

# Denoising with Generative Models

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# Problem + Why is it important ?

- Record high quality audio                      => perfect conditions necessary
- Can we do this in software ?                      => implement in e.g. smartphones

If successful, can be applied to other types of signals (radar, lidar, radio, ...)



# Precise problem statement

- Input : low quality audio (noise, low resolution, reverberation, ...)
- Output : better audio (less noise, higher resolution, ..)

Sounds better => success !

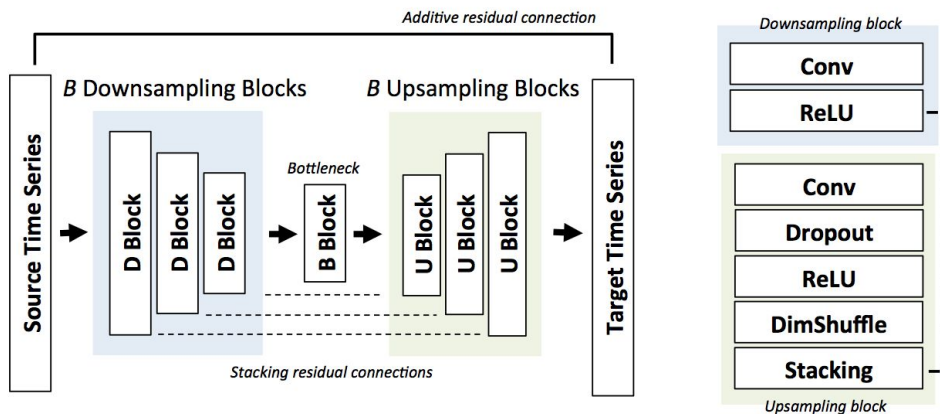


# Previous work

Kuleshov, Enam, Ermon : **Audio Super Resolution with Neural Networks**, 2017

- Audio super-resolution applied on speech
- One of the only paper that also tried with music (piano)
- Convolutional neural network that works directly on the audio signal
- Bottleneck architecture

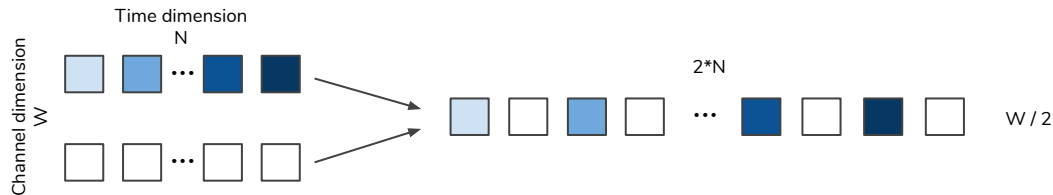
# One previous work



## Bottleneck architecture

- $N$  Downsampling blocks
- $N$  Upsampling blocks
- Skip connections (stacking and additive)

## DimShuffle : Sub-pixel operation

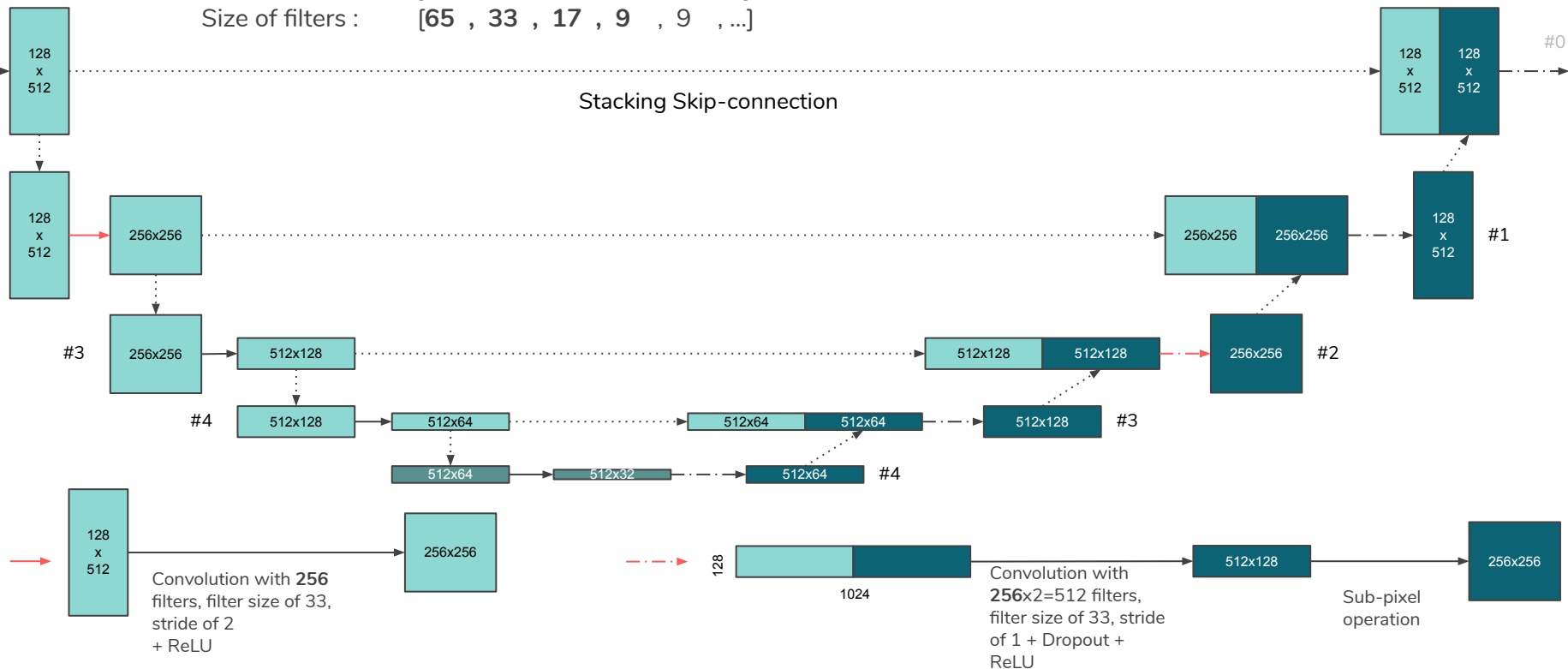


# Stacking skip-connection

Number of filters : [128, 256, 512, 512, 512, ...]

Size of filters : [65, 33, 17, 9, 9, ...]

Stacking Skip-connection





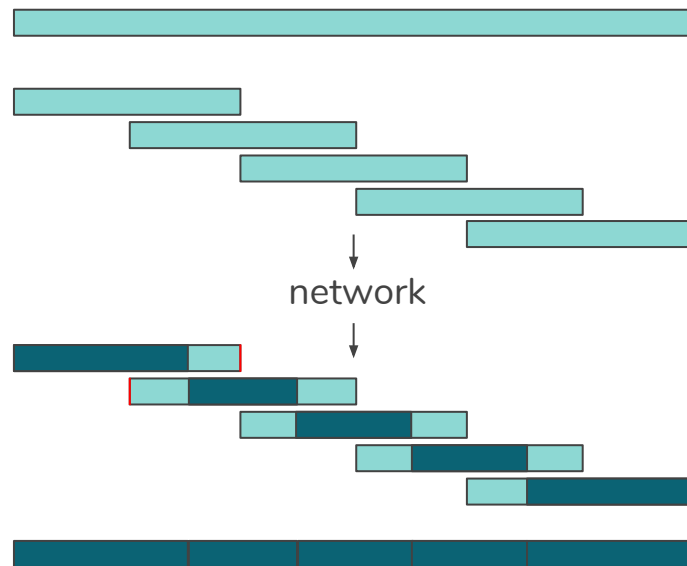
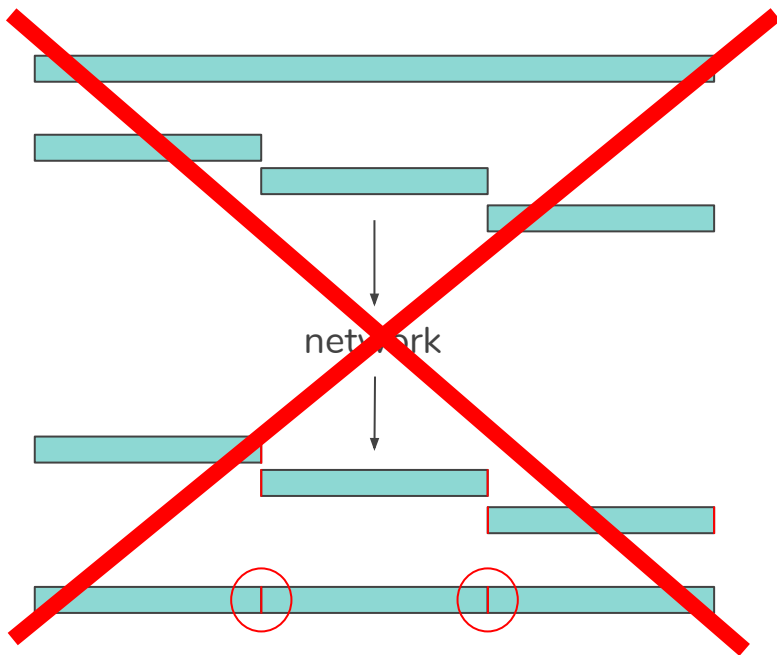
# Proposed method

- Implement the network proposed by Kuleshov, Enam, Ermo
- Once it is working as expected, try to improve it using GANs
  - Sung Kim, Visvesh Sathe, **Adversarial Audio Super-Resolution with Unsupervised Feature Losses**
- Not only for super-resolution, but also for denoising and reverberation removal



# Re-constructing the audio

We want it to work for any audio file







# Experiment

## Datasets

- Maestro dataset : 122GB 44.1kHz 1411 kbps wav
- Beethoven dataset : 330MB 44.1kHz 112 kbps ogg

## Metrics

- SNR (Signal to noise ratio), higher = better
- LSD (Log-spectral distance), lower = better
- MOS (Mean opinion score)

$$\text{SNR}(x, x_{ref}) = 10 \log_{10} \frac{\|x_{ref}\|_2^2}{\|x - x_{ref}\|_2^2}$$

$$\text{LSD}(X, X_{ref}) = \frac{1}{W} \sum_{w=1}^W \sqrt{\frac{1}{K} \sum_{k=1}^K \left( \log_{10} \frac{|X(w,k)|^2}{|X_{ref}(w,k)|^2} \right)^2}$$



# Work done so far

- Implementation using pytorch
- Complete pipeline
  - Data preparation
  - Training
  - Generate improved audio file
  - Evaluation with metrics
- Modular code
  - Any dataset
  - Any preprocessing on the audio
- Trained on noisy, downsampled and reverberated audio



# Results Super-resolution

5kHz to 10kHz, 10 epochs, mini-batch of 32 samples, sliding window of 1024 with stride 512, depth 4, dropout of 0.5. Metrics on 1'000'000 points.

	SNR		LSD	
	Low quality	Improved	Low quality	Improved
Default	27.42	1.65	2.32	<b>1.39</b>
4 -> 8 layers	27.42	1.65	2.32	<b>1.34</b>
Dropout 0.5 -> 0.2	27.42	1.65	2.32	<b>1.41</b>

SNR worse, but LSD better

When listening : hear saturation, but also hear frequencies back



# Results Denoising and Dereverberation

	SNR		LSD	
	Low quality	Improved	Low quality	Improved
Denoising	6.02	3.55	1.05	1.22
Dereverberation	7.26	3.08	0.41	0.90

Always worse, probably need more training data / deeper network



# Planned work

Using inspiration from other papers (**Adversarial Audio Super-Resolution with Unsupervised Feature Losses**)

- Turn into a GAN
- Add loss computed by features of another network
- Try more realistic pre-processing



# References

- Volodymyr Kuleshov, S. Zayd Enam, and Stefano Ermon, **Audio Super-resolution using neural networks**, [arXiv:1708.00853](https://arxiv.org/abs/1708.00853), 2017
- Sung Kim, Visvesh Sathe, **Adversarial Audio Super-Resolution with Unsupervised Feature Losses**, ICLR 2019 Conference Blind Submission, <https://openreview.net/forum?id=H1eH4n09KX>
- François G. Germain, Qifeng Chen, and Vladlen Koltun, **Speech Denoising with Deep Feature Losses**, [arXiv:1806.10522](https://arxiv.org/abs/1806.10522), 2018
- Santiago Pascual , Antonio Bonafonte , Joan Serra, **SEGAN: Speech Enhancement Generative Adversarial Network**, [arXiv:1703.09452](https://arxiv.org/abs/1703.09452), 2017
- Curtis Hawthorne, Andriy Stasyuk, Adam Roberts, Ian Simon, Cheng-Zhi Anna Huang, Sander Dieleman, Erich Elsen, Jesse Engel, and Douglas Eck. "**Enabling Factorized Piano Music Modeling and Generation with the MAESTRO Dataset**." In International Conference on Learning Representations, 2019, [\[link\]](#)