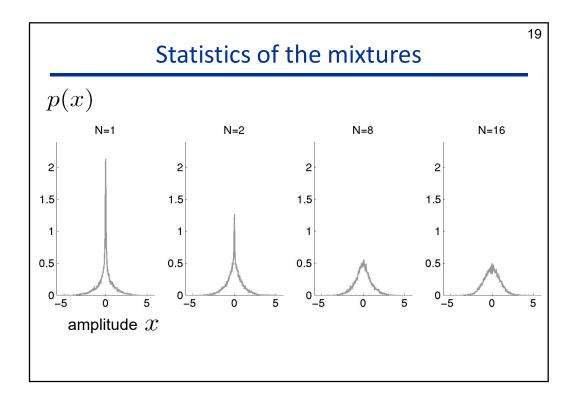
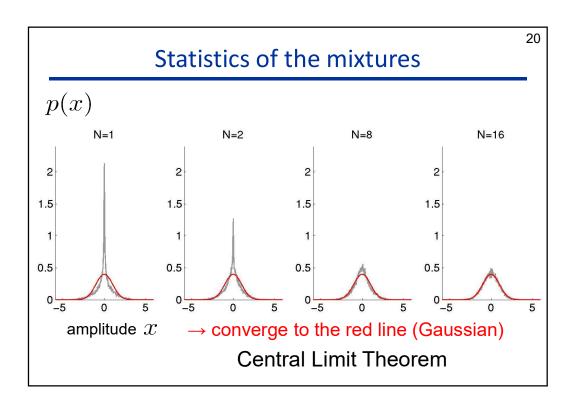
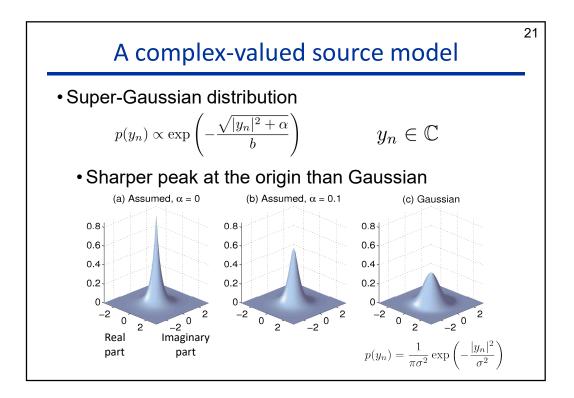


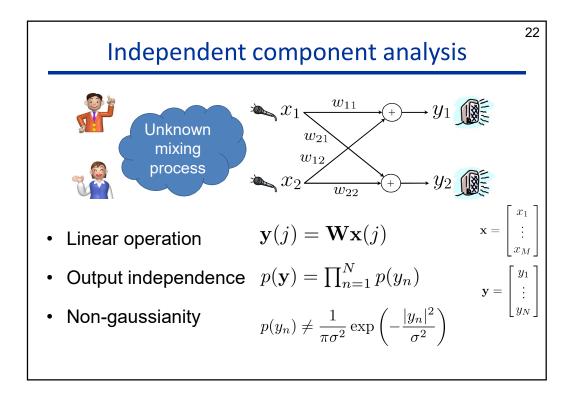
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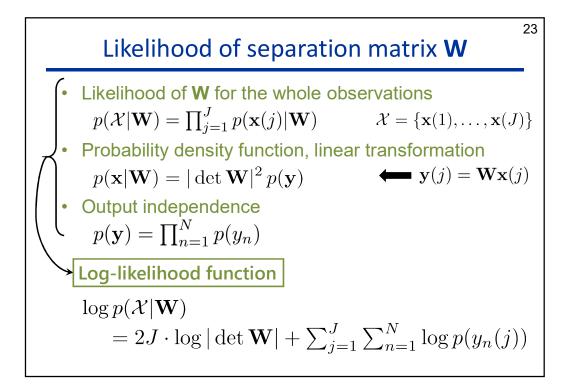


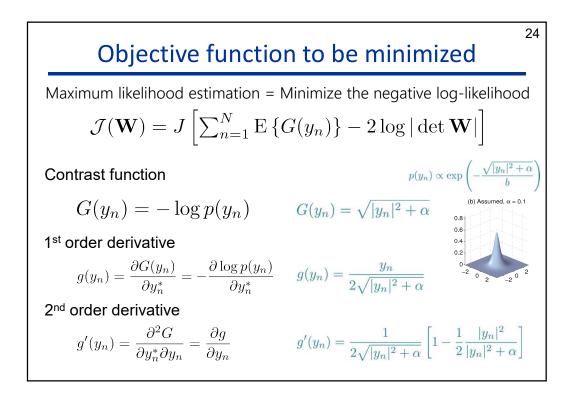


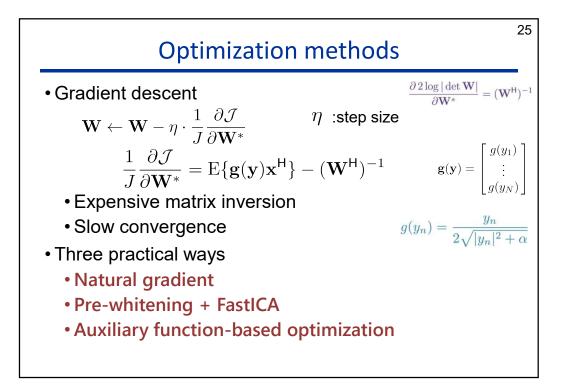
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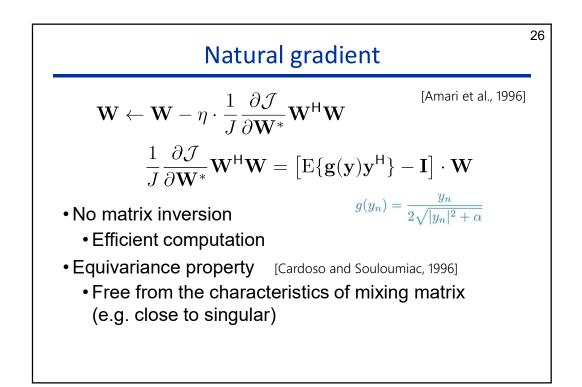


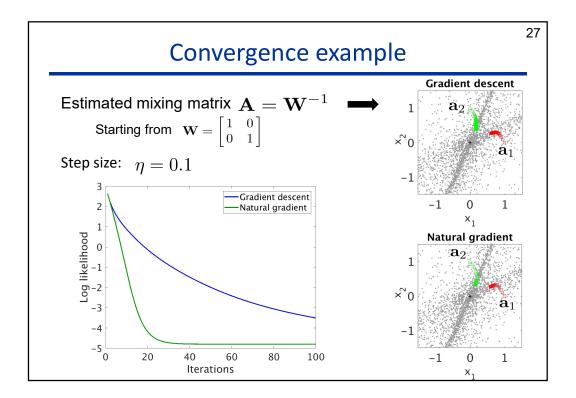


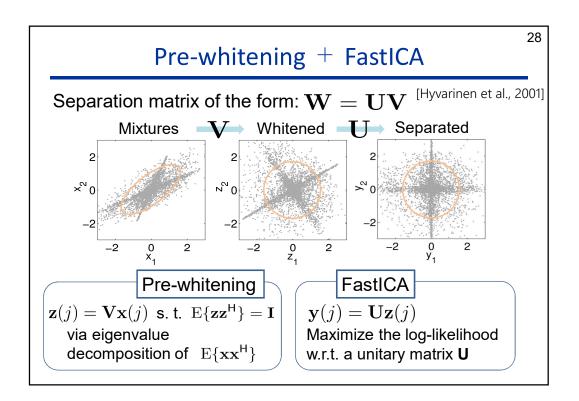


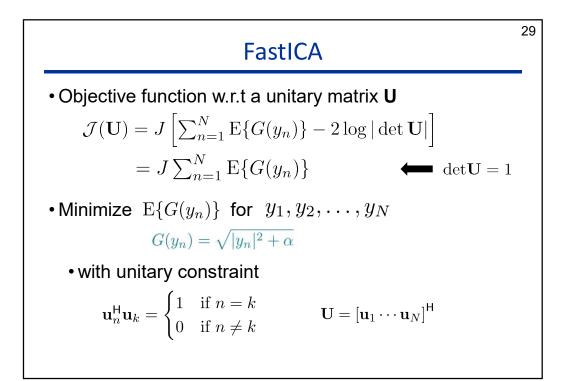


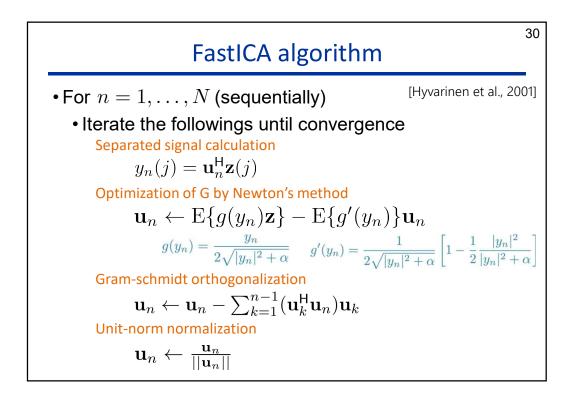


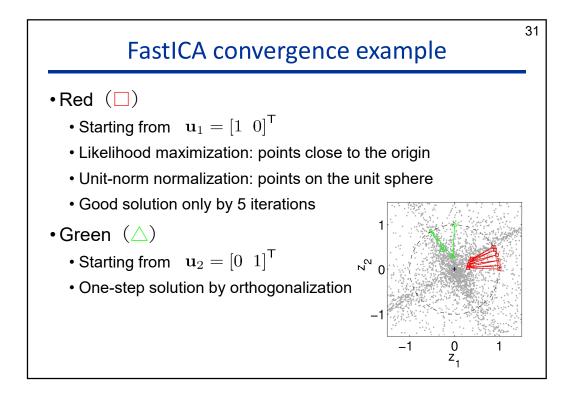


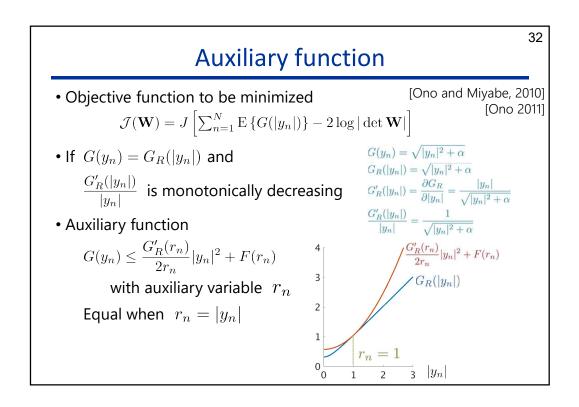


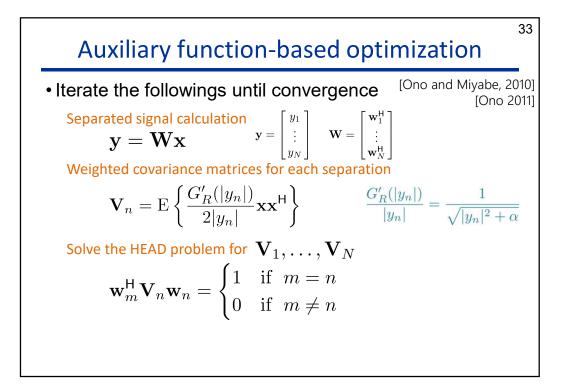


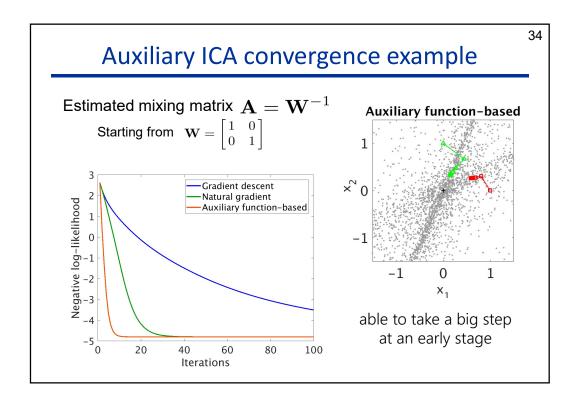




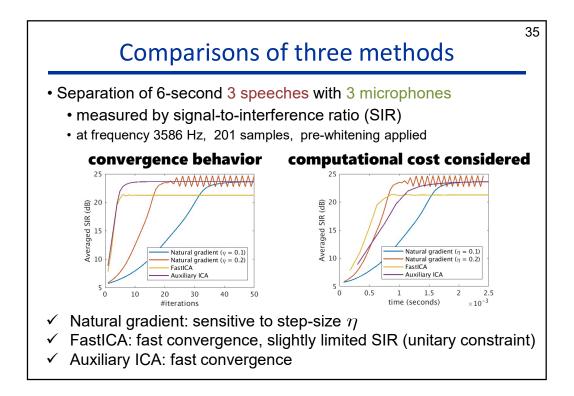


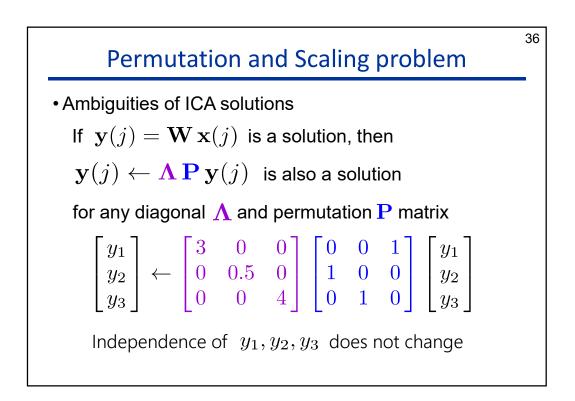


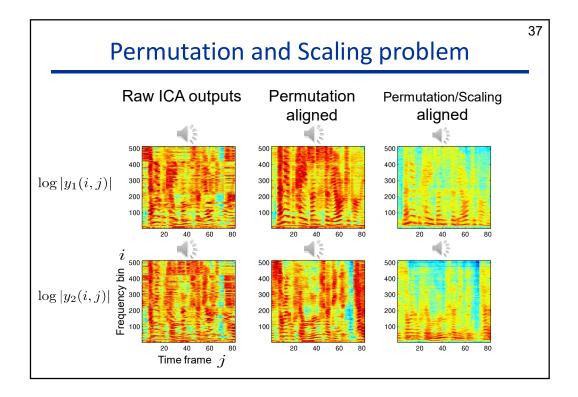


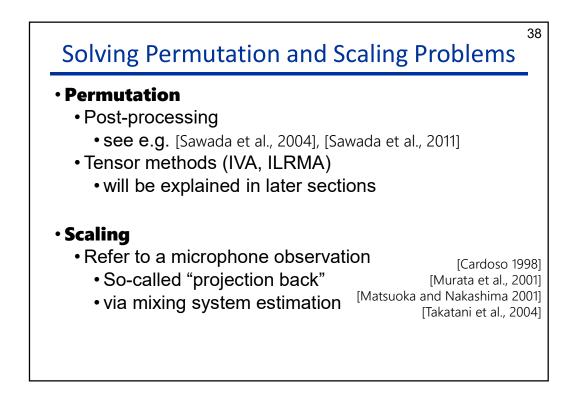


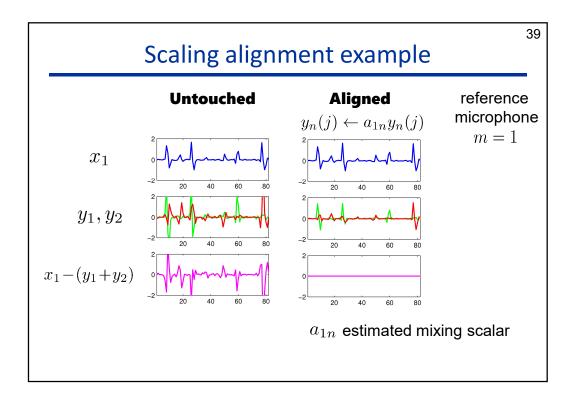
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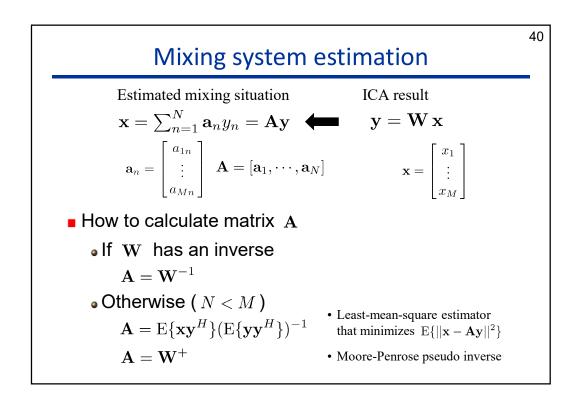


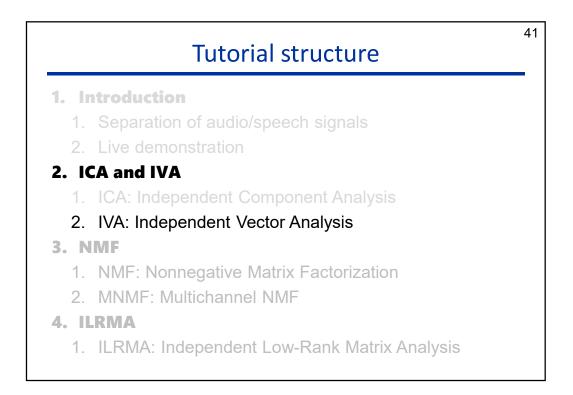


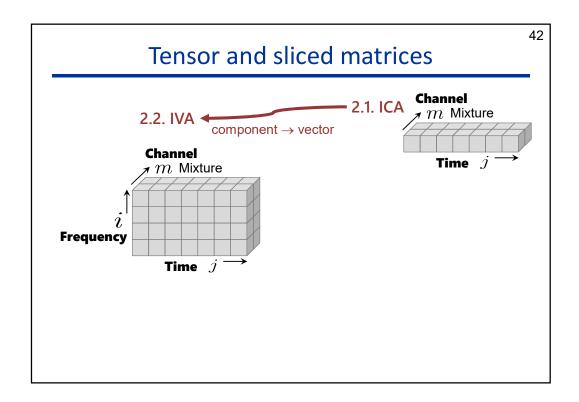


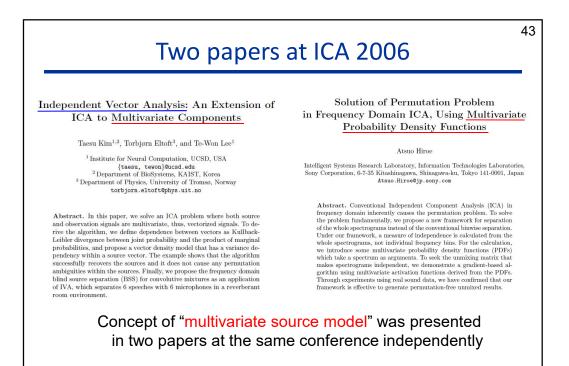


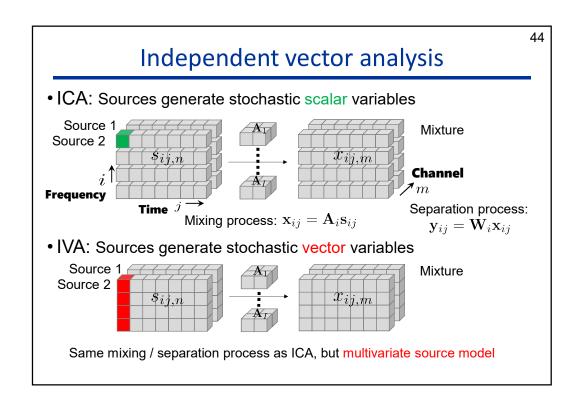


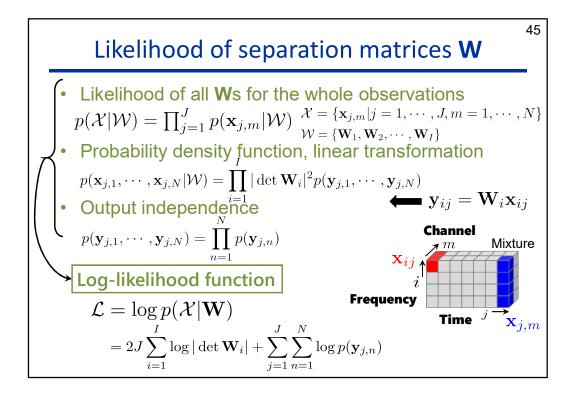


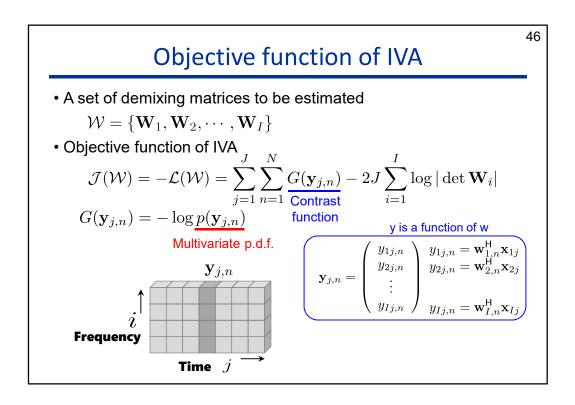


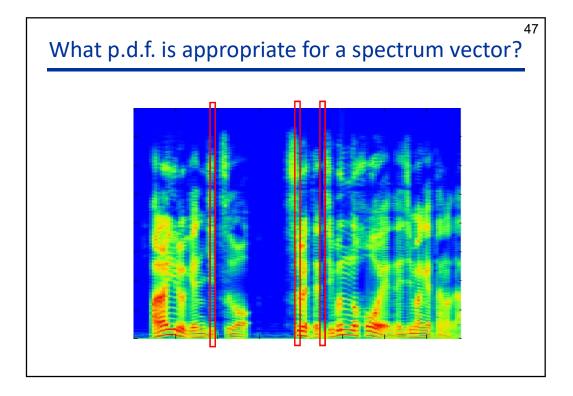


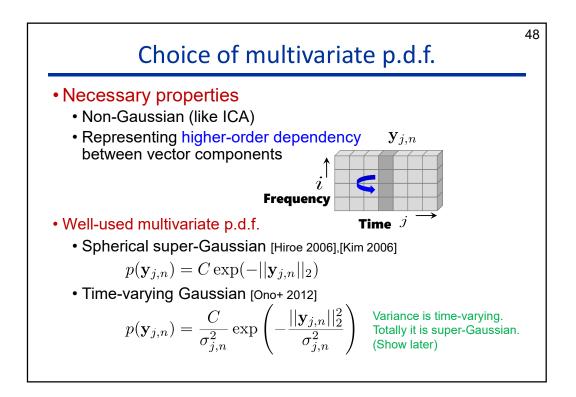


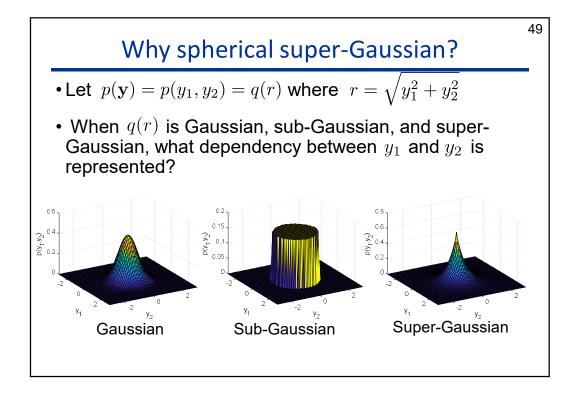


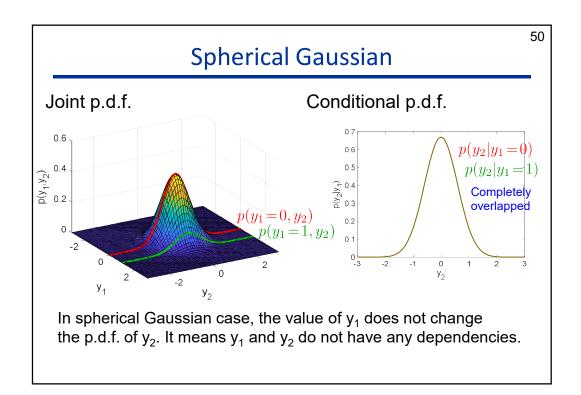


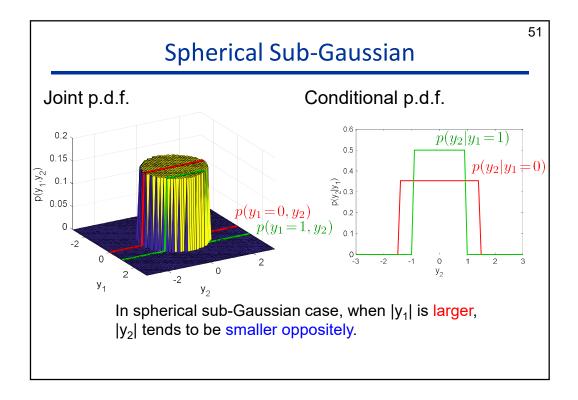


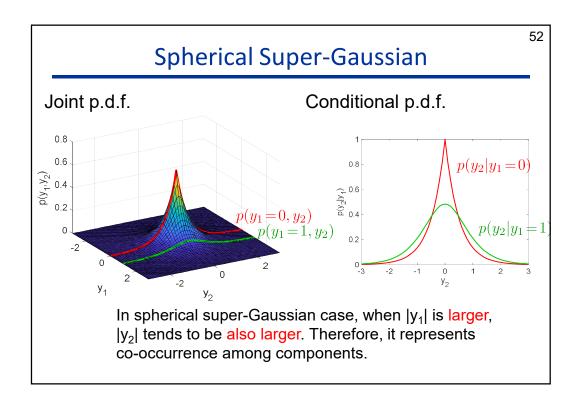




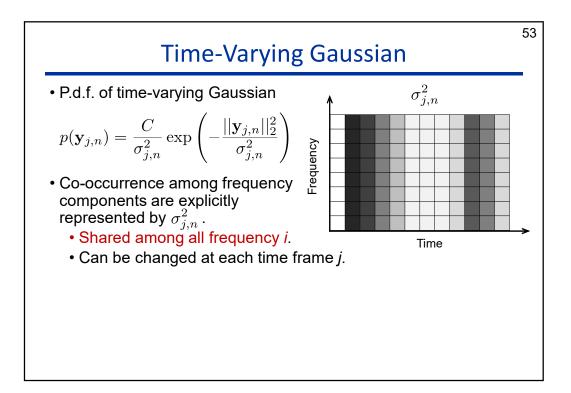


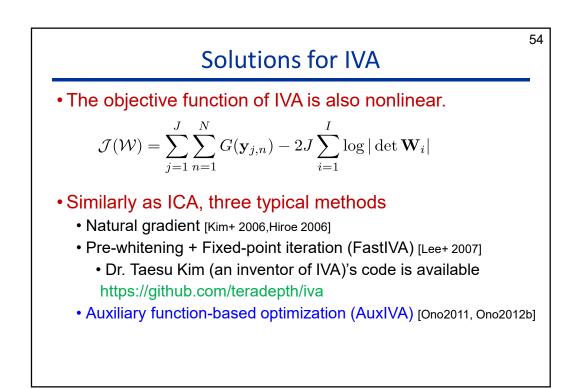


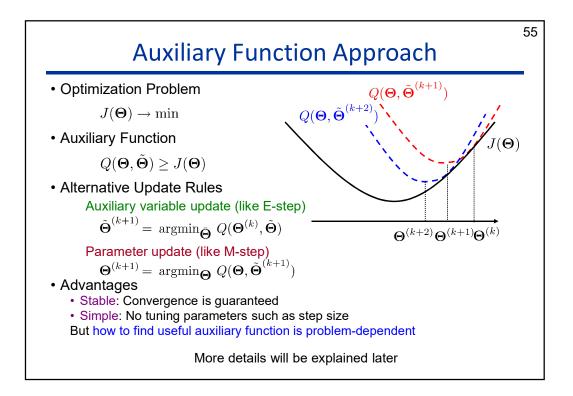


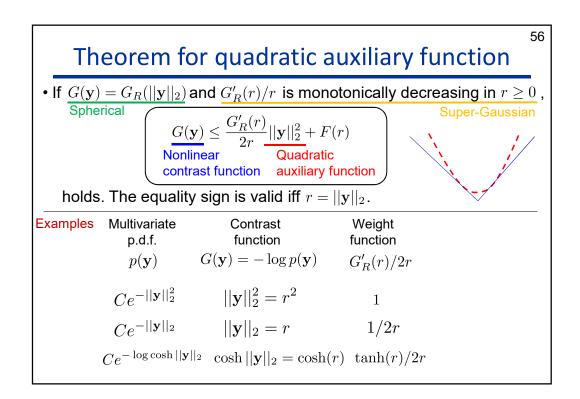


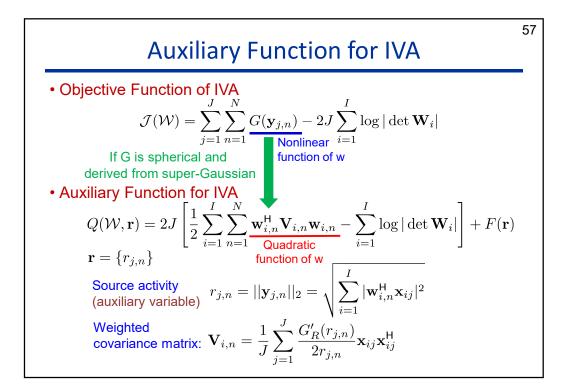
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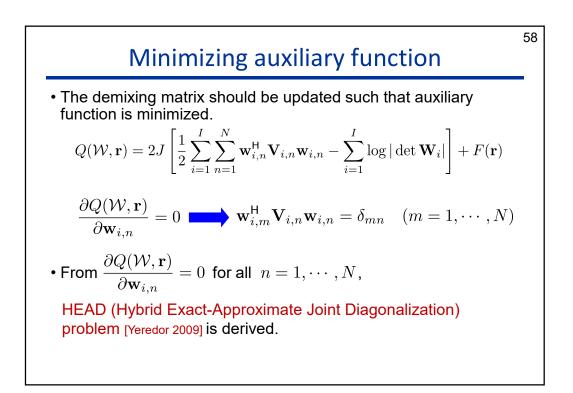


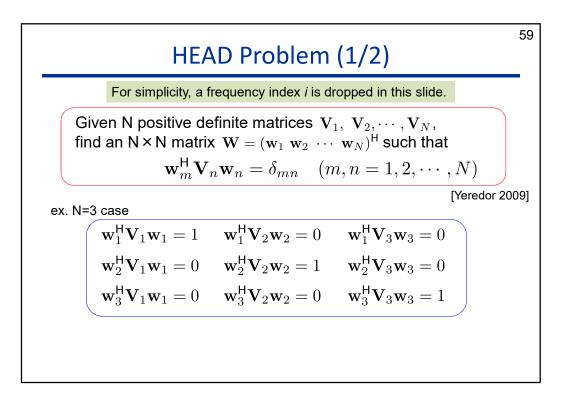


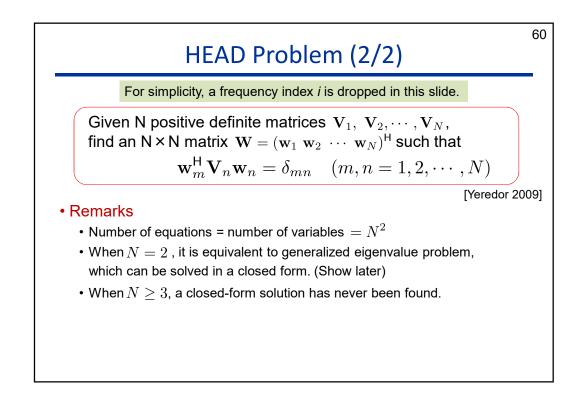


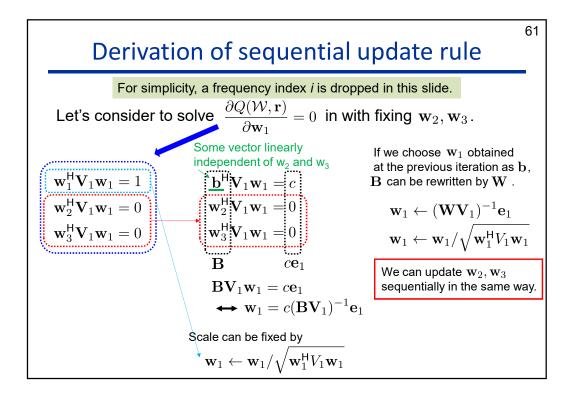


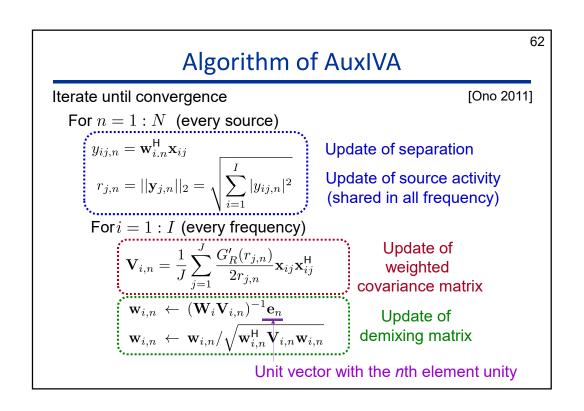


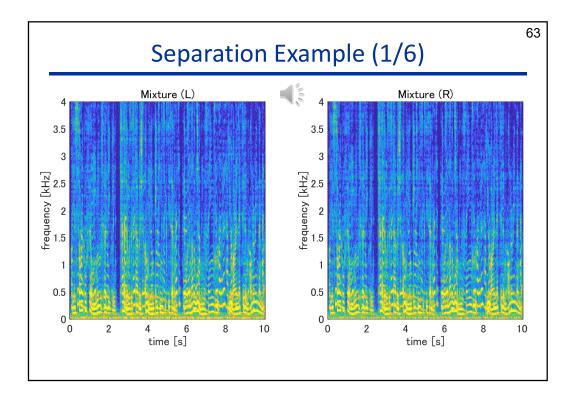


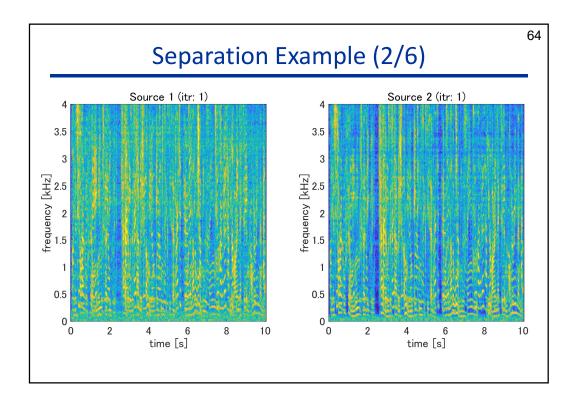




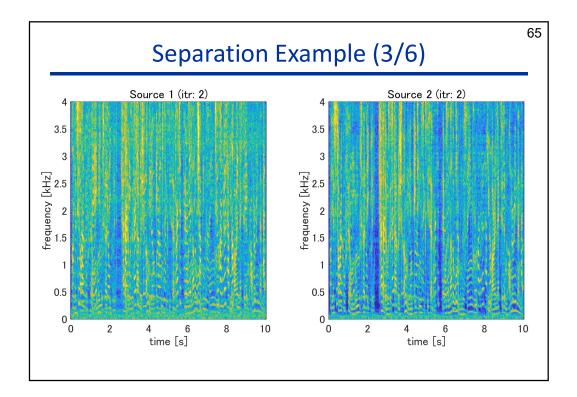


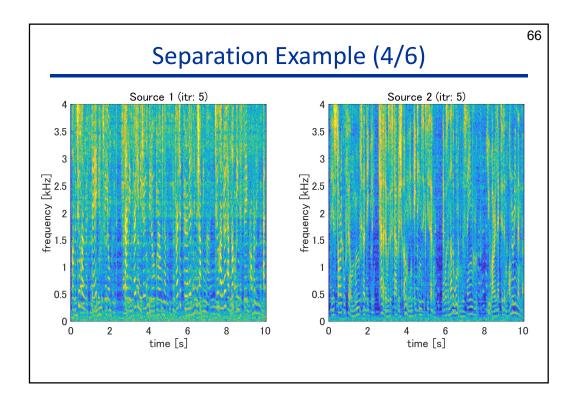




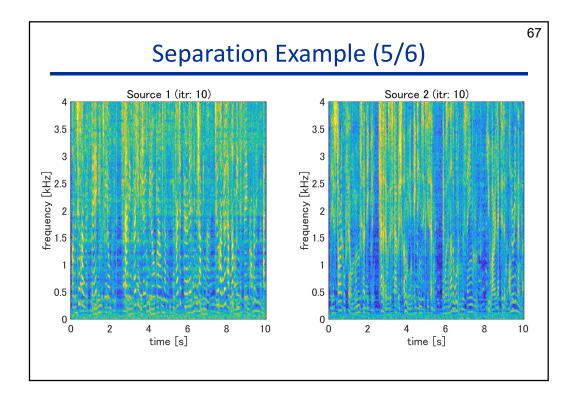


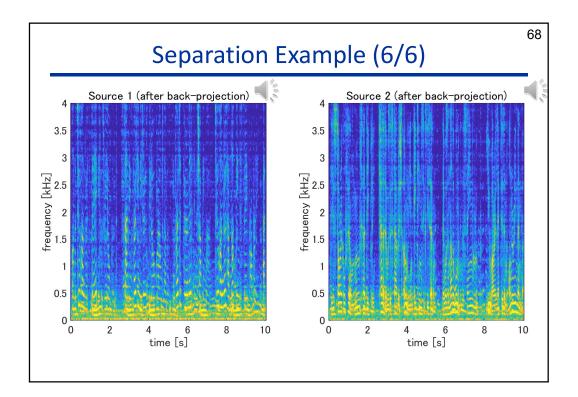
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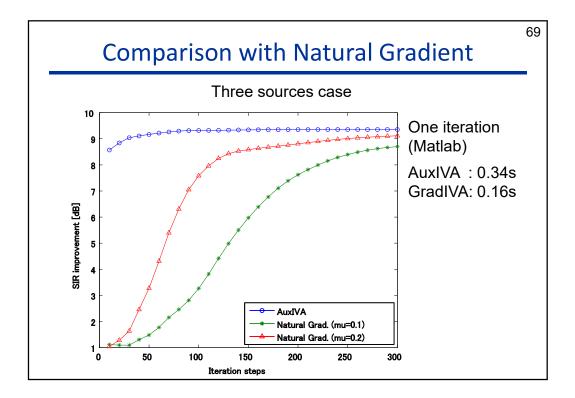


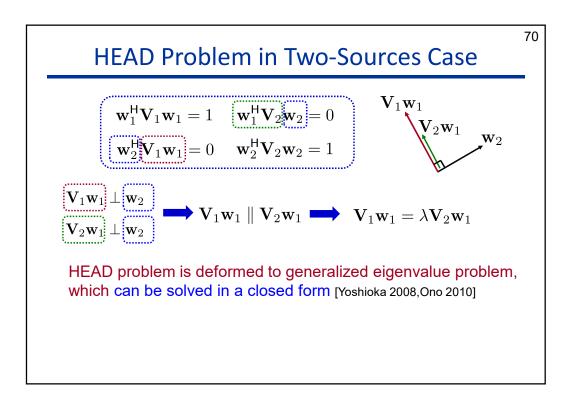
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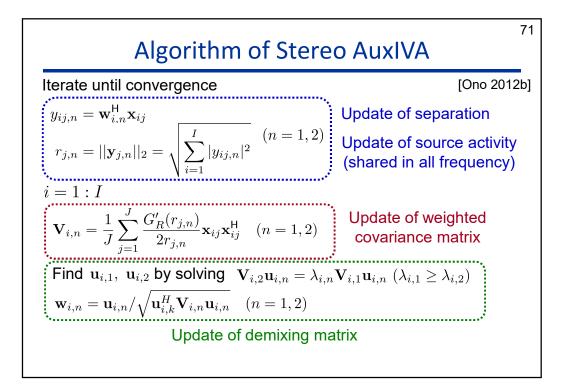


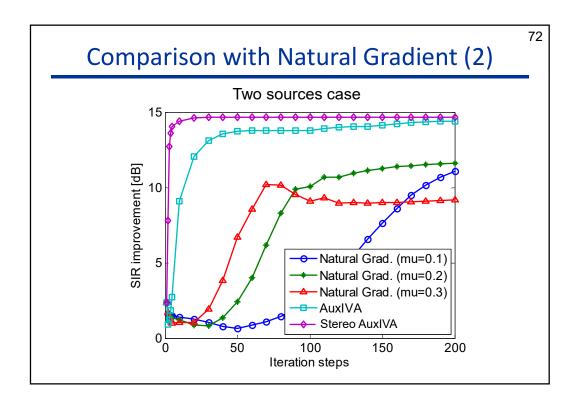


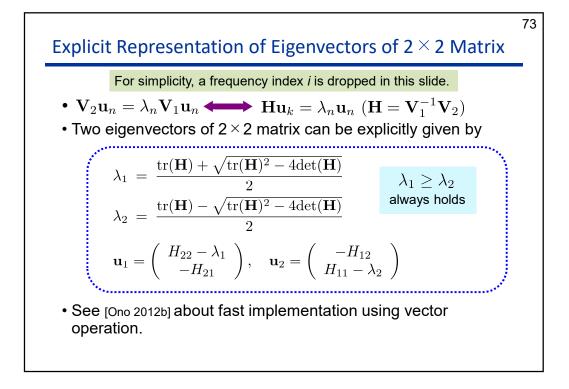
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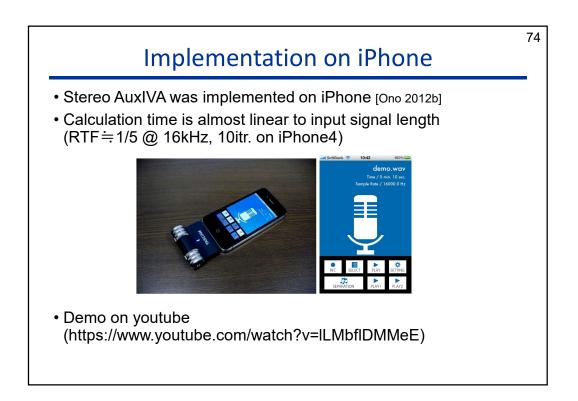


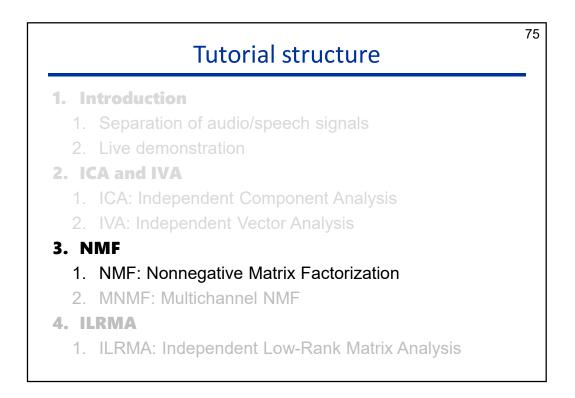


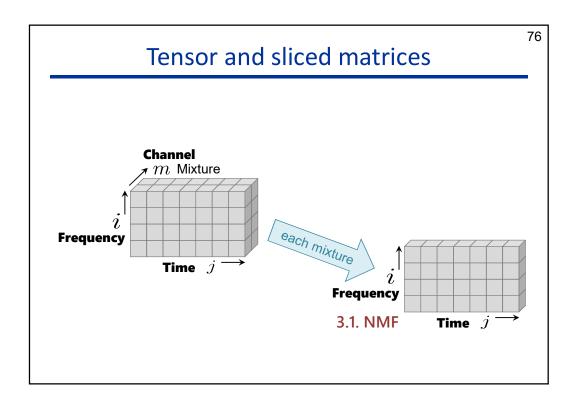


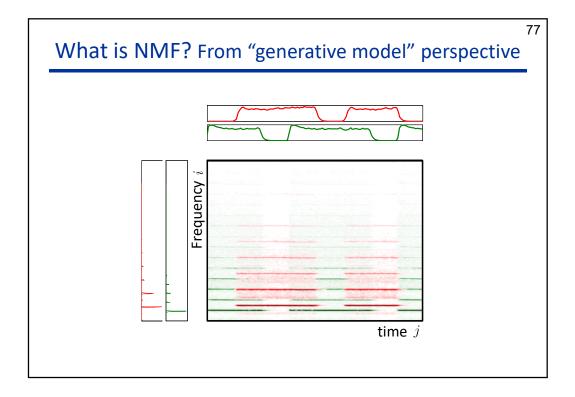


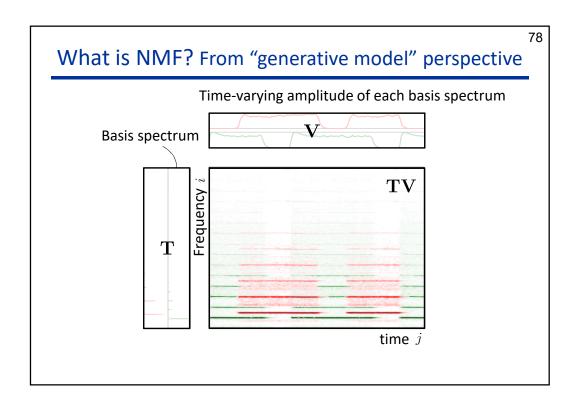


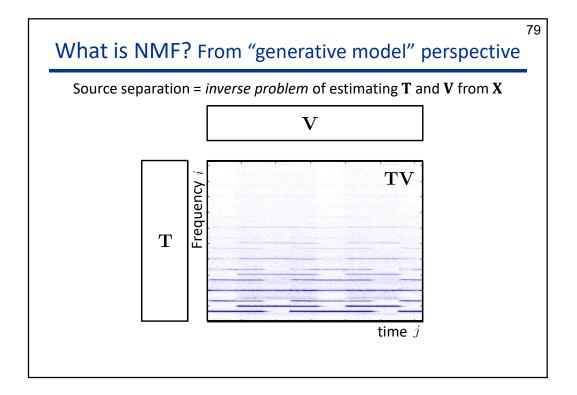


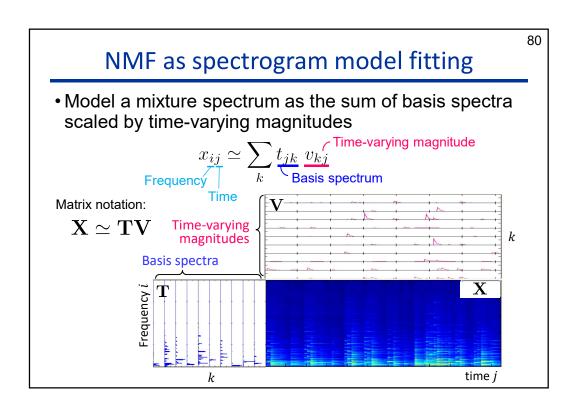


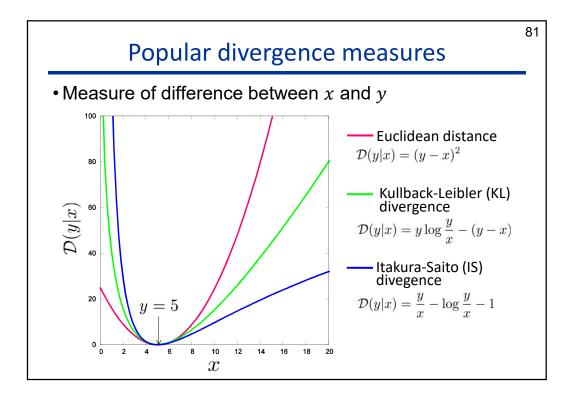


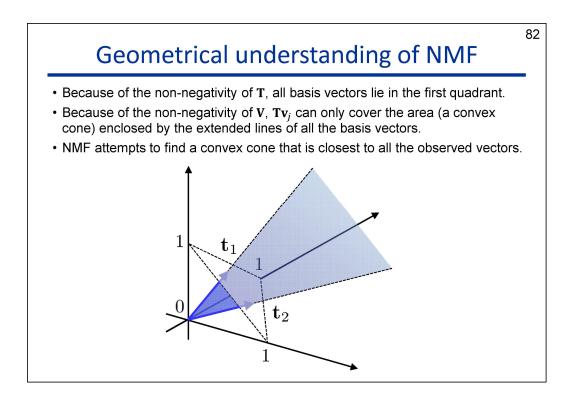


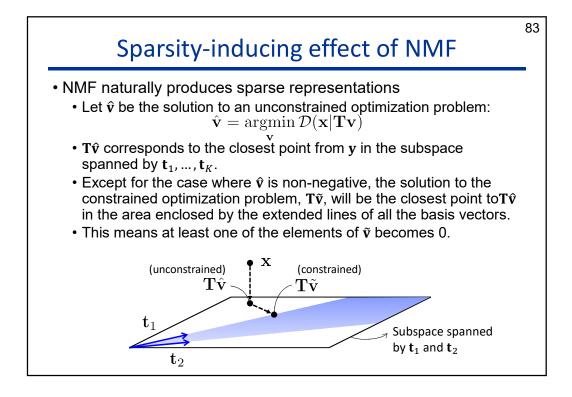


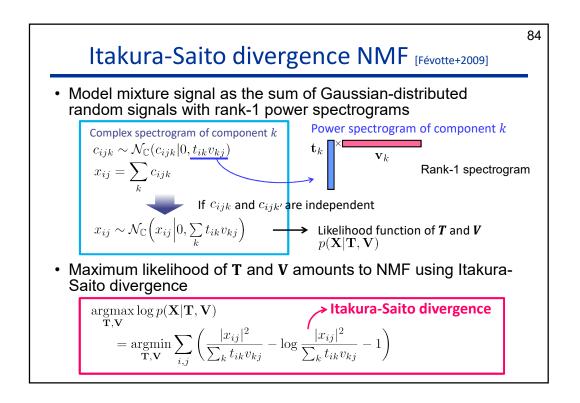


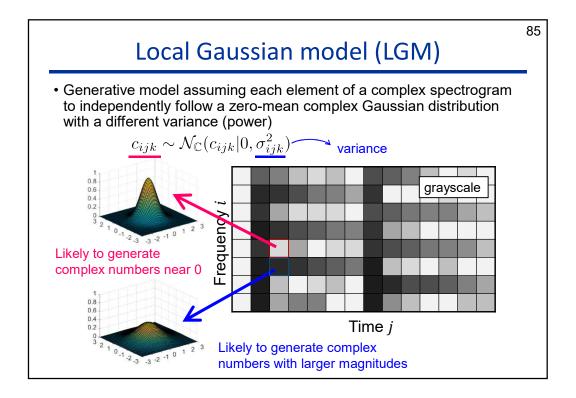


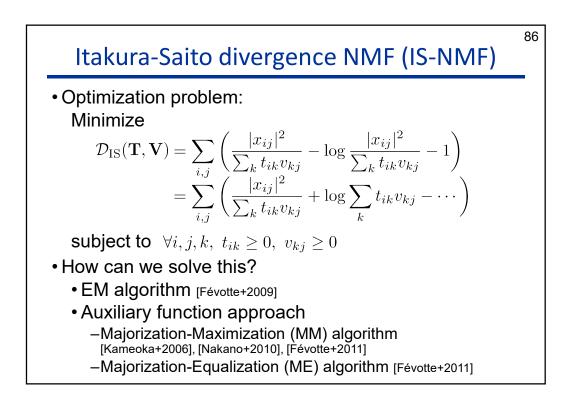


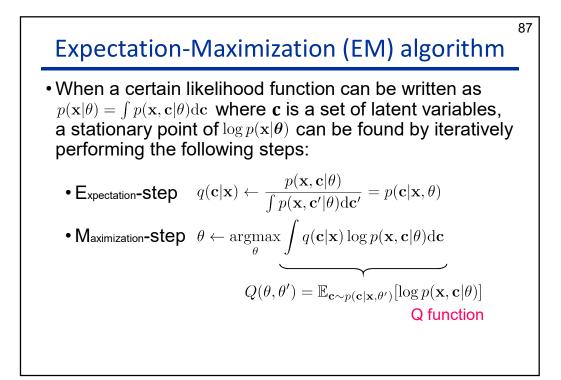


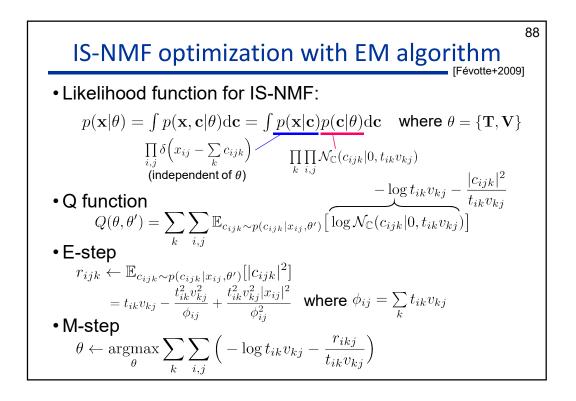


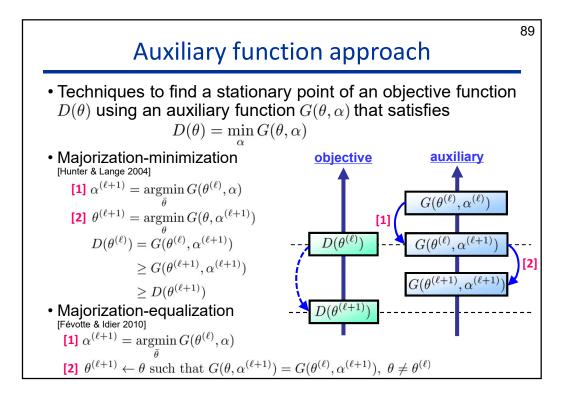


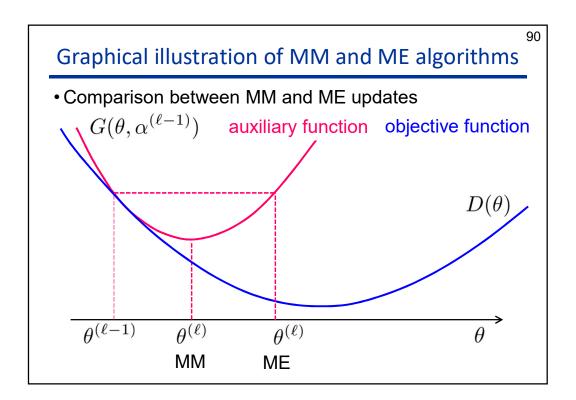


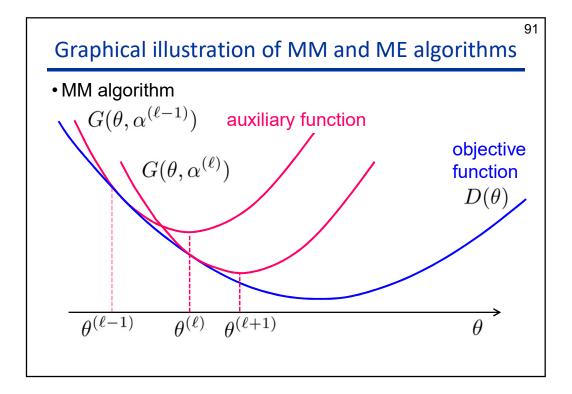


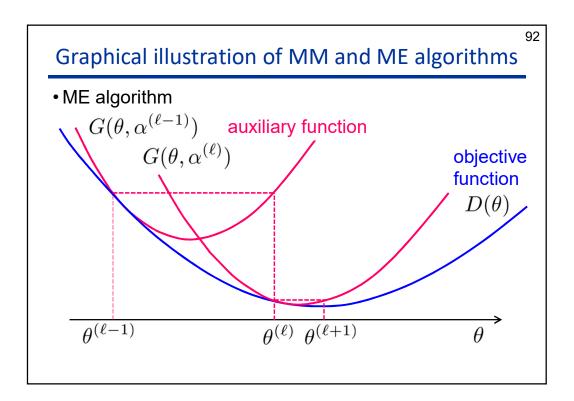


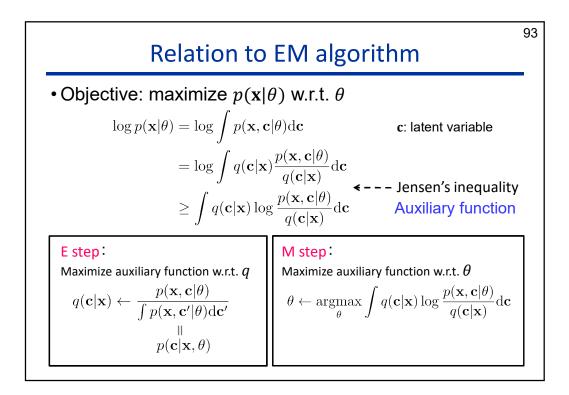


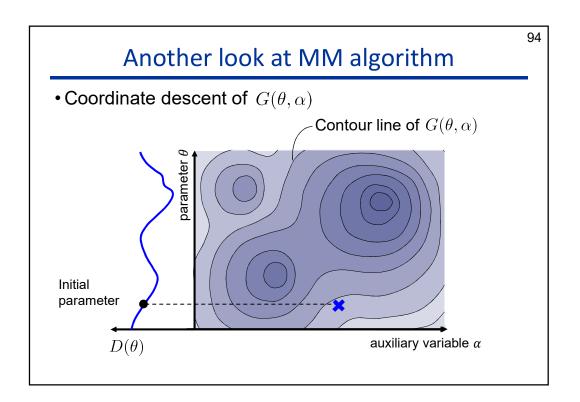


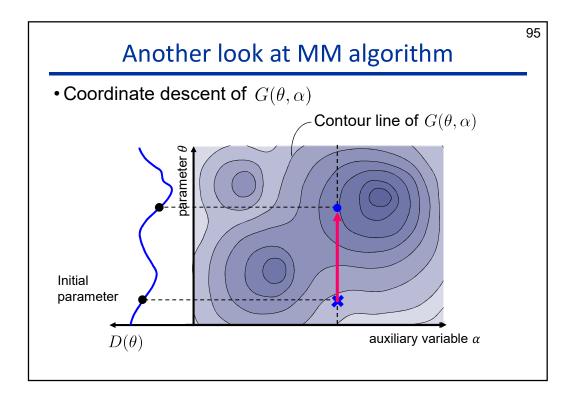


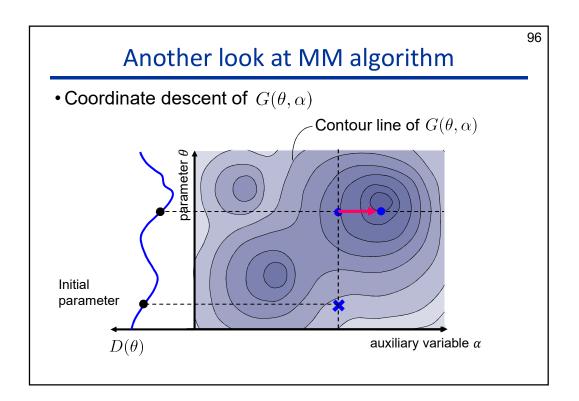


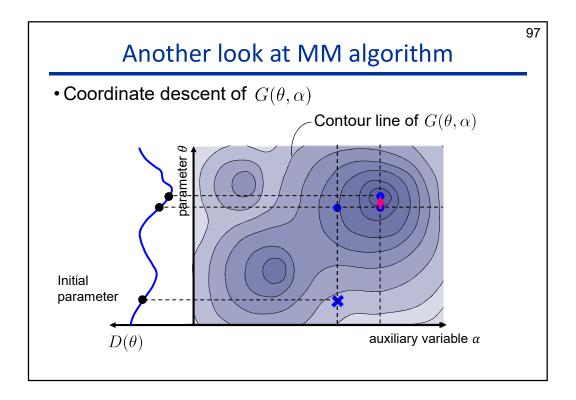


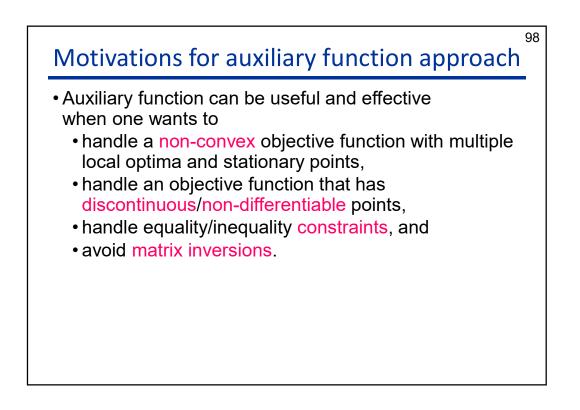


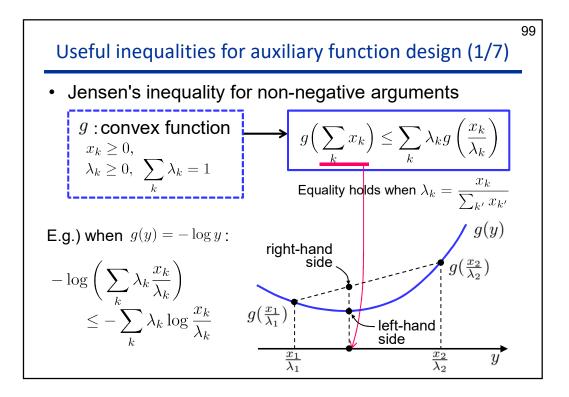


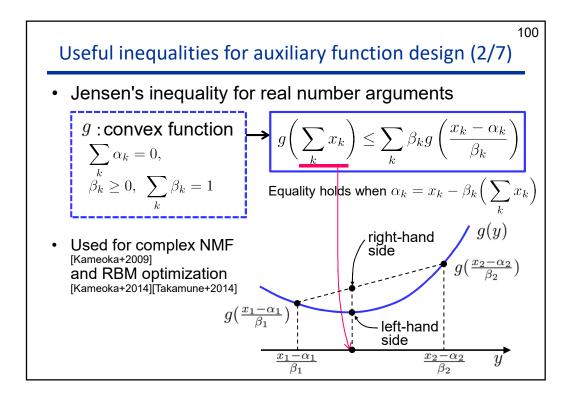


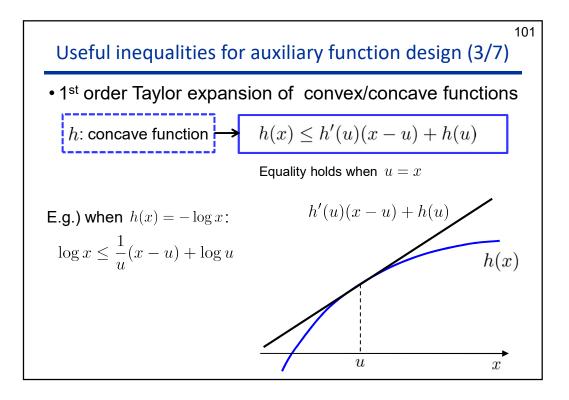


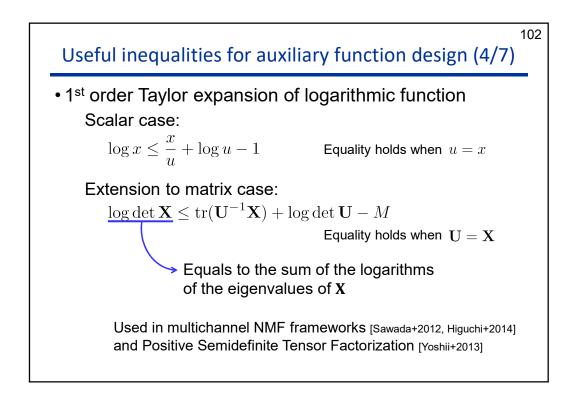


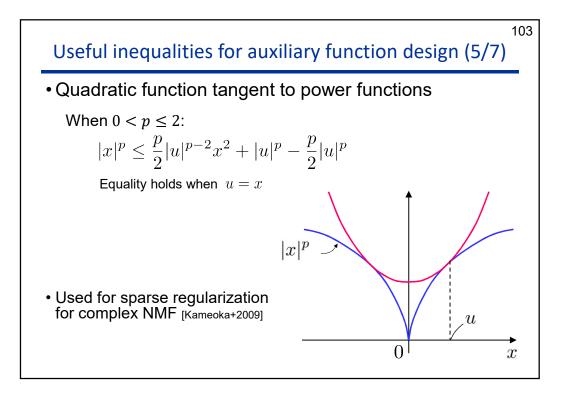


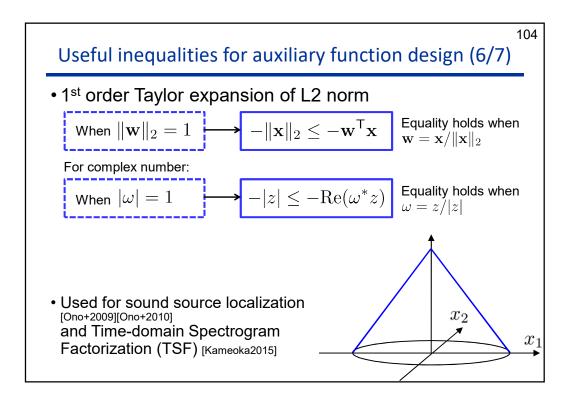


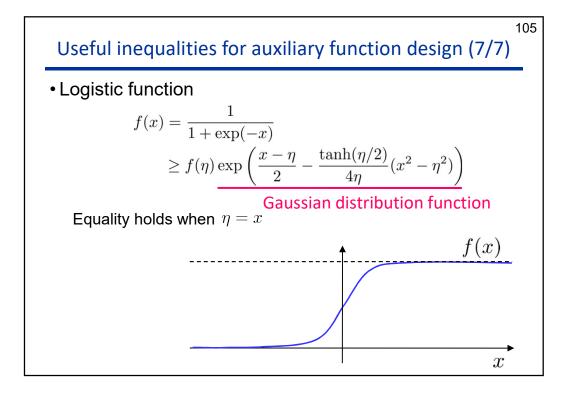


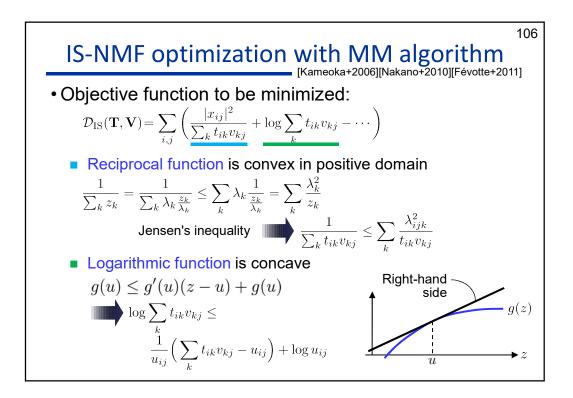


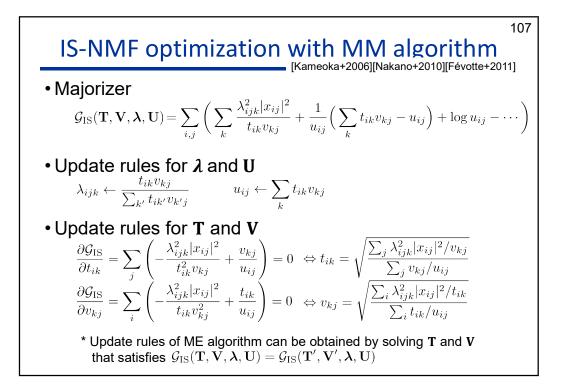


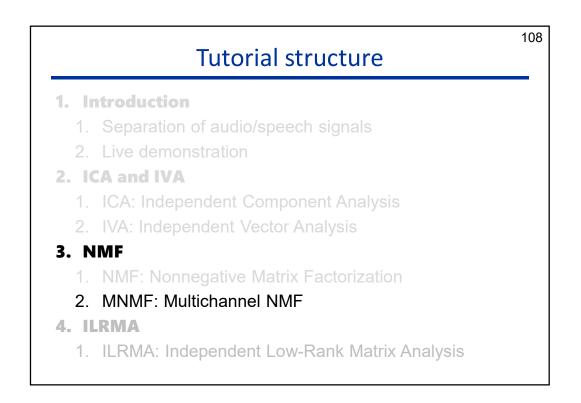


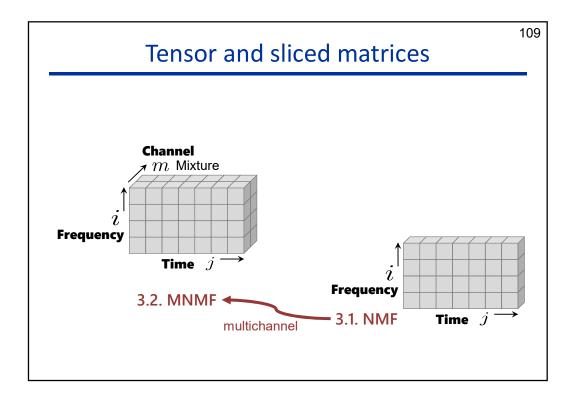


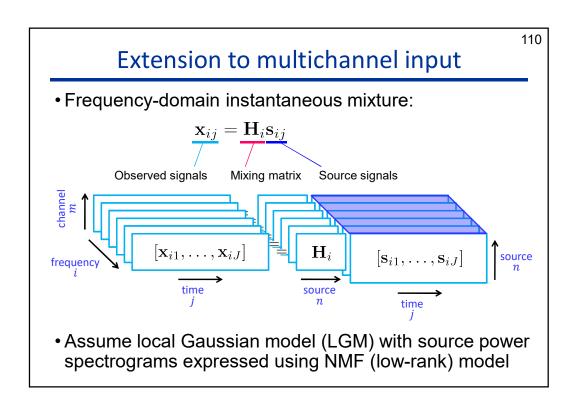




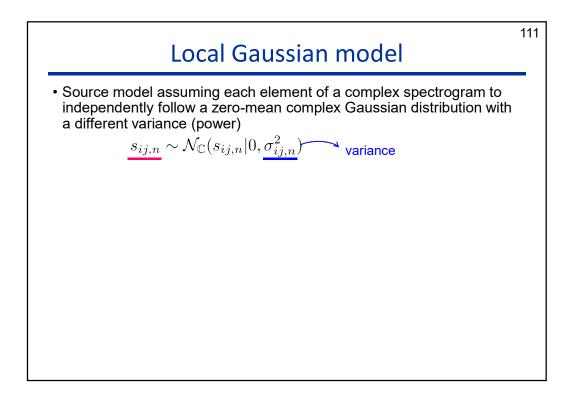


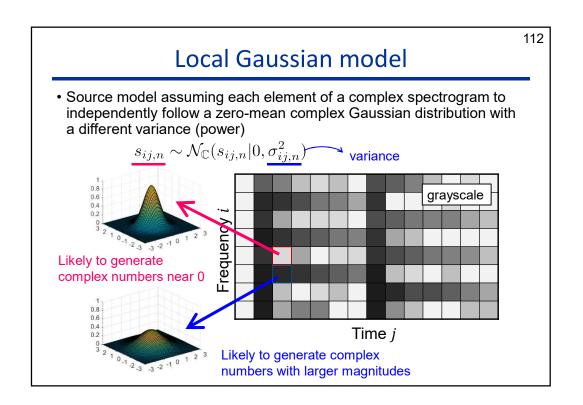


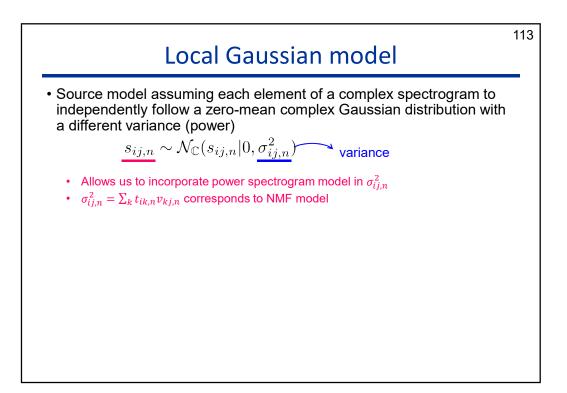


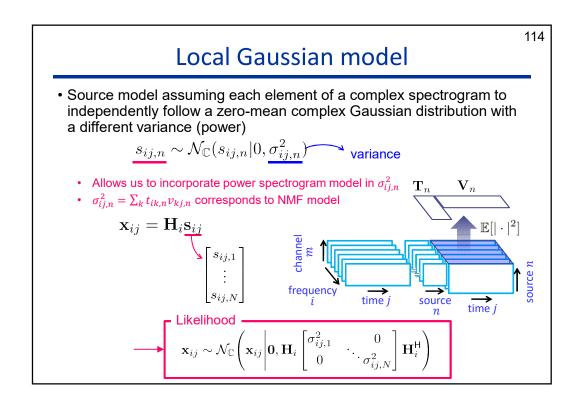


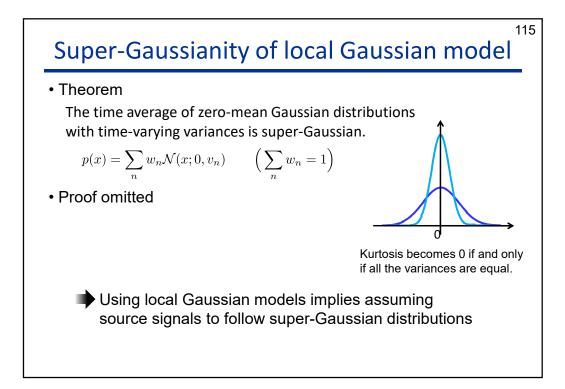
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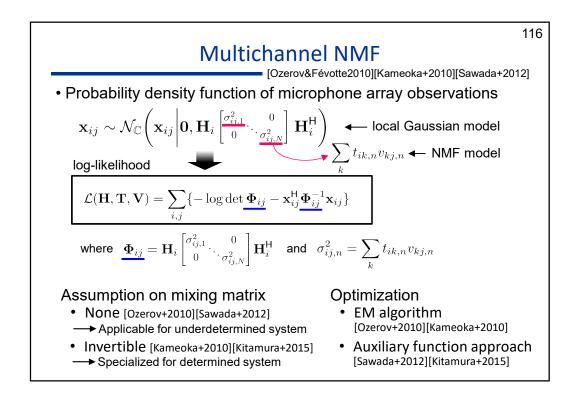


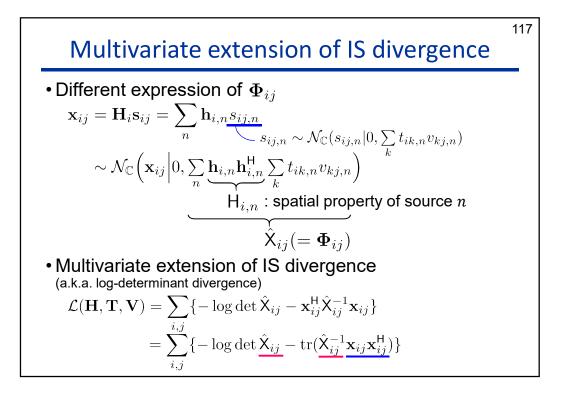


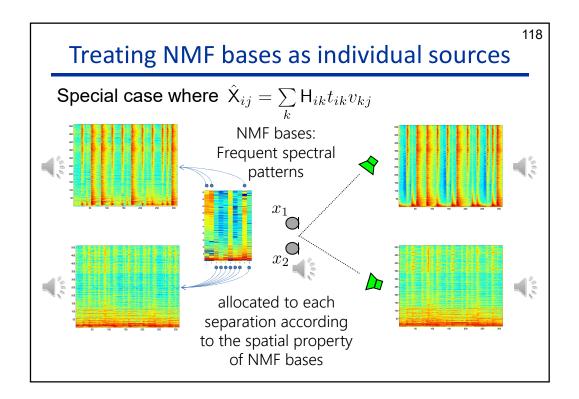


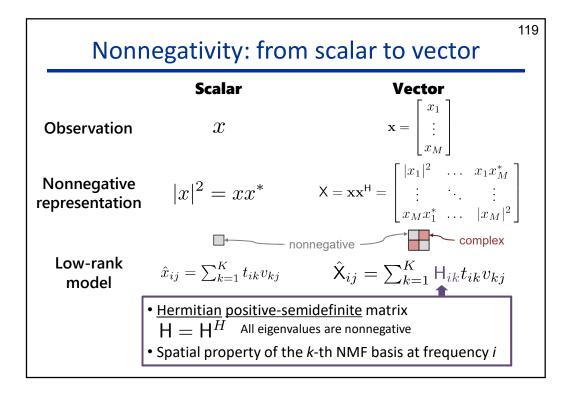


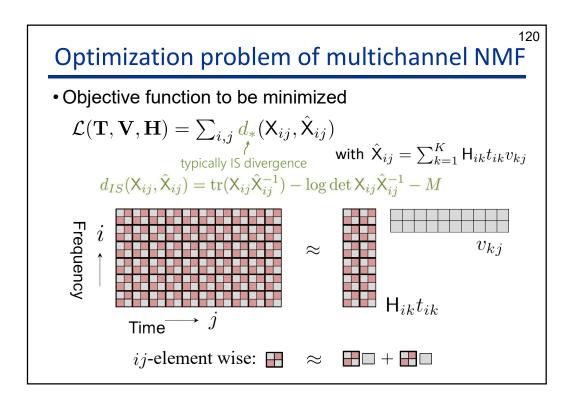


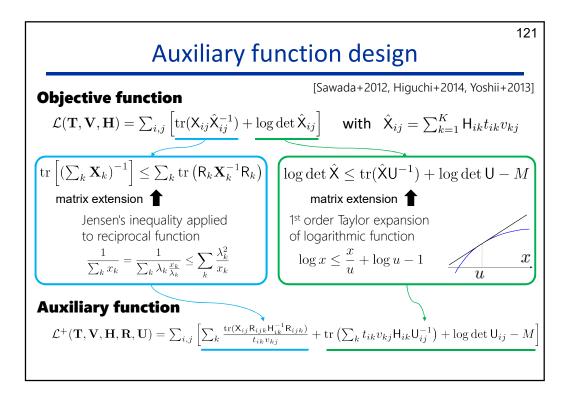


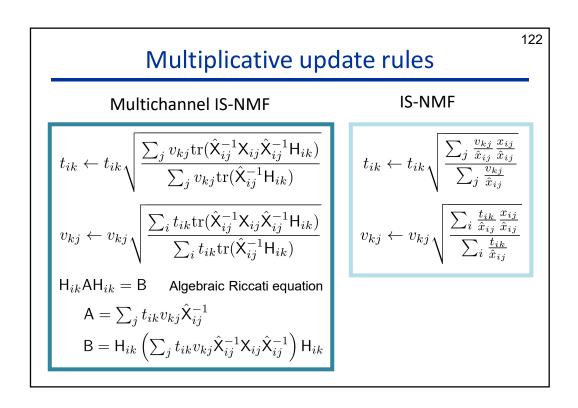




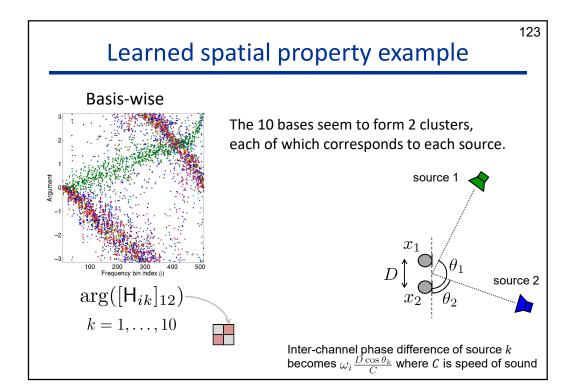


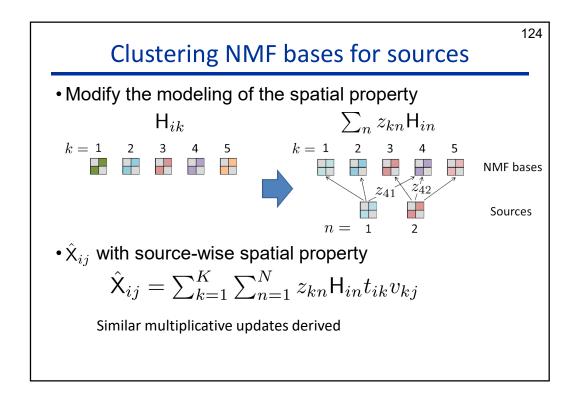


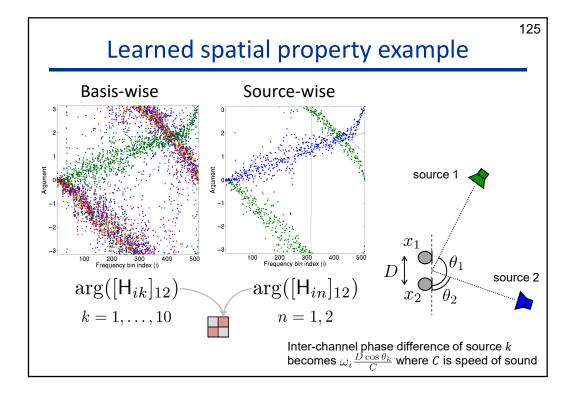


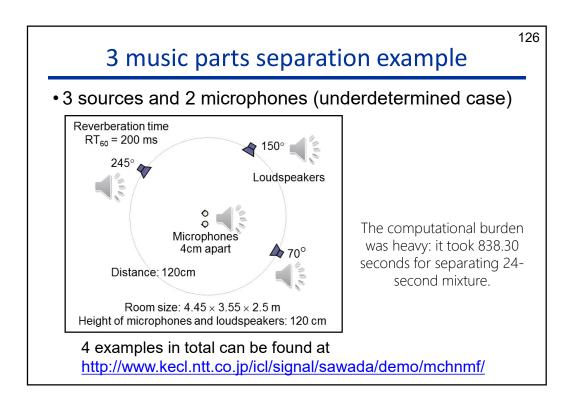


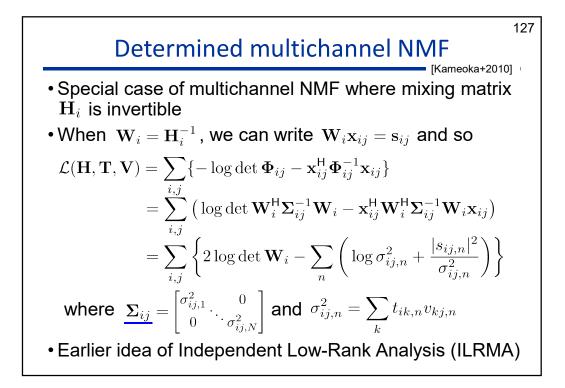
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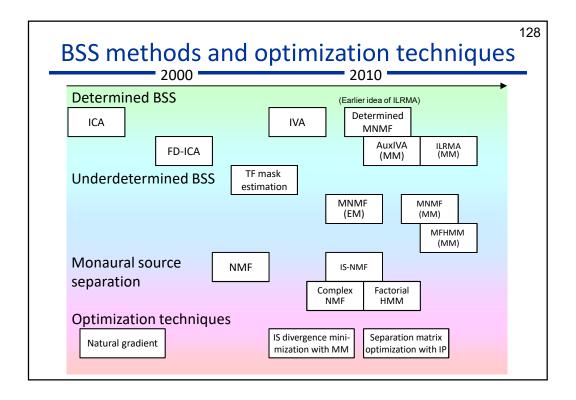


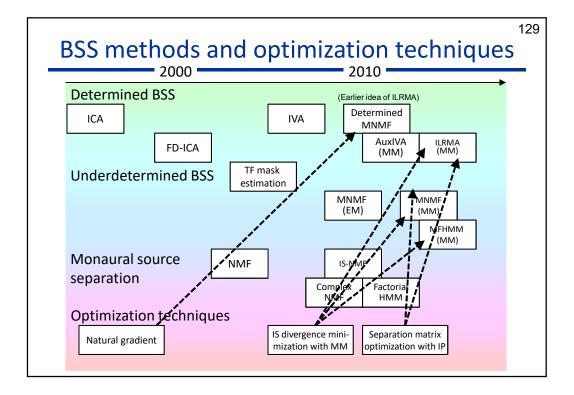








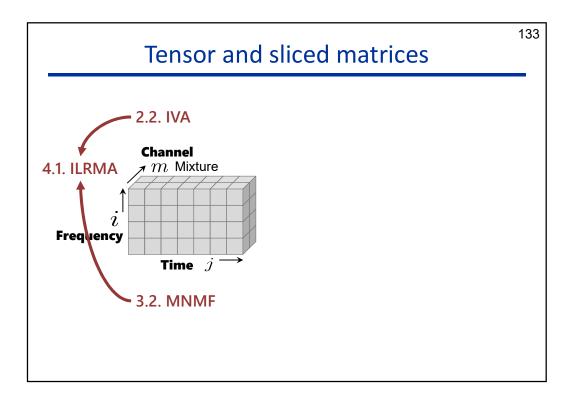


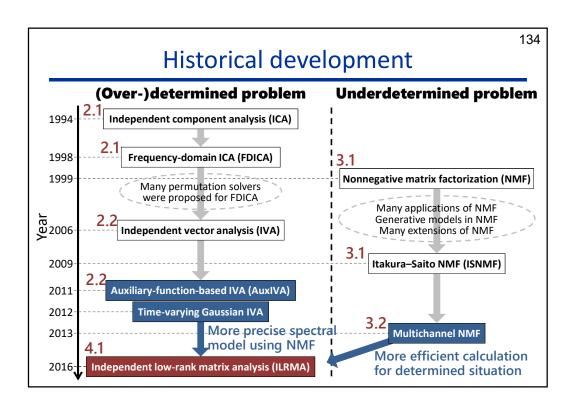


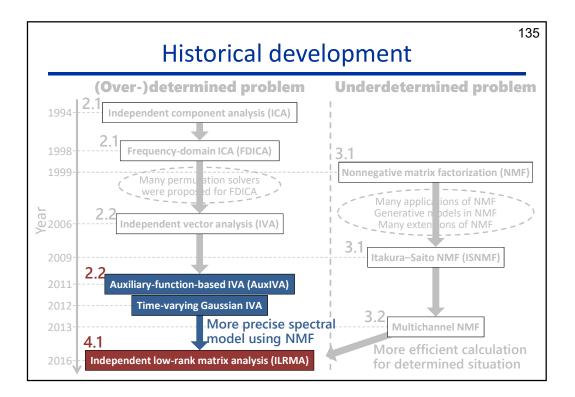
Categorization of LGM-based BSS methods			
Generative model $\mathbf{x}_{i,j} \sim \mathcal{N}_{\mathbb{C}}(\mathbf{x}_{i,j}   oldsymbol{\mu}_{i,j}, oldsymbol{\Phi}_{i,j})$			
Method	$oldsymbol{\mu}_{i,j}$	$ oldsymbol{\Phi}_{i,j} $	
Attias (2003)	0	$\sum_{n \sum_{j'} \sigma_{i,j-j',n}^2 H_{i,j',n}}$	
Ozerov & Févotte (2010)	0	$\sum_{n} \sigma_{i,j,n}^2 H_{i,n} + B_i$	
Duong et al. (2010)	0	$\sum_{n} \sigma_{i,j,n}^2 H_{i,n}$	
Kameoka et al. (2010)	$\mathbf{W}_{i}^{-1}\sum_{j'=1}^{J'-1}\mathbf{F}_{i,j'}\mathbf{x}_{i,j-j'}$	$\left  \mathbf{W}_{i}^{-1} \mathbf{\Sigma}_{i,j} \mathbf{W}_{i}^{-1}  ight $	
Yoshioka et al. (2011)	$\mathbf{W}_{i}^{-1} \sum_{j'=1}^{J'-1} \mathbf{F}_{i,j'} \mathbf{x}_{i,j-j'}$	$\left  \mathbf{W}_{i}^{-1} \mathbf{\Sigma}_{i,j} \mathbf{W}_{i}^{-1} \right $	
Ono et al. (2012)	0	$\left  \mathbf{W}_{i}^{-1} \mathbf{\Sigma}_{i,j} \mathbf{W}_{i}^{-1}  ight $	
Sawada et al. (2013)	0	$\sum_{n} \sigma_{i,j,n}^2 H_{i,n}$	
Higuchi et al. (2014)	0	$\sum_{n}\sum_{j'}\sigma_{i,j-j',n}^2H_{i,j',n}$	
Kitamura et al. (2015)	0	$\left\  \mathbf{W}_{i}^{-1} \mathbf{\Sigma}_{i,j} \mathbf{W}_{i}^{-1}  ight.$	
López et al. (2015)	0	$\left\  \mathbf{W}_{i}^{-1} \mathbf{\Sigma}_{i,j} \mathbf{W}_{i}^{-1}  ight.$	
Adiloğlu & Vincent (2016)	0	$\sum_{n} \sigma_{i,j,n}^2 H_{i,n}$	
Kounades-Bastian et al. (2016)	0	$\sum_n \sigma_{i,j,n}^2 H_{i,j,n}$	

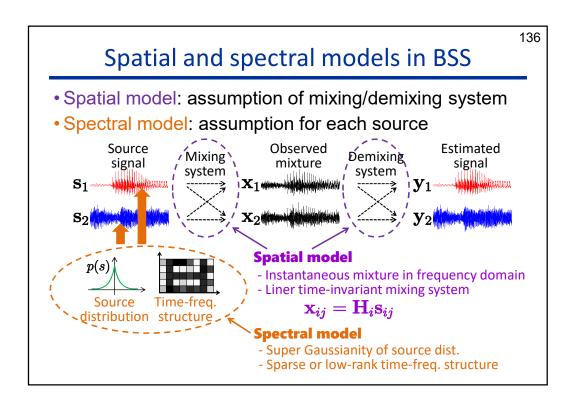
## 131 Categorization of LGM-based BSS methods Constraints on $\sigma_{i,j,n}^2$ $\sigma_{i,j,n}^2$ Method $t_{i,k_{j,n},n}$ where $k_{j,n} \sim \pi$ Attias (2003) Ozerov & Févotte (2010) $\sum_{k} t_{i,k,n} v_{k,j,n}$ $\sigma_{i,j,n}^2$ Duong et al. (2010) $\sum_{k \mid l} t_{i,k,n} a_{i,l,n} v_{k,j,n}$ Kameoka et al. (2010) $v_{j,n}a_{i,j,n}$ Yoshioka et al. (2011) Ono et al. (2012) $v_{j,n}$ $\sum_k z_{k,n} t_{i,k} v_{k,j}$ Sawada et al. (2013) $t_{i,k_{j,n},n}v_{j,n}$ where $k_{j,n} \sim \pi$ Higuchi et al. (2014) $\sum_{k} z_{k,n} t_{i,k} v_{k,j}$ Kitamura et al. (2015) $\sigma_{i,j,n}^2$ López et al. (2015) $\left(\sum_{k} t_{i,k,n}^{\mathrm{e}} v_{k,j,n}^{\mathrm{e}}\right) \left(\sum_{k'} t_{i,k',n}^{\mathrm{f}} v_{k',j,n}^{\mathrm{f}}\right)$ Adiloğlu & Vincent (2016) Kounades-Bastian et al. (2016) $\sum_{k} t_{i,k,n} v_{k,j,n}$

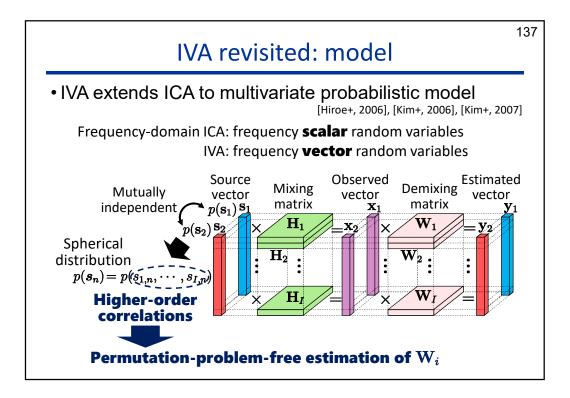
Tutorial structure	13
1. Introduction	
1. Separation of audio/speech signals	
2. Live demonstration	
2. ICA and IVA	
1. ICA: Independent Component Analysis	
2. IVA: Independent Vector Analysis	
3. NMF	
1. NMF: Nonnegative Matrix Factorization	
2. MNMF: Multichannel NMF	
4. ILRMA	
1. ILRMA: Independent Low-Rank Matrix Analysis	

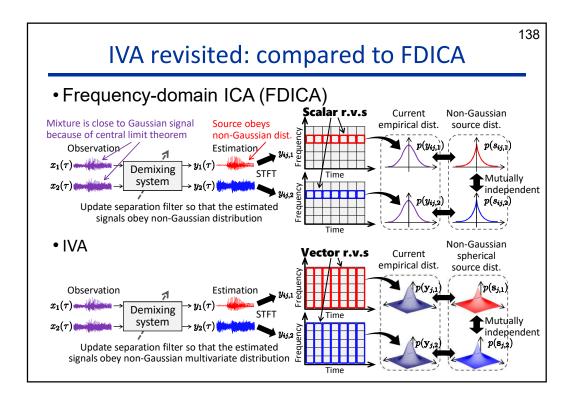


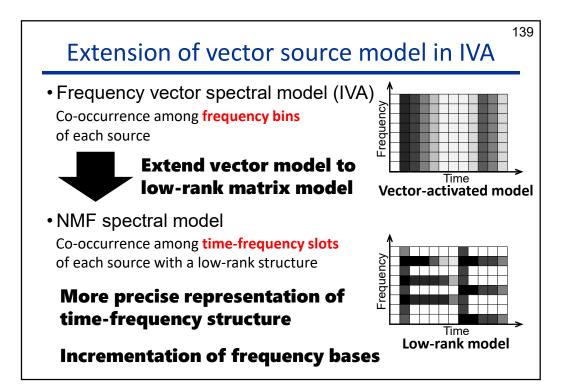


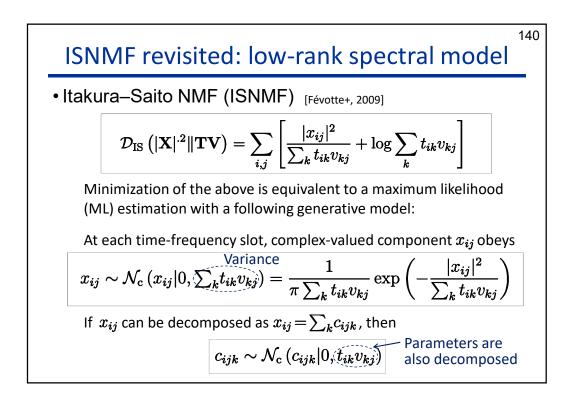


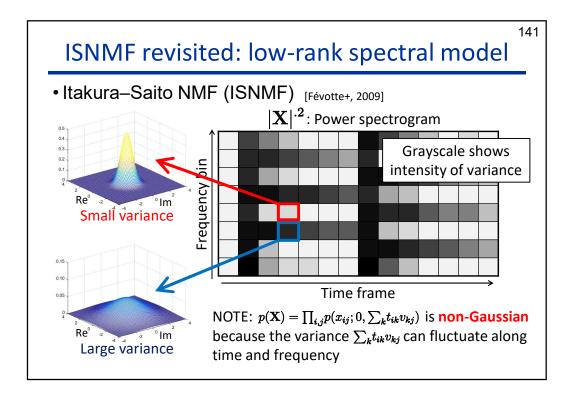


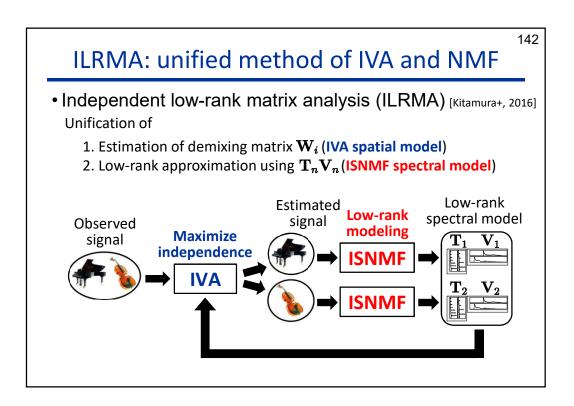


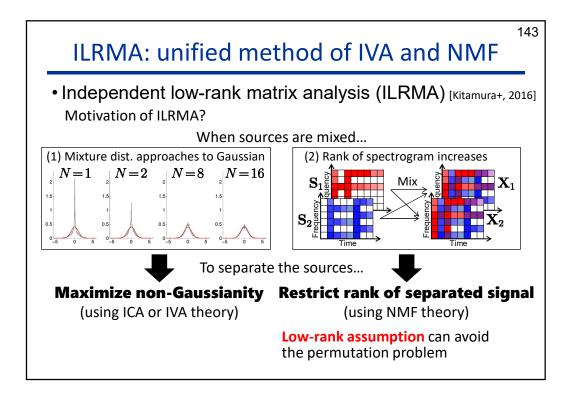


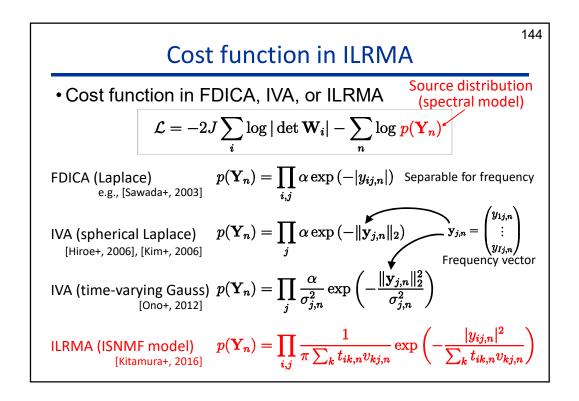


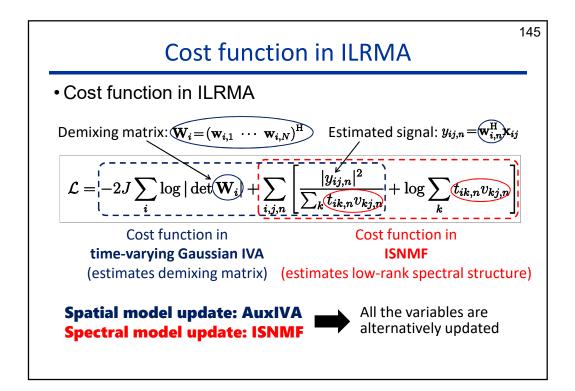


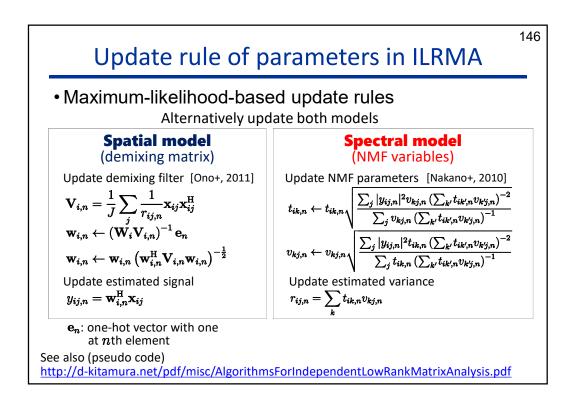


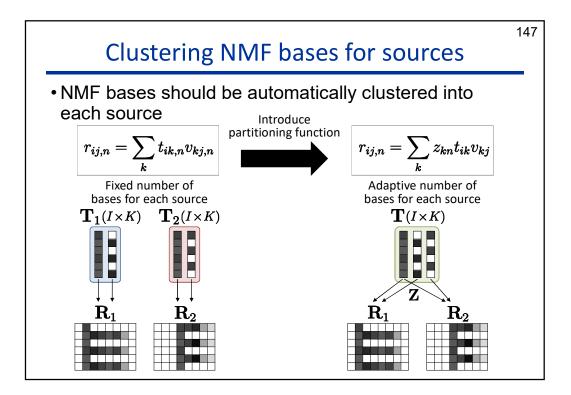


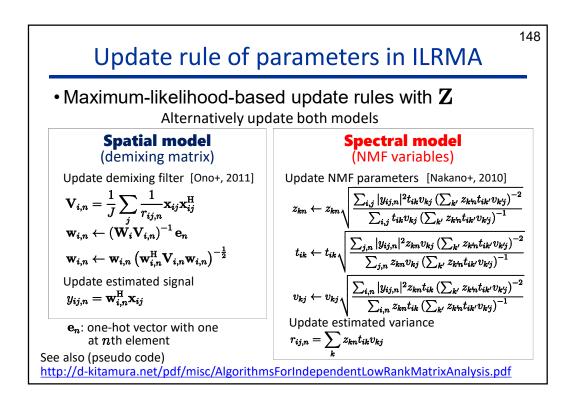


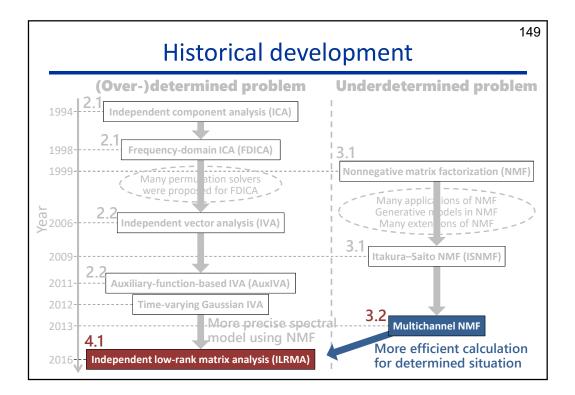


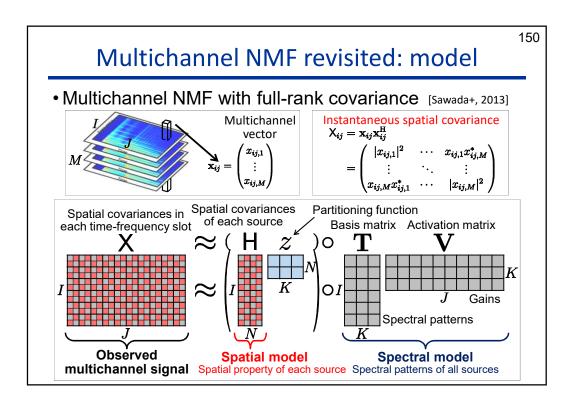


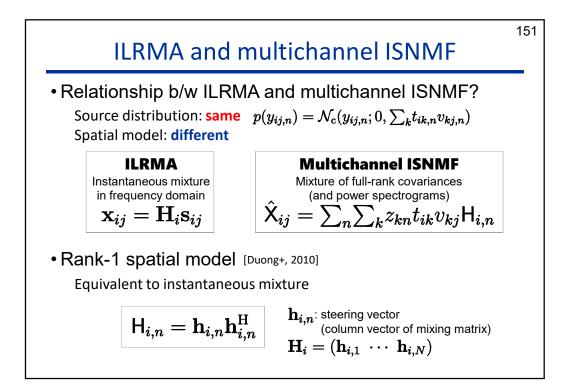


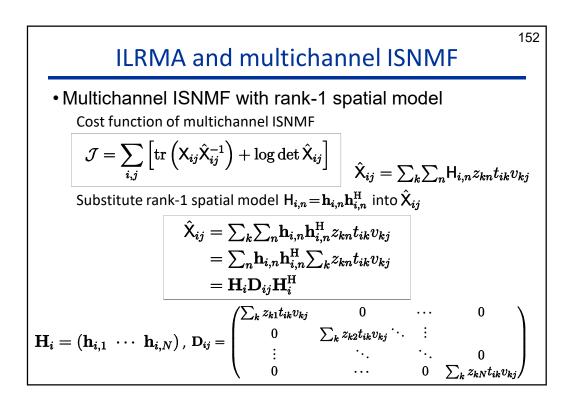


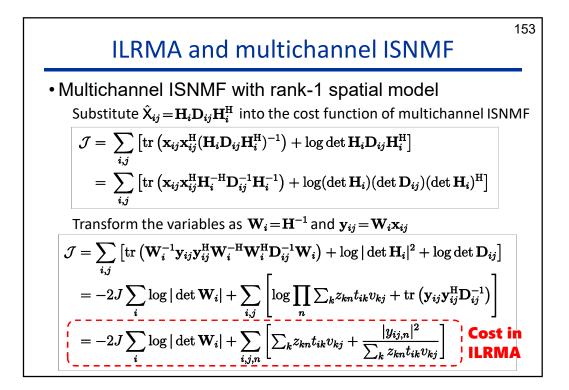


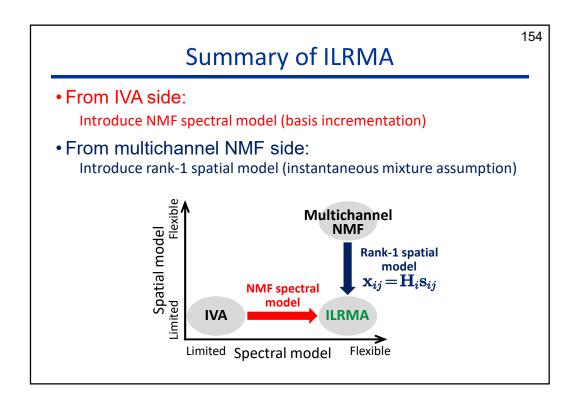


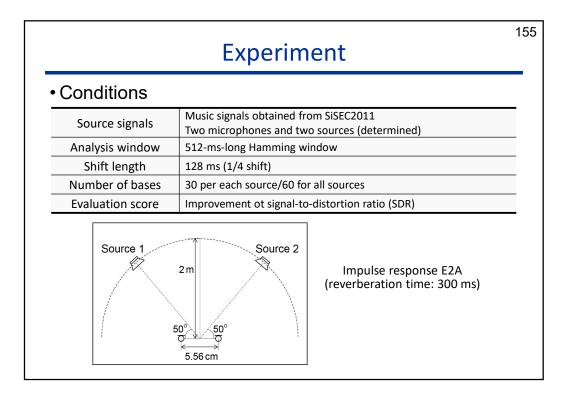


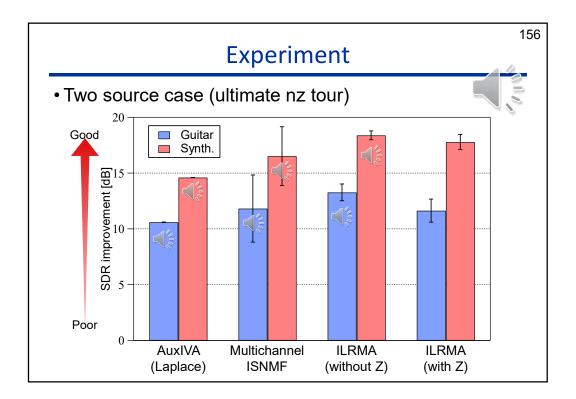


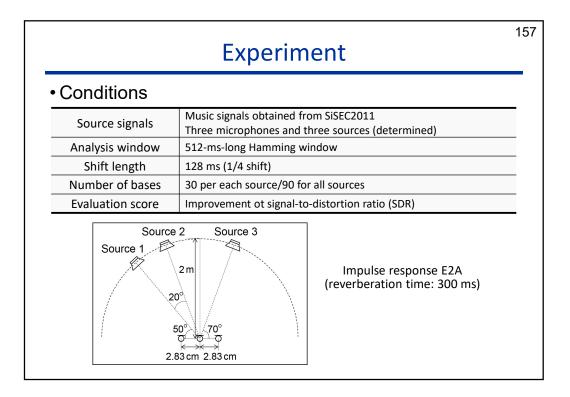


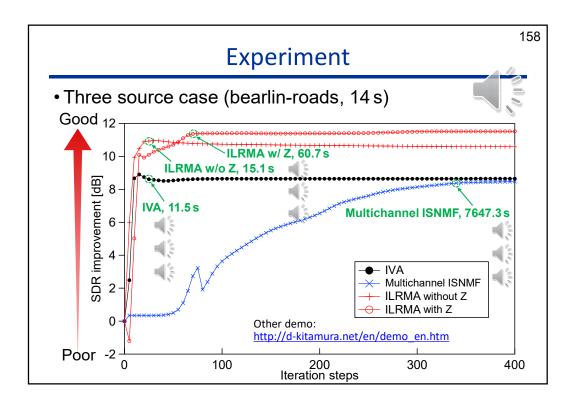




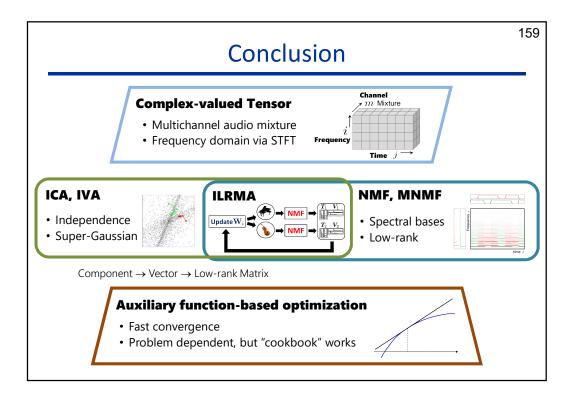








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# Thank you very much for attending this tutorial !

Hiroshi Sawada





Hirokazu Kameoka





Daichi Kitamura

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Frequency-domain approach to convolutive BSS

- [Smaragdis, 1998], [Parra and Spence, 2000], [Schobben and Sommen, 2002], [Anemüller and Kollmeier, 2000], [Asano et al., 2003], [Saruwatari et al., 2006], [Yoshioka et al., 2008], [Duong et al., 2010]
- Permutation alignment [Sawada et al., 2004], [Sawada et al., 2007], [Sawada et al., 2011]
- Scaling adjustment to microphone observations [Cardoso, 1998], [Murata et al., 2001],
  - [Matsuoka and Nakashima, 2001], [Takatani et al., 2004], [Mori et al., 2006]

#### ICA: Independent Component Analysis

- Information-maximization approach [Bell and Sejnowski, 1995]
- Maximum likelihood (ML) estimation [Cardoso, 1997]
- Natural gradient [Amari et al., 1996], [Cichocki and Amari, 2002]
- Equivariance property [Cardoso and Souloumiac, 1996]
- FastICA [Hyvärinen et al., 2001]
- Complex-valued ICA [Bingham and Hyvärinen, 2000], [Sawada et al., 2003]
- Auxiliary function based ICA [Ono and Miyabe, 2010]

### IVA: Independent Vector Analysis

- Multivariate p.d.f. [Hiroe, 2006], [Kim et al., 2006], [Kim et al., 2007]
- FastIVA [Lee et al., 2006], [Lee et al., 2007]
- HEAD problem [Yeredor, 2009]
- Auxiliary function based IVA (AuxIVA) [Ono, 2011], [Ono, 2012b], [Ikeshita et al., 2017]
- Time-varying Gaussian p.d.f. [Ono, 2012a], [Ono et al., 2012]
- Supervised or model-based IVA [Ono et al., 2012], [Lopez et al., 2015], [Nesta and Koldovský, 2017]
- Online IVA [Kim, 2010], [Taniguchi et al., 2014], [Sunohara et al., 2017]

#### NMF: Nonnegative Matrix Factorization

- Auxiliary function based optimization for Euclidean distance NMF and generalized Kullback-Leibler divergence NMF [Lee and Seung, 1999], [Lee and Seung, 2001]
- EM-based optimization for Itakura-Saito divergence NMF [Févotte et al., 2009]
- Auxiliary function based optimization for Itakura-Saito divergence NMF [Kameoka et al., 2006], [Nakano et al., 2010], [Févotte and Idier, 2011]
- Auxiliary function based optimization for  $\beta$  divergence NMF [Nakano et al., 2010], [Févotte and Idier, 2011]
- Auxiliary function based optimization for sparse NMF [Kameoka et al., 2009]

#### Multi-channel NMF

- EM-based optimization [Ozerov and Févotte, 2010]
- Auxiliary function based optimization [Sawada et al., 2012], [Sawada et al., 2013], [Higuchi and Kameoka, 2014]

#### ILRMA: Independent Low-Rank Matrix Analysis

- Earlier idea (determined multi-channel NMF) [Kameoka et al., 2010]
- Multichannel NMF with rank-1 spatial model [Kitamura et al., 2015a]
- ILRMA [Kitamura et al., 2016], [Kitamura et al., 2018]
- Relaxation of Rank-1 spatial model [Kitamura et al., 2015b]
- Maximization-equalization algorithm [Mitsui et al., 2017b]
- Optimal window length [Kitamura et al., 2017]
- Based on Student's t-distribution [Mogami et al., 2017]
- With sparse regularization [Mitsui et al., 2017a]
- With spatial regularization [Mitsui et al., 2018]

Other references related to auxiliary function based optimization

 [Lange et al., 2000], [Hunter and Lange, 2000], [Hunter and Lange, 2004], [Ono et al., 2009], [Ono and Sagayama, 2010], [Févotte and Idier, 2011], [Yoshii et al., 2013], [Kameoka and Takamune, 2014], [Sun et al., 2017]

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