

Exploiting knowledge for model-based deep music generation

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MBZUAI Workshop 2025

*work with collaborators and in particular **K. Schulze-Forster, M. Agarwal, T. Baoueb, X. Bie, C. Wang, C. Doire, L. Kelley, B. Torres, P. Chouteau, R. Badeau**


Content

- **Context and motivation**
- **Towards hybrid (or model-based) deep learning**
 - Some examples in other domains
 - Hybrid deep learning in audio
- **Specific examples in**
 - Unsupervised music source separation
 - Symbolic music generation with transformers
 - Music timbre transfer
- **Conclusion**

Context and motivation

- Machine learning: a growing trend towards pure “Data-driven” deep learning approaches
- High performances but some main limitations:
 - *“Knowledge” is learned (only) from data*
 - *Complexity: overparametrized models (>> 100 millions parameters)*
 - Overconsumption regime
 - Non-interpretable/non-controllable

Context and motivation

- Machine learning: a growing trend towards pure “Data-driven” deep learning approaches
- High performances but some main limitations:
 - “Knowledge” is learned (only) from data
 - Complexity: overparametrized models (> 100 millions parameters)
 - Overconsumption regime
 - Non-interpretable/non-controllable
- The main goal of the project : 

Main goal : To build controllable and frugal machine listening models based on expressive generative modelling

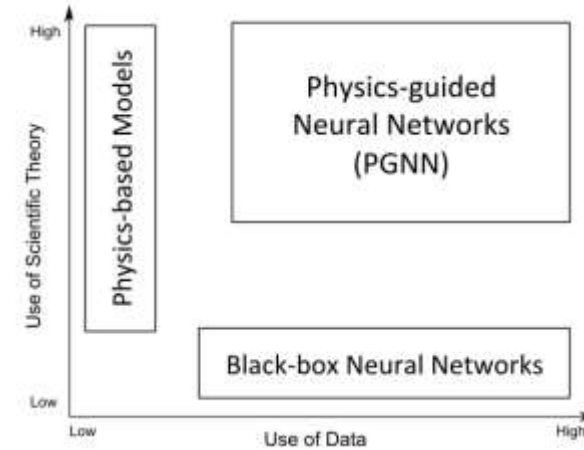
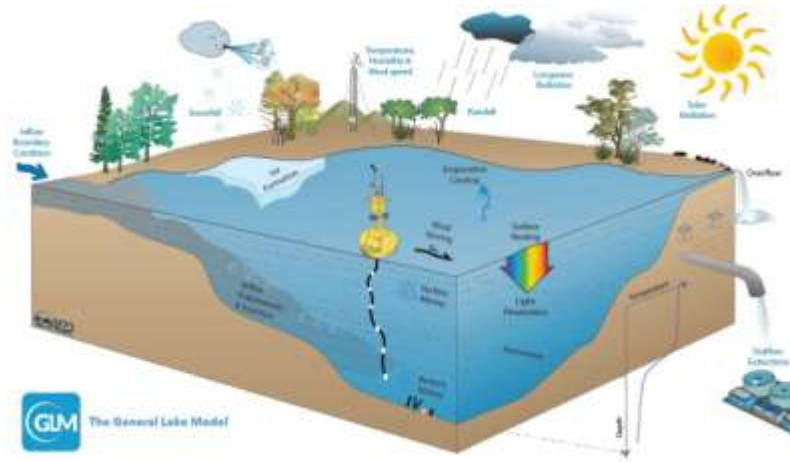
Approach: to build *Hybrid deep learning models*, by **integrating our prior knowledge** about the nature of the processed data.



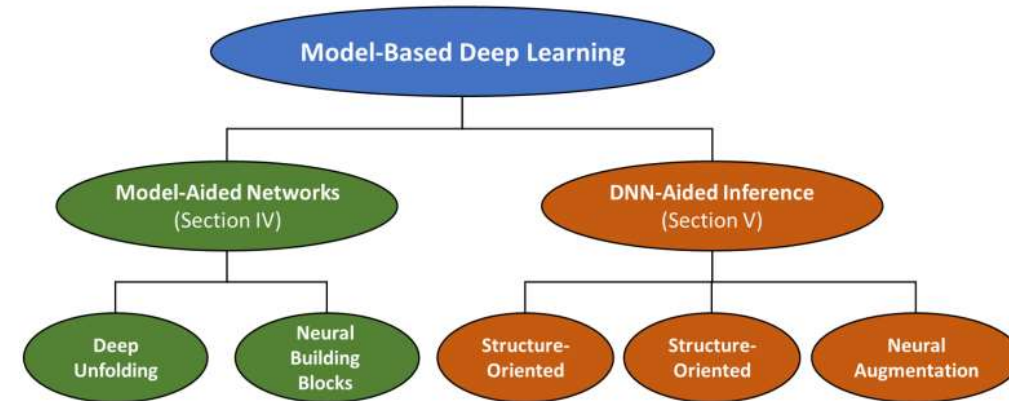
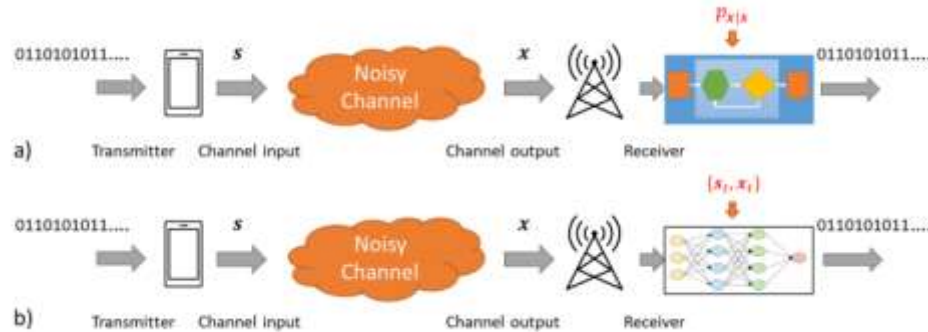
Towards Hybrid (or model-based) deep learning

... some prior works.

- Physics-guided neural networks in remote sensing [1],



- Digital communication and Image restoration [2,3]

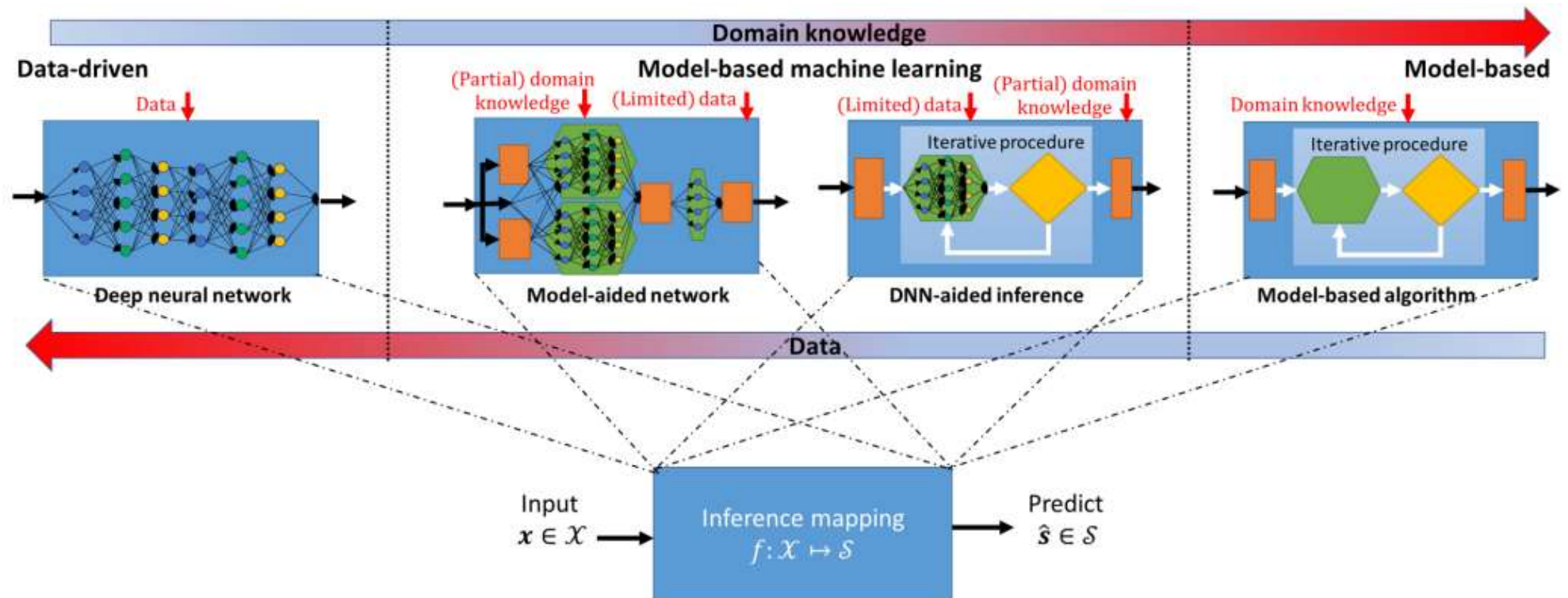


[1] A. Karpatne & al. "Physics-guided Neural Networks (PGNN): An Application in Lake Temperature Modeling," arXiv, 1710.11431, 2017.
 [2] B. Lecouat & al., "Fully Trainable and Interpretable Non-Local Sparse Models for Image Restoration.," 2020. (hal-02414291v2).
 [3] N. Shlezinger, & al., "Model-Based Deep Learning," in *Proceedings of the IEEE*, vol. 111, no. 5, pp. 465-499, May 2023,

Exploiting knowledge for model-based deep music generation

Towards Hybrid (or model-based) deep learning ... some prior works.

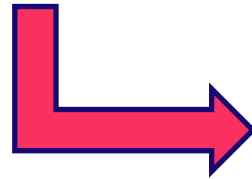
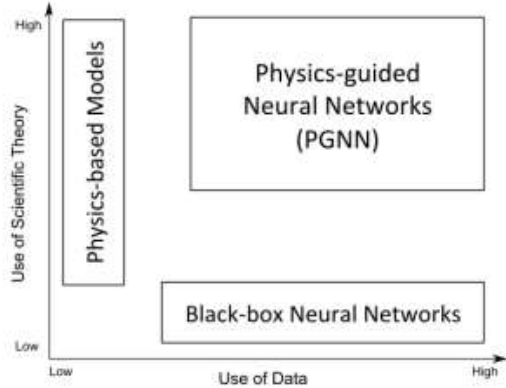
- Illustration of model-based versus data-driven inference (from [3])



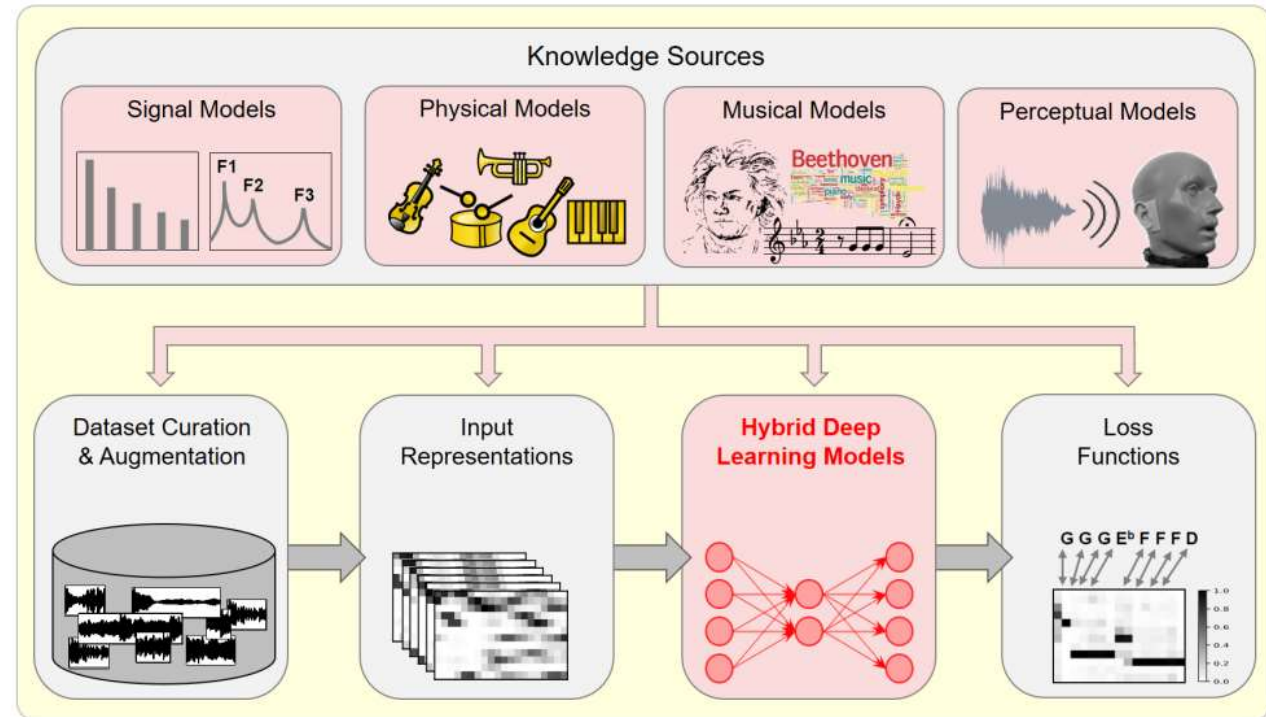
Towards model-based deep learning approaches

- Coupling model-based and deep learning:

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Example with Hybrid deep model for Music signals



Towards model-based deep learning

... some prior works in audio

➤ Use of a model-based feature representation

- Non-Negative Matrix Factorization (NMF) models with CNNs for audio scene classification [1, 2]

➤ Exploit the concept of deep unrolling

- **Deep NMF** : Converting one iteration of NMF (iterative algorithm) into one layer of a DNN [3]

➤ Use DNN as noise estimator

- **Deep Griffin-Lim**: Each iteration of an iterative phase retrieval algorithm is « denoised » by DNN [4]

➤ ... Many other examples

[1] V. Bisot & al., "Feature Learning with Matrix Factorization Applied to Acoustic Scene Classification", ACM/IEEE Trans. on ASLP, vol. 25, no. 6, 2017

[2] V. Bisot & al., Leveraging deep neural networks with nonnegative representations for improved environmental sound classification *IEEE International Workshop on Machine Learning for Signal Processing MLSP, Sep 2017, Tokyo*,

[3] J. L. Roux & al., "Deep NMF for speech separation,," in *IEEE Int. Conf. on Acous., Speech and Signal Proc. (ICASSP)*, 2015

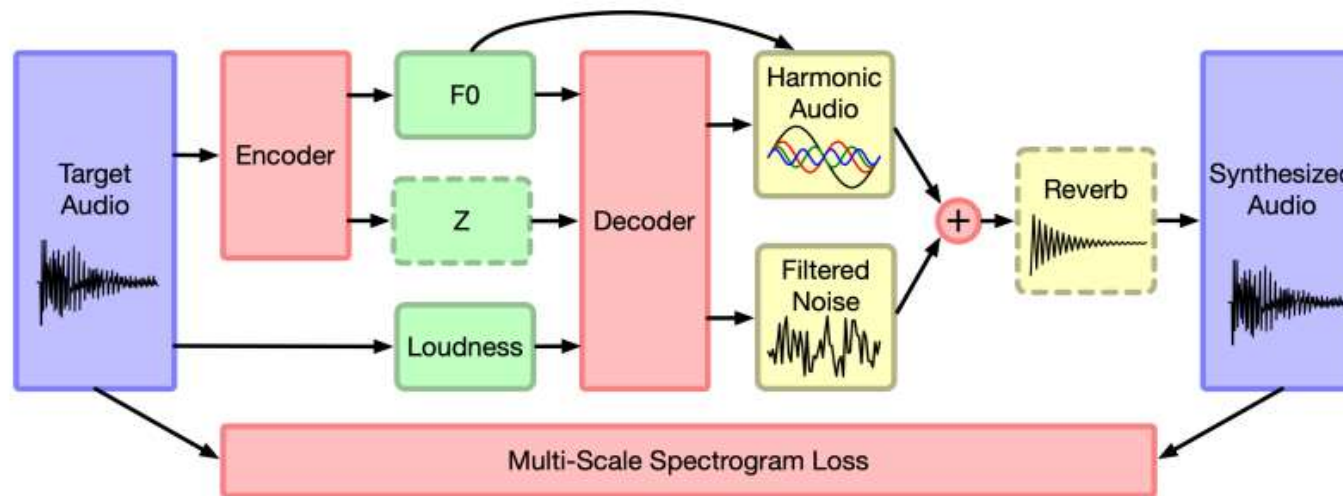
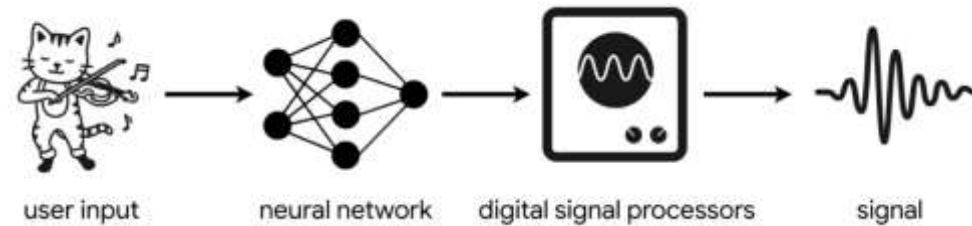
[4] Y. Masuyama, K. Yatabe, Y. Koizumi, Y. Oikawa and N. Harada, "Deep Griffin-Lim Iteration: Trainable Iterative Phase Reconstruction Using Neural Network," in *IEEE Journal of Selected Topics in Signal Processing*, vol. 15, no. 1, pp. 37-50, Jan. 2021,



Towards model-based deep learning

... some prior works in audio

- Coupling signal processing modules with deep learning for audio synthesis
- The example of DDSP (Engel & al.)

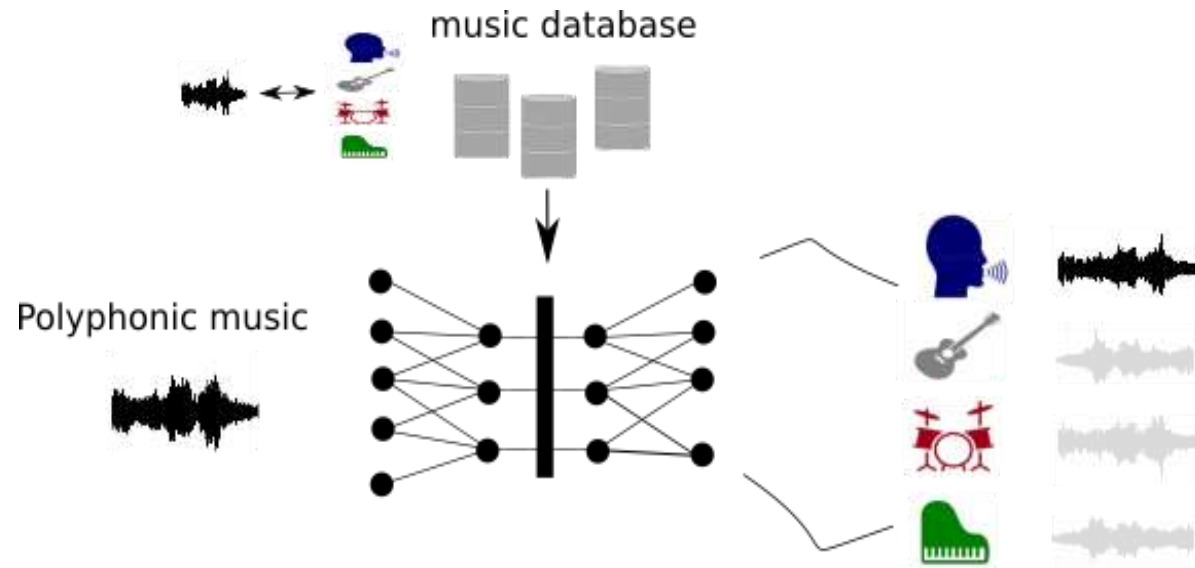


X. Wang & al. "Neural Source-Filter Waveform Models for Statistical Parametric Speech Synthesis," in IEEE/ACM Trans. on ASLP Proc., vol. 28, 2020.
 J. Engel & al., "DDSP: Differentiable Digital Signal Processing," in Int. Conf. on Learning Representations (ICLR), 2020.

Towards model-based deep learning

... by integrating our prior knowledge about the nature of the processed data.

- For example in music source separation



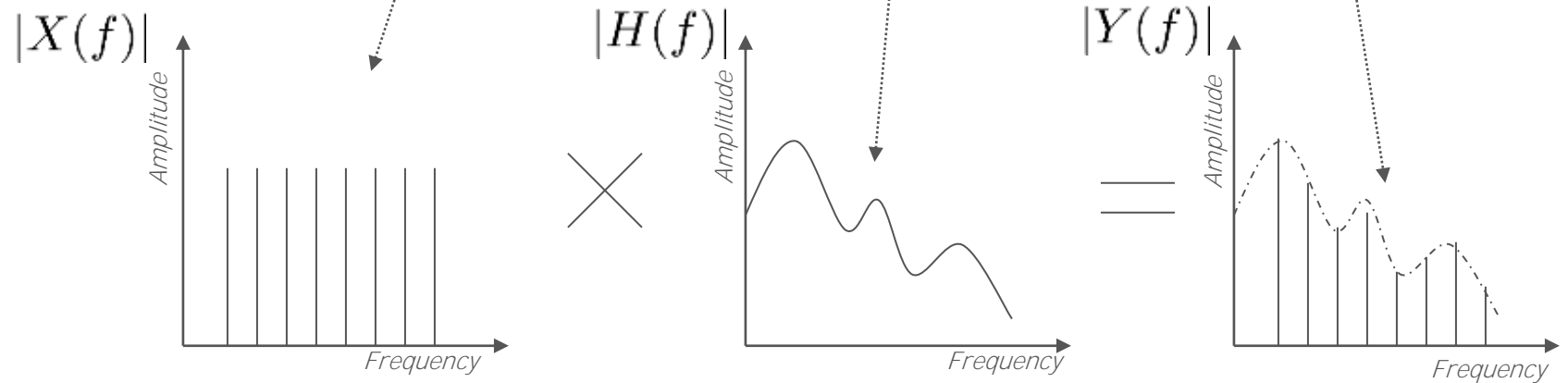
Main limitations:

- Difficulty to obtain « aligned » data
- Knowledge learned (only) from data
- Complexity: overparametrized models
- Overconsumption regime
- **Non-interpretable/non-controllable**

The source filter model

an efficient speech production model

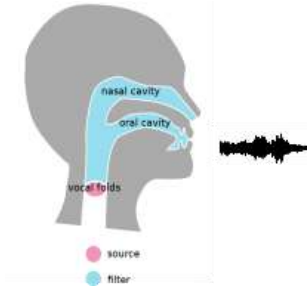
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Towards model-based deep learning

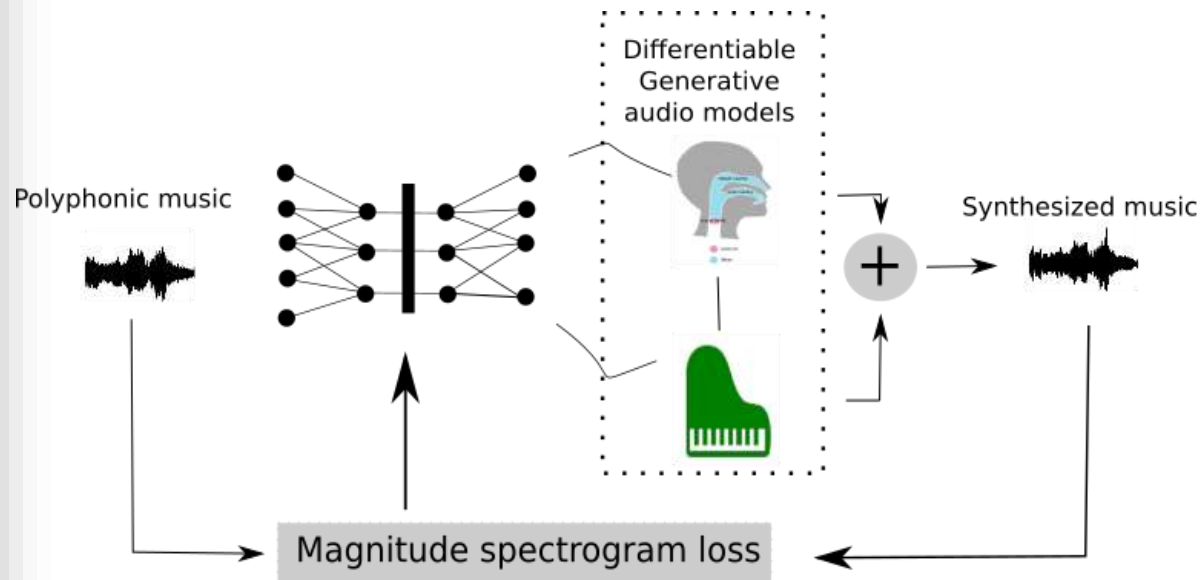
... by integrating our prior knowledge about the nature of the processed data.

Knowledge about « how the sound is produced » (e.g. sound production models)



Singing voice as a source / filter model :

- source = vibration of vocal folds
- Filter = resonances of vocal/nasal cavities



Exploiting knowledge for model-based deep music generation

Towards model-based deep learning

... by integrating our prior knowledge about the nature of the processed data.

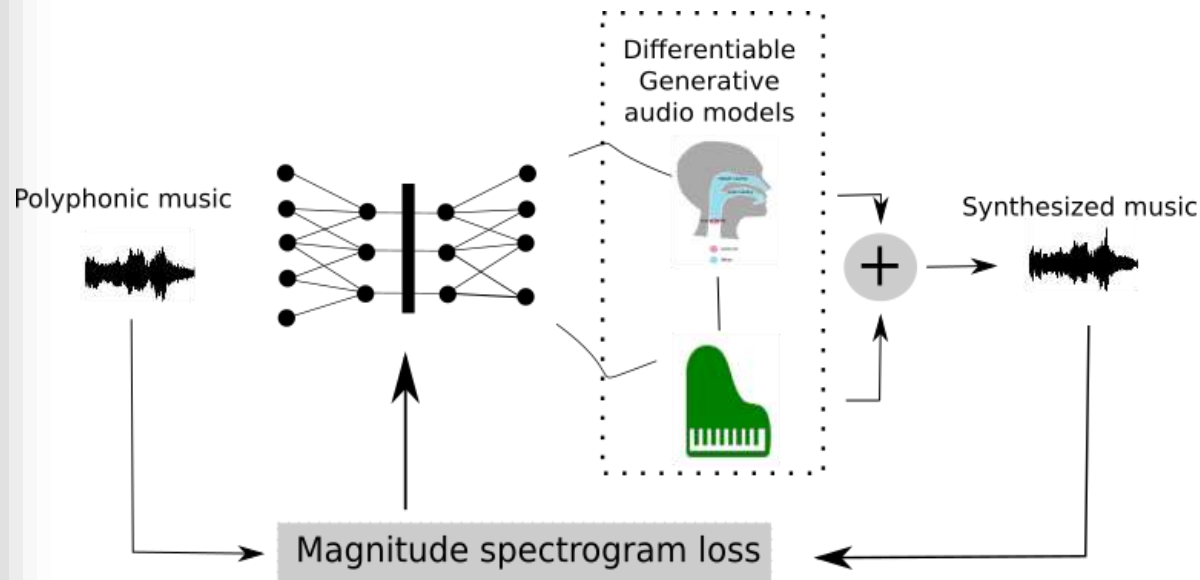
Exploiting knowledge for model-based deep music generation

Knowledge about « how the sound is produced » (e.g. sound production models)



Singing voice as a source / filter model :

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A new paradigm

- Model is at the « core » of neural architecture
- Source separation **by synthesis** (*no interference from other sources*)
- Learning only from the polyphonic recording (*no need of the true individual tracks*)

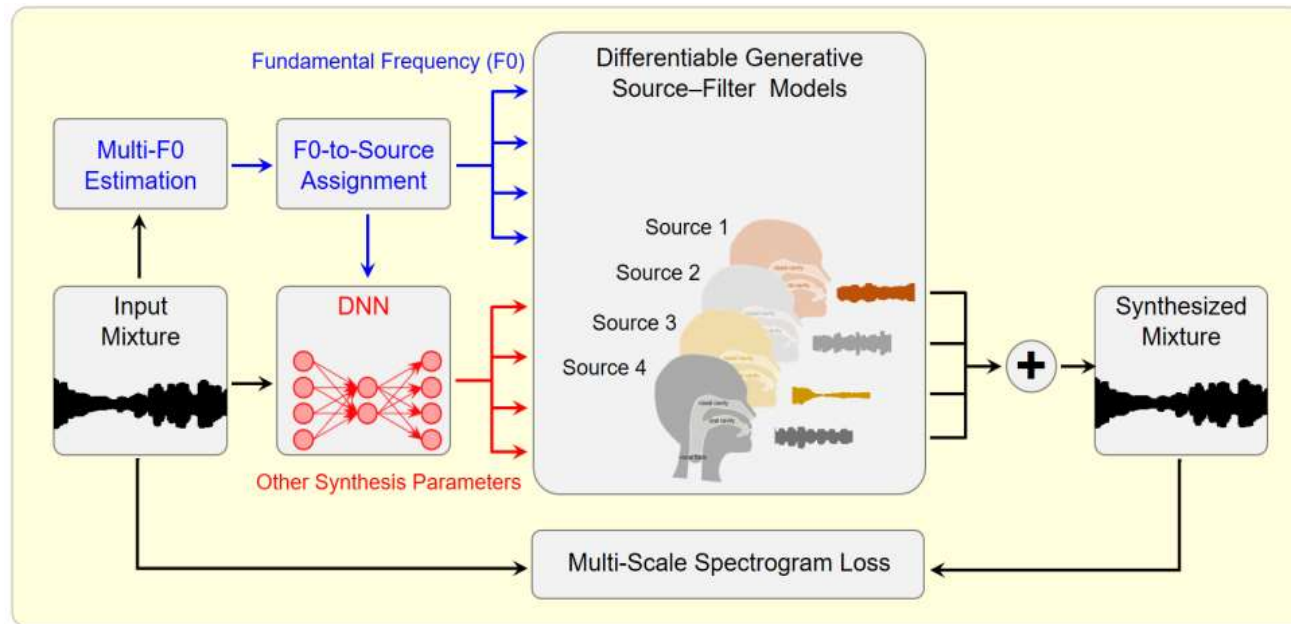
Novel sound transformation capabilities:

- Timbre/melody of the voice,
- Lyrics, translation
- Re-harmonization

Towards model-based deep learning

... by integrating our prior knowledge about the nature of the processed data.

- An example for unsupervised singing voice separation



Highlights

- Unsupervised :
 - Learning only from the polyphonic recording (*no need of the true individual tracks*)
- Homogeneous sources :
 - All sources have similar acoustic properties

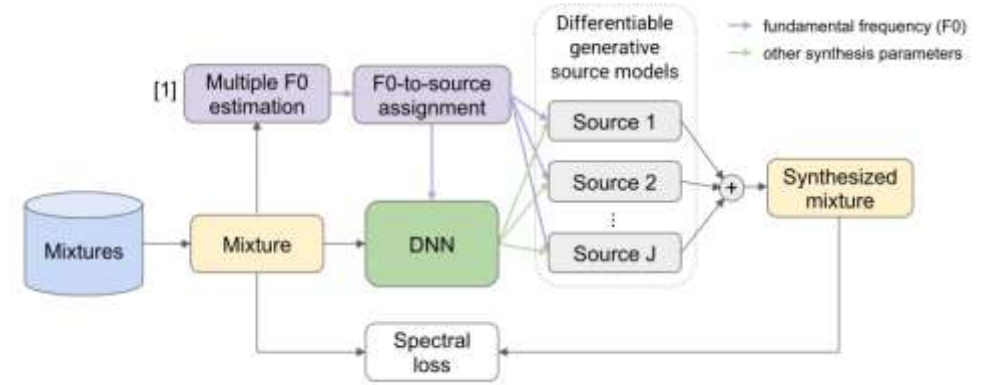
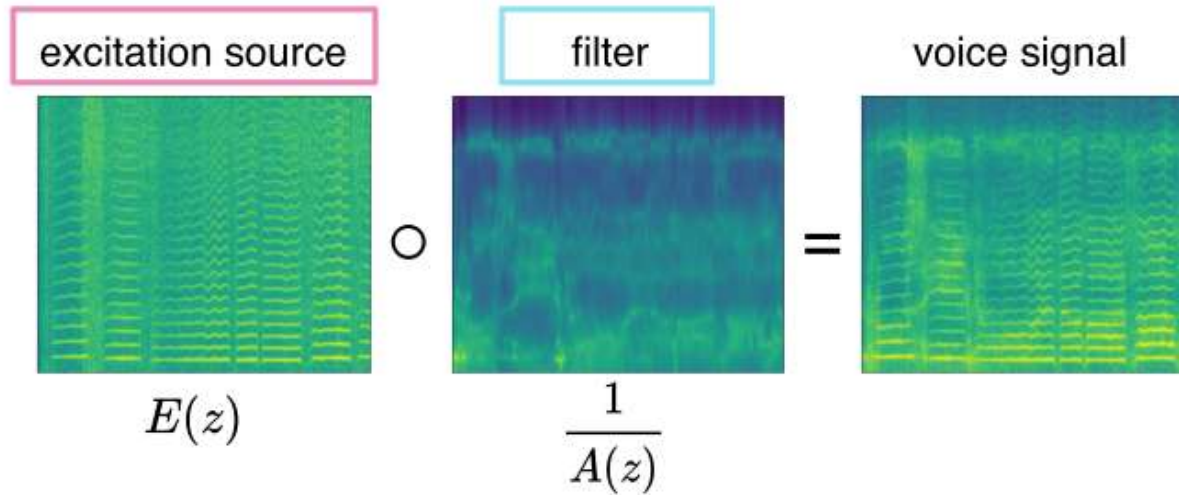


Parametric source models

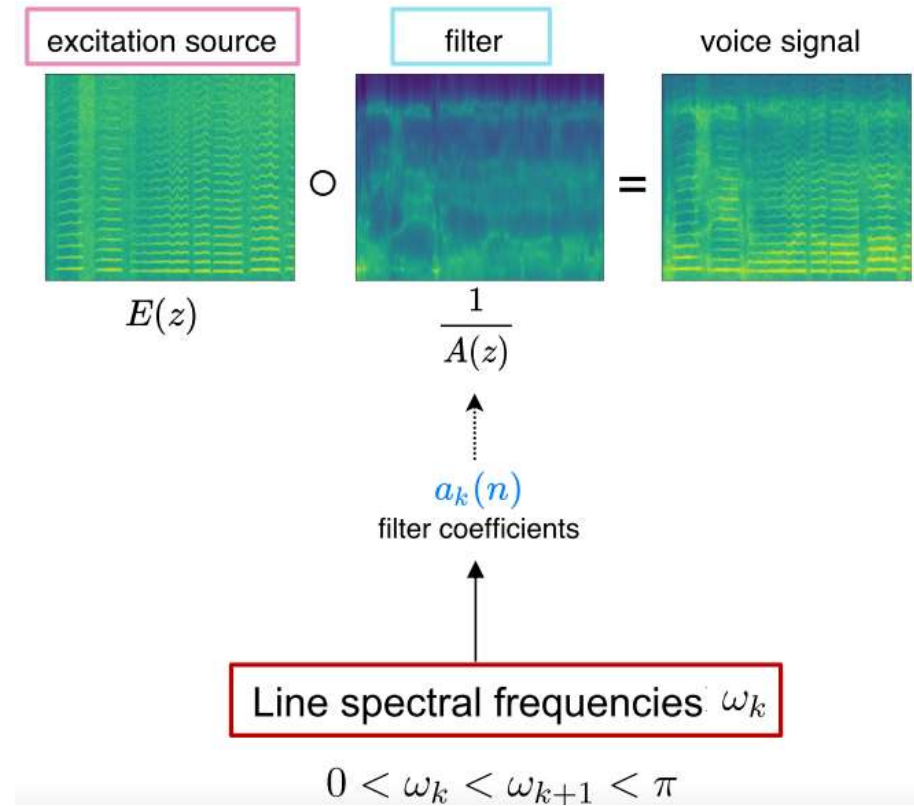
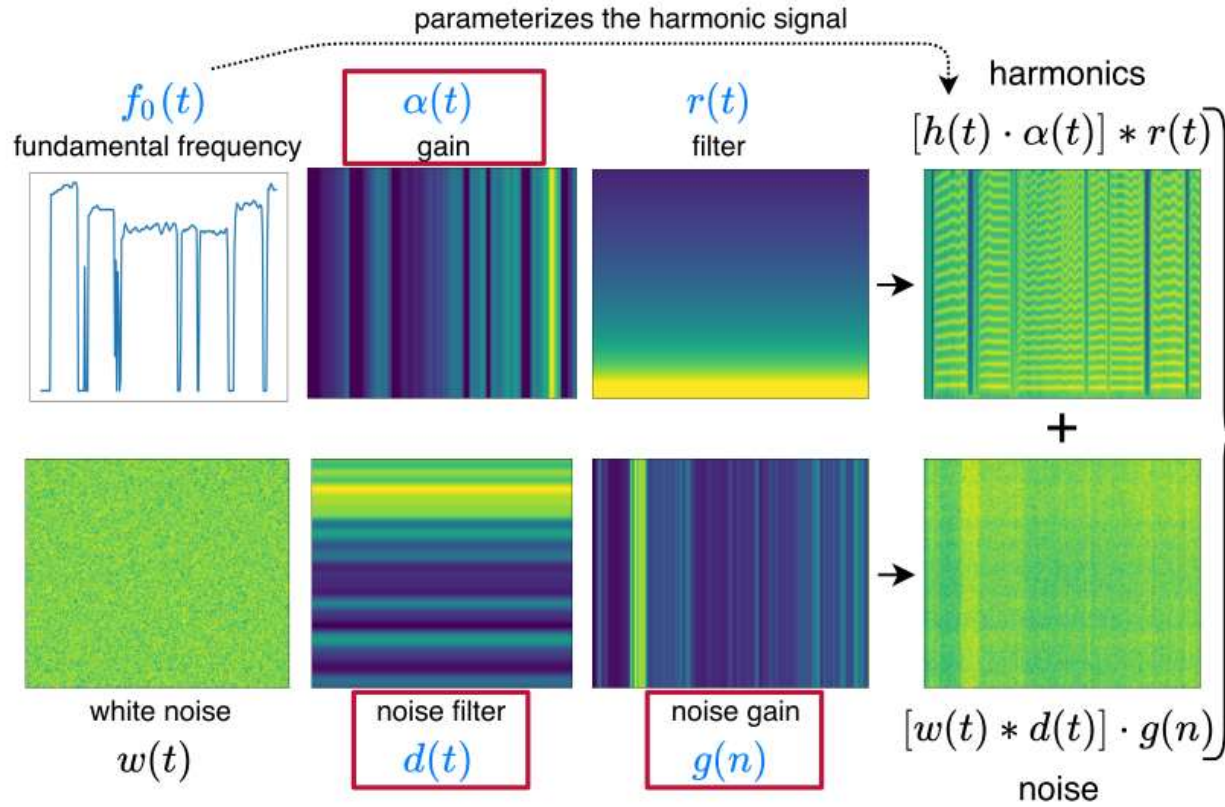
Singing voice as a source / filter model :



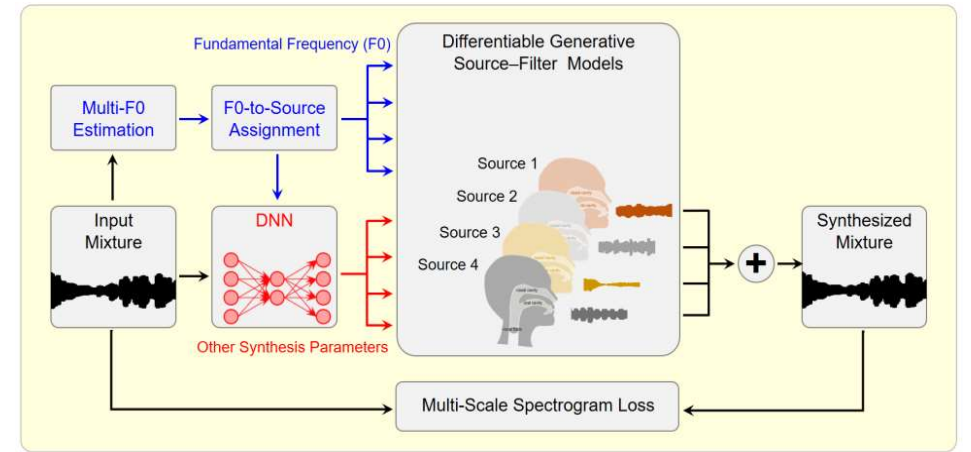
- source = vibration of vocal folds
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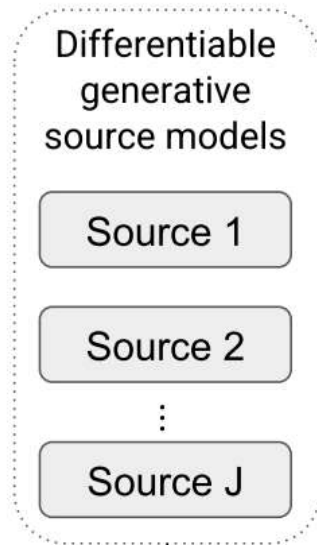
Parametric source models



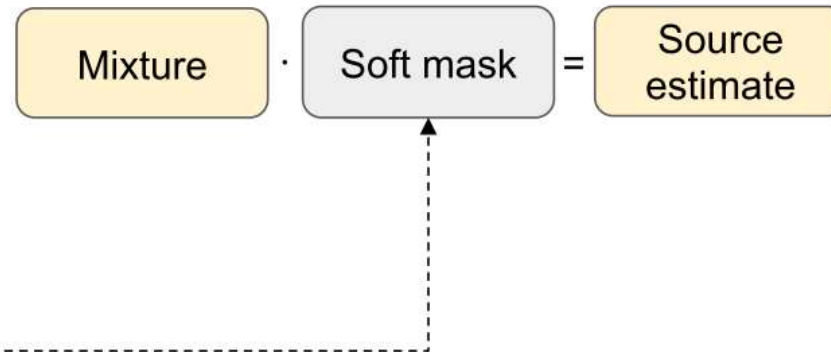
Synthesis or filtering



Synthesis

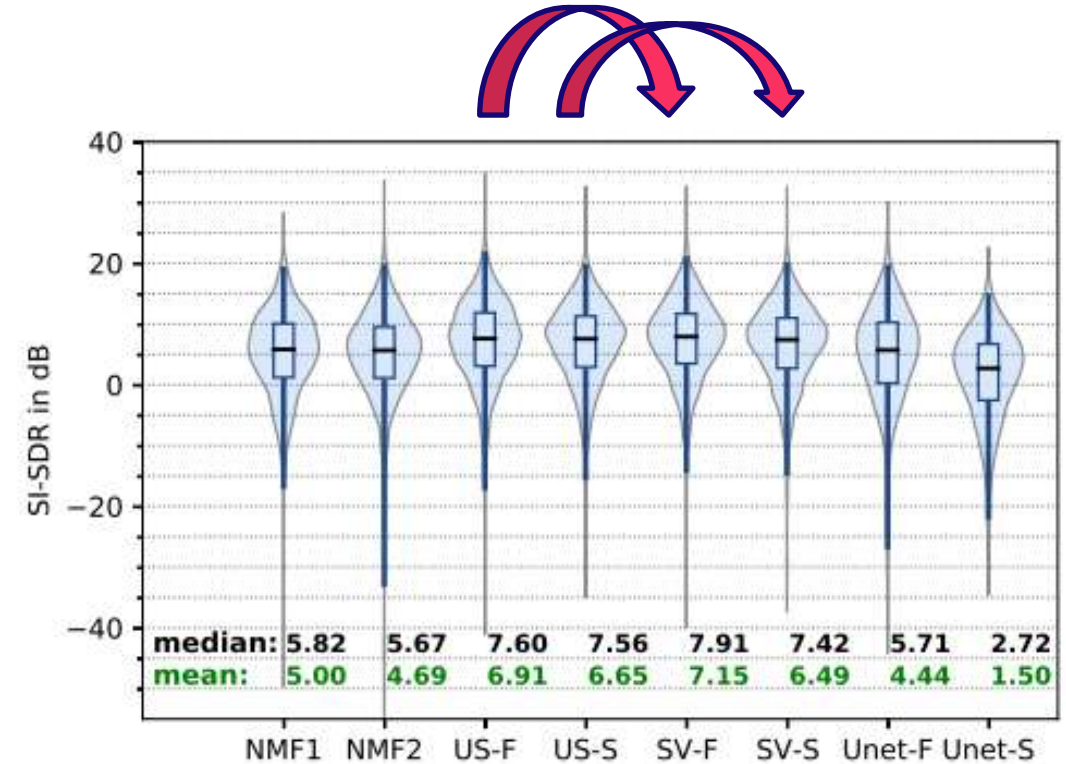


Filtering



Some results

- Unsupervised (US) \approx supervised (SV)



(b) $J = 4$ sources



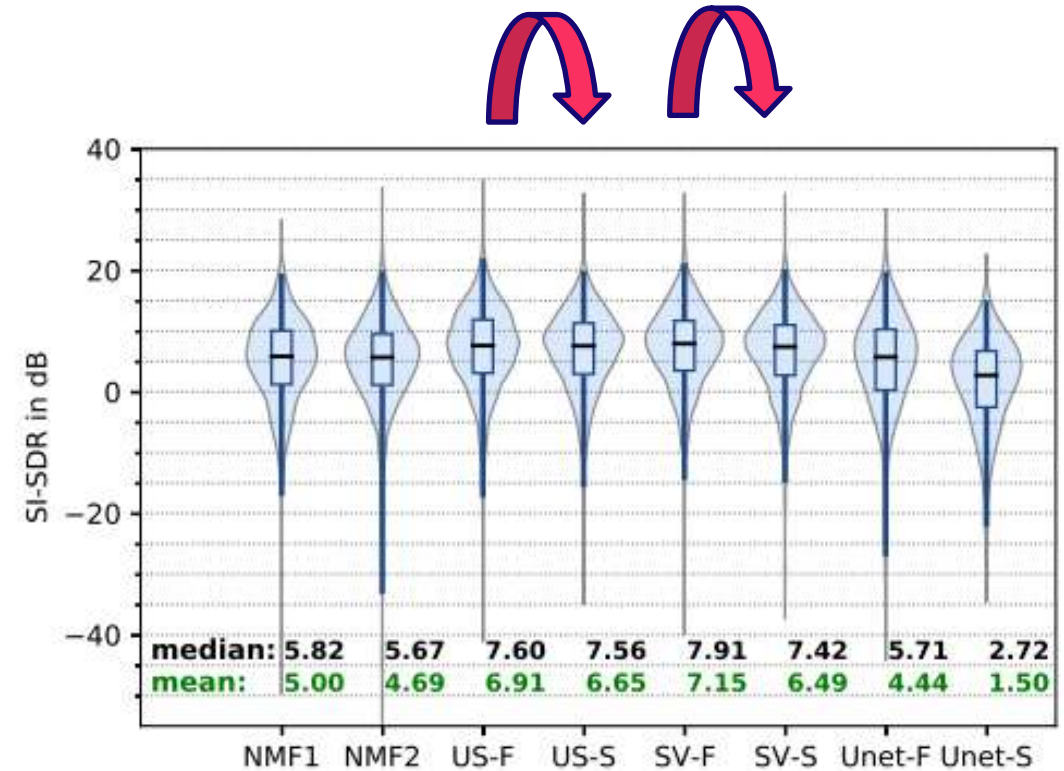
NMF1: S. Ewert and M. Müller, "Using score-informed constraints for NMF-based source separation," in Proc. IEEE Int. Conf. on Acoustics, Speech and Signal Processing. IEEE, 2012, pp. 129–132.

NMF2: J.-L. Durrieu, B. David, and G. Richard, "A musically motivated mid-level representation for pitch estimation and musical audio source separation," IEEE J. Selected Topics in Signal Processing, vol. 5, no. 6, pp. 1180–1191, 2011.

UNET: D. Petermann, P. Chandna, H. Cuesta, J. Bonada, and E. Gomez, "Deep learning based source separation applied to choir ensembles," in Proc. Int. Soc. Music Inf. Retrieval Conf., 2020, pp. 733–739.

Some results

- Unsupervised (US) \approx supervised (SV)
- Almost no drop of performances when using only 3% of the training data (US-F vs. US-S and SV-F vs. SV-S)



(b) $J = 4$ sources



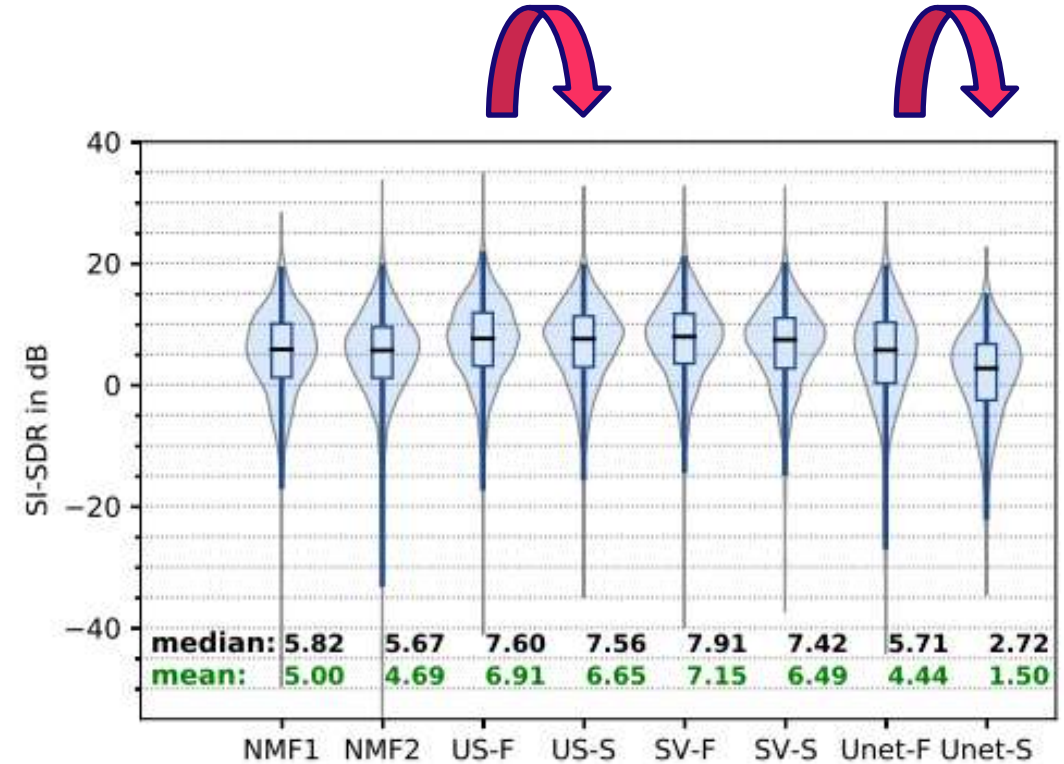
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Some results

- Unsupervised (US) \approx supervised (SV)
- Almost no drop of performances when using only 3% of the training data (US-F vs. US-S and SV-F vs. SV-S)
- ..much larger drop of performances of the supervised baseline model (Unet)



(b) $J = 4$ sources



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A short audio demo and some take aways

- **A short demo at**
- <https://schufo.github.io/umss/>
 - Or [local link](#)
- **Some take aways**
 - Only a small amount of data needed
 - Filtering the mixture better than synthesis
 - Differentiable stable all-pole filter
 - Parameterization of the mixture is provided
 - Extension possible to a fully end-to-end approach [1]

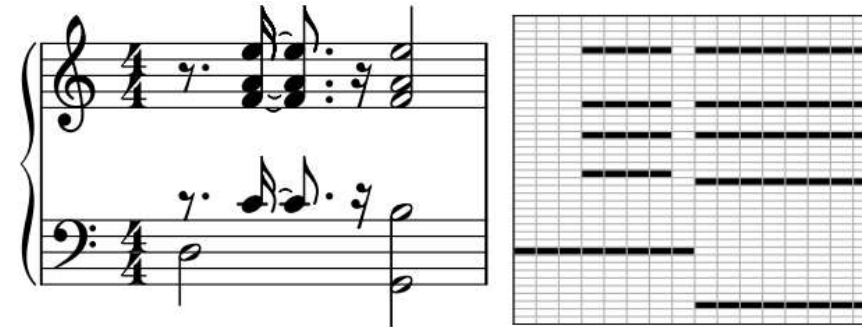


Symbolic music generation with transformers

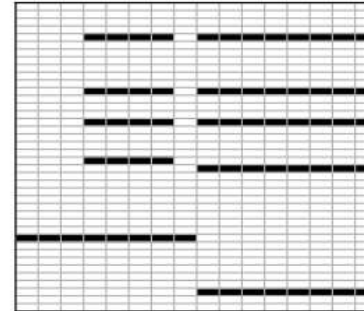
Symbolic music generation with transformers

- Symbolic music
 - Input: Tokens (text) of pianoroll

Music score



MIDI representation (or piano roll)



```
NoteOn(50) TimeShift(9) NoteOn(60) NoteOn(65)
NoteOn(69) NoteOn(76) TimeShift(12) NoteOff(60)
NoteOff(65) NoteOff(69) NoteOff(76) TimeShift(3)
NoteOff(50) NoteOn(43) NoteOn(59) NoteOn(65)
NoteOn(69) NoteOn(76) TimeShift(24) NoteOff(All)
```

Representation as sequence of tokens

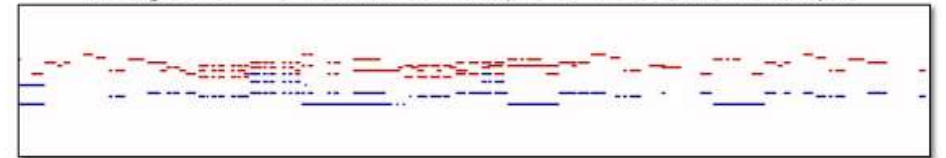
Symbolic music generation with transformers

- **Data-driven Symbolic Music Generation is hard!**

- ✓ Inconsistency in melody and rhythm, absence of multi-scale structures found in real music [1]
- ✓ Practical limitations: limited dataset sizes (compared to, e.g., language, vision) feeds into limited model sizes



ImageNet - 14 million samples, 65k dims/sample



LakhMIDI - 600k samples, 65k dims/sample

- **Possible solution to do more with less: hybrid deep models**

- ✓ Add knowledge about musical structure to data-driven models

Symbolic music generation with transformers

- **Already many possibilities to exploit musical structure ...**
 - Long line of research to include structure in music generation systems [1]
 - Particularly for Transformers:
 - Tokens for musical structures [2-4]
 - Positional Encoding using musical structure [5,6]
- **... But can we improve how Transformers represent and use structural information?**

[1] Bhandari & Colton, "Motifs, Phrases, and Beyond: The Modelling of Structure in Symbolic Music Generation", EvoMUSART, 2024.

[2] Ren, et al., "PopMAG: Pop music accompaniment generation", ACM MM, 2020.

[3] Huang & Yang, "Pop Music Transformer: Beat-based modelling and generation of expressive pop piano compositions", ACM MM, 2020.

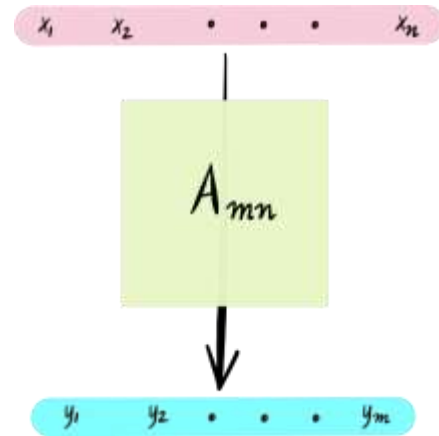
[4] Hsiao, et al., "Compound word transformer: Learning to compose full-song music over dynamic directed hypergraphs", AAAI, 2021.

[5] Liu, et al., "Symphony Generation with permutation invariant language model", arXiv, 2022.

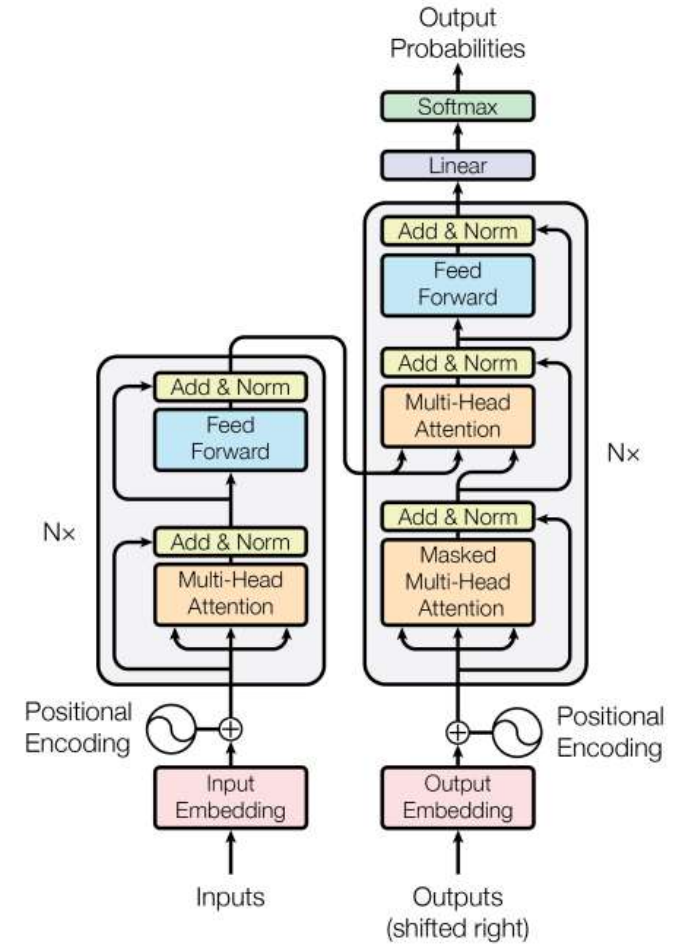
[6] Guo, Kang & Herremans, "A domain-knowledge-inspired music embedding space and a novel attention mechanism for symbolic music modeling", AAAI, 2022.

Symbolic Music Generation

- For example with transformers :
 - Attention: Invariance to temporal order of inputs



- **Role of the PE:** to provide the information about which element of the input sequence comes in which order.



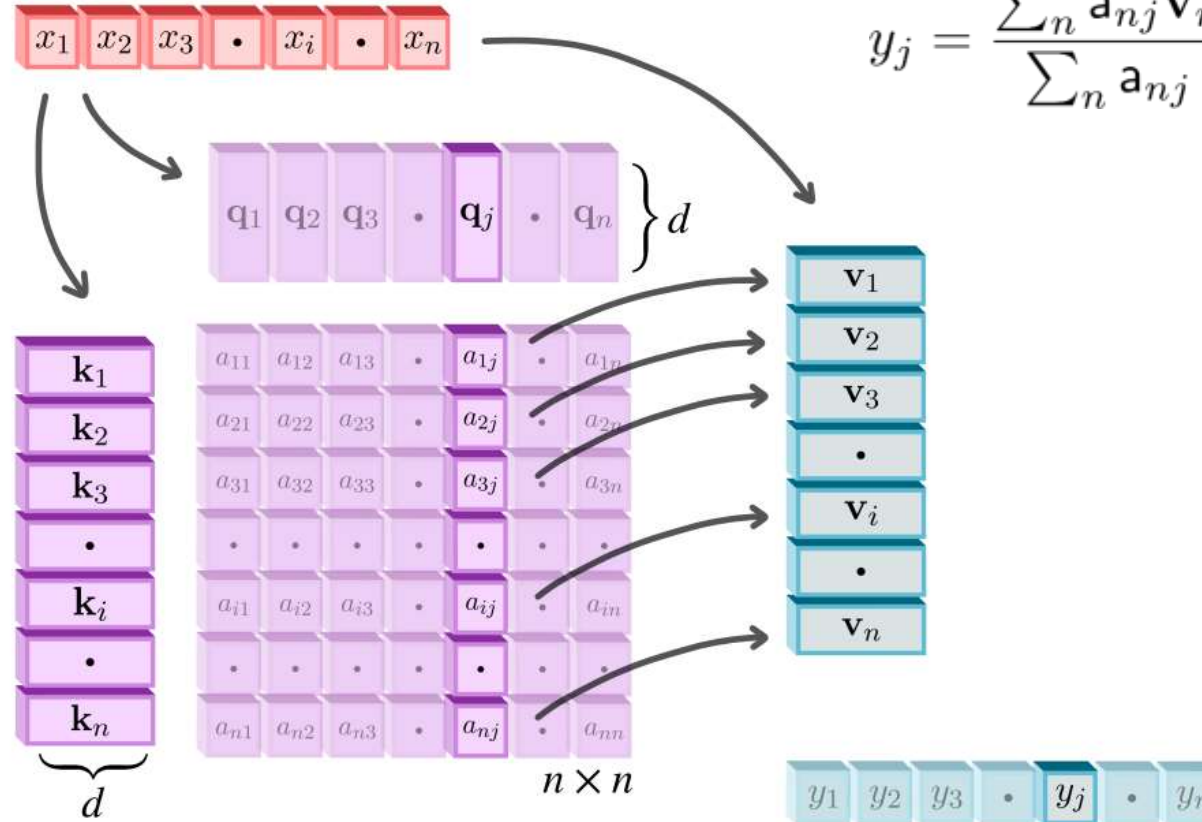
Towards exploiting « musical structured informed » Position Encoding (PE)



Symbolic Music Generation

- Attention

Exploiting knowledge for model-based deep music generation



$$y_j = \frac{\sum_n a_{nj} \mathbf{v}_n}{\sum_n a_{nj}} \text{ with } a_{nj} = \exp\left(\frac{a_{nj}}{\sqrt{D}}\right)$$

$$a_{nj} = \mathbf{q}_j \mathbf{k}_n^\top$$



Symbolic Music Generation

- Classic Positional Encoding

Absolute positional encoding (APE)

$$x_1 \ x_2 \ x_3 \ \dots \ x_i \ \dots \ x_n + p_1 \ p_2 \ p_3 \ \dots \ p_i \ \dots \ p_n \quad p_i = g(i)$$

Relative positional encoding (RPE)

$$f \left(\begin{matrix} a_{11} & a_{12} & a_{13} & \dots & a_{1j} & \dots & a_{1n} \\ a_{21} & a_{22} & a_{23} & \dots & a_{2j} & \dots & a_{2n} \\ a_{31} & a_{32} & a_{33} & \dots & a_{3j} & \dots & a_{3n} \\ \dots & \dots & \dots & \dots & \dots & \dots & \dots \\ a_{i1} & a_{i2} & a_{i3} & \dots & a_{ij} & \dots & a_{in} \\ \dots & \dots & \dots & \dots & \dots & \dots & \dots \\ a_{n1} & a_{n2} & a_{n3} & \dots & a_{nj} & \dots & a_{nn} \end{matrix} , \begin{matrix} p_{11} & p_{12} & p_{13} & \dots & p_{1j} & \dots & p_{1n} \\ p_{21} & p_{22} & p_{23} & \dots & p_{2j} & \dots & p_{2n} \\ p_{31} & p_{32} & p_{33} & \dots & p_{3j} & \dots & p_{3n} \\ \dots & \dots & \dots & \dots & \dots & \dots & \dots \\ p_{i1} & p_{i2} & p_{i3} & \dots & p_{ij} & \dots & p_{in} \\ \dots & \dots & \dots & \dots & \dots & \dots & \dots \\ p_{n1} & p_{n2} & p_{n3} & \dots & p_{nj} & \dots & p_{nn} \end{matrix} \right) \quad p_{ij} = g(i - j)$$

- Structure Positional Encoding

Absolute positional encoding (APE)

$$x_1 \ x_2 \ x_3 \ \dots \ x_i \ \dots \ x_n + p_1 \ p_2 \ p_3 \ \dots \ p_i \ \dots \ p_n \quad p_i = g(s_i)$$

Relative positional encoding (RPE)

$$f \left(\begin{matrix} a_{11} & a_{12} & a_{13} & \dots & a_{1j} & \dots & a_{1n} \\ a_{21} & a_{22} & a_{23} & \dots & a_{2j} & \dots & a_{2n} \\ a_{31} & a_{32} & a_{33} & \dots & a_{3j} & \dots & a_{3n} \\ \dots & \dots & \dots & \dots & \dots & \dots & \dots \\ a_{i1} & a_{i2} & a_{i3} & \dots & a_{ij} & \dots & a_{in} \\ \dots & \dots & \dots & \dots & \dots & \dots & \dots \\ a_{n1} & a_{n2} & a_{n3} & \dots & a_{nj} & \dots & a_{nn} \end{matrix} , \begin{matrix} p_{11} & p_{12} & p_{13} & \dots & p_{1j} & \dots & p_{1n} \\ p_{21} & p_{22} & p_{23} & \dots & p_{2j} & \dots & p_{2n} \\ p_{31} & p_{32} & p_{33} & \dots & p_{3j} & \dots & p_{3n} \\ \dots & \dots & \dots & \dots & \dots & \dots & \dots \\ p_{i1} & p_{i2} & p_{i3} & \dots & p_{ij} & \dots & p_{in} \\ \dots & \dots & \dots & \dots & \dots & \dots & \dots \\ p_{n1} & p_{n2} & p_{n3} & \dots & p_{nj} & \dots & p_{nn} \end{matrix} \right) \quad p_{ij} = g(s_i, s_j)$$

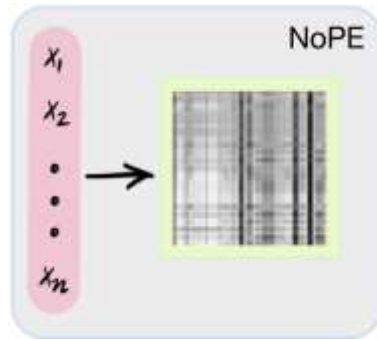
Exploiting knowledge for model-based deep music generation



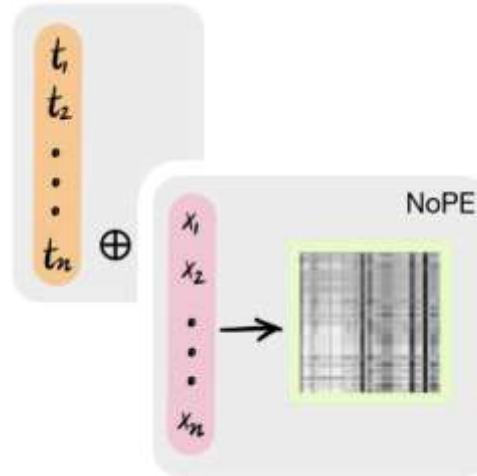
Symbolic Music Generation

« musical structure-informed » Position Encoding (PE)

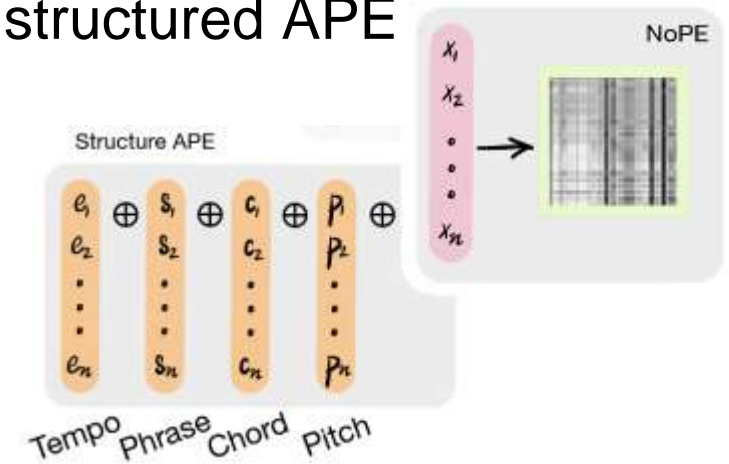
- From No Positional Encoding



...to Absolute PE



...to structured APE



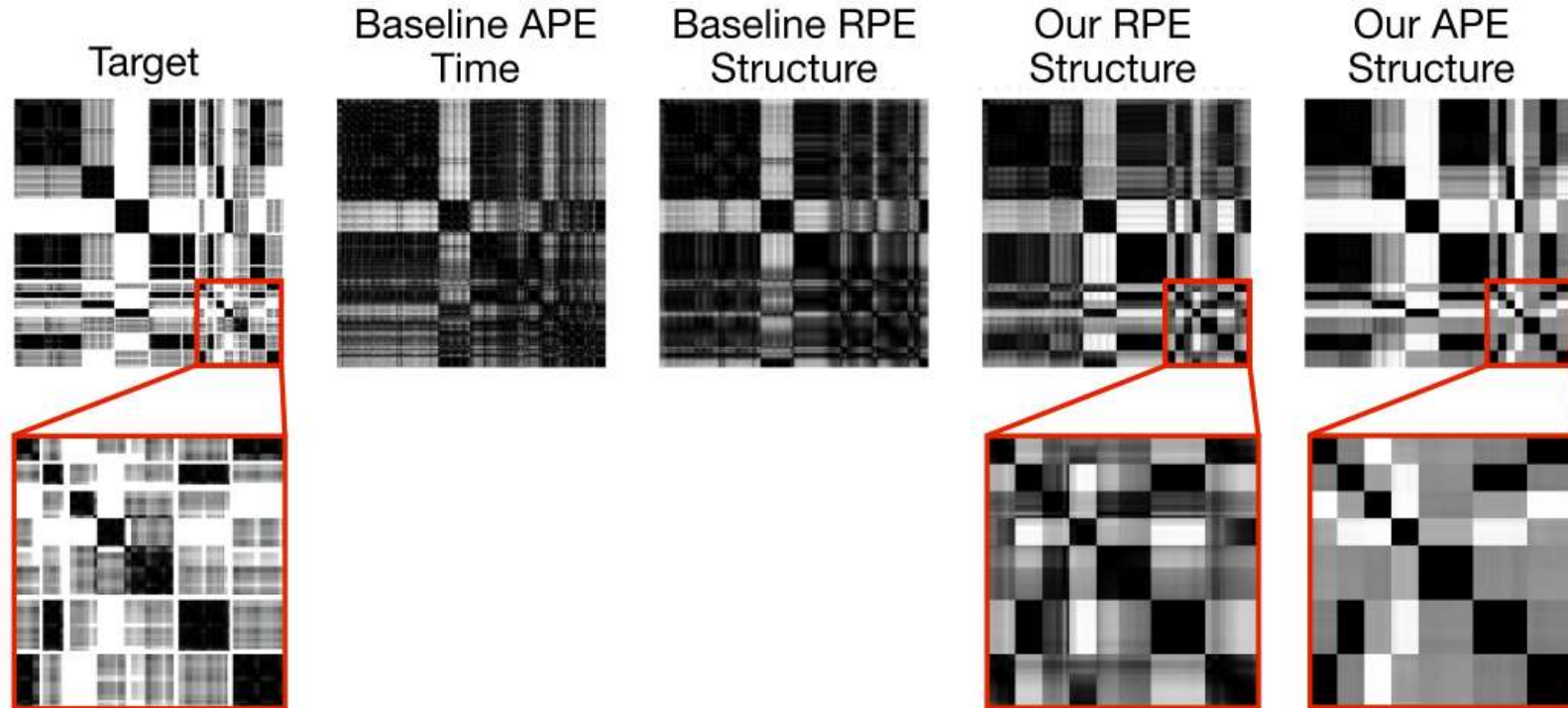
Results show that better music generation can be achieved by using knowledge about musical structure in data-driven Transformers through Positional Encoding



Accompaniment generation from melody tracks

Illustration

- Our structure-informed positional encoding captures large-scale and small-scale structures :
 - ✓ Self-similarity matrices of chroma profiles (*chroma is a feature representation capturing chords information*)



Extension for linear complexity structure-informed PE

- **Exploiting a kernelized form of attention [1,2]**

$$a_{mn} = \mathcal{K}(\mathbf{q}_m, \mathbf{k}_n) = \mathbb{E} \left[\phi(\mathbf{q}_m) \phi(\mathbf{k}_n)^\top \right]$$

- With multiple instantiations, ϕ captures, on average, the relationship between \mathbf{q}_m and \mathbf{k}_n

➡ leads to linear-complexity Transformers.

- Applicable for Absolute Position Encoding

- **Stochastic Position Encoding [3]** => Applicable to Relative PE with linear complexity

- Key ideas:

- Express the Attention matrix with position kernels $\mathbf{A} = \exp \left(\left[\sum_{d=1}^D q_{md} \mathcal{P}_d(m, n) k_{nd} \right]_{mn} / \sqrt{D} \right)$
- Express the position kernel as a covariance matrix $(\forall \mathcal{M}, \mathcal{N}) (\forall m, n) \mathcal{P}_d(m, n) = \mathbb{E} [\overline{Q}_d(m) \overline{K}_d(n)]$

- **Extension to structure-informed stochastic Position Encoding [4]**

[1] Y.-H. H. Tsai & al. Transformer Dissection: An Unified Understanding for Transformer's Attention via the Lens of Kernel," EMNLP, 2019

[2] K. M. Choromanski, & al. Rethinking Attention with Performers," ICML, 2021

[3] A. Liutkus & al. Relative Positional Encoding for Transformers with Linear Complexity," ICML, 2021

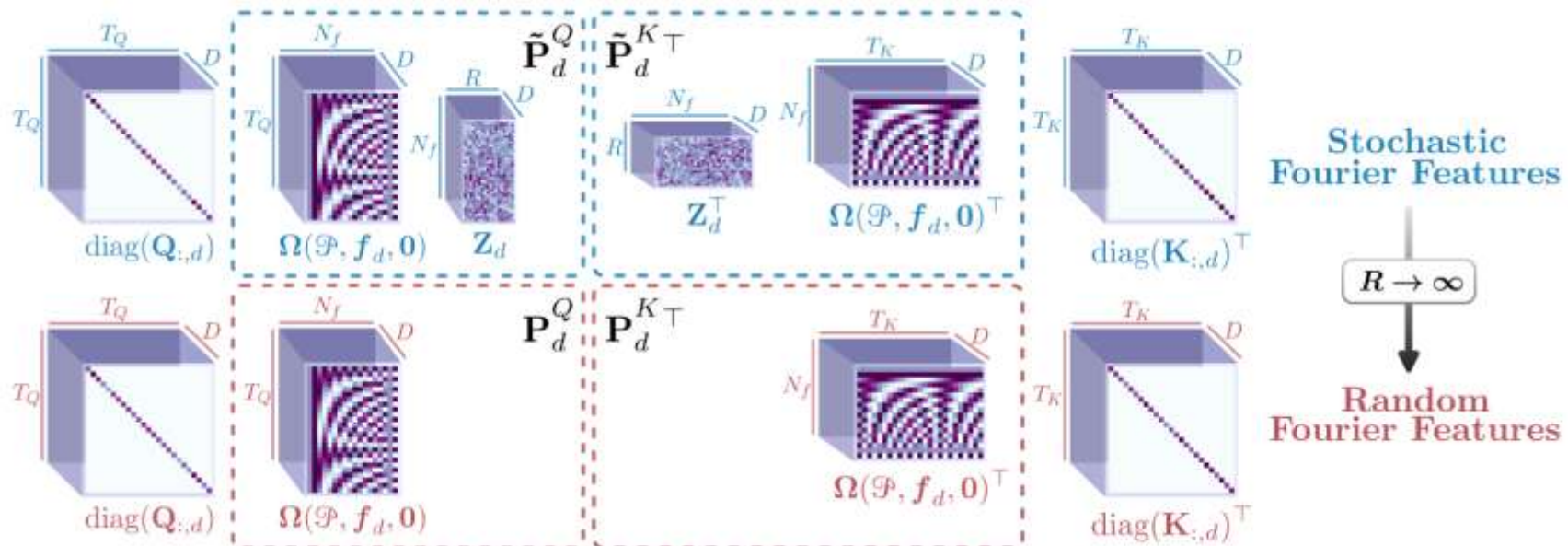
[4] M. Agarwal & al, F-StrIPE: Fast Structure-Informed Positional Encoding for Symbolic Music Generation, ICASSP 2025.

Extension for linear complexity structure-informed PE

- **F-StrIPE: Structure informed stochastic Position Encoding [4]**

The positional matrix \mathbf{P}_d captures the relationship between pairs (m, n) of timesteps from the positional index sequences $\mathcal{P}_Q = \{1, \dots, m, \dots, T_Q\}$ and $\mathcal{P}_K = \{1, \dots, n, \dots, T_K\}$.

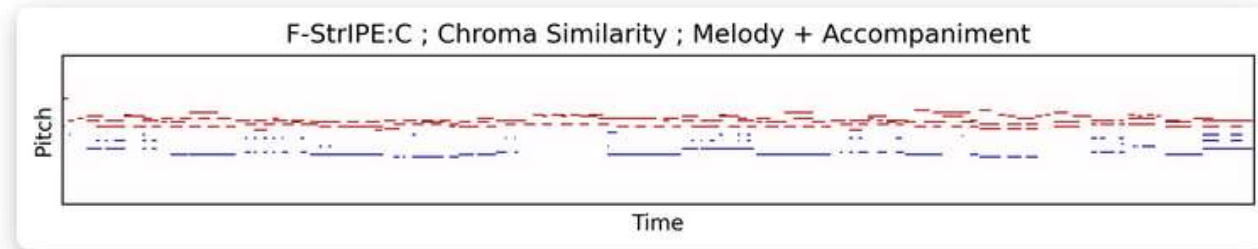
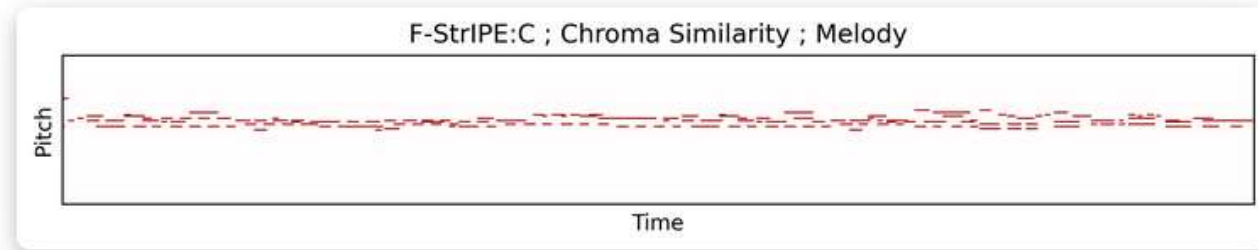
F-StrIPE: exploiting structure-aware positional indices $p_i = \mathbf{s}(i)$ instead of classic time indices $p_i = i$



[4] M. Agarwal & al, F-StrIPE: Fast Structure-Informed Positional Encoding for Symbolic Music Generation, ICASSP 2025.

F-StrIPE: Structure informed stochastic Position Encoding [4]

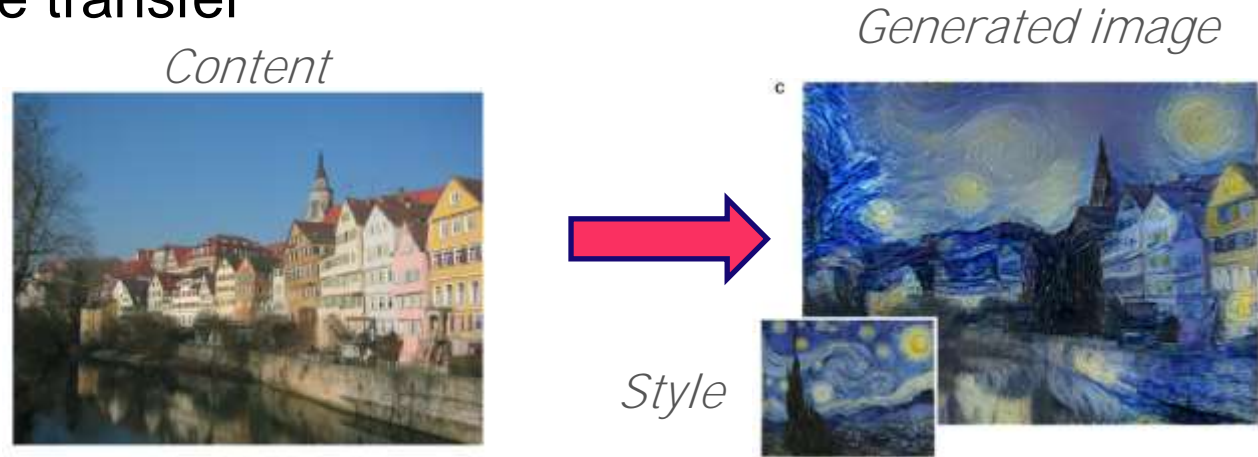
- Demo page at : bit.ly/faststructurepe
- Best example for « Chroma similarity » metric (*training 16 bars of melody – generation 16 bars of accompaniment*)



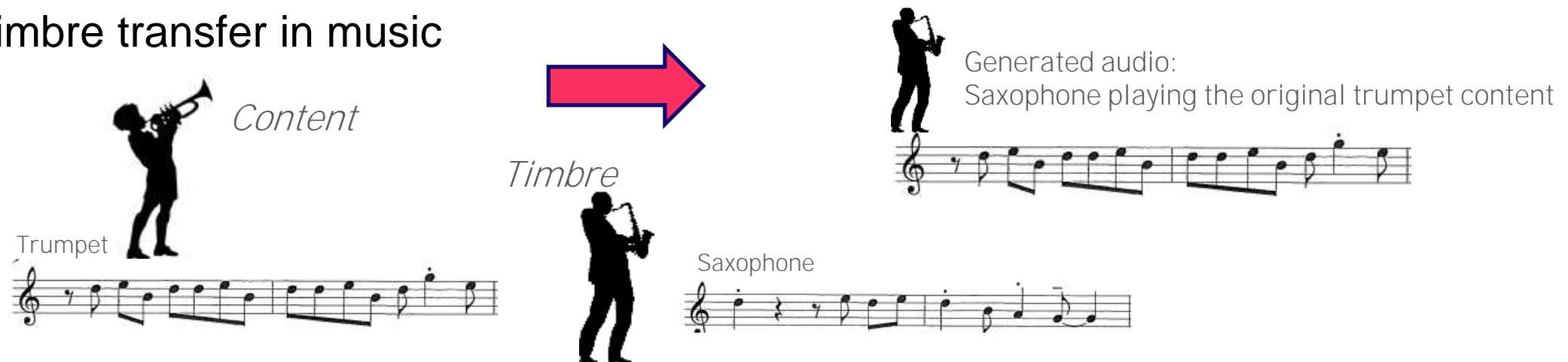
Musical Timbre transfer

Timbre transfer : a specific application of style transfer to music

- Image style transfer



- Timbre transfer in music



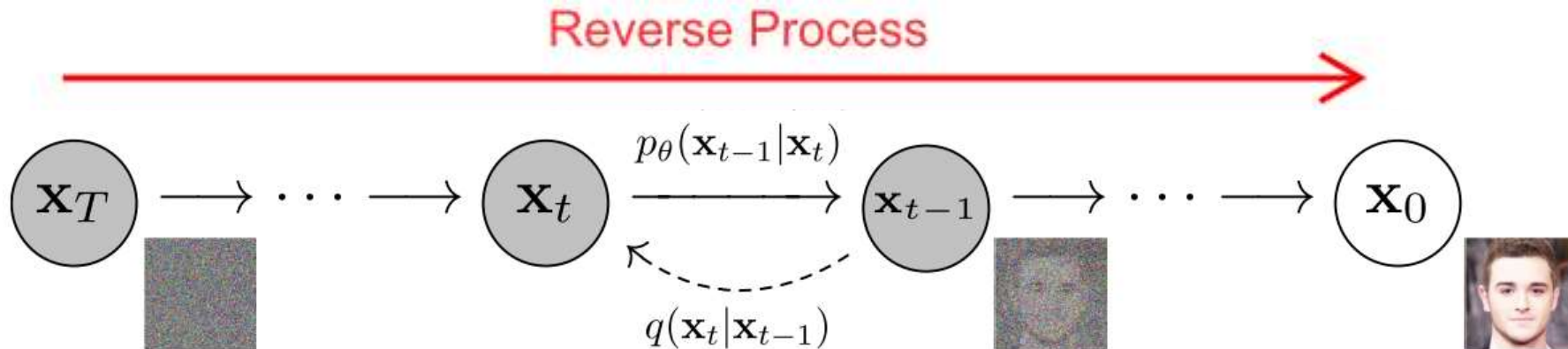
WavTransfer: A Flexible End-to-end Multi-instrument Timbre Transfer with Diffusion

- **Timbre Transfer:**
 - Essential for distinguishing sounds with the same pitch and loudness
 - Modifies the tonal quality while preserving pitch and structure
- *Common models* : Need for separate models for each pair of instrument for timbre transfer
- **WaveTransfer[1]:**
 - Works for audio mixtures and individual instruments
 - Generates audio waveforms directly
 - Operates at multiple sampling frequencies 16 kHz and 44.1 kHz

Background

Denoising diffusion probabilistic models (DDPMs)

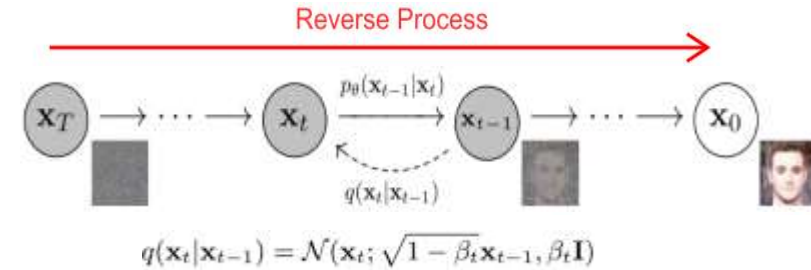
- Characterising a data distribution by gradually introducing noise into samples for T steps and then learning the process of reversing it



$$q(\mathbf{x}_t|\mathbf{x}_{t-1}) = \mathcal{N}(\mathbf{x}_t; \sqrt{1 - \beta_t}\mathbf{x}_{t-1}, \beta_t\mathbf{I})$$

Background

Denoising diffusion probabilistic models (DDPMs)



- $\mathbf{x}_0 \sim q(\mathbf{x}_0)$: an initial sample, $\{\beta_t\}_{t=1}^T$: a noise schedule
- Let $\alpha_t = 1 - \beta_t$ and $\bar{\alpha}_t = \prod_{u=1}^t \alpha_u$. \mathbf{x}_t can be sampled at any arbitrary time step t :

$$\text{Forward Process: } \mathbf{x}_t = \sqrt{\bar{\alpha}_t} \mathbf{x}_0 + \sqrt{1 - \bar{\alpha}_t} \boldsymbol{\varepsilon}, \quad \boldsymbol{\varepsilon} \sim \mathcal{N}(\boldsymbol{\varepsilon}; \mathbf{0}, \mathbf{I}) \quad (1)$$

- The training loss is given by:

$$\mathcal{L}_\theta = \min_{\theta} \mathbb{E} \left[\|\boldsymbol{\varepsilon}_\theta(\mathbf{x}_t, t) - \boldsymbol{\varepsilon}\|_2^2 \right], \quad (2)$$

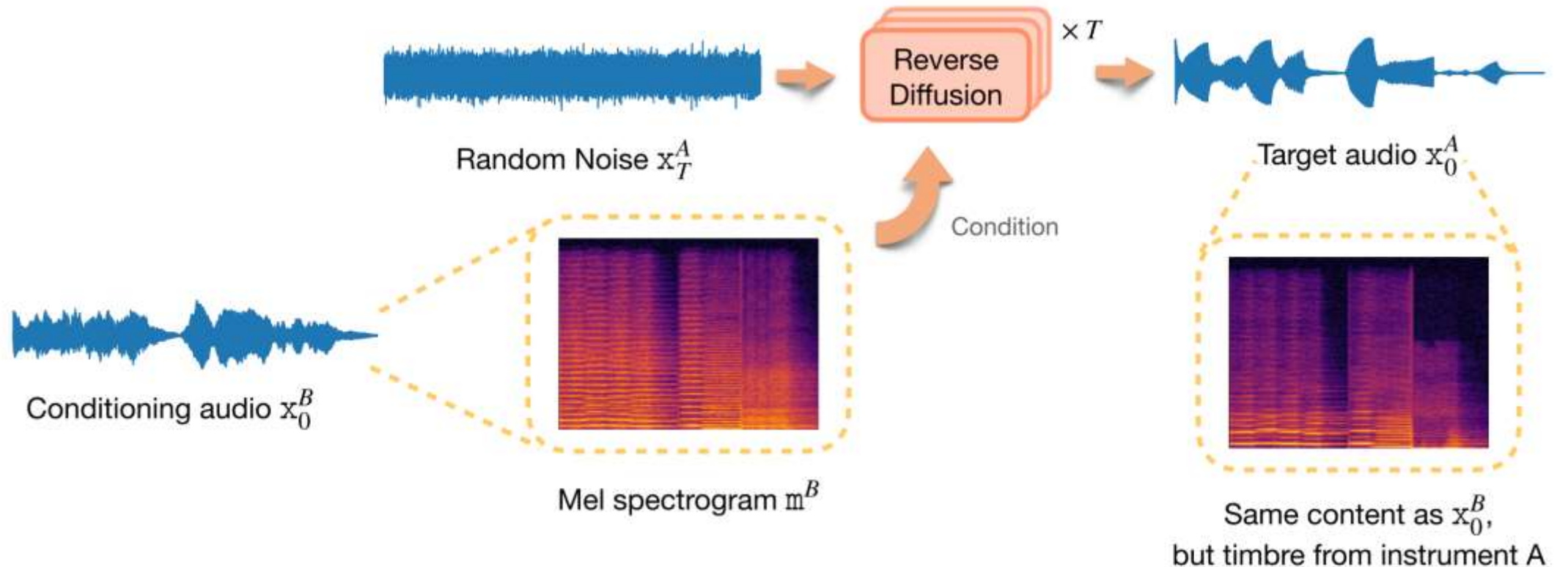
- During inference, we can iteratively sample the data from $\mathbf{x}_T \sim \mathcal{N}(\mathbf{x}_T; \mathbf{0}, \mathbf{I})$ to \mathbf{x}_0 via:

$$\mathbf{x}_{t-1} = \mathcal{N}\left(\mathbf{x}_{t-1}; \frac{1}{\sqrt{\alpha_t}} \left(\mathbf{x}_t - \frac{1 - \alpha_t}{\sqrt{1 - \bar{\alpha}_t}} \boldsymbol{\varepsilon}_\theta(\mathbf{x}_t, t) \right), \sigma_t^2 \mathbf{I} \right), \quad (3)$$

where σ_t is a time dependent constant.

Timbre transfer : principle of WaveTransfer [1]

- Extending Wavgrad [2] for timbre transfer.
- Timbre transfer objective: generate a target audio x_0^A from a random noise x_T^A and conditioning audio x_0^B

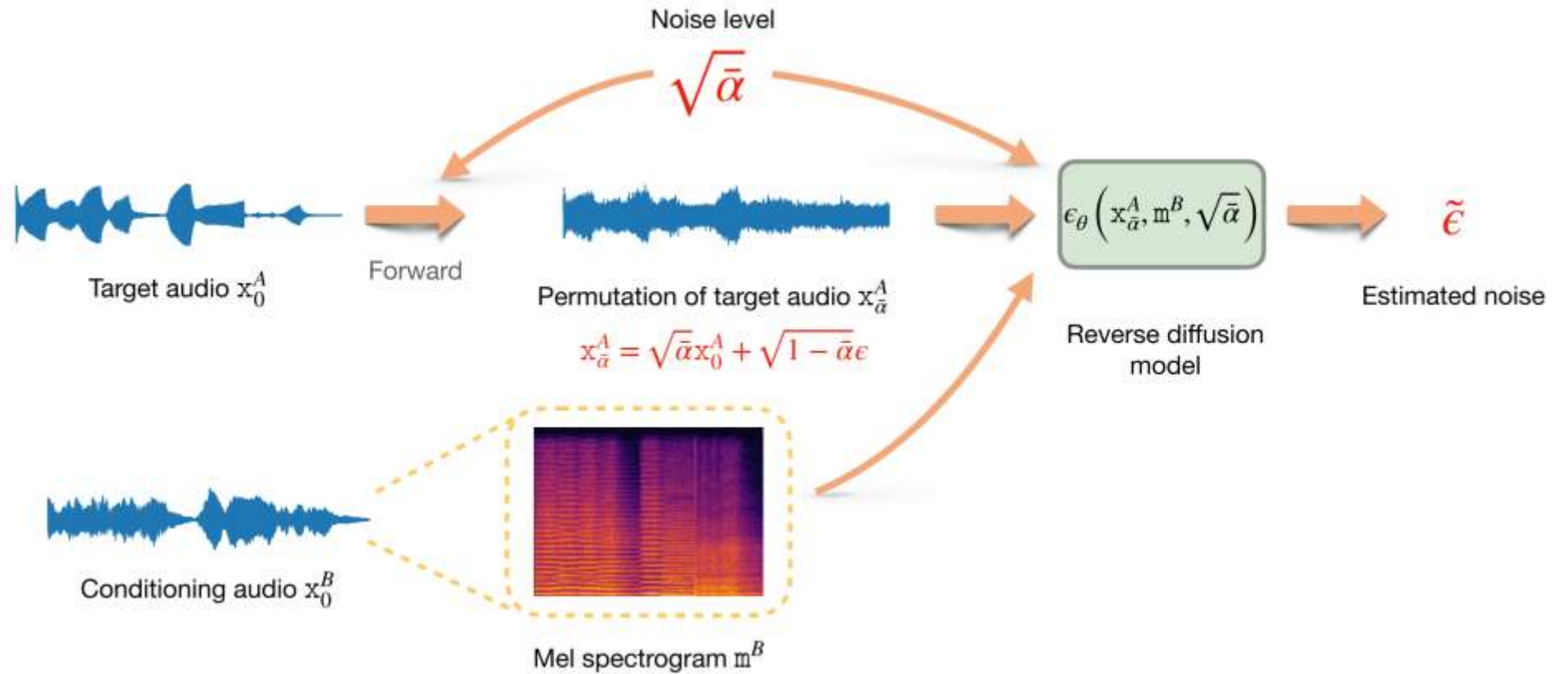


[1] T. Baoueb, X. Bie, G. Richard, WaveTransfer: A Flexible End-to-end Multi-instrument Timbre Transfer with Diffusion, ICASSP 2025

[2] Nanxin Chen, Yu Zhang, Heiga Zen, Ron J Weiss, Mohammad Norouzi, and William Chan, "Wavegrad: Estimating gradients for waveform generation," in Proc. ICLR, 2021

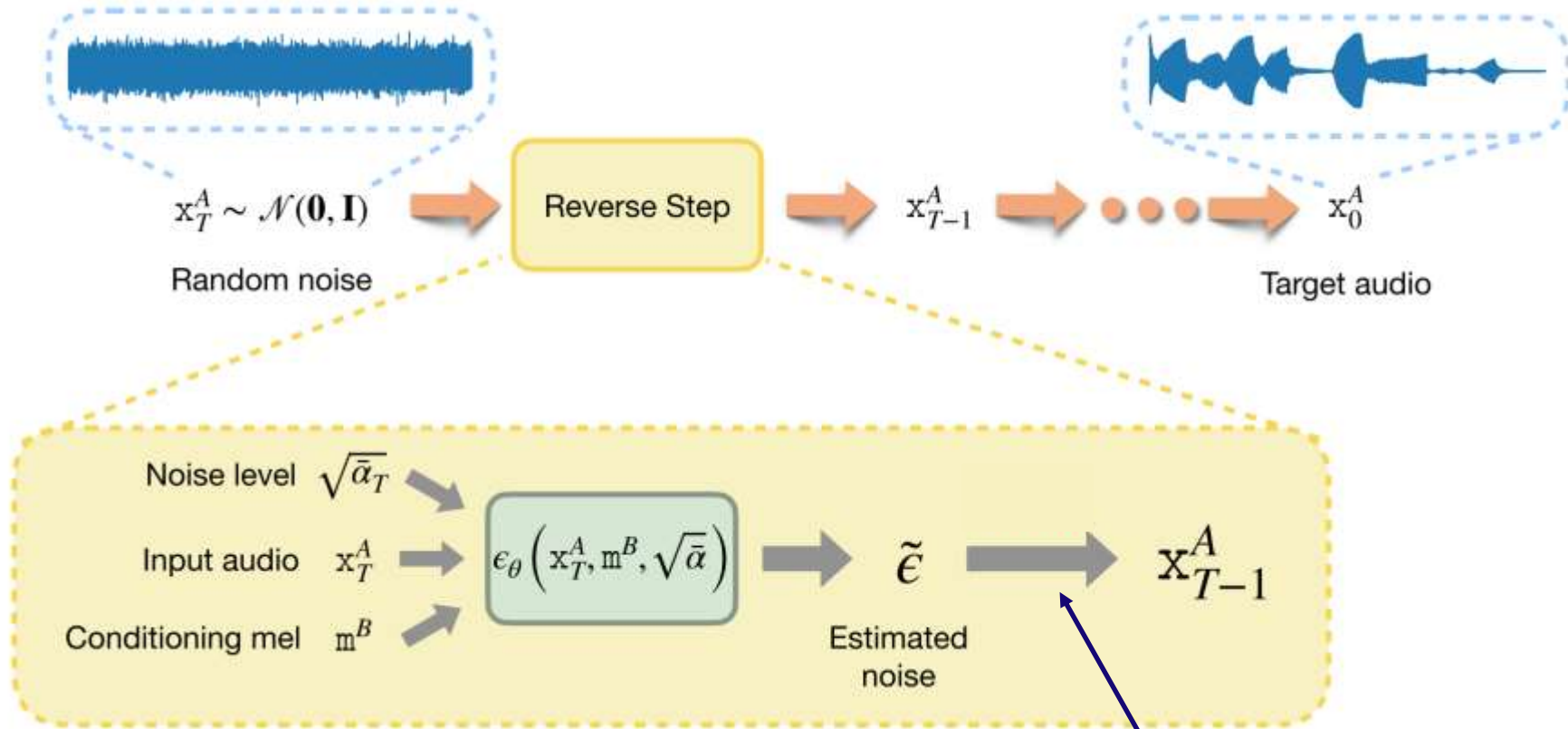
WaveTransfer: Training process

- Supervised Training: Aligned dataset
- \mathbf{x}_0^A and \mathbf{x}_0^B : same content, \neq instruments



$$\mathcal{L}_\theta = \min_{\theta} \mathbb{E} \left[\left\| \epsilon_\theta \left(\sqrt{\bar{\alpha}}\mathbf{x}_0^A + \sqrt{1-\bar{\alpha}}\epsilon, \mathbf{m}^B, \sqrt{\bar{\alpha}} \right) - \epsilon \right\|_1 \right]$$

WaveTransfer: Inference



$$x_{n-1} = \frac{1}{\sqrt{\alpha_n}} \left(x_n - \frac{1 - \alpha_n}{\sqrt{1 - \bar{\alpha}_n}} \epsilon_\theta \left(x_n, m^B, \sqrt{\bar{\alpha}_n} \right) \right) + \sigma_n z,$$

Wavetransfer: Timbre transfer demo

<https://wavetransfer.github.io/>

- Timbre transfer : piano to vibraphone (16 kHz)

Name	Input (ground truth)	Target (ground truth)	Music-STAR	DiffTransfer	WT ¹⁶ _{global} with WG-6	WT ¹⁶ _{global} with BDDM-20
Pirates of Caribbean						

- Mixture of Timbre transfer : piano+strings -> vibraphone + clarinet

Name	Input (ground truth)	Target (ground truth)	Music-STAR	DiffTransfer	WT ¹⁶ _{global} with WG-6	WT ¹⁶ _{global} with BDDM-20	WT ¹⁶ _{mix} with WG-6	WT ¹⁶ _{mix} with BDDM-20
Beethoven								



Only trained on the specific mixtures

WaveTransfer

- **Capabilities of the model**

- Handles timbre transfer for both audio mixtures and individual instruments in one model
- Eliminates the requirement for separate model training for each timbre transfer

- **Current Limitations**

- Relies on an aligned dataset
- Limited instrument diversity in timbre transfer

To conclude

- The potential for hybrid deep learning ...
 - **Interpretability, Controllability, Explainability**
 - Hybrid model becomes controllable by human-understandable parameters
 - New audio capabilities: perceptually meaningful sound transformation
 - **Frugality: gain of several orders of magnitude** in the need of data and model complexity
 - **Towards a more resource efficient and sustainable AI**
 - **Applicable to many audio processing problems**
 - Exploiting room acoustics for Audio dereverberation [1],
 - Exploiting physical/signal models for music synthesis [2],
 - Exploiting “audio class specific” codebooks for audio compression and separation [3]
 - Exploiting key speech attributes for controlled speech synthesis and transformation [4]
 - ...

[1] Louis Bahrman, Mathieu Fontaine, Gaël Richard. A Hybrid Model for Weakly-Supervised Speech Dereverberation. *IEEE ICASSP 2025*, [\(hal-04931672\)](#)

[2] Lenny Renault, Rémi Mignot, Axel Roebel. Differentiable Piano Model for MIDI-to-Audio Performance Synthesis. *Int. Conf.on Digital Audio Effects (DAFx20in22)*, Sep 2022, Vienna,

[3] Xiaoyu Bie, Xubo Liu, Gaël Richard. Learning Source Disentanglement in Neural Audio Codec. *IEEE ICASSP 2025*, [\(hal-04902131\)](#)

[4] Samir Sadok, Simon Leglaive, Laurent Girin, Gaël Richard, Xavier Alameda-Pineda. AnCoGen: Analysis, Control and Generation of Speech with a Masked Autoencoder. *IEEE ICASSP 2025*, [\(hal-04891286\)](#)