

# **Analyse, transformation et reconnaissance des signaux sonores**

« Analysis, transformation and  
recognition of audio signals »

**Gaël RICHARD**

Professeur à Télécom Paris



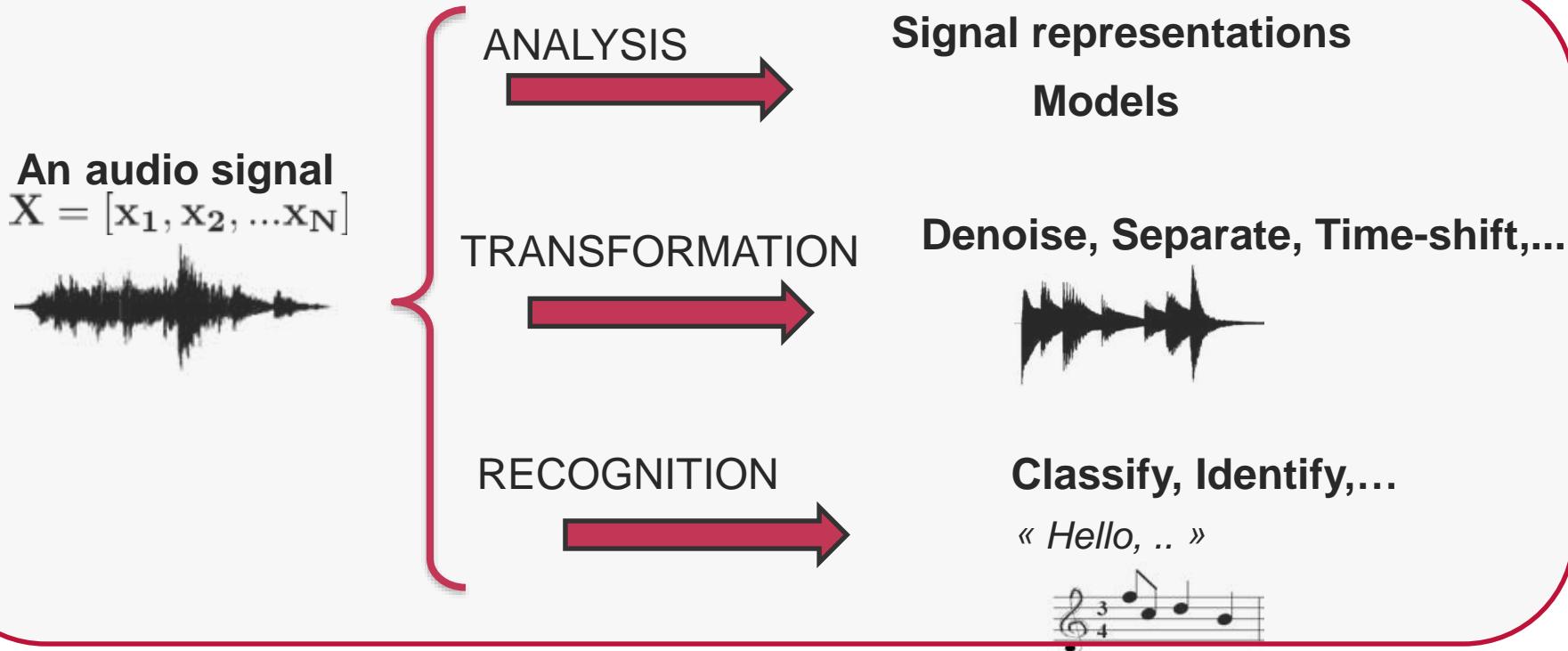
Cérémonie Prix IMT-Académie des sciences, 1<sup>er</sup> décembre 2020

# A research done in collaboration....

■ An immense thanks to all ....



# Analysis, transformation, recognition ... of audio signals



**Text or symbols**

*« Hello, .. »*



**SYNTHESIS**



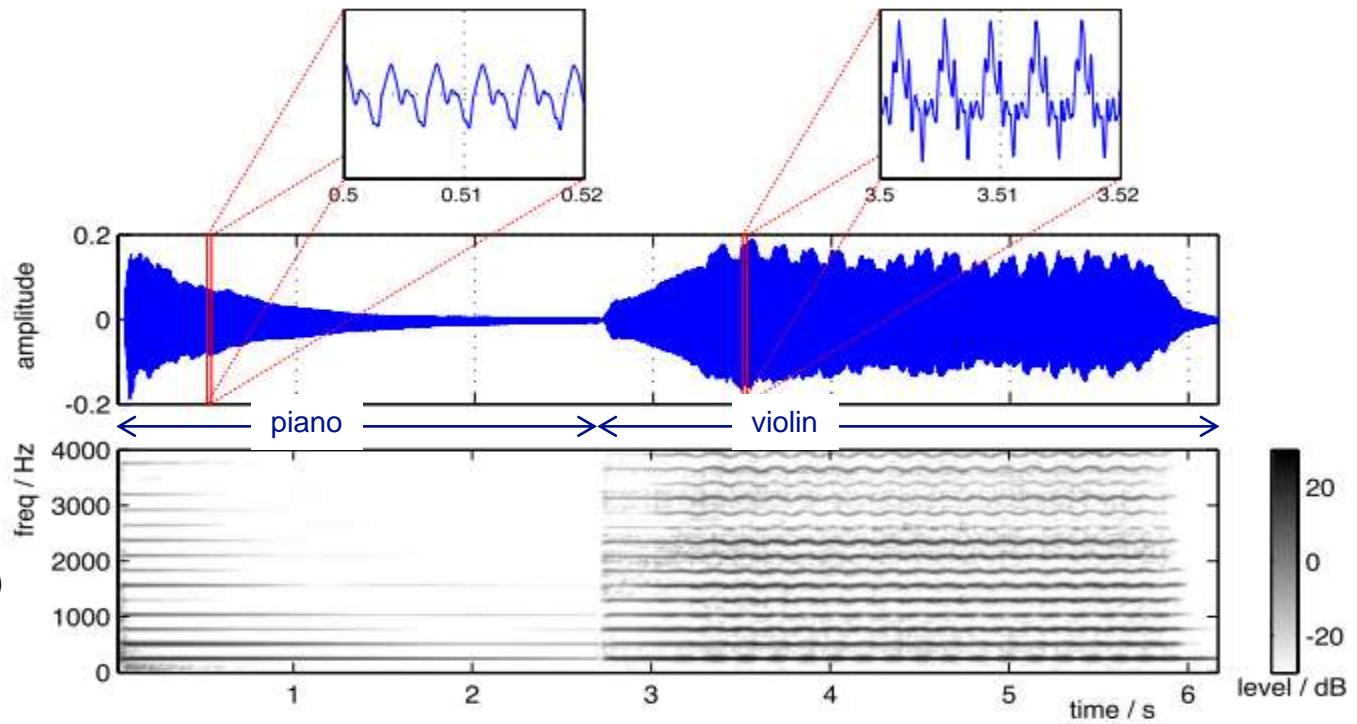
**Audio signal**



# Analysis: Time-frequency representations

- Example with a music signal : note C (Do), with fundamental frequency of 262 Hz, played on a piano and then on a violin.

Time-domain signal



Images from M. Mueller, D. Ellis, A. Klapuri, G. Richard « Signal Processing for Music Analysis, IEEE Trans. On Selected topics of Signal Processing, oct. 2011



# Some generic signal models

## ■ Sum of sinusoids + noise

$$x(n) = \sum_{i=1}^I A_i \cdot \sin(2\pi\nu_i n + \phi_i) + b(n)$$



# Some generic signal models

## ■ Sum of sinusoids + noise

$$x(n) = \sum_{i=1}^I A_i(\textcolor{red}{n}).\sin(2\pi\nu_i n + \phi_i) + \textcolor{red}{m(n)}.b(n)$$



- Modulated noise models for speech synthesis and modification [2]
- Damped sinusoids models for parametric audio coding [3]



[2] G. Richard, C. d'Alessandro, "Analysis/synthesis and modification of the speech aperiodic component", Speech Communication, Vol. 19, Issue 3, September 1996, Pages 221–244

[3] O.Derrien, R. Badeau, G. Richard, "A Parametric Audio Coding with Exponentially Damped Sinusoids, IEEE Trans. on Audio, Speech and Language Processing, Vol 21, N° 7, July 2013.



# Some generic signal models

## ■ Sum of sinusoids + noise

$$x(n) = \sum_{i=1}^I A_i(n) \cdot \sin(2\pi\nu_i(n)n + \phi_i) + b(n)$$


- Modulated noise models for speech synthesis and modification [2]
- Damped sinusoids models for parametric audio coding [3]
- Adaptive Signal subspace tracking [4]
- Frequency estimation in Amplitude/Frequency modulated models [5]



[2] G. Richard, C. d'Alessandro, "Analysis/synthesis and modification of the speech aperiodic component", Speech Communication, Vol. 19, Issue 3, September 1996, Pages 221–244

[3] O.Derrien, R. Badeau, G. Richard, "A Parametric Audio Coding with Exponentially Damped Sinusoids, IEEE Trans. on Audio, Speech and Language Processing, Vol 21, N° 7, July 2013.

[4] R. Badeau, B. David and G. Richard, "Fast Approximated Power Iteration Subspace Tracking", IEEE Trans. on Signal Processing, Vol. 53, Issue 8, Part 1, Aug. 2005 Page(s):2931 – 2941

[5] M. Betser, P. Collen, G. Richard and B. David « Estimation of frequency for AM/FM models using the phase vocoder framework», IEEE Trans. on Signal Processing, Vol. 56, N°. 2, February 2008., Page(s):505 - 517.

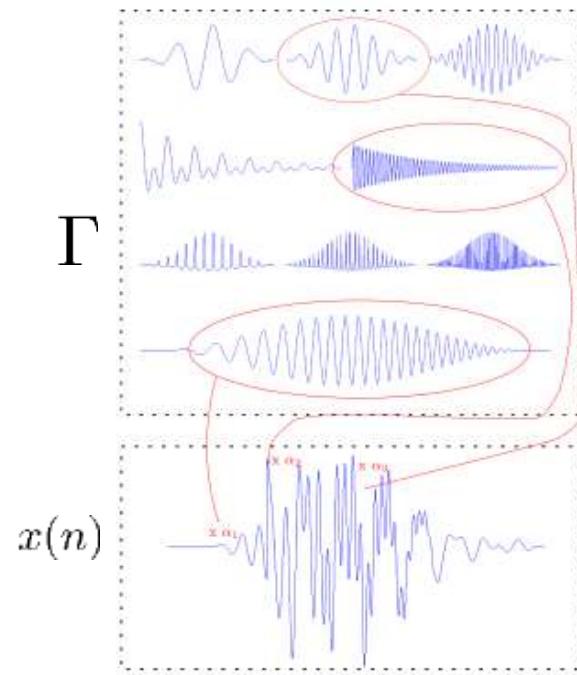


## More specific models ...

### ■ Signal « decomposition » methods

$$x(n) = \sum_{\lambda \in \Gamma} \alpha_\lambda h_\lambda(n)$$

→ The signal is a linear combination of atoms  $h_\lambda(n)$  taken in a dictionary  $\Gamma$



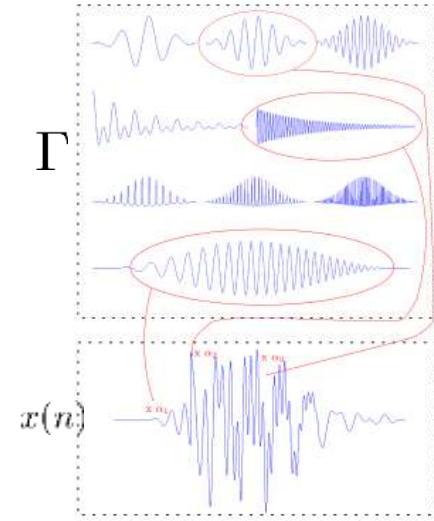
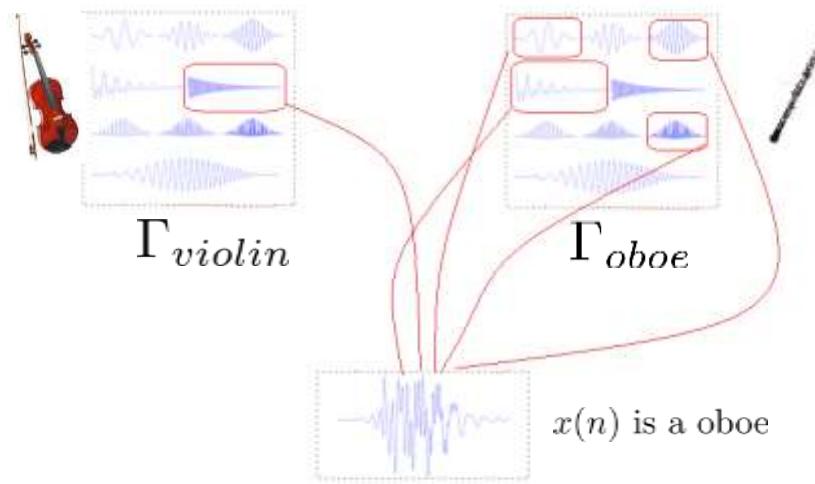
## More specific models ...

### ■ Signal « decomposition » methods

$$x(n) = \sum_{\lambda \in \Gamma} \alpha_\lambda h_\lambda(n)$$

→ With « informed » atoms :

- *Source identification and separation [6]*



[6] P. Leveau, E. Vincent, G. Richard and L. Daudet, « Instrument-Specific Harmonic Atoms for Mid-Level Musical Audio Representation » IEEE Trans. on ASLP, Volume 16, N°1 Jan. 2008 Page(s):116 – 128



# More specific models ...

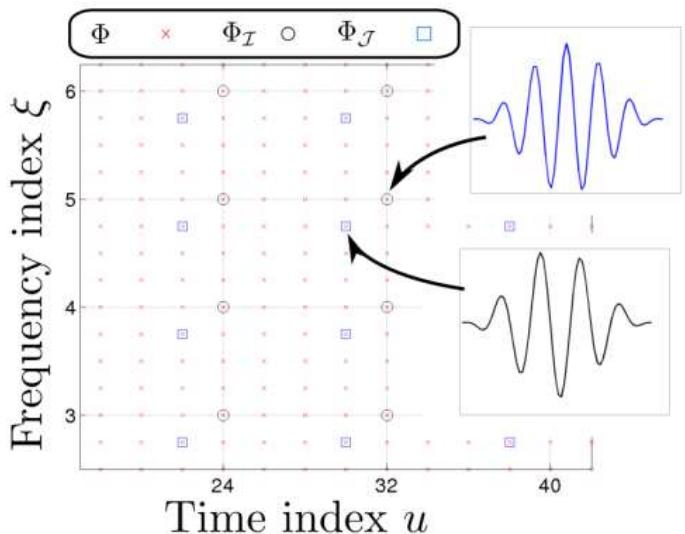
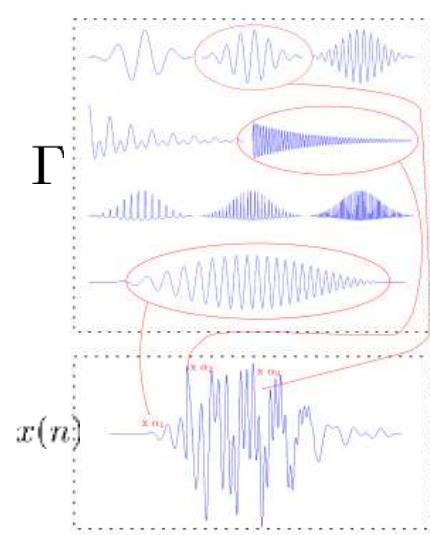
## ■ Signal « decomposition » methods

$$x(n) = \sum_{\lambda \in \Gamma} \alpha_\lambda h_\lambda(n)$$

- With multi-resolution time-frequency atoms :
- *Music signal compression* [7]

With sequence of random dictionaries :

- *Efficiency, denoising (audio, EEG signals)* [8]



[6] P. Leveau, E. Vincent, G. Richard and L. Daudet, « Instrument-Specific Harmonic Atoms for Mid-Level Musical Audio Representation » IEEE Trans. on ASLP, Volume 16, N°1 Jan. 2008 Page(s):116 – 128

[7] E. Ravelli, G. Richard, L. Daudet, Union of MDCT bases for audio coding, IEEE Trans. on Audio, Speech and Language Processing, Vol. 16, Issue 8, pp 1361-1372, Nov. 2008.

[8] M. Moussallam, L. Daudet, G. Richard, "Matching pursuits with random sequential subdictionaries", Signal Processing, 2012.

# More specific models ...

## ■ Non-negative Matrix factorization (NMF)<sup>[9]</sup>

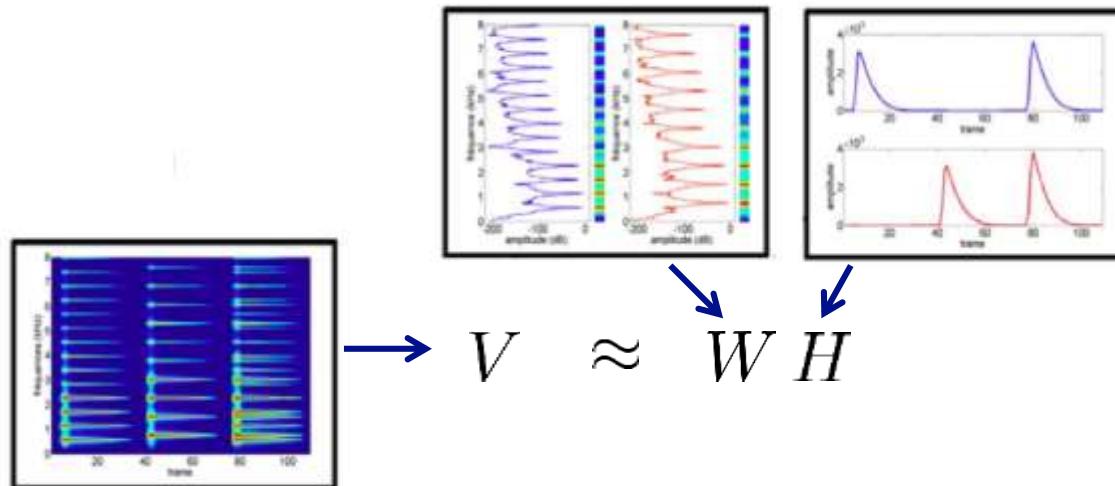


Figure from R. Hennequin



[9] D. Lee, H., Seung, Learning the parts of objects by non-negative matrix factorization. *Nature* **401**, 788–791 (1999).

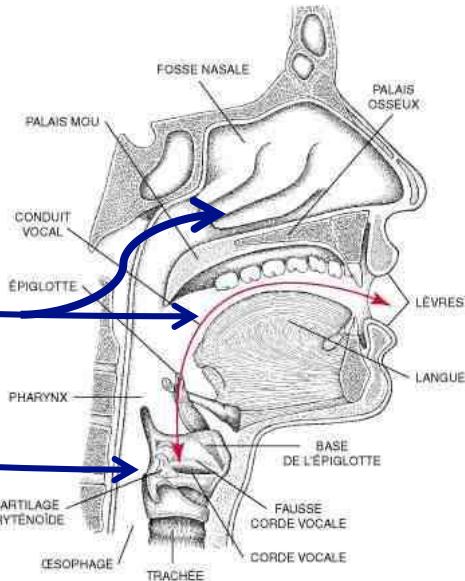
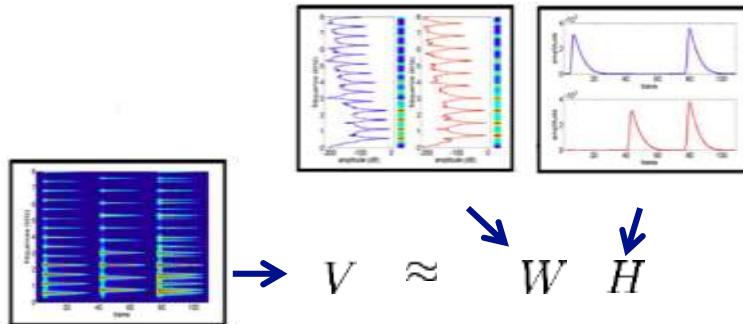
# More specific models ...

## ■ Non-negative Matrix factorization (NMF)<sup>[9]</sup>

- Underdetermined source separation
- Un example on singing voice separation

$$\underbrace{\mathbf{X}}_{\text{Recording}} = \underbrace{\mathbf{V}}_{\text{Voice}} + \underbrace{(\mathbf{W}^M \mathbf{H}^M)}_{\text{music}},$$

$$\underbrace{\mathbf{V}}_{\text{Voice}} = \underbrace{\mathbf{S}}_{\text{source}} \bullet \underbrace{\mathbf{F}}_{\text{filter}}$$



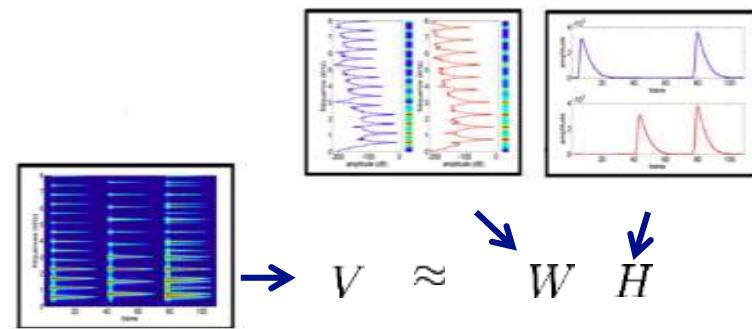
[9] D. Lee, H., Seung, Learning the parts of objects by non-negative matrix factorization. *Nature* **401**, 788–791 (1999).

[10] J-L Durrieu, B. David, G. Richard, A musically motivated mid-level representation for pitch estimation and musical audio source separation, *IEEE Journal on Selected Topics in Signal Processing*, October 2011.

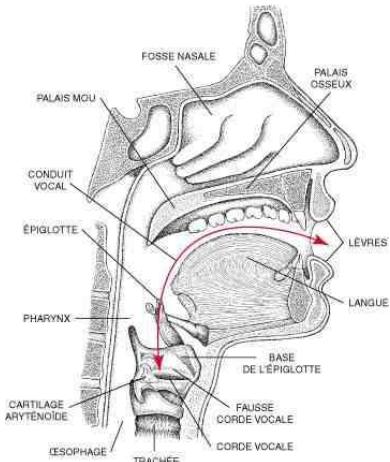
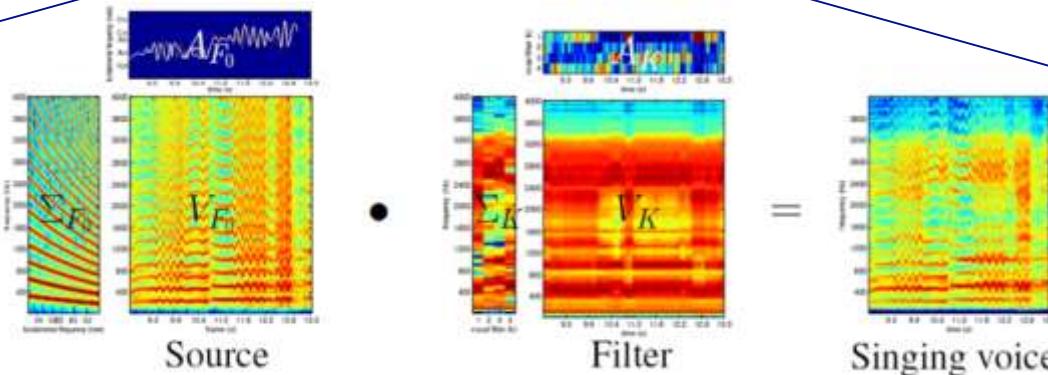
# More specific models ...

## ■ Non-negative Matrix factorization (NMF)<sup>[9]</sup>

- Underdetermined source separation
- An example on singing voice separation



$$\underbrace{\mathbf{X}}_{\text{Recording}} = \underbrace{\mathbf{V}}_{\text{Voice}} + \underbrace{(\mathbf{W}^M \mathbf{H}^M)}_{\text{music}},$$



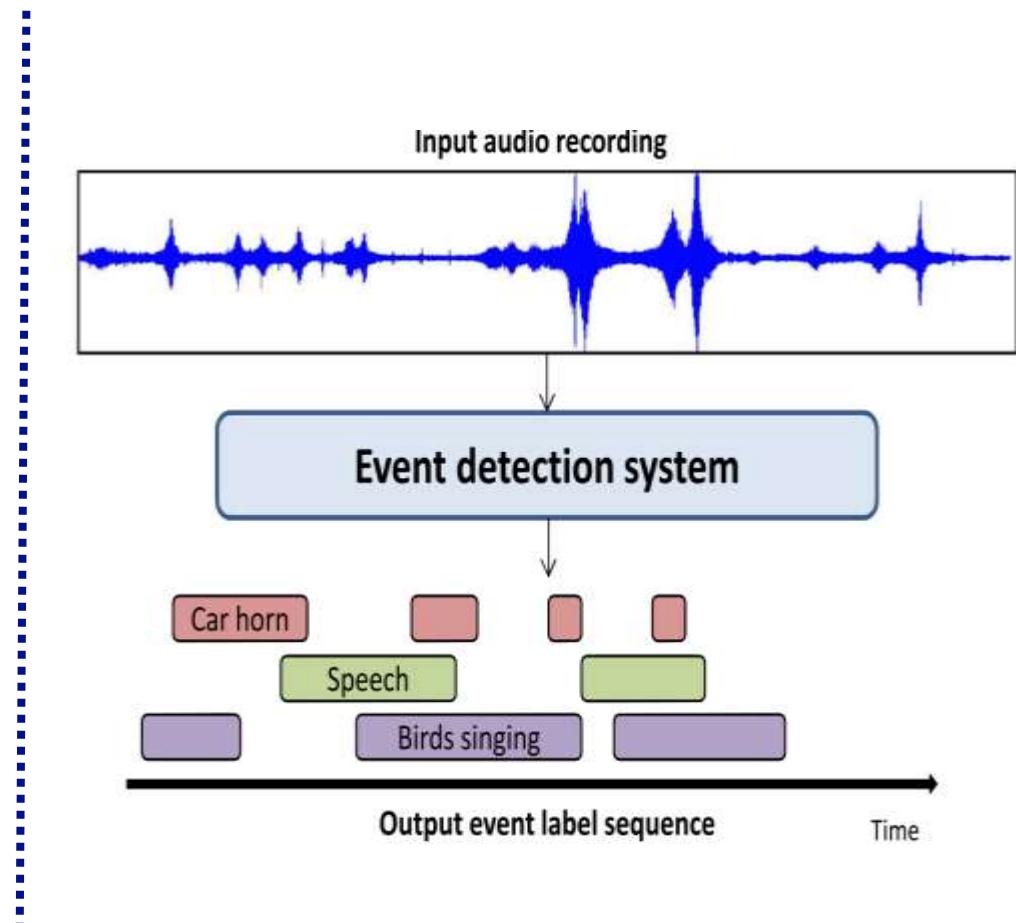
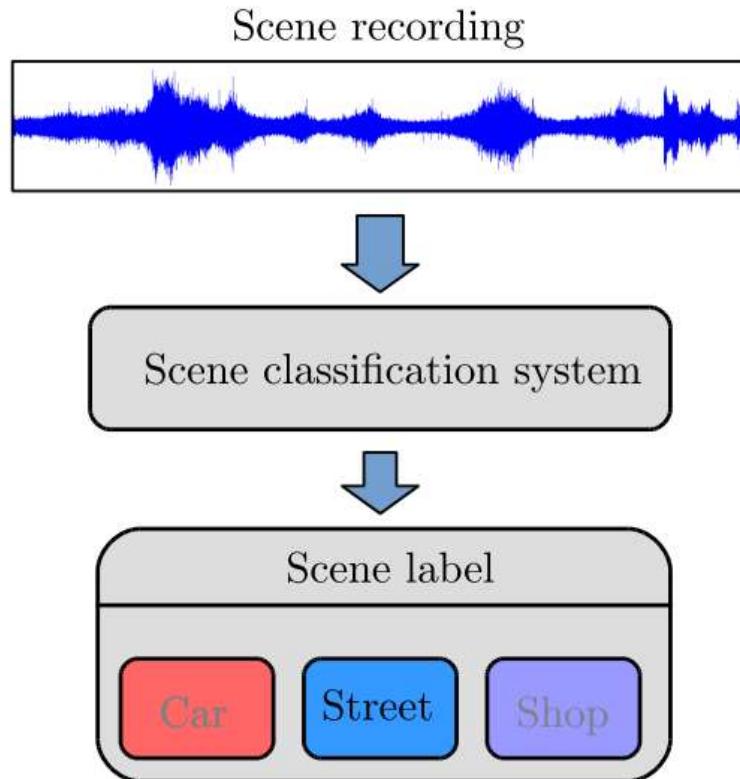
[9] D. Lee, H., Seung, Learning the parts of objects by non-negative matrix factorization. *Nature* **401**, 788–791 (1999).

[10] J-L Durrieu, B. David, G. Richard, A musically motivated mid-level representation for pitch estimation and musical audio source separation, *IEEE Journal on Selected Topics in Signal Processing*, October 2011.

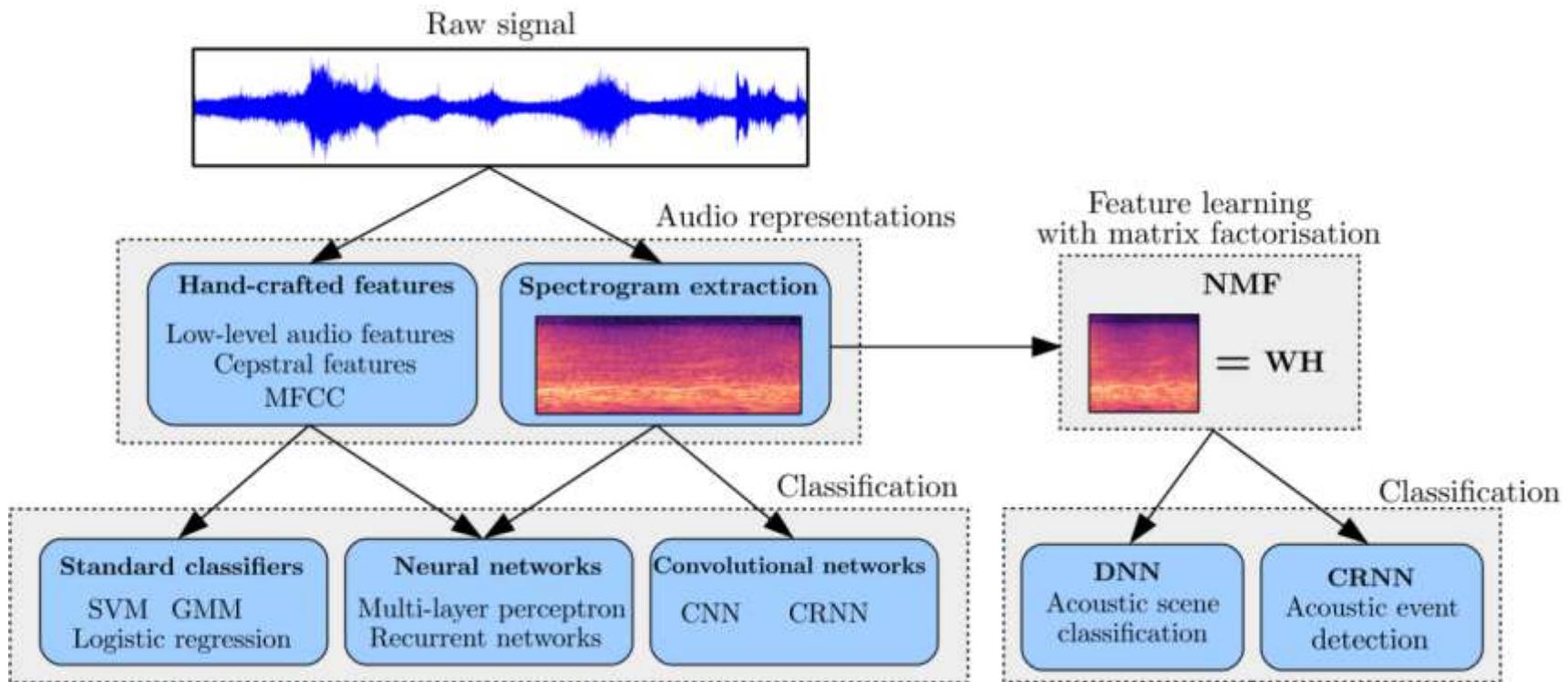


# To recognize

## *Sound events and acoustic scene recognition*



# Associating signal models and deep learning

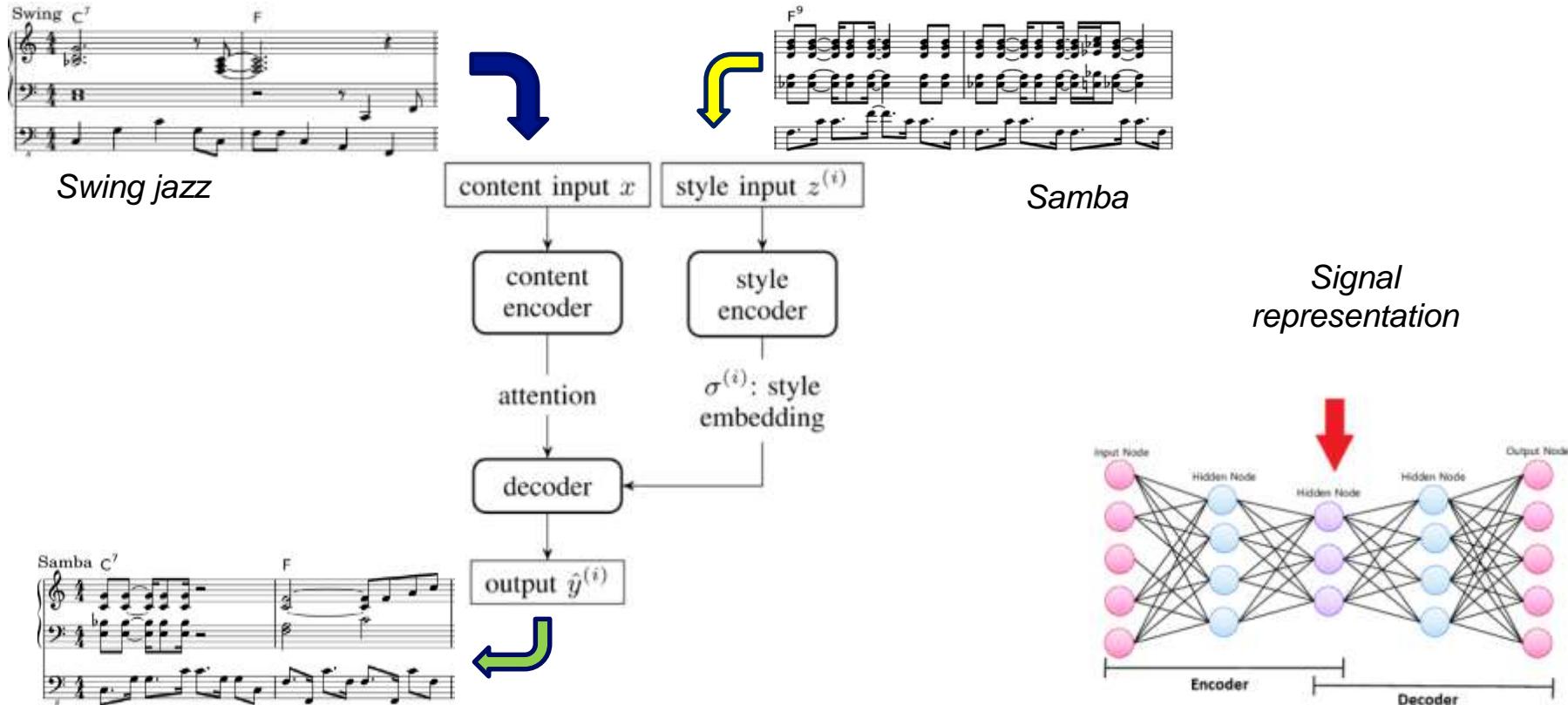


[11] V. Bisot, R. Serizel, S. Essid, G. Richard, "Feature Learning with Matrix Factorization Applied to Acoustic Scene Classification", IEEE/ACM Transactions on Audio, Speech, and Language Processing, (2017), Special Issue on Sound Scene and Event Analysis.

# Recognize, Transform, Synthesize ...

## Symbolic music style transfer

- ... Or playing a given music file in the style of another music excerpt.



[13] Ondrej Cifka, Umut Simsekli, Gaël Richard, "Groove2Groove: One-Shot Music Style Transfer with Supervision from Synthetic Data", IEEE/ACM Transactions on Audio, Speech, and Language Processing, (preprint) accepted for publication, 2020

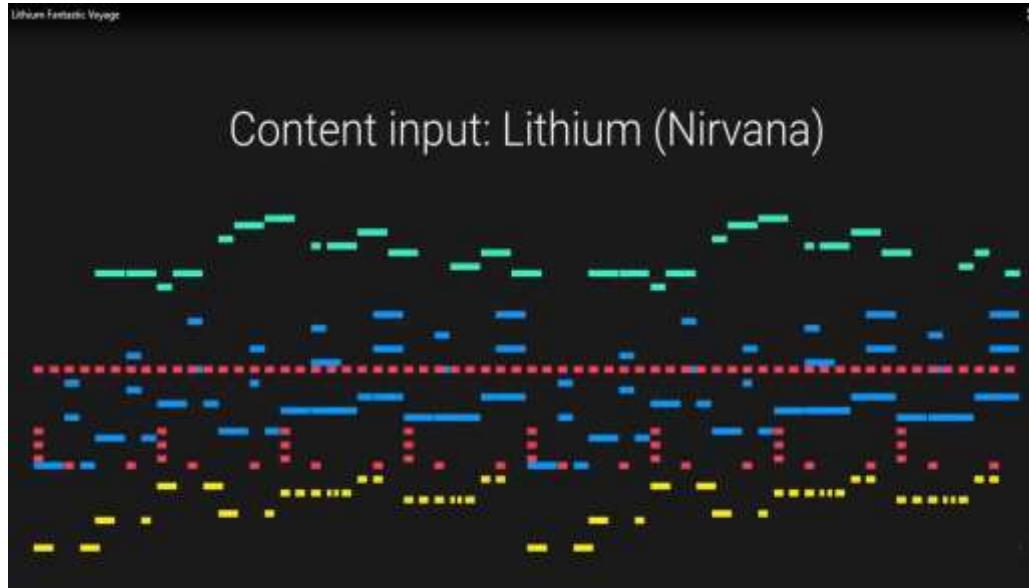
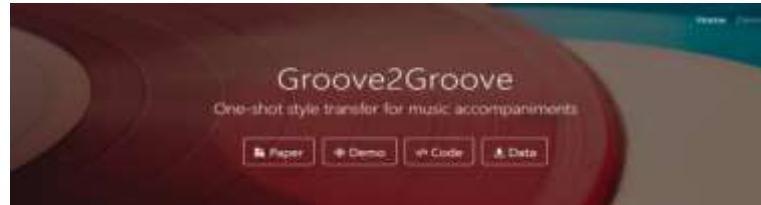
Sound examples at : <https://groove2groove.telecom-paris.fr>

# Recognize, Transform, Synthesize ...

## Symbolic music style transfer

- ... Or playing a given music file in the style of another music excerpt.

[A short demo](#)



[13] Ondrej Cifka, Umut Simsekli, Gaël Richard, "Groove2Groove: One-Shot Music Style Transfer with Supervision from Synthetic Data", IEEE/ACM Transactions on Audio, Speech, and Language Processing, vol. 28, 2020

Sound examples at : <https://groove2groove.telecom-paris.fr>

# Conclusion

## ■ ... towards hybrid models

- Models are **parameter efficient** and **interpretable**,
- Deep Neural networks are **very powerful** but needs **huge amount of data** and **computing power**, and are not always easily interpretable.
- Hybrid models: promise for more explanability, frugality and efficiency

