Speech Denoising via Nonnegative Matrix Factorization

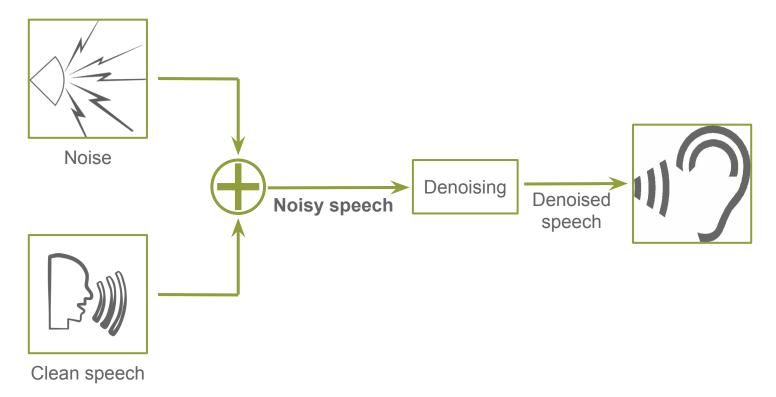


Your Answer

Oleg Alenkin Artyom Chashchin Vladimir Chernykh German Novikov

December 2016

Problem





Applications

Automatic Speech Recognition



Telephone conversations



Hearing aids





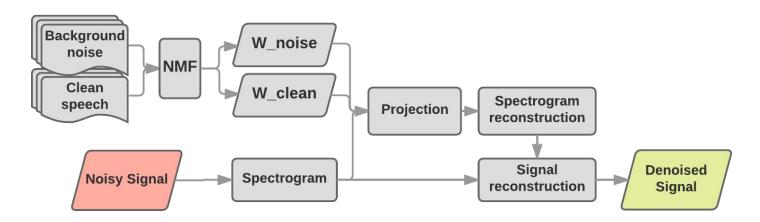
Data

- CHiME Speech Separation and Recognition Challenge
 http://spandh.dcs.shef.ac.uk/chime_challenge/chime_download.html
 Recordings of WSJ utterances + 8 hours of noise
- Berlin Database of Emotional Speech
 http://www.emodb.bilderbar.info/download/
 Clean utterances
- Aurora noising
 http://aurora.hsnr.de/download.html



Solution

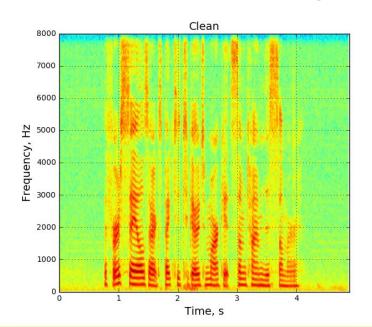
- Learn frequency patterns from speech and noise via NMF
- Decompose new signal on joint "dictionary" of patterns
- Take only projection corresponding to "clean speech"

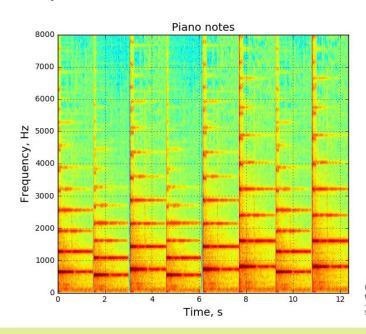




Spectrogram

- Represent signal in Time-Frequency domain
- Built via Short-Term Fourier Transform FFT with sliding overlapped window
- STFT Complex spectrogram S ⇒ Amplitudes V

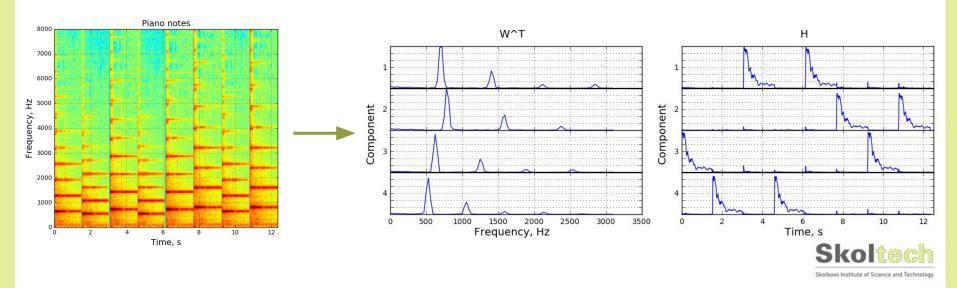






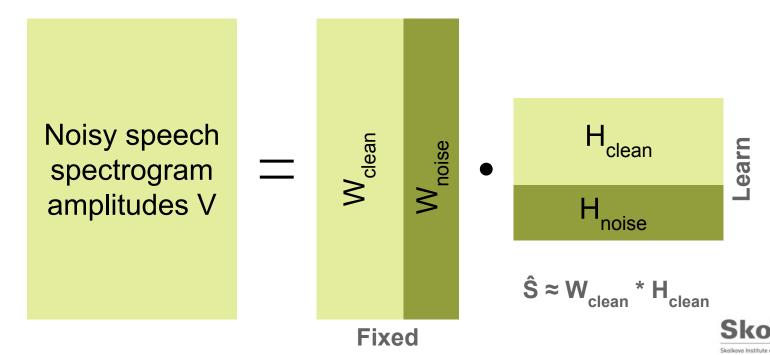
NMF

- Factorization V ≈ W * H, where V, W, H real nonnegative
- Interpretation: W frequency patterns, H time-activation matrix
- Hidden dimension ≈ number of phonemes ≈ 40
- Learn speech and noise "building blocks"



Projection

- Join "dictionaries" concatenate matrices W_{noise} and W_{clean}
- Project signal onto them



NMF: how to compute

Optimization problem:

$$(W^*, H^*) = \operatorname{argmin}_{W \ge 0, H \ge 0} D(V, WH)$$

$$D(P,Q) = \sum_{i=1}^{m} \sum_{j=1}^{n} d(p_{ij}, q_{ij})$$

The most popular metrics:

$$d(p,q) = (p-q)^2$$
 Frobenius norm

$$d(p,q) = p \ln(\frac{p}{q}) - p + q \quad \text{KL divergence}$$

Multiplicative Update Method:

$$[\nabla_H]_{kj} = \frac{\partial D(V, WH)}{\partial h_{kj}} = [\nabla_H^+]_{kj} - [\nabla_H^-]_{kj}$$

$$h_{kj} \leftarrow h_{kj} - \frac{h_{kj}}{[\nabla_H^+]_{kj}} \left([\nabla_H^+]_{kj} - [\nabla_H^-]_{kj} \right)$$



NMF: how to compute

Alternating Nonnegative Least Squares:

1) Initialize
$$W_{ia}^1 \geq 0, H_{bj}^1 \geq 0, \forall a, i, b, j.$$

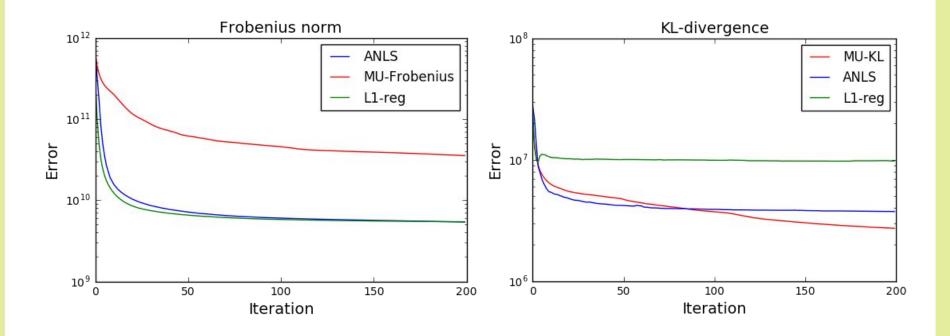
2) For
$$k=1,2,...$$

$$W^{k+1} = \operatorname{argmin}_{W \ge 0} D\left(V, WH^k\right),\,$$

$$H^{k+1} = \operatorname{argmin}_{H>0} D\left(V, W^{k+1}H\right).$$



Methods' Convergence





Quasi-Newton method

$$(W^*, H^*) = \operatorname{argmin}_{W>0, H>0} D_{KL}(V, WH)$$

$$W \leftarrow \max(\varepsilon, W - H_W^{-1} \nabla_W D_{KL})$$

$$\nabla_W D_{KL} = H^T J_{M \times K} - H^T (V \oslash (WH))$$

$$H_W = \operatorname{diag}\{h_{W,m}, \ m = 1, \dots, M\}$$

$$h_{W,m} = H \operatorname{diag}\{[V \oslash (Q \otimes Q)]_{m,:}\}H^T$$

 $H \leftarrow \max(\varepsilon, H - H_H^{-1} \nabla_H D_{KL})$ $\nabla_H D_{KL} = J_{M \times K} W^T - (V \oslash (WH)) W^T$ $H_H = \operatorname{diag}\{h_{H,k}, \ k = 1, \dots, K\}$ $h_{H,k} = W^T \operatorname{diag}\{[V \oslash (Q \otimes Q)]_{:,k}\} W$

Problem: hard to compute inverse Hessian



Signal reconstruction

Naive method with zero phase:

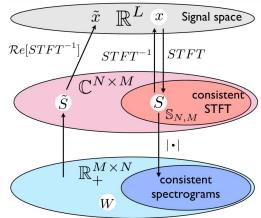
$$\hat{X} = STFT^{-1}(\hat{S})$$

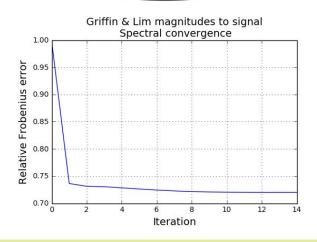
Noisy signal phases:

$$\hat{X} = STFT^{-1}(\hat{S} \times \exp\left(i \angle STFT(X_{noisy})\right))$$

Griffin & Lim iterative method:

$$\hat{X}_n = STFT^{-1}(\hat{S} \times \exp(i \angle STFT(\hat{X}_{n-1})))$$





Demo

