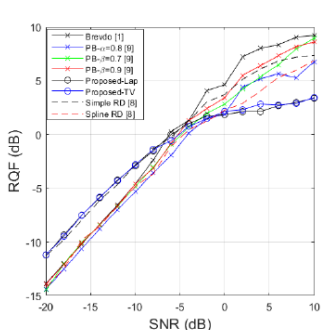


Figure 1: Spectrogram of the analyzed multicomponent signals with overlapping components.



## Finite Rate of Innovation (FRI) approach for mode extraction

- Partners : Paris
- Involved Tasks : T1.2 : New ridge extraction technique

### Contributions

- FRI-based approach for mode extraction and IF estimation (submitted to SPL [Legros, Fourer, 2022])
- Possibly combined with synchrosqueezing
- Application on both synthesized and real-world signals

### Motivation

- Robust IF estimation and Ridge detection method

## Main Idea

- Observation model :

$$s_{n,m} \approx \sum_{k=0}^{K-1} a_k(n) g(m - \phi'_k(n)) \quad (21)$$

where  $a_k(n) = \alpha_k^2(n)$  and  $g(m) = e^{-\left(\frac{2\pi m L}{M}\right)^2}$

- Assumes the signal to retrieve as a stream of Dirac Pulses :

$$f_n(m) = \sum_{k=0}^{K-1} a_k(n) \delta(m - \phi'_k(n)) \quad . \quad (22)$$

- Estimation of  $\hat{f}_n = \mathbf{D}_g^{-1} \mathbf{V}^{-1} \mathbf{s}_n$  where  $[\mathbf{V}]_{m,\lambda} = e^{j2\pi\left(\frac{m\lambda}{M}\right)}$  is a  $(M \times 2M_0 + 1)$  matrix and  $\mathbf{D}_g$  is a diagonal matrix gathering the discrete time Fourier series coefficients of  $g$ .

- Estimation of the mode location by estimating the annihilating filter  $h$  such as :

$$\begin{aligned}
 (\hat{f}_n * \mathbf{h})(l) &= \sum_{i \in \mathbb{Z}} h(i) \hat{f}_n(l - i) = 0 \\
 &= \sum_{k=0}^{K-1} a_k(n) e^{-\frac{j2\pi l \phi'_k(n)}{M}} \underbrace{\sum_{i \in \mathbb{Z}} h(i) e^{\frac{j2\pi i \phi'_k(n)}{M}}}_{H\left(e^{-\frac{j2\pi \phi'_k(n)}{M}}\right)} = 0 \quad (23)
 \end{aligned}$$

with  $H(z)$  the Z-transform of  $\mathbf{h}$ , whose roots are  $e^{-\frac{j2\pi \phi'_k(n)}{M}}$ .

- In presence of noise,  $h$  minimizes  $\|\hat{f}_n * \mathbf{h}\|^2$  (Total Least Squares method)
- The FRI-TLS method can be combined with vertical synchrosqueezing (FRI-SST)

## Numerical results

(b) SNR=0dB

|                        | C1                        | C2                        | C3                        | Average                   |
|------------------------|---------------------------|---------------------------|---------------------------|---------------------------|
| Brevido [10]           | 4.06<br>$\pm 0.82$        | 4.34<br>$\pm 0.94$        | -0.22<br>$\pm 0.51$       | 2.71<br>$\pm 0.76$        |
| PB, $\beta = 0.7$ [9]  | 3.45<br>$\pm 5.54$        | 3.60<br>$\pm 3.77$        | 1.17<br>$\pm 1.84$        | 2.74<br>$\pm 3.72$        |
| PB, $\alpha = 0.5$ [9] | 3.66<br>$\pm 3.78$        | 4.18<br>$\pm 1.36$        | 0.59<br>$\pm 1.87$        | 2.81<br>$\pm 2.33$        |
| RD [8]                 | 9.00<br>$\pm 1.72$        | 8.21<br>$\pm 1.04$        | 6.56<br>$\pm 1.41$        | 7.92<br>$\pm 1.39$        |
| FRI                    | 3.71<br>$\pm 1.47$        | 1.43<br>$\pm 1.09$        | 0.13<br>$\pm 0.76$        | 1.76<br>$\pm 1.11$        |
| FRI TLS (proposed)     | 8.81<br>$\pm 1.08$        | <b>8.28</b><br>$\pm 2.37$ | <b>7.37</b><br>$\pm 0.98$ | <b>8.15</b><br>$\pm 1.48$ |
| FRI SST (proposed)     | <b>9.16</b><br>$\pm 2.65$ | 7.53<br>$\pm 2.09$        | 6.03<br>$\pm 0.87$        | 7.57<br>$\pm 1.87$        |

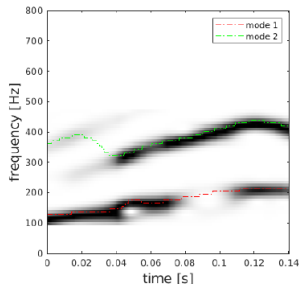


Fig. 2. Estimation of the first  $K = 2$  signal components of the speech signal using the proposed TLS method.

## Malab Codes

<https://codeocean.com/capsule/7022037/tree/v1>

## Harmonic/Percussive Source Separation

- Partners : Paris
- Involved Tasks : T1.2 : New ridge extraction technique, T4.2 : application of SST to audio signals

### Contributions

- Computes AM-FM estimator in the TF plane for separating harmonic / percussive sources [Fourer, 22]
- Combines Machine-learning with time-frequency analysis
- Application to music audio signals

### Motivation

- Disentangling the harmonic deterministic part from the percussive and noisy part
- Enhancing audio signal for MIR applications



## Main Idea

- Assumes an instantaneous mixture

$$x(t) = s_h(t) + s_p(t) \quad (24)$$

with

$$x(t) = e^{\lambda_x(t) + j\phi_x(t)}, \quad \text{with } j^2 = -1, \quad (25)$$

- Uses previously proposed estimators in [Fourer, Auger et al. 2018]

$$\hat{q}_x^{(tn)}(t, \omega) = \frac{F_x^{\mathcal{D}^n} h F_x^h - F_x^{\mathcal{D}^{n-1}} h F_x^{\mathcal{D}^n} h}{F_x^{\mathcal{T}^n} h F_x^{\mathcal{D}^{n-1}} h - F_x^{\mathcal{T}^n} h F_x^{\mathcal{D}^n} h} \quad (26)$$

$$\hat{\psi}_x^{(tn)}(t, \omega) = \frac{F_x^{\mathcal{D}^n} h F_x^{\mathcal{T}^n} h - F_x^{\mathcal{T}^n} h F_x^{\mathcal{D}^n} h}{F_x^{\mathcal{T}^n} h F_x^{\mathcal{D}^{n-1}} h - F_x^{\mathcal{T}^n} h F_x^{\mathcal{D}^n} h} + j\omega \quad (27)$$

- 

$$\hat{\nu}_x(t, \omega) = \text{Re}(\hat{q}_x(t, \omega)), \quad \hat{\alpha}_x(t, \omega) = \text{Im}(\hat{q}_x(t, \omega)) \quad (28)$$

$$\hat{\lambda}_x(t, \omega) = \text{Re}(\hat{\psi}_x(t, \omega)), \quad \hat{\phi}_x(t, \omega) = \text{Im}(\hat{\psi}_x(t, \omega)) \quad (29)$$

- AM-FM Parameters (after discretization) :

- AM :  $\hat{\lambda}_x[k, m]$

- FM :  $\hat{\alpha}_x[k, m]$

- AM-FM :  $C_x[k, m] = \sqrt{\hat{\lambda}_x[k, m]^2 + \hat{\alpha}_x[k, m]^2}$

## Main Idea

### Training based on Linear Discriminant Analysis

- Reference Separation Mask using ground truth :

$$M_h^{(true)}[k, m] = \begin{cases} 1 & \text{if } |F_{s_h}^h[k, m]|^2 > |F_{s_p}^h[k, m]|^2 \\ 0 & \text{otherwise} \end{cases}, \quad (30)$$

$$M_p^{(true)}[k, m] = 1 - M_h^{(true)}[k, m] \quad (31)$$

- Compute the centroid of each source (i.e.  $\mu_h$  ou  $\mu_p$ ) in the discriminant space.

⇒ The trained model corresponds to eigenvectors and  $\mu_h$  et  $\mu_p$ .

### Separation

- For each TF point, Compute the descriptors  $\mathcal{Q}_x[k, m]$
- Compute the linear projection  $P_{\mathcal{Q}}$
- Separation masks :

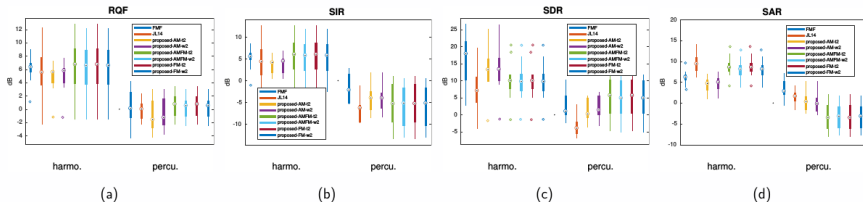
$$M_h[k, m] = \begin{cases} 1 & \text{if } \|P_{\mathcal{Q}}[k, m] - \mu_h\| < \|P_{\mathcal{Q}}[k, m] - \mu_p\| \\ 0 & \text{otherwise} \end{cases}, \quad M_p[k, m] = 1 - M_h[k, m]. \quad (32)$$

- Reconstruction :

$$\hat{s}_h = \text{TFCT}^{-1}(F_x^h[k, m]M_h[k, m]) \quad (33)$$

$$\hat{s}_p = \text{TFCT}^{-1}(F_x^h[k, m]M_p[k, m]) \quad (34)$$

## Numerical results



## Audio results

<https://fourer.fr/publi/gretsi22/>

## Biblio : Objective 1 : New approaches for the study of MCSs

- Fourer, D., & Auger, F. (2019, September). Second-order time-reassigned synchrosqueezing transform : Application to Draupner wave analysis. In 2019 27th European Signal Processing Conference (EUSIPCO) (pp. 1-5). IEEE.
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To appear :

- Legros, Q., & Fourer, Time-Frequency Ridge Estimation of Multi-Component Signals using Sparse Modeling of Signal Innovation.
- Legros, Q., & Fourer, Pseudo-Bayesian Approach for Robust Mode Detection and Extraction based on the STFT

# Plan

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  - Contribution 1 : EEG Signal Analysis
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## HVS detection from EEG signals using recursive synchrosqueezing transform

- Partners : Paris, Tsing Hua University (Taiwan)
- Involved Tasks : T1.3 and T4.1 : multivariate SST and application to the study of ECG and EEG signals

### Contributions

- A new HVS method designed to ECG signals
- Combines recursive synchrosqueezing with a detector
- Application on real-world signals

### Motivation

- Fast Prediction of Parkinson high voltage spindles (HVS)

## Numerical results

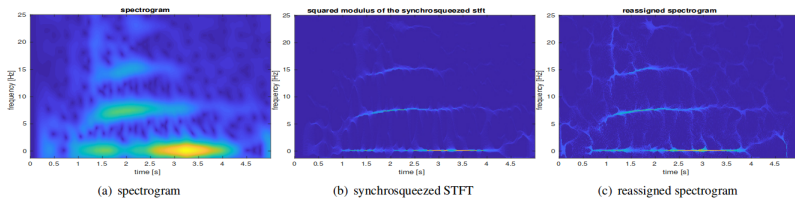


FIGURE 1 – Comparisons of the different recursive TFRs computed for an EEG signal with a HVS.

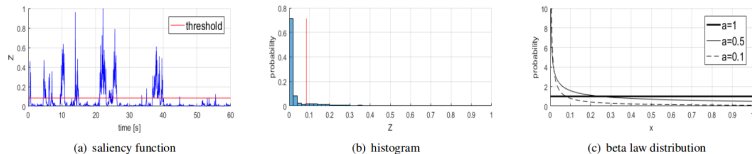


FIGURE 2 – Resulting saliency function (a) and its histogram (b) computed from the recursive synchrosqueezed STFT of an EEG signal. Its probability density function can be compared to a beta distribution (c) with parameter  $a \in \{0.1, 0.5, 1\}$ .

## Application of the recursive and fft-based reassignment method to radar signal analysis

- Partners : Paris, Warsaw Univ. (Poland)
- Involved Tasks : T4.3 : software development

### Contributions

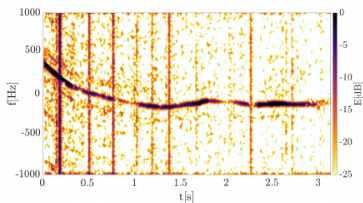
- A complexity comparison between classical and recursive reassignment
- Application to radar signals

### Motivation

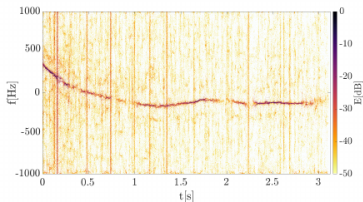
- Future application to radar signal processing



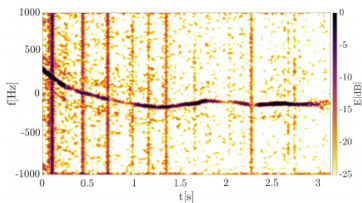
## Numerical results



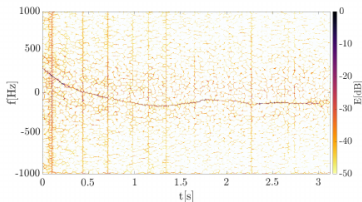
(c) Recursive spectrogram of the echo originating from a drone



(d) Recursive reassigned spectrogram of the echo originating from a drone



(c) FFT-based spectrogram of the echo originating from a drone

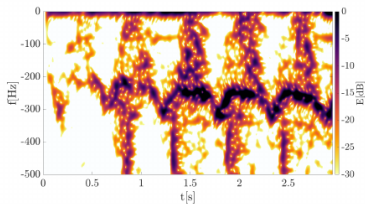


(d) FFT-based reassigned spectrogram of the echo originating from a drone

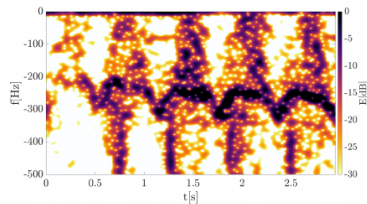
Fig. 2: Results for the recursive method.

Fig. 3: Results for the classical FFT-based method.

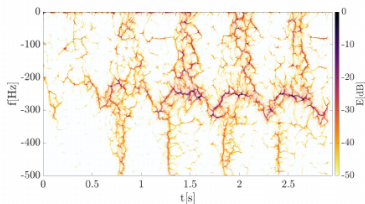
## Numerical results



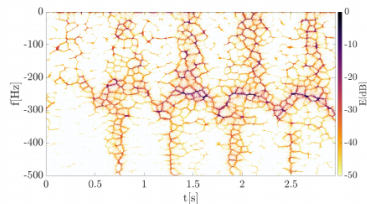
(a) Recursive spectrogram of the echo originating from a walking human



(a) FFT-based spectrogram of the echo originating from a walking human



(b) Recursive reassigned spectrogram of the echo originating from a walking human



(b) FFT-based reassigned spectrogram of the echo originating from a walking human

## Smart Beekeeping based on TF analysis and deep learning

- Partners : Paris
- Involved Tasks : T4.2 : application of SST to audio signals

### Contributions

- Combine TF analysis and deep neural networks for audio classification
- Application to beehive signal analysis for bee queen detection [Orlowska, Fourer, 21]
- Analysis of piping signals [Fourer, Orlowska, 22]

### Motivation

- Predicting the health state of a beehive (smart Beekeeping)

## Numerical results : piping signals analysis

Table 4: Experiment 3: Simultaneously Detection and classification classification comparative results.

| Method   | Feat. dimension | Label      | Recall      | Precision   | F - score   | Accuracy    |
|----------|-----------------|------------|-------------|-------------|-------------|-------------|
| TTB+SVM  | 164             | Tooting    | 0.88        | 0.78        | 0.83        | 0.82        |
|          |                 | Quacking   | 0.03        | 0.12        | 0.05        |             |
|          |                 | Non-piping | 0.99        | 0.89        | 0.94        |             |
| 1D-CNN   | 11,025          | Tooting    | 0.93        | 0.84        | 0.88        | 0.85        |
|          |                 | Quacking   | 0.10        | 0.54        | 0.16        |             |
|          |                 | Non-piping | 0.99        | 0.86        | 0.92        |             |
| MFCC+CNN | 17×47           | Tooting    | 0.88        | 0.81        | 0.84        | 0.84        |
|          |                 | Quacking   | 0.18        | 0.45        | 0.26        |             |
|          |                 | Non-piping | <b>0.99</b> | <b>0.90</b> | <b>0.95</b> |             |
| STFT+CNN | 512×42          | Tooting    | <b>0.94</b> | <b>0.97</b> | <b>0.95</b> | <b>0.91</b> |
|          |                 | Quacking   | <b>0.50</b> | <b>0.76</b> | <b>0.60</b> |             |
|          |                 | Non-piping | <b>0.99</b> | 0.89        | 0.94        |             |

## Numerical results

**Table 4** Comparison of the classification results in Experiment 2 (4-fold hive-independent cross-validation).

| Method           | Features | Label    | Precision   | Recall      | F - score   | Accuracy    |
|------------------|----------|----------|-------------|-------------|-------------|-------------|
| MFCCs+CNN [11]   | 20x44    | Queen    | 0.36        | 0.44        | 0.40        | 0.31        |
|                  |          | No queen | 0.22        | 0.16        | 0.19        |             |
| STFT+CNN         | 513x44   | Queen    | 0.77        | 0.76        | 0.66        | 0.55        |
|                  |          | No queen | 0.33        | 0.20        | 0.33        |             |
| CQT+CNN          | 513x44   | Queen    | 0.10        | 0.07        | 0.08        | 0.25        |
|                  |          | No queen | 0.32        | 0.41        | 0.36        |             |
| mean-CQT+CNN     | 27x44    | Queen    | 0.25        | 0.11        | 0.16        | 0.38        |
|                  |          | No queen | 0.41        | 0.65        | 0.50        |             |
| mean-STFT+CNN    | 27x44    | Queen    | 0.71        | 0.86        | 0.78        | 0.75        |
|                  |          | No queen | 0.81        | 0.64        | 0.71        |             |
| mean-STFT+CNN+DA | 27x44    | Queen    | <b>0.96</b> | <b>0.99</b> | <b>0.96</b> | <b>0.96</b> |
|                  |          | No queen | <b>0.99</b> | <b>0.94</b> | <b>0.96</b> |             |

### Python codes

[https://github.com/agniorlowska/beequeen\\_prediction](https://github.com/agniorlowska/beequeen_prediction)

<https://fourer.fr/dcse22>

## Speech Emotion Recognition using TF Analysis

- Partners : Paris
- Involved Tasks : T4.2 : application of SST to audio signals

### Contributions

- Combine TF analysis and deep neural networks for audio classification
- Efficient data augmentation technique for audio classification
- Investigation of several deep CNN architectures originally designed for image classification
- Application to speech signal for emotion recognition [Xia, Fourer, 21]

### Motivation

- Speech Emotion Recognition

## Main Idea

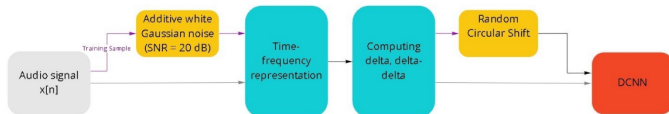
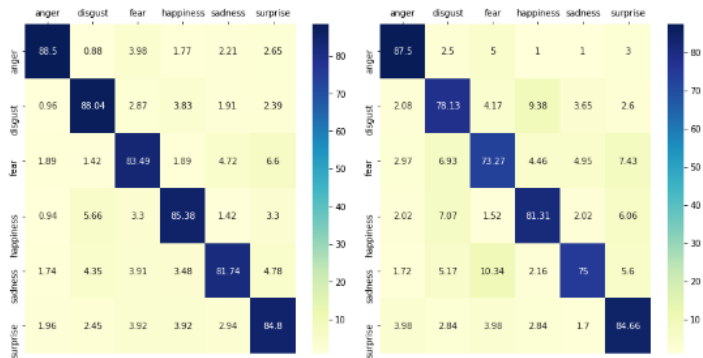


Figure 1: Overview of the proposed approach.

## Python codes

<https://github.com/llnanis/SER-RCS>

## Numerical results

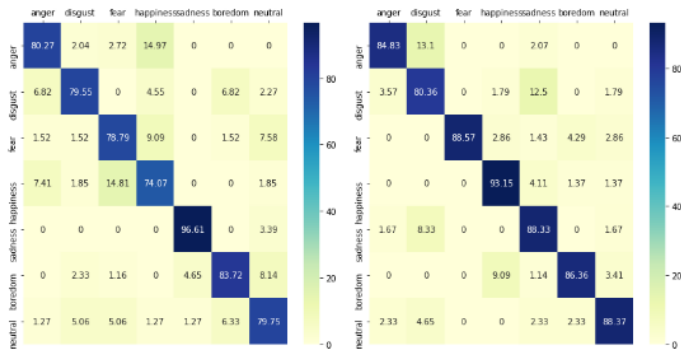


(a) proposed, STFT-Alex+RCS41 (Acc. 85.33%) (b) DCNN-DTPM [12] (Acc. 79.25%)

Figure 4: *eNTERFACE05* confusion matrices obtained using our proposed method STFT-Alexnet + RCS41 (a) and DCNN-DTPM [12].



## Numerical results



(a) proposed, STFT-Alex+RCS19 (Acc. 81.82%) (b) DCNN-DTPM [12] (Acc. 87.31%)

Figure 5: *EMO-DB* confusion matrices obtained using our proposed method *STFT-Alexnet + RCS19* (a) and *DCNN-DTPM* [12].

## Objective 4 : Applications and Software developments

- Souriau, R., Fourer, D., Chen, H., Lerbet, J., Maaref, H., & Vigneron, V. (2019, August). High-Voltage Spindles detection from EEG signals using recursive synchrosqueezing transform. In GRETSI.
- Abratkiewicz, K., Samczynski, P., & Fourer, D. (2020, September). A Comparison of the Recursive and FFT-based Reassignment Methods in micro-Doppler Analysis. In 2020 IEEE Radar Conference (RadarConf20) (pp. 1-6). IEEE.
- Orłowska, A., Fourer, D., Gavini, J. P., & Cassou-Ribehart, D. (2022). Honey Bee Queen Presence Detection from Audio Field Recordings using Summarized Spectrogram and Convolutional Neural Networks. In International Conference on Intelligent Systems Design and Applications (pp. 83-92). Springer, Cham.
- Xia, S., Fourer, D., Audin, L., Rouas, J. L., & Shochi, T. (2022, May). Speech Emotion Recognition using Time-frequency Random Circular Shift and Deep Neural Networks. In Speech Prosody 2022.

To appear :

- D. Fourer and A. Orłowska, Detection and Identification of Beehive Piping Audio Signals. Proc. DCASE 2022. Nancy, France. (accepted for publication)

# Plan

- 1 Introduction
- 2 Objective 1 : New approaches for the study of MCSs
  - Contribution 1 : Second-order time-reassigned synchrosqueezing
  - Contribution 2 : New mode extraction methods
  - Contribution 3 : Harmonic/Percussive separation using AM-FM
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## On the use of concentrated TFR with DNN

- Partners : Paris, Nantes
- Involved Tasks : T4.3 and T2.2 : development of original DNN-based signal processing tools

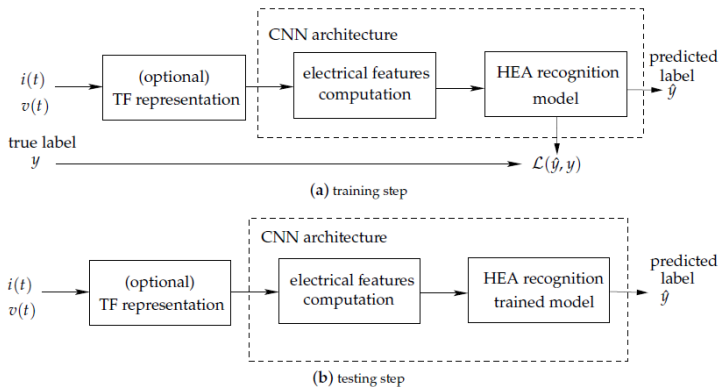
### Contributions

- First study on the relevance of concentrated TFR applied to DNN-based signal classification
- Combination of 2D-CNN architecture with reassigned spectrogram and synchrosqueezed STFT
- Application to non-intrusive load monitoring (NILM)

### Motivation

- Improving the accuracy of existing NILM methods
- Investigating the relevance of concentrated TFR for signal classification

# Main Idea



## Numerical results

Table 1. Comparative results (in percentage) of the different HEA recognition methods applied to the PLAID dataset. The window parameter  $L$  is empirically chosen to provide the best results.

|  | Acc  | $F_M$ | Rec  | Pre  |
|--|------|-------|------|------|
| P, Q + Random Forest [15,19]   | 97.8 | 97.7  | 97.6 | 97.9 |
| STFT ( $L = 60$ , single-input CNN)                                  | 87.1 | 87.2  | 87.3 | 88.4 |
| STFT ( $L = 600$ , CNN with two channels)                            | 97.7 | 97.5  | 97.5 | 97.9 |
| STFT ( $L = 600$ , CNN concatenated)                                 | 95.6 | 95.7  | 95.5 | 96.1 |
| Synchrosqueezing ( $L = 600$ , single-input CNN)                     | 91.9 | 92.1  | 92.4 | 93.1 |
| Synchrosqueezing ( $L = 60$ , CNN with two channels)                 | 85.4 | 85.0  | 85.4 | 86.1 |
| Synchrosqueezing ( $L = 60$ , CNN concatenated)                      | 87.2 | 87.3  | 87.4 | 87.9 |
| Time-reassigned synchrosqueezing ( $L = 60$ , single-input CNN)      | 85.8 | 86.1  | 86.4 | 85.9 |
| Time-reassigned synchrosqueezing ( $L = 60$ , CNN with two channels) | 91.4 | 91.2  | 90.9 | 92.1 |
| Time-reassigned synchrosqueezing ( $L = 60$ , CNN concatenated)      | 92.3 | 92.3  | 92.4 | 91.9 |
| Reassigned spectrogram ( $L = 600$ , single-input CNN)               | 74.4 | 75.0  | 74.1 | 77.3 |

## DNN-based reassignment

- Partners : Paris
- Involved Tasks : T2.2 : development of original DNN-based signal processing tools

### Contributions

- Improving the readability of a TFR
- Comparison of classical and DNN-based reassignment methods

### Motivation

- Improving the robustness to noise with a data-driven approach
- Reassignment can be viewed as an image post-processing operation

## Main Idea

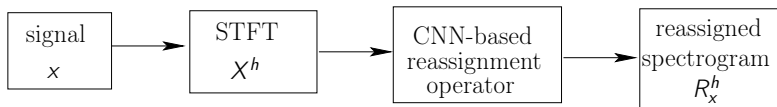


Figure : Proposed DNN-based reassignment method

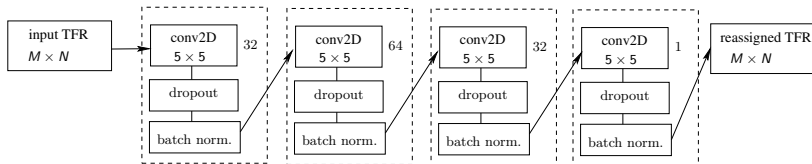


Figure : Proposed architecture based on 2D CNN.

- Uses 2D convolutional neurons with a  $5 \times 5$  kernel
- Activation function : REctified Linear Unit (RELU)
- Dropout : Randomly discard 10% of the computed coefficients
- Optimizer : RMSProp<sup>1</sup>

1. Bengio, Yoshua. "Rmsprop and equilibrated adaptive learning rates for nonconvex optimi-



## Unitary test on a sinusoidal signal 1/2

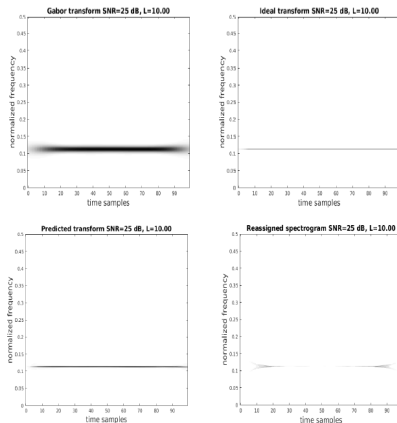


Figure : Comparison between  $|X^h|$ , ITFR, DNN1 estimation and classical reassigned spectrogram.

## Unitary test on a sinusoidal signal 2/2

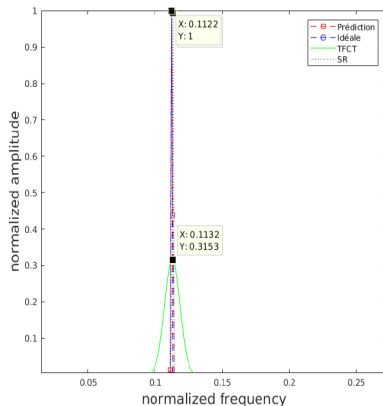


Figure : Comparison between  $|X^h|$ , ITFR, DNN1 estimation and classical reassigned spectrogram at a given time instant  $n = 50$ .

## Unitary test on an impulse signal 1/2

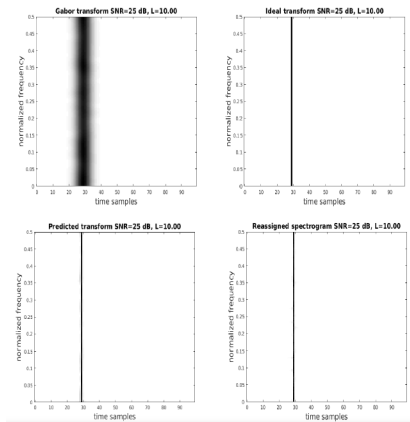


Figure : Comparison between  $|X^h|$ , ITFR, DNN2 estimation and classical reassigned spectrogram.

## Unitary test on an impulse signal 2/2

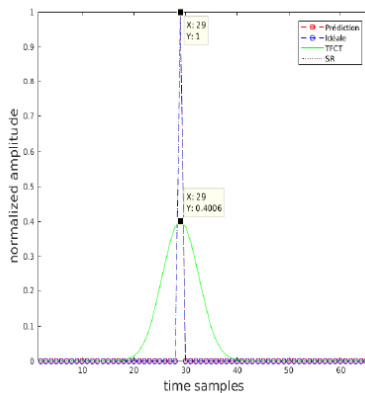


Figure : Comparison between  $|X^h|$ , ITFR, DNN2 estimation and classical reassigned spectrogram at a given normalized frequency  $\lambda = 0.3$ .

## Noisy signal (SNR=5dB)

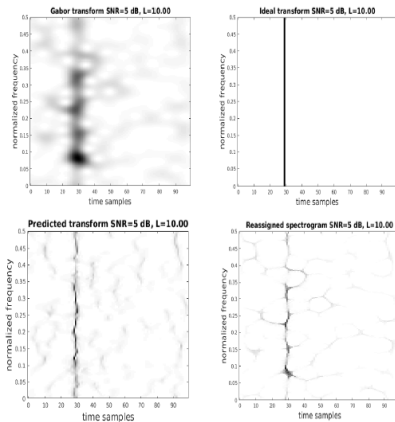


Figure : Comparison between  $|X^h|$ , ITFR, DNN2 estimation and classical reassigned spectrogram.

## Noisy signal (SNR=5dB)

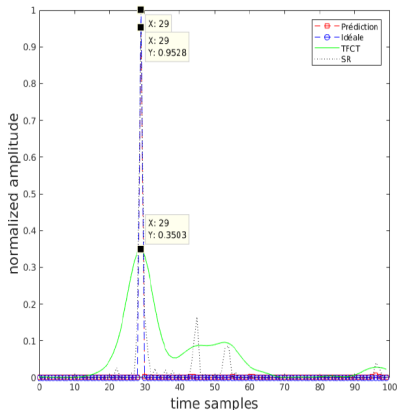


Figure : Comparison between  $|X^h|$ , ITFR, DNN2 estimation and classical reassigned spectrogram at a given normalized frequency  $\lambda = 0.3$ .

# Multi-component noisy signal 1/2

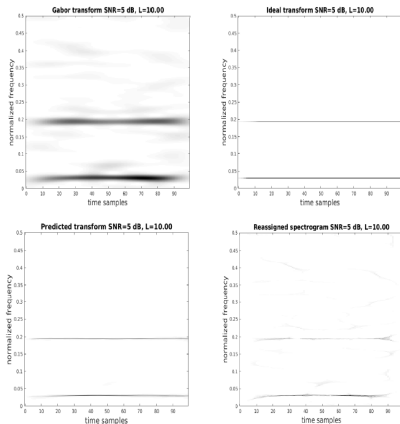


Figure : Comparison between  $|X^h|$ , ITFR, DNN2 estimation and classical reassigned spectrogram.

## Multi-component noisy signal 2/2

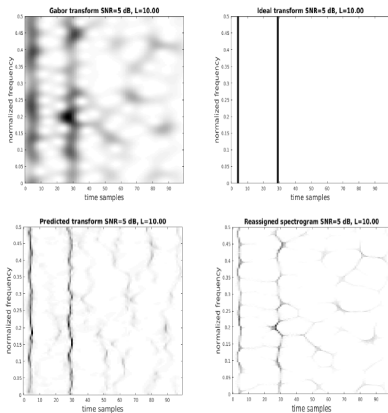


Figure : Comparison between  $|X^h|$ , ITFR, DNN2 estimation and classical reassigned spectrogram.



# Overlapping components 1/2

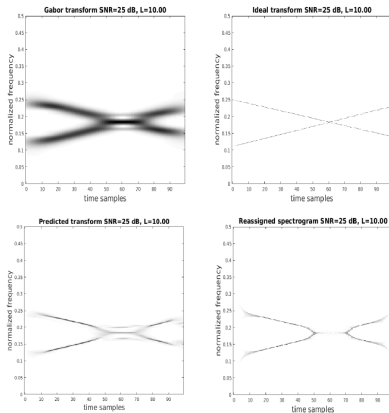


Figure : Comparison between  $|X^h|$ , ITFR, DNN2 estimation and classical reassigned spectrogram.

## Overlapping components 2/2

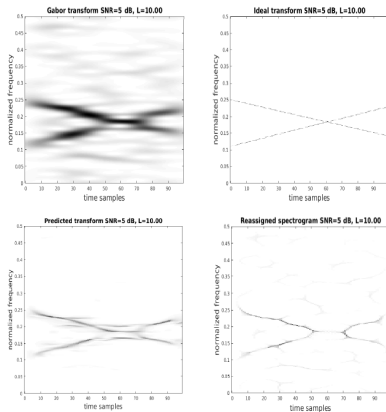


Figure : Comparison between  $|X^h|$ , ITFR, DNN2 estimation and classical reassigned spectrogram.

## Objective 2 : Improving signal representations using data-driven

- Houidi, S., Fourer, D., & Auger, F. (2020). On the use of concentrated time-frequency representations as input to a deep convolutional neural network : Application to non intrusive load monitoring. Entropy, 22(9), 911.

In preparation :

- D.Fourer, Q. Legros and F. Auger. Improving the readability of Time-frequency and Time-Scale Representations using Deep Convolutional Neural Networks

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## Outlook

### 2022

- oct (1 month) : EM-based IF and CR estimation methods
- nov-dec (2 months) : Finalizing and submitting work on DNN-based reassignment

### With extension for Quentin (6 extra months)

- jan-mar (3 months) :
  - Mode-extraction DNN-based methods (image-based segmentation using advanced neural network architectures)
  - TFR information estimation with F. Auger (Nantes)
- apr-jun (3 months) : Adaptive representations learning (investigation based on recurrent CNN with constraint)

### Other ideas :

- spectrogram/scalogram phase reconstruction
- spectrogram/scalogram zeros/peaks statistical analysis
- overlapping modes reconstruction

## Future work directions / Ideas for future projects and collaborations

- A public Benchmarking for mode extraction, denoising and IF/CR estimations methods (with Juan)
- TFR structure information extraction from a machine learning point of view (with Francois)
- New academic project proposal on signal processing for smart beekeeping (with all bees friends)
- Green IA : using TF analysis for computing low-dimension signal models for efficient computation, embedded systems and energy saving
- ...