Source Separation Tutorial Mini-Series III: Extensions and Interpretations to Non-Negative Matrix Factorization

> Nicholas Bryan Dennis Sun

Center for Computer Research in Music and Acoustics, Stanford University

> DSP Seminar April 9th, 2013

Roadmap of Talk





3 Supervised and Semi-Supervised Separation

4 Probabilistic Interpretation









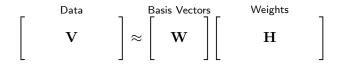
Roadmap of Talk

1 Review

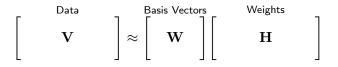
- **2** Further Insight
- **3** Supervised and Semi-Supervised Separation
- **4** Probabilistic Interpretation
- **5** Extensions
- 6 Evaluation
- **7** Future Research Directions



Non-Negative Matrix Factorization

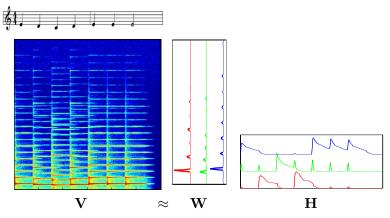


Non-Negative Matrix Factorization



- A matrix factorization where everything is non-negative
- $\mathbf{V} \in \mathbf{R}^{F imes T}_+$ original non-negative data
- $\mathbf{W} \in \mathrm{R}^{F imes K}_+$ matrix of basis vectors, dictionary elements
- $\mathbf{H} \in \mathrm{R}^{K imes T}_+$ matrix of activations, weights, or gains
- K < F < T (typically)
 - A compressed representation of the data
 - A low-rank approximation to ${\bf V}$

NMF With Spectrogram Data

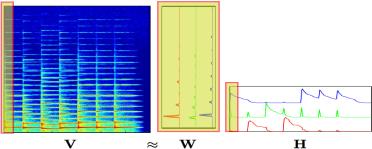


NMF of Mary Had a Little Lamb with K = 3 play stop

- The basis vectors capture prototypical spectra [SB03]
- The weights capture the gain of the basis vectors

Factorization Interpretation I

Columns of $\mathbf{V}\approx$ as a weighted sum (mixture) of basis vectors

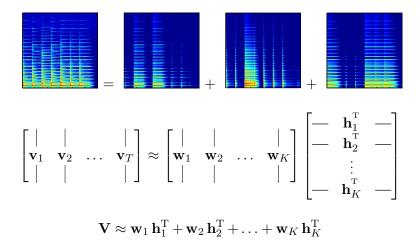


 \approx

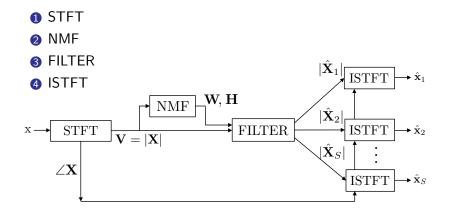
$$\begin{bmatrix} | & | & | \\ \mathbf{v}_1 & \mathbf{v}_2 & \dots & \mathbf{v}_T \\ | & | & | \end{bmatrix} \approx \begin{bmatrix} K \\ \sum_{j=1}^K \mathbf{H}_{j1} \mathbf{w}_j & \sum_{j=1}^K \mathbf{H}_{j2} \mathbf{w}_j & \dots & \sum_{j=1}^K \mathbf{H}_{jT} \mathbf{w}_j \end{bmatrix}$$

Factorization Interpretation II

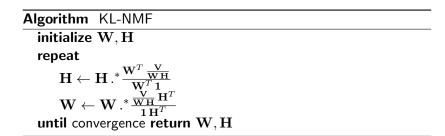
 ${\bf V}$ is approximated as sum of matrix "layers"



General Separation Pipeline



An Algorithm for NMF



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Non-Negativity

• Question: Why do we get a 'parts-based' representation of sound?

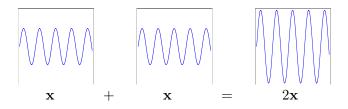
Non-Negativity

• Question: Why do we get a 'parts-based' representation of sound?

• Answer: Non-negativity avoids destructive interference

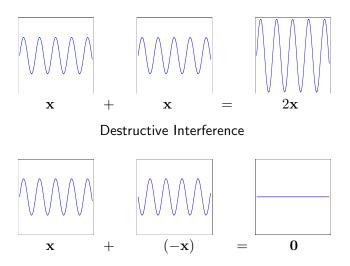
Constructive and Destructive Interference

Constructive Interference



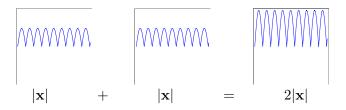
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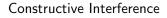


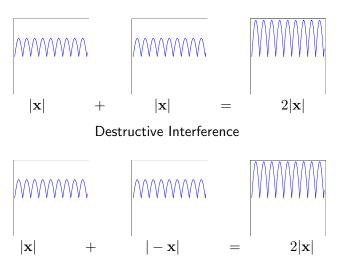
Non-Negative Constructive and Destructive Interference

Constructive Interference



Non-Negative Constructive and Destructive Interference





Non-negativity Avoids Destructive Interference

• With non-negativity, destructive interference cannot happen

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• Everything must cumulatively add to explain the original data

Non-negativity Avoids Destructive Interference

• With non-negativity, destructive interference cannot happen

• Everything must cumulatively add to explain the original data

• But ...

In doing so, we violate the superposition property of sound

$$\mathbf{x} = \mathbf{x}_1 + \mathbf{x}_2 + \ldots + \mathbf{x}_N$$

and actually solve

$$|\mathbf{X}| \approx |\mathbf{X}_1| + |\mathbf{X}_2| + \ldots + |\mathbf{X}_N|$$

Approximation II

Alternatively, we can see this approximation via:

$$\mathbf{x} = \mathbf{x}_1 + \mathbf{x}_2 + \ldots + \mathbf{x}_N$$
$$|\mathbf{X}| e^{j\phi} = |\mathbf{X}_1| e^{j\phi_1} + |\mathbf{X}_2| e^{j\phi_2} + \ldots + |\mathbf{X}_N| e^{j\phi_N}$$
$$|\mathbf{X}| e^{j\phi} \approx (|\mathbf{X}_1| + |\mathbf{X}_2| + \ldots + |\mathbf{X}_N|) e^{j\phi}$$
$$|\mathbf{X}| \approx |\mathbf{X}_1| + |\mathbf{X}_2| + \ldots + |\mathbf{X}_N|$$

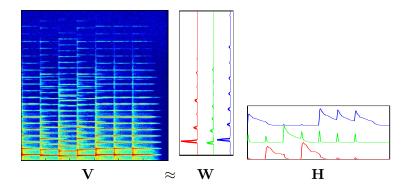
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Unsupervised Separation I

Single, simultaneously estimation of ${\bf W}$ and ${\bf H}$ from a mixture ${\bf V}$



What we've seen so far

Unsupervised Separation II

• Complex sounds need more than one basis vector

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• Difficult to control which basis vector explain which source

Unsupervised Separation II

• Complex sounds need more than one basis vector

• Difficult to control which basis vector explain which source

• No way to control the factorization other than F, T, and K

General idea:

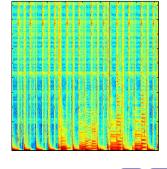
 Use isolated training data of each source within a mixture to pre-learn individual models of each source [SRS07] General idea:

 Use isolated training data of each source within a mixture to pre-learn individual models of each source [SRS07]

2 Given a mixture, use the pre-learned models for separation

Supervised Separation I

Example:

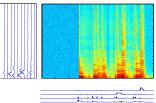


Drum and Bass Loop play stop

Supervised Separation II

Use isolated training data to learn factorization for each source

Bass Loop play stop

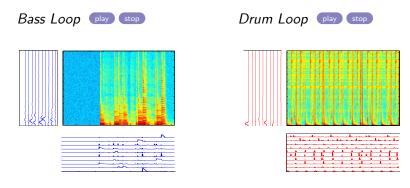


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 $\mathbf{V}_1 \approx \mathbf{W}_1 \, \mathbf{H}_1$

Supervised Separation II

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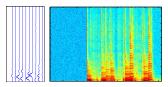
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Supervised Separation III

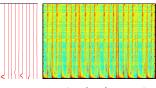
Throw away the activations \mathbf{H}_1 and \mathbf{H}_2

Bass Loop play stop



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Drum Loop play stop



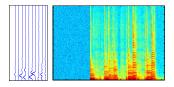
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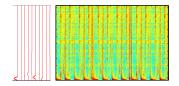
 $\mathbf{V}_1 \approx \mathbf{W}_1 \, \mathbf{H}_1$ 

 $\mathbf{V}_2 \approx \mathbf{W}_2 \, \mathbf{H}_2$ 

# Supervised Separation IV

Concatenate basis vectors of each source for complete dictionary





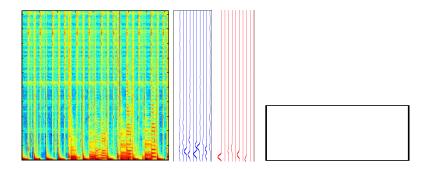
$$\mathbf{W} \approx \begin{bmatrix} \mathbf{W}_1 & \mathbf{W}_2 \end{bmatrix} =$$



# Supervised Separation V

V

Now, factorize the mixture with  $\mathbf{W}$  fixed (only estimate  $\mathbf{H}$ )

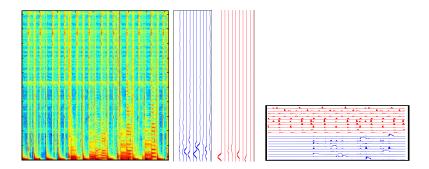


$$pprox \mathbf{W} = \mathbf{H} \ pprox \mathbf{W} = [\mathbf{W}_1 \ \mathbf{W}_2] \qquad \begin{bmatrix} \mathbf{H}_1^{\mathrm{T}} \\ \mathbf{H}_2^{\mathrm{T}} \end{bmatrix}$$

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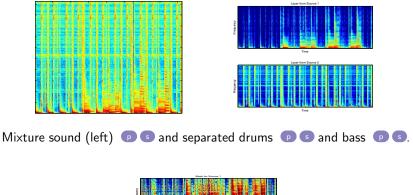
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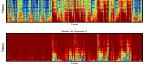
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- Hold W fixed, and factorize unknown mixture of sources V (only estimate H)

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- **5** Once complete, use **W** and **H** as before to filter and separate each source

## Sound Examples





Masking filters used to process mixture into the separated sources.



• What if you don't have isolated training data for each source?



• What if you don't have isolated training data for each source?

• And unsupervised separation still doesn't work?

## Semi-Supervised Separation

General Idea:

 Learn supervised dictionaries for as many sources as you can [SRS07]

# Semi-Supervised Separation

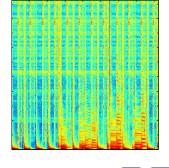
General Idea:

 Learn supervised dictionaries for as many sources as you can [SRS07]

 Infer remaining unknown dictionaries from the mixture (only fix certain columns of W)

# Semi-Supervised Separation I

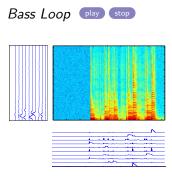
#### Example:



Drum and Bass Loop play stop

# Semi-Supervised Separation II

Use isolated training data to learn factorization for as many sources as possible (e.g. one source)

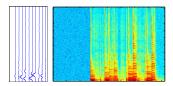


 $\mathbf{V}_1 \approx \mathbf{W}_1 \, \mathbf{H}_1$ 

## Semi-Supervised Separation III

Throw away the activations  $\mathbf{H}_1$ 

Bass Loop play stop

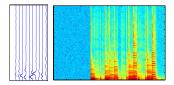




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# Semi-Supervised Separation IV

Concatenate *known* basis vectors with *unknown* basis vectors (initialized randomly) for complete dictionary



Known bass basis vectors



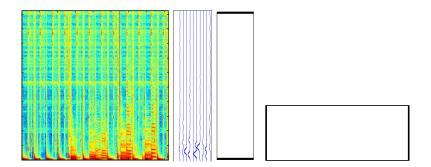
Unknown drum basis vectors (initialized randomly)

$$\mathbf{W} \approx \begin{bmatrix} \mathbf{W}_1 \, \mathbf{W}_2 \end{bmatrix} =$$

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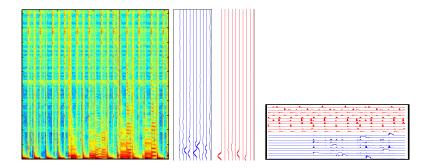


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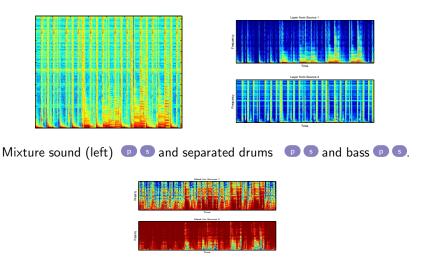
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# Sound Examples

Supervised the bass.



Masking filters used to process mixture into the separated sources.

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#### Some notation:

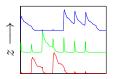
 $\boldsymbol{z}$  indexes basis vectors,  $\boldsymbol{f}$  frequency bins, and  $\boldsymbol{t}$  time frames.

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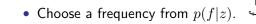
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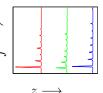
For each time frame t, repeat the following:

• Choose a component from p(z|t).



 $t \longrightarrow$ 



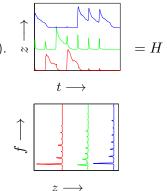


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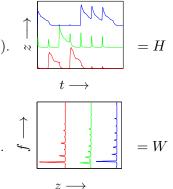
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= H

= W

 $t \longrightarrow$ 

 $z \longrightarrow$ 

• Choose a frequency from p(f|z).

The spectrogram  $V_{ft}$  are the counts that we obtain at the end of the day. We want to estimate p(z|t) and p(f|z).

Is this realistic?

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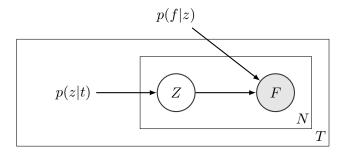
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- In audio, this model is called probabilistic latent component analysis, or PLCA [SRS06]

#### Latent Variable Model

We only observe the outcomes  $V_{ft}$ . But the full model involves unobserved variables Z.

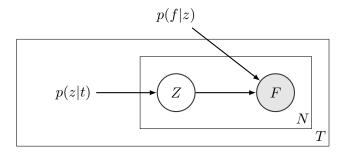
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The **Expectation-Maximization (EM) algorithm** is used to fit latent variable models. It is also used in estimating Hidden Markov Models, Gaussian mixture models, etc.

To fit the parameters, we choose the parameters that maximize the likelihood of the data. Let's zoom in on a single time frame:

$$p(v_1, ..., v_F) = \frac{(\sum_f v_f)!}{v_1! ... v_F!} \prod_{f=1}^F p(f|t)^{v_f}$$

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$$p(f|t) = \sum_{z} p(z|t)p(f|z)$$

Putting it all together, we obtain:

$$p(v_1, ..., v_F) = \frac{(\sum_f v_f)!}{v_1! ... v_F!} \prod_{f=1}^F \left( \sum_z p(z|t) p(f|z) \right)^{v_f}$$

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- We want to maximize this over p(z|t) and p(f|z).
- In general, with probabilities it is easier to maximize the log than the thing itself:

$$\log p(v_1, ..., v_F) = \sum_{f=1}^F v_f \log \left( \sum_z p(z|t) p(f|z) \right) + \text{const.}$$

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- We want to maximize this over p(z|t) and p(f|z).
- In general, with probabilities it is easier to maximize the log than the thing itself:

$$\log p(v_1, ..., v_F) = \sum_{f=1}^F v_f \log \left( \sum_z p(z|t) p(f|z) \right) + \text{const.}$$

 Remember from last week: First thing you should always try is differentiate and set equal to zero. Does this work here?

• Last week, we talked about minimizing the KL divergence between V and WH.

$$D(V||WH) = -\sum_{f,t} V_{ft} \log\left(\sum_{z} W_{fz} H_{zt}\right) + \sum_{f,t} \sum_{z} W_{fz} H_{zt} + \text{const.}$$

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Now watch what we do with the log-likelihood....

• Suppose we observed the latent component for a frequency quanta. Then we wouldn't need to average over the components; its log-likelihood would be:

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In summary, we've replaced

$$\log\left(\sum_{z}p(z|t)p(f|z)\right) \quad \text{by} \quad \sum_{z}p(z|f,t)\log p(z|t)p(f|z)$$
 Look familiar?

**E-step**: Calculate  $p(z|f,t) = \frac{p(z|t)p(f|z)}{\sum_{z} p(z|t)p(f|z)}$  M-step: Maximize  $\sum_{f,t} V_{ft} \sum_{z} p(z|f,t) \log p(z|t)p(f|z)$ 

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M-step: Maximize

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Majorization: Calculate

Minimization: Minimize

$$\phi_{ftz} = \frac{W_{fz}H_{zt}}{\sum_{z}W_{fz}H_{zt}}$$

$$-\sum_{f,t} V_{ft} \sum_{z} \phi_{zft} \log W_{fz} H_{zt} + \sum_{f,t,z} W_{fz} H_{zt}$$

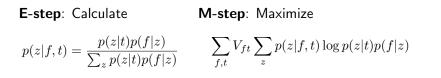
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The EM updates are exactly the multiplicative updates for NMF, up to normalization!



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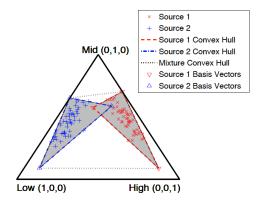
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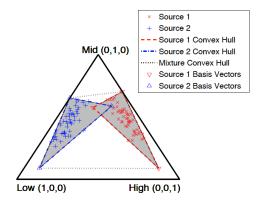
The EM algorithm is a special case of MM, where the minorizing function is the expected conditional log likelihood.

### Geometric Interpretation



• We can think of the basis vectors p(f|z) as lying on a probability simplex.

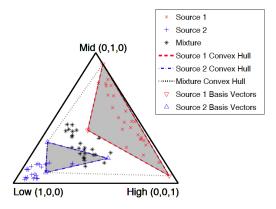
### Geometric Interpretation



- We can think of the basis vectors p(f|z) as lying on a probability simplex.
- The possible sounds for a given source is the convex hull of the basis vectors for that source.

#### Geometric Interpretation

In supervised separation, we try to explain time frames of the mixture signal as combinations of the basis vectors of the different sources.



# Roadmap of Talk

- 1 Review
- 2 Further Insight
- **3** Supervised and Semi-Supervised Separation
- **4** Probabilistic Interpretation
- **5** Extensions
- 6 Evaluation
- **7** Future Research Directions
- 8 Matlab

#### Extensions

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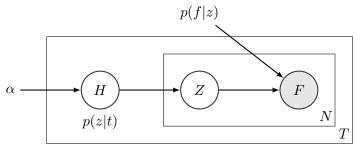
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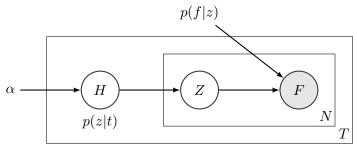
- The number of parameters that need to be estimated is huge: FK + KT.
- In high-dimensional settings, it is useful to impose additional structure.
- We will look at two ways to do this: **priors** and **regularization**.

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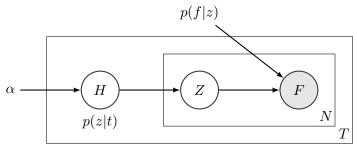


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• Estimate the **posterior** distribution  $p(H|\alpha, V)$ .

• Bayes' rule: 
$$p(H|\alpha, V) = \frac{p(H, V|\alpha)}{p(V|\alpha)} = \frac{p(H|\alpha)p(V|H)}{p(V|\alpha)}$$

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• We can choose priors that encode structural assumptions, like sparsity.

## Regularization Viewpoint

• Another way is to add another term to the objective function:

$$\underset{W,H\geq 0}{\text{minimize}} \ D(V||WH) + \lambda \Omega(H)$$

 $\Omega$  encodes the desired structure,  $\lambda$  controls the strength.

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- Some common choices for Ω(H):
  - sparsity:  $||H||_1 = \sum_{z,t} |H_{zt}|$
  - smoothness:  $\sum_{z,t} (H_{z,t} H_{z,t-1})^2$

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#### **Evaluation Measures**

- Signal-to-Interference Ratio (SIR)
- Signal-to-Artifact Ratio (SAR)
- Signal-to-Distortion Ratio (SDR)

We want all of these metrics to be as high as possible [VGF06]

#### **Evaluation Measures**

To compute these three measures, we must obtain:

- $\mathbf{s} \in R^{T \times N}$  original unmixed signals (ground truth)
- $\hat{\mathbf{s}} \in R^{T \times N}$  estimated separated sources

Then, we decompose these signals into

- $s_{target}$  actual source estimate
- $e_{interf}$  interference signal (i.e. the unwanted source)
- $e_{artif}$  artifacts of the separation algorithm

#### **Evaluation Measures**

To compute  $s_{target}$ ,  $e_{interf}$ , and  $e_{artif}$ 

- $s_{target} = P_{s_j} \hat{s}_j$
- $e_{interf} = P_{\mathbf{s}}\hat{s}_j P_{s_j}\hat{s}_j$

• 
$$e_{artif} = \hat{s}_j - P_{\mathbf{s}}\hat{s}_j$$

where  $P_{s_i}$  and  $P_{\mathbf{s}}$  are  $T \times T$  projection matrices

# Signal-to-Interference Ratio (SIR)

#### A measure of the suppression of the unwanted source

$$\mathsf{SIR} = 10\log_{10} \frac{||s_{target}||^2}{||e_{interf}||^2}$$

### Signal-to-Artifact Ratio (SAR)

A measure of the artifacts that have been introduced by the separation process

$$\mathsf{SAR} = 10 \log_{10} \frac{||s_{target} + e_{interf}||^2}{||e_{artif}||^2}$$

### Signal-to-Distortion Ratio (SDR)

#### An overall measure that takes into account both the SIR and SAR

$$\mathsf{SDR} = 10 \log_{10} \frac{||s_{target}||^2}{||e_{artif} + e_{interf}||^2}$$

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- One problem with NMF is the need to specify the number of basis vectors *K*.
- Even more parameters if you include regularization.
- BSS eval metrics give us a way to learn the optimal settings for source separation.
- Generate synthetic mixtures, try different parameter settings, and choose the parameters that give the best BSS eval metrics.

A Matlab tool box for source separation evaluation [VGF06]:

http://bass-db.gforge.inria.fr/bss_eval/

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#### **Research Directions**

- Score-informed separation sheet music
- Interactive separation user-interaction
- Temporal dynamics how sounds change over time
- Unsupervised separation grouping basis vectors, clustering
- Phase estimation complex NMF, STFT constraints, etc.
- Universal models big data for general models of sources

#### Demos

• Universal Speech Models

- Interactive Source Separation
  - Drums + Bass
  - Guitar + Vocals + AutoTune
  - Jackson 5 Remixed

# STFT

```
x1 = wavread('bass');
x2 = wavread('drums');
[xm fs] = wavread('drums+bass');
FFTSIZE = 1024;
HOPSIZE = 256;
WINDOWSIZE = 512;
X1 = myspectrogram(x1,FFTSIZE,fs,hann(WINDOWSIZE),-HOPSIZE);
V1 = abs(X1(1:(FFTSIZE/2+1),:));
X2 = myspectrogram(x2,FFTSIZE,fs,hann(WINDOWSIZE),-HOPSIZE);
V2 = abs(X2(1:(FFTSIZE/2+1),:));
Xm = myspectrogram(xm,FFTSIZE,fs,hann(WINDOWSIZE),-HOPSIZE);
Vm = abs(Xm(1:(FFTSIZE/2+1),:)); maxV = max(max(db(Vm)));
```

```
F = size(Vm,1);
T = size(Vm,2);
```

- https://ccrma.stanford.edu/~jos/sasp/Matlab_ listing_myspectrogram_m.html
- https://ccrma.stanford.edu/~jos/sasp/Matlab_ listing_invmyspectrogram_m.html

#### NMF

```
K = [25 25]; % number of basis vectors
MAXITER = 500; % total number of iterations to run
[W1, H1] = nmf(V1, K(1), [], MAXITER, []);
[W2, H2] = nmf(V2, K(2), [], MAXITER, []);
[W, H] = nmf(Vm, K, [W1 W2], MAXITER, 1:sum(K));
function [W, H] = nmf(V, K, W, MAXITER, fixedInds)
F = size(V,1); T = size(V,2);
rand('seed'.0)
if isempty(W)
    W = 1 + rand(F, sum(K));
end
H = 1 + rand(sum(K), T):
inds = setdiff(1:sum(K).fixedInds);
ONES = ones(F,T):
for i=1:MAXITER
    % update activations
    H = H .* (W'*(V./(W*H+eps))) ./ (W'*ONES);
    % update dictionaries
    W(:,inds) = W(:,inds) .* ((V./(W*H+eps))*H(inds,:)') ./(ONES*H(inds,:)');
end
% normalize W to sum to 1
sumW = sum(W);
W = W * diag(1./sumW);
H = diag(sumW) * H;
```

# FILTER & ISTFT

```
% get the mixture phase
phi = angle(Xm);
c = [1 cumsum(K)];
for i=1:length(K)
    % create masking filter
    Mask = W(:,c(i):c(i+1))*H(c(i):c(i+1),:)./(W*H);
    % filter
    XmagHat = Vm.*Mask;
    % create upper half of frequency before istft
    XmagHat = [XmagHat; conj( XmagHat(end-1:-1:2,:))];
    % Multiply with phase
    XHat = XmagHat.*exp(1i*phi);
    % create upper half of frequency before istft
    xhat(:,i) = real(invmyspectrogram(XmagHat.*exp(1i*phi))
end
```

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