





## **Audio source separation**

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TSIA 206 - Speech and audio processing

#### Part I

#### Introduction



Introduction

- Source separation
  - ► Art of estimating "source" signals, assumed independent, from the observation of one or several "mixtures" of these sources
- Application examples:
  - ▶ Denoising (cocktail party, suppression of vuvuzela, karaoke)

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- ► Separation of the instruments in polyphonic music
- ► Remix, transformations, re-spatialization

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#### Typology of the mixture models

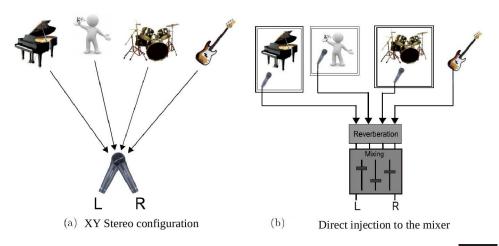
- ▶ Definition of the problem
  - ▶ Observations: M mixtures  $x_m(t)$ , concatenated in a vector  $\mathbf{x}(t)$
  - ▶ Unknowns: K sources  $s_k(t)$ , concatenated in a vector  $\mathbf{s}(t)$
  - ▶ General mixture model: function  $\mathscr{A}$  which transforms  $\mathbf{s}(t)$  into  $\mathbf{x}(t)$
- ► Linearity: 𝒜 is a linear map
- ► Memory:
  - Convolutive mixtures
  - Instantaneous mixtures:  $\mathbf{x}(t) = \mathbf{A}\mathbf{s}(t)$ 
    - $\blacktriangleright$   $\mathscr{A}$  is defined by the "mixture matrix" **A** (of dimension  $M \times K$ )
- ► Inversibility:
  - $\triangleright$  Determined mixtures: M = K
  - Over-determined mixtures: M > K
  - ► Under-determined mixtures: *M* < *K*

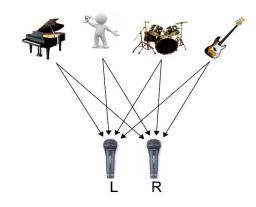












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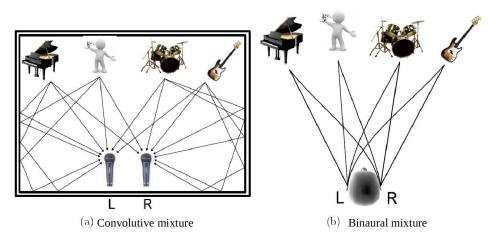
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#### **Convolutive linear mixtures**



Part II

Mathematical reminders





#### Real random vectors

- Notation:  $\phi[x]$  denotes a function of  $\rho(x)$
- Mean vector:  $\mu_{\mathsf{x}} = \mathbb{E}[\mathsf{x}]$
- Covariance matrix:  $\mathbf{\Sigma}_{xx} = \mathbb{E}[(\mathbf{x} \mu_x)(\mathbf{x} \mu_x)^T]$
- Characteristic function:  $\phi_{\mathsf{x}}(\mathbf{f}) = \mathbb{E}[e^{-2i\pi\mathbf{f}^{\mathsf{T}}\mathbf{x}}] = \int_{\mathbb{D}} p(\mathbf{x})e^{-2i\pi\mathbf{f}^{\mathsf{T}}\mathbf{x}}d\mathbf{x}$
- ► Probability distribution:  $p(\mathbf{x}) = \int_{\mathbb{R}} \phi_{\mathbf{x}}(\mathbf{f}) e^{+2i\pi \mathbf{f}^T \mathbf{x}} d\mathbf{f}$
- Cumulants:
  - ► Definition:  $\ln(\phi_X(\mathbf{f})) = \sum_{n=1}^{+\infty} \frac{(-2i\pi)^n}{n!} \sum_{k_1=1}^K \sum_{k_n=1}^K \kappa_{k_1...k_n}^n[\mathbf{x}] f_{k_1} ... f_{k_n}$
  - $\triangleright \kappa^n[\mathbf{x}]$  is an *n*-th order tensor
  - $ightharpoonup \kappa^1[\mathbf{x}]$  is the mean vector,  $\kappa^2[\mathbf{x}]$  is the covariance matrix
  - If p(x) is symmetric (p(-x) = p(x)),  $\kappa^n[x] = 0$  for any odd value n
  - the ratio  $\kappa_{k,k,k,k}^4[\mathbf{x}]/(\kappa_{k,k}^2[\mathbf{x}])^2$  is called "kurtosis"



- ▶ The Gaussian distribution is the one such that all cumulants of order n > 2 are zero
- Characteristic function

$$\phi_{\mathsf{x}}(\mathbf{f}) = \exp(-2i\pi\mathbf{f}^{\mathsf{T}}\boldsymbol{\mu}_{\mathsf{x}} - 2\pi^{2}\mathbf{f}^{\mathsf{T}}\boldsymbol{\Sigma}_{\mathsf{xx}}\mathbf{f})$$

ightharpoonup Probability density function (defined if  $\Sigma_{xx}$  is invertible)

$$p(\mathbf{x}) = \frac{1}{(2\pi)^{\frac{K}{2}}\det(\mathbf{\Sigma}_{xx})^{\frac{1}{2}}}\exp\left(-\frac{1}{2}(\mathbf{x} - \boldsymbol{\mu}_{x})^{T}\mathbf{\Sigma}_{xx}^{-1}(\mathbf{x} - \boldsymbol{\mu}_{x})\right)$$





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#### WSS vector processes

- ▶ Definition: the cumulants of orders 1 et 2 are translation-invariant
- $\triangleright$  Covariance matrices of 2 centered WSS processes  $\mathbf{x}(t)$  and  $\mathbf{y}(t)$ :
  - ▶ Definition:  $\mathbf{R}_{xy}(\tau) = \mathbb{E}\left[\mathbf{x}(t+\tau)\mathbf{y}(t)^T\right]$
  - Property:  $\mathbf{R}_{xx}(0) = \mathbf{\Sigma}_{xx}$  is Hermitian and positive semi-definite.
- $\triangleright$  PSD matrices of a WSS process  $\mathbf{x}(t)$ :
  - ▶ Definition:  $\mathbf{S}_{xx}(v) = \sum_{\tau \in \mathcal{T}} \mathbf{R}_{xx}(\tau) e^{-2i\pi v \tau}$
  - Property:  $\forall v$ ,  $\mathbf{S}_{xx}(v)$  is Hermitian and positive semi-definite

## Information theory

- ► Shannon entropy
  - ▶ Definition:  $\mathbb{H}[\mathbf{x}] = -\mathbb{E}[\ln(p(\mathbf{x}))]$
  - ightharpoonup  $\mathbb{H}[x]$  is not necessarily non-negative for a continuous r.v.
- ► Kullback-Leibler divergence
  - $D_{KL}(p||q) = \int p(\mathbf{x}) \ln \left( \frac{p(\mathbf{x})}{q(\mathbf{x})} \right) d\mathbf{x}$
  - Property:  $D_{KL}(p||q) \ge 0$ ,  $D_{KL}(p||q) = 0$  if and only if p = q
- Mutual information
  - ► Definition:  $\mathbb{I}[\mathbf{x}] = \mathbb{E}\left[\ln\left(\frac{p(\mathbf{x})}{p(\mathbf{x}_1)...p(\mathbf{x}_K)}\right)\right] = D_{KL}(p(\mathbf{x})||p(\mathbf{x}_1)...p(\mathbf{x}_K))$
  - Property:  $\mathbb{I}[\mathbf{x}] = 0$  if and only if  $x_1 \dots x_K$  are mutually independent
  - ▶ Relationship with entropy:  $\mathbb{I}[\mathbf{x}] = \sum_{k=1}^{K} \mathbb{H}[x_k] \mathbb{H}[\mathbf{x}]$





#### Part III

#### Linear instantaneous mixtures



- Observation model:
  - $\forall t, \mathbf{x}(t) = \mathbf{A}\mathbf{s}(t)$  where  $\mathbf{A} \in \mathbb{R}^{M \times K}$  is called the "mixture"
  - Sources are assumed IID:  $p(\{s_k(t)\}_{k,t}) = \prod_{k=1}^K \prod_{t=1}^T p_k(s_k(t))$
- Problem: estimate **A** and sources  $\mathbf{s}(t)$  given  $\mathbf{x}(t)$
- ▶ Definition: non-mixing matrix
  - $\triangleright$  a matrix **C** of dimension  $K \times K$  is non-mixing if and only if it has a unique non-zero entry in each row and each column
- ▶ If  $\widetilde{\mathbf{s}}(t) = \mathbf{C}\mathbf{s}(t)$  and  $\widetilde{\mathbf{A}} = \mathbf{A}\mathbf{C}^{-1}$ , then  $\mathbf{x}(t) = \widetilde{\mathbf{A}}\widetilde{\mathbf{s}}(t)$  is another admissible decomposition of the observations
  - ► Sources can be recovered up to a permutation and a multiplicative factor



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#### Linear separation of sources

- ▶ Let  $\mathbf{y}(t) = \mathbf{B}\mathbf{x}(t)$ , where  $\mathbf{B} \in \mathbb{R}^{K \times M}$  is referred to as the "separation matrix"
- Linear separation is feasible if **A** has rank *K*:
  - We get  $\mathbf{y}(t) = \mathbf{s}(t)$  by defining:
    - ▶  $\mathbf{B} = \mathbf{A}^{-1}$  in the determined case (M = K)
    - **B** =  $\mathbf{A}^{\dagger}$  in the over-determined case (M > K)
  - ightharpoonup the pseudo-inverse  $\mathbf{A}^{\dagger} = (\mathbf{A}^T \mathbf{A})^{-1} \mathbf{A}^T$  is such that  $\mathbf{A}^{\dagger} \mathbf{A} = \mathbf{I}_K$
- ▶ In the under-determined case (M < K), separation is not feasible

### Part IV

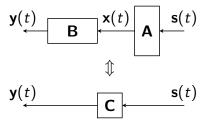
Independent component analysis





### Independent component analysis (ICA)

- ► In practice matrix **A** is unknown:
  - $\triangleright$  We look for a matrix **B** that makes the  $v_k$  independent (ICA)
  - ▶ We then get equation  $\mathbf{y}(t) = \mathbf{C}\mathbf{s}(t)$ , where  $\mathbf{C} = \mathbf{B}\mathbf{A}$
  - ► The problem is solved if matrix **C** is non-mixing



- Theorem (identifiability)
  - $\triangleright$  Let  $s_k$  be K IID sources, among which at most one is Gaussian, and  $\mathbf{y}(t) = \mathbf{C}\mathbf{s}(t)$  with **C** invertible ((over)-determined case). If signals  $y_k(t)$  are independent, then **C** is non-mixing.



- We now suppose that the sources are centered:  $\mathbb{E}[\mathbf{s}(t)] = \mathbf{0}$ and that the mixture is (over-)determined
- ► Canonical problem: we can assume without loss of generality that  $\mathbf{s}(t)$  is spatially white  $(\mathbf{\Sigma}_{ss} = \mathbb{E}[\mathbf{s}(t)\mathbf{s}(t)^T] = \mathbf{I}_K)$
- ▶ Then  $\Sigma_{xx} = A\Sigma_{ss}A^T = AA^T$ : A is a matrix square root of
- ▶ We first aim to whiten (decorrelate) the mixture:
  - $ightharpoonup \Sigma_{xx}$  is diagonalizable in an orthonormal basis:  $\Sigma_{xx} = \mathbf{Q} \mathbf{\Lambda}^2 \mathbf{Q}^T$ where  $\Lambda = \mathrm{diag}(\lambda_1 \dots \lambda_M)$  with  $\lambda_1 \ge \lambda_K > \lambda_{K+1} = \lambda_M = 0$  (the rank of  $\Sigma_{xx}$  is equal to K)
  - ▶ Let  $\mathbf{S} = \mathbf{Q}_{(:,1:K)} \mathbf{\Lambda}_{(1:K,1:K)} \in \mathbb{R}^{M \times K}$
  - **S** is a matrix square root of  $\Sigma_{xx}$ :  $\Sigma_{xx} = SS^T$
  - ► Let  $\mathbf{W} = \mathbf{S}^{\dagger}$  and  $\mathbf{z}(t) = \mathbf{W}\mathbf{x}(t)$
  - ► Then  $\mathbf{z}(t)$  is white  $(\mathbb{E}[\mathbf{z}(t)] = \mathbf{0}$  and  $\mathbf{\Sigma}_{zz} = \mathbf{W}\mathbf{\Sigma}_{xx}\mathbf{W}^T = \mathbf{I})$



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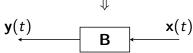
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#### Whitening

- $\blacktriangleright$  We conclude without loss of generality that  $\mathbf{U} \triangleq \mathbf{W}\mathbf{A}$  is a rotation matrix ( $\mathbf{U}\mathbf{U}^T = \mathbf{I}$ ).
- ► Then  $\mathbf{y}(t) = \mathbf{U}^T \mathbf{z}(t) = \mathbf{U}^T \mathbf{W} \mathbf{x}(t) = (\mathbf{W} \mathbf{A})^{-1} (\mathbf{W} \mathbf{A}) \mathbf{s}(t) = \mathbf{s}(t)$ .
- ightharpoonup We can thus assume  $\mathbf{B} = \mathbf{U}^T \mathbf{W}$  where  $\mathbf{U}$  is a rotation matrix.

$$\begin{array}{c|c}
\mathbf{y}(t) & \mathbf{z}(t) & \mathbf{w} \\
\downarrow & & \downarrow \\
\end{array}$$



## **Higher order statistics**

- $\triangleright$  One can estimate  $\Sigma_{xx}$  from the observations and get **W**
- ▶ The whiteness property (second order cumulants) determines W and leaves U unknown.
- ▶ If sources are Gaussian, the  $z_k$  are independent and **U** cannot be determined.
- ▶ In order to determine rotation **U**, we need to exploit the non-Gaussianity of sources and characterize the independence property by using cumulants of order greater than 2.





#### **Contrast functions**

- ▶ Definition:  $\phi$  is a "contrast function" if and only if  $\phi[\mathbf{C}\mathbf{s}(t)] \ge \phi[\mathbf{s}(t)] \ \forall \mathbf{C}$  and if  $\phi[\mathbf{C}\mathbf{s}(t)] = \phi[\mathbf{s}(t)] \Leftrightarrow \mathbf{C}$  is non-mixing.
- ► Separation is performed by minimizing  $\phi[\mathbf{y}(t) = \mathbf{C}\mathbf{s}(t)]$  with respect to  $\mathbf{U}$  (or  $\mathbf{B}$ )
- ▶ "Canonical" contrast function:  $\phi_{IM}[\mathbf{y}(t)] = \mathbb{I}[\mathbf{y}(t)]$
- ► Orthogonal contrasts: to be minimized under the constraint  $\mathbb{E}[\mathbf{y}(t)\mathbf{y}(t)^T] = \mathbf{I}$ . For instance,  $\phi_{IM}^{\circ}[\mathbf{y}(t)] = \sum_{k=1}^{K} \mathbb{H}(y_k(t))$
- ▶ Order 4 approximation of  $\phi_{IM}^{\circ}$ :  $\phi_{ICA}^{\circ}[\mathbf{y}(t)] = \sum_{ijkl \neq iiii} (\kappa_{ijkl}^{4}[\mathbf{y}(t)])^{2}$
- **D**escent algorithms for minimizing  $\phi$  with respect to **B** or **U**:
  - ► Gradient algorithm applied to matrix **B**
  - ► Parameterization of **U** with Givens rotations and coordinate descent



- 1. Estimation of the covariance matrix  $\Sigma_{xx}$
- 2. Diagonalization of  $\Sigma_{xx}$ :  $\Sigma_{xx} = \mathbf{Q} \mathbf{\Lambda}^2 \mathbf{Q}^T$  where  $\mathbf{\Lambda} = \operatorname{diag}(\lambda_1 \dots \lambda_M)$  with  $\lambda_1 \geq \dots \geq \lambda_M \geq 0$
- 3. Computation of  $\mathbf{S} = \mathbf{Q}_{(:,1:K)} \mathbf{\Lambda}_{(1:K,1:K)}$
- 4. Computation of the whitening matrix  $\mathbf{W} = \mathbf{S}^{\dagger}$
- 5. Data whitening:  $\mathbf{z}(t) = \mathbf{W}\mathbf{x}(t)$
- 6. Estimation of **U** by minimizing the contrast function  $\phi^{\circ}$
- 7. Estimation of source signals via  $\mathbf{y}(t) = \mathbf{U}^T \mathbf{z}(t)$





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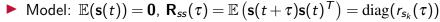
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# Temporal coherence of sources

#### Part V

Second order methods



- ightharpoonup Canonical problem: we assume that  $\Sigma_{ss} = R_{ss}(0) = I$
- ▶ We first aim to spatially whiten the mixture:
  - Let **S** be a matrix square root of  $Σ_{xx}$
  - ► Let  $\mathbf{W} = \mathbf{S}^{\dagger}$  and  $\mathbf{z}(t) = \mathbf{W}\mathbf{x}(t)$
- ▶ Since  $\Sigma_{xx} = \mathbf{A} \mathbf{A}^T$ ,  $\mathbf{U} \triangleq \mathbf{W} \mathbf{A}$  is a rotation matrix
- ▶ However,  $\forall \tau \in \mathbb{Z}$ ,  $\mathbf{R}_{zz}(\tau) = \mathbf{U}\mathbf{R}_{ss}(\tau)\mathbf{U}^T$
- The joint diagonalization of matrices  $\mathbf{R}_{zz}(\tau)$  for various values of  $\tau$  permits us to identify rotation  $\mathbf{U}$





23/48

#### Joint diagonalization

SOBI algorithm

- Unicity theorem :
  - Let a set of matrices  $\mathbf{R}_{zz}(\tau)$  of dimension  $K \times K$  and of the form  $\mathbf{R}_{zz}(\tau) = \mathbf{U}\mathbf{R}_{ss}(\tau)\mathbf{U}^T$  with  $\mathbf{U}$  unitary and  $\mathbf{R}_{ss}(\tau) = \mathrm{diag}(r_{s_k}(\tau))$ . Then  $\mathbf{U}$  is unique (up to a non-mixing matrix) if and only if  $\forall 1 \leq k \neq l \leq K$ , there is  $\tau$  such that  $r_{s_k}(\tau) \neq r_{s_l}(\tau)$
- ▶ Joint diagonalization methods: minimize the criterion

$$J(\mathbf{U}) = \sum_{\tau} \|\mathbf{U}^{T} \mathbf{R}_{zz}(\tau) \mathbf{U} - \operatorname{diag}(\mathbf{U}^{T} \mathbf{R}_{zz}(\tau) \mathbf{U})\|_{F}^{2}$$

► Parameterization of **U** with Givens rotations and coordinate descent

- ► Second Order Blind Identification (SOBI)
  - 1. Estimation and diagonalization of  $\Sigma_{xx}$ :  $\Sigma_{xx} = \mathbf{Q} \mathbf{\Lambda}^2 \mathbf{Q}^T$  where  $\mathbf{\Lambda} = \operatorname{diag}(\lambda_1 \dots \lambda_M)$  with  $\lambda_1 \geq \dots \geq \lambda_M \geq 0$
  - 2. Computation of  $S = Q_{(:,1:K)} \Lambda_{(1:K,1:K)}$
  - 3. Computation of the whitening matrix  $\mathbf{W} = \mathbf{S}^{\dagger}$
  - 4. Data whitening:  $\mathbf{z}(t) = \mathbf{W}\mathbf{x}(t)$
  - 5. Estimation of covariance matrices  $\mathbf{R}_{zz}(\tau)$  for various delays  $\tau$
  - 6. Approximate joint diagonalization of matrices  $\mathbf{R}_{zz}(\tau)$  in a common basis  $\mathbf{U}$
  - 7. Estimation of source signals via  $\mathbf{y}(t) = \mathbf{U}^T \mathbf{z}(t)$





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#### Non-stationarity of sources

- ▶ Model:  $\mathbb{E}(\mathbf{s}(t)) = \mathbf{0}$ ,  $\mathbf{\Sigma}_{ss}(t) \triangleq \mathbb{E}(\mathbf{s}(t)\mathbf{s}(t)^T) = \operatorname{diag}(\sigma_k^2(t))$
- ▶ Then  $\forall t \in \mathbb{Z}$ ,  $\mathbf{\Sigma}_{\mathsf{xx}}(t) = \mathbf{A}\mathbf{\Sigma}_{\mathsf{ss}}(t)\mathbf{A}^{\mathsf{T}}$
- ▶ Joint diagonalization methods: minimize the criterion

$$J(\mathbf{B}) = \sum_{t} \|\mathbf{B} \mathbf{\Sigma}_{xx}(t) \mathbf{B}^{T} - \operatorname{diag}(\mathbf{B} \mathbf{\Sigma}_{xx}(t) \mathbf{B}^{T})\|_{F}^{2}$$

- ► Gradient descent algorithm applied to matrix **B**
- In the over-determined case, **B** must be constrained to span the principal subspace of all matrices  $\Sigma_{xx}(t)$
- ► Variant of the SOBI algorithm:
  - 1. Segmentation of source signals and estimation of covariance matrices  $\Sigma_{xx}(t)$  on windows centered at different times t
  - 2. Joint diagonalization of matrices  $\Sigma_{xx}(t)$  in a common basis B
  - 3. Estimation of source signals via  $\mathbf{y}(t) = \mathbf{B}\mathbf{x}(t)$

## Conclusion of the first part

- ► The use of higher order cumulants is only necessary for the non-Gaussian IID source model
- ▶ Second order statistics are sufficient for sources that are:
  - stationary but not IID (→ spectral dynamics)
  - ▶ non stationary (→ temporal dynamics)
- ▶ Remember that classical tools (based on second order statistics) are appropriate for blind separation of independent (and possibly Gaussian) sources, on condition that the spectral / temporal source dynamics is taken into account.





#### Part VI

## Time-frequency methods



- Motivations
  - Spectral and temporal dynamics are highlighted by a time-frequency (TF) representation of signals
  - ► TF representations are appropriate to process convolutive and/or under-determined mixtures
- ▶ Use of a filter bank (examples: STFT, MDCT):
  - $\blacktriangleright$  Decomposition in F sub-bands and decimation of factor  $T \le F$
  - $\blacktriangleright$  Analysis filters  $h_f$  and synthesis filters  $g_f$
  - ► TF representation of sources:  $s_k(f, n) = (h_f * s_k)(nT)$

  - ► TF representation of mixtures:  $x_m(f,n) = (h_f * x_m)(nT)$ ► Perfect reconstruction:  $s_k(t) = \sum_{f=1}^F \sum_{n \in \mathbb{Z}} g_f(t-nT) s_k(f,n)$
- ▶ Then  $\forall f, n, \mathbf{x}(f, n) = \mathbf{A}\mathbf{s}(f, n)$  (same linear instantaneous mixture)





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#### Non-stationary source model

- Assumption: independent and centered second order processes
  - Model of non-stationary sources:
    - $\blacktriangleright$  if the time frames  $n_1$  and  $n_2$  are disjoint, then  $s_k(., n_1)$  and  $s_k(., n_2)$  are uncorrelated and of distinct variances
  - Model of WSS sources:
    - if sub-bands  $f_1$  and  $f_2$  are disjoint  $(h_{f_1} * \widetilde{h}_{f_2} = 0)$ , then  $s_k(f_1, .)$ and  $s_k(f_2,.)$  are WSS, uncorrelated and of distinct variances  $\sigma_k^2(f_1) = (h_{f_1} * \widetilde{h}_{f_1} * r_{s_k})(0)$  and  $\sigma_k^2(f_2) = (h_{f_2} * \widetilde{h}_{f_2} * r_{s_k})(0)$
  - ► Time-frequency source model:
    - ightharpoonup all  $s_k(f,n)$  are uncorrelated for all n and f, of distinct variances  $\sigma_{k}^{2}(f,n)$  ( $\Rightarrow$  time-frequency dynamics)

## Separation method

► Separation by joint matrix diagonalization:

Let 
$$\Sigma_{ss}(f, n) = \mathbb{E}[\mathbf{s}(f, n)\mathbf{s}(f, n)^T]$$
 and  $\Sigma_{xx}(f, n) = \mathbb{E}[\mathbf{x}(f, n)\mathbf{x}(f, n)^T]$ 

- ► Then  $\Sigma_{xx}(f,n) = \mathbf{A}\Sigma_{ss}(f,n)\mathbf{A}^T$  where  $\Sigma_{ss}(f,n) = \operatorname{diag}(\sigma_{k}^{2}(f,n))$
- ► Variant of the SOBI algorithm:
  - 1. TF analysis of the mixtures:  $x_k(f,n) = (h_f * x_k)(nT)$
  - 2. Estimation of covariance matrices  $\Sigma_{xx}(f,n)$
  - 3. Joint diagonalization of matrices  $\Sigma_{xx}(f,n)$  in a basis **B**
  - 4. Estimation of the source signals via  $\mathbf{y}(f,n) = \mathbf{B}\mathbf{x}(f,n)$
  - 5. TF synthesis of the sources:  $y_k(t) = \sum_{f=1}^F \sum_{n \in \mathbb{Z}} g_f(t - nT) y_k(f, n)$





31/48

#### Part VII

#### Convolutive mixtures





- Instantaneous mixture model unsuitable for real acoustic mixtures
- ▶ Let  $\mathbf{x}_k(f,n) \in \mathbb{R}^M$  be the **image** of source  $s_k(f,n)$ 
  - received multichannel signal if only source  $s_k(f, n)$  was active
- ► Mixture model:  $\mathbf{x}(f,n) = \sum_{k=1}^{K} \mathbf{x}_k(f,n)$
- ▶ Decomposition of the source separation problem
  - **separation**: estimate  $\mathbf{x}_k(f,n)$  from the mixture  $\mathbf{x}(f,n)$
  - **deconvolution**: estimate  $s_k(f, n)$  from  $\mathbf{x}_k(f, n)$
- Mixture model:  $x_m(t) = \sum_{k=1}^K (a_{mk} * s_k)(t)$ , i.e.  $\mathbf{x}(t) = \mathbf{A} * \mathbf{s}(t)$
- ► Theorem (identifiability)
  - Let  $s_k$  be K IID sources, among which at most one is Gaussian, and  $\mathbf{y}(t) = \mathbf{C} * \mathbf{s}(t)$  with  $\mathbf{C}$  invertible ((over)-determined case). If signals  $y_k(t)$  are independent, then  $\mathbf{C}$  is non-mixing.



33/48

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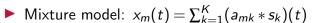


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#### Time-frequency approach



- Assumptions:
  - ▶ the filter bank corresponds to an STFT
  - ightharpoonup the IR of  $a_{mk}$  is short compared with the window length
  - $ightharpoonup \forall m, k, f, a_{mk}(v)$  varies slowly compared with  $h_f(v)$
  - $ightharpoonup \Rightarrow (h_f * a_{mk})(t) \approx a_{mk}(f) h_f(t)$
- ► Approximation of the convolutive mixture model:

$$x_m(f,n) = \sum_{k=1}^K a_{mk}(f) s_k(f,n)$$

- Matrix form:  $\mathbf{x}(f,n) = \mathbf{A}(f)\mathbf{s}(f,n)$ 
  - F instantaneous mixture models in every sub-band
  - ▶ ⇒ we can use any ICA method in every sub-band

## Independent component analysis

- ▶ Let  $\mathbf{y}(f,n) = \mathbf{B}(f)\mathbf{x}(f,n)$ , where  $\mathbf{B}(f) \in \mathbb{C}^{K \times M}$
- ▶ Linear separation is feasible if A(f) has rank K:
  - ▶ We get  $\mathbf{y}(f, n) = \mathbf{s}(f, n)$  by defining:
    - ▶  $\mathbf{B}(f) = \mathbf{A}(f)^{-1}$  in the determined case (M = K)
    - ▶  $\mathbf{B}(f) = \mathbf{A}(f)^{\dagger}$  in the over-determined case (M > K)
- ▶ In the under-determined case (M < K), separation remains impossible
- ▶ In practice matrix  $\mathbf{A}(f)$  is unknown:
  - We look for B(f) that makes the  $y_k(f,n)$  independent (ICA)
  - We then get  $\mathbf{y}(f,n) = \mathbf{C}(f)\mathbf{s}(f,n)$ , where  $\mathbf{C}(f) = \mathbf{B}(f)\mathbf{A}(f)$
  - ightharpoonup C(f) is non-mixing







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#### **Indeterminacies**

- ▶ Problem: indeterminacies (permutations and multiplicative factors) in matrices C(f)
  - $\blacktriangleright$   $\forall k$ , identify indexes  $k_f$  such that  $\forall f$ ,  $y_{k_f}(f,n) = c_{k_f,k} s_k(f,n)$
  - ightharpoonup identify the multiplicative factors  $c_{k_f,k}$
- ▶ Infinitely many solutions ⇒ need to constrain the model:
  - Assumptions on the mixture
    - ightharpoonup continuity of the frequency responses  $a_{mk}(f)$  with respect to f
    - ▶ → beamforming model and anechoic model
  - Assumptions on the sources
    - $\blacktriangleright$  similarity of the temporal dynamics of  $\sigma_k^2(f,n)$



- **Beamforming** model:
  - Assumptions: plane waves, far field, no reverberation, linear antenna
  - ► Model:  $a_{mk}(f) = e^{-2i\pi f \tau_{mk}}$  where  $\tau_{mk} = \frac{d_m}{c} \sin(\theta_k)$
  - Parameters: positions  $d_m$  of the sensors and angles  $\theta_k$  of the sources
- ► Anechoic model:
  - Assumptions: punctual sources, no reverberation
  - Model:  $a_{mk}(f) = \alpha_{mk}e^{-2i\pi f \tau_{mk}}$  where  $\alpha_{mk} = \frac{1}{\sqrt{4\pi r_{mk}}}$  and  $\tau_{mk} = \frac{r_{mk}}{c}$
  - $\triangleright$  Parameters: distances  $r_{mk}$  between the sensors and sources





7/48 Une école de l'IMT

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#### Part VIII

#### Under-determined mixtures

### **Under-determined convolutive mixtures**

- ▶ Usual case in audio: monophonic (M = 1) or stereophonic (M = 2) mixtures, with a number of sources K > M
- ► Convolutive mixture model:  $\mathbf{x}(f,n) = \mathbf{A}(f)\mathbf{s}(f,n)$  with M < K
- Assumption: the mixture model  $\mathbf{A}(f)$  and the source model  $\mathbf{\Sigma}_{ss}(f,n)$  are known
- Even in this case, the exact separation is not feasible, because there is no matrix  $\mathbf{B}(f)$  such that  $\mathbf{B}(f)\mathbf{A}(f)=\mathbf{I}_K$

$$\mathbf{y}(f,n)$$
  $\mathbf{B}(f)$   $\mathbf{x}(f,n)$   $\mathbf{A}(f)$   $\mathbf{s}(f,n)$ 





## Separation via non-stationary filtering

- Let  $\mathbf{v}(f,n) = \mathbf{B}(f,n)\mathbf{x}(f,n)$  where  $\mathbf{B}(f,n) \in \mathbb{C}^{K \times M}$  depends on n
- ▶ Minimum Mean Square Error (MMSE) estimator: we look for  $\mathbf{B}(f,n)$  which minimizes  $\mathbb{E}[\|\mathbf{y}(f,n)-\mathbf{s}(f,n)\|_2^2]$
- ► Solution:  $\mathbf{B}(f,n) = \mathbf{\Sigma}_{sx}(f,n)\mathbf{\Sigma}_{xx}(f,n)^{-1}$ where  $\mathbf{\Sigma}_{xx}(f,n) = \mathbf{A}(f)\mathbf{\Sigma}_{ss}(f,n)\mathbf{A}(f)^H$  and  $\Sigma_{sx}(f,n) = \Sigma_{ss}(f,n)\mathbf{A}(f)^H((.)^H \text{ denotes the Hermitian}$ conjugate)
- ▶ Remark:  $\mathbf{x}(f,n) = \mathbf{A}(f)\mathbf{y}(f,n)$  (exact reconstruction)
- ► Particular case: monophonic mixtures
  - ▶ Without loss of generality, we define  $\mathbf{A}(f) = [1, ..., 1]$
  - ► We get  $y_k(f, n) = \frac{\sigma_k^2(f, n)}{\sum_{l=1}^K \sigma_l^2(f, n)} x(f, n)$
  - ▶ ⇒ similar to Wiener filtering

### Separation method

- 1. TF analysis of the mixtures:  $x_k(f,n) = (h_f * x_k)(f,nT)$
- 2. Estimation of  $\mathbf{A}(f)$  and  $\sigma_{k}^{2}(f,n)$ 
  - instantaneous mixture model
  - sparse source model
- 3. Computation of  $\mathbf{B}(f,n) = \mathbf{\Sigma}_{ss}(f,n)\mathbf{A}(f)^{H} \left(\mathbf{A}(f)\mathbf{\Sigma}_{ss}(f,n)\mathbf{A}(f)^{H}\right)^{-1}$
- 4. Estimation of the source signals via  $\mathbf{y}(f,n) = \mathbf{B}(f,n)\mathbf{x}(f,n)$
- 5. TF synthesis of the sources:  $y_k(t) = \sum_{f=1}^F \sum_{n \in \mathbb{Z}} g_f(t - nT) y_k(f, n)$





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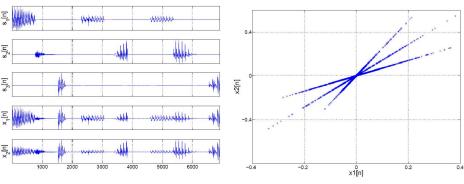
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## Stereophonic mixtures: temporal sparsity

#### Case of a linear instantaneous stereophonic mixture: $\mathbf{x}(t) = \mathbf{A}\mathbf{s}(t)$



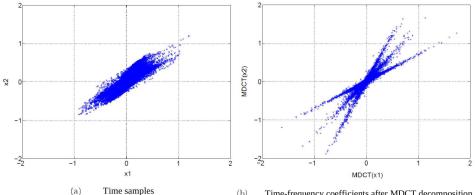
(a) Temporal source signals and corresponding stereo mixture

Dispersion diagrams  $(x_1, x_2)$  over time

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## **Sparsity** in a transformed domain

Case of a linear instantaneous stereophonic mixture:  $\mathbf{x}(f,n) = \mathbf{A}\mathbf{s}(f,n)$ 



Time-frequency coefficients after MDCT decomposition







#### **DUET** method

- ► Degenerate Unmixing Estimation Technique (DUET)
- Linear instantaneous stereophonic mixture model:  $\mathbf{x}(f,n) = \mathbf{A}\mathbf{s}(f,n)$ 
  - ▶ Without loss of generality, we assume  $\mathbf{A}_{(:,k)} = \begin{bmatrix} \cos(\theta_k) \\ \sin(\theta_k) \end{bmatrix} \forall k$
- ► **Sparse** source model:
  - ▶  $\forall f, n, \exists ! \ k_{(f,n)}$  such that  $\sigma^2_{k_{(f,n)}}(f,n) > 0$ , and  $\forall l \neq k_{(f,n)}$ ,  $\sigma^2_l(f,n) = 0$
- ▶ If only source k is active at (f,n), then  $\mathbf{x}(f,n) = \mathbf{a}_k s_k(f,n)$





- 1. TF analysis of the mixtures:  $x_k(f,n) = (h_f * x_k)(nT)$
- 2. Estimation of parameters  $\theta_k$  and of the active source  $k_{(f,n)}$ 
  - $\triangleright$  computation of the histogram of the angles of vectors  $\mathbf{x}(f,n)$
  - $\triangleright$  peak detection in order to estimate parameters  $\theta_k$
  - determination of the active source at (f, n) by proximity with  $\theta_{\nu}$
- 3. Source separation: for all k,
  - estimation of source images via binary masking:  $\mathbf{y}_k(f,n) = \mathbf{x}(f,n) \ \forall \ (f,n)$  such that  $k_{(f,n)} = k$  and  $\mathbf{y}_k(f,n) = 0$  for the other time-frequency bins (f,n)
  - MMSE estimation of the sources:  $y_k(f, n) = \hat{\mathbf{a}}_k(f)^{\dagger} \mathbf{y}_k(f, n)$
- 4. TF synthesis of the sources:

$$y_k(t) = \sum_{f=1}^F \sum_{n \in \mathbb{Z}} g_f(t - nT) y_k(f, n)$$



5/48 Une école de l'IMT

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## Part IX

#### Conclusion

## Conclusion

#### Summary

- Source separation requires to make assumptions about the mixture and sources
- ► For an (over-)determined instantaneous linear mixture, the assumption of independent sources is sufficient
- ► In all other cases, we need to model the mixture and/or the sources
- Perspectives
  - Non-stationary mixtures (adaptive algorithms)
  - ▶ Informed source separation (e.g. from music score)
  - ► Deep learning techniques
  - ▶ Objective assessment of audio source separation







47/48

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