



Hunting Echoes *for* *Auditory Scene Analysis*

Diego Di Carlo • 27.05.2019

Supervised by
Nancy BERTIN, Antoine DELEFORGE



Outline

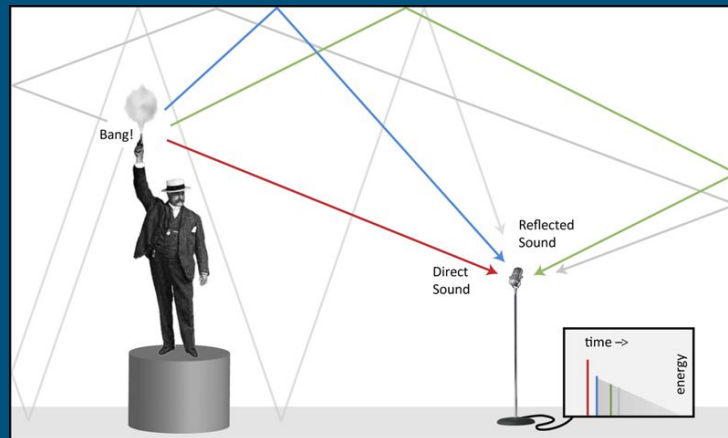
1. What are acoustic echoes?
 2. If you know them, we can (1st year)
 - a. *Sound Source Separation and **SEPARAKE***
Presented at ICASSP18
 - b. *Sound Source Localization and **MIRAGE***
Presented at ICASSP19
 3. How to know them? (2nd year)
 - a. Continuous Dictionary and **BRAIRE**
Work in Progress
 - b. Learning-based approach and **MIRAGE**
Presented at ICASSP19
 4. Applications
 - a. *Honda Haru*
-

ACOUSTIC ECHOES

(Room) Impulse Response: the linear filtering effect due to the propagation of sound from a source to a microphone in a room.

It accounts for ...

- ... the **geometry** of the audio scene:
 - Room shape and size
 - Source position
 - Microphone position
 - ... other objects (e.g. furnitures) sizes and shape.
- ... the **acoustic properties** of the audio scene:
 - Wall materials
 - ... Type of other objects (e.g. furnitures materials)



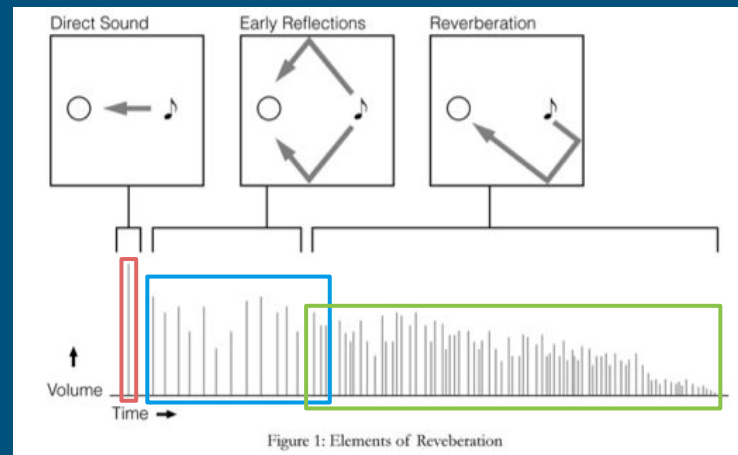
ACOUSTIC ECHOES

RIR can be subdivided in:

- Direct (or anechoic) path
- Early reflections (**Echoes**)
- Late Reverberation

For some audio inverse problems, the sound propagation is typically...

- ... ignored [Le Roux et al. 2015, DC et al. 2017];
- ... assumed as a single anechoic path [Rickard 2007];
- ... modelled entirely [Ozerov et al. 2010, Nugraha et al. 2016];
- ... assumed as late reverberation [Leglaive et al. 2016].



If we know them?



Reverberation has a detrimentally affects typical **Audio Inverse Problem** algorithm.

Can Echoes help?

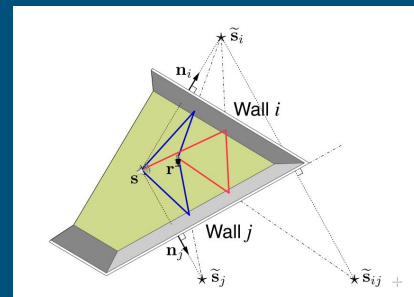
Recent **echo-aware** methods showed that the knowledge of early echoes increases their performances.

HP : Assume we know echo's
coefficients and locations



Inverse Problems with Echo-aware methods

- Speech enhancement
 - Sound Source Localization
 - [Riberio et al. 2010, DC 2019, ...]
 - Sound Source Separation
 - [Scheibler et al. 18, Leglaive et al. 2016]
 - Beamforming
 - [Dockmanic et al. 2015]
- Microphone calibration
 - [Salvati et al. 16, Dockmanic et al. 2013]
- Dereverberation and Room Equalization
 - [Krishnan et al. 2018]
- Room Geometry Estimation
 - [Crocco et al. 2016, Dockmanic et al. 2013, ...]



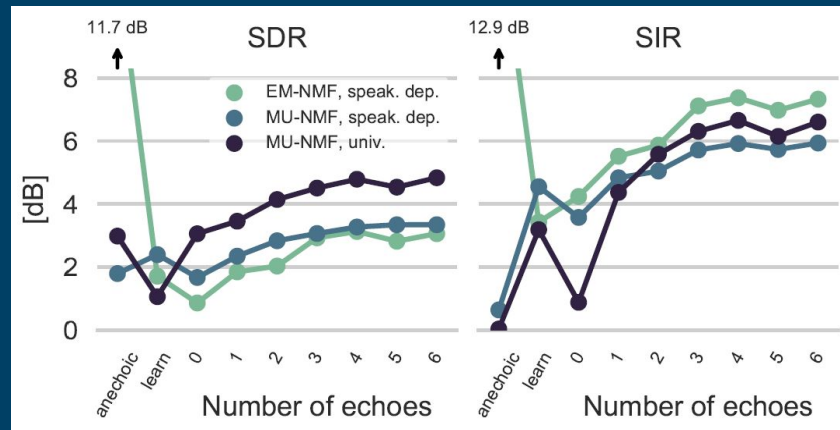
SEPARAKE

*Sound source separation with
a little help from echoes*

Presented at ICASSP 2018

[Robin Scheibler, Diego Di Carlo,
Antoine Deleforge, Ivan Dokmanic]

Use transfer function models taking into account early echoes in NMF-based sound source separation methods



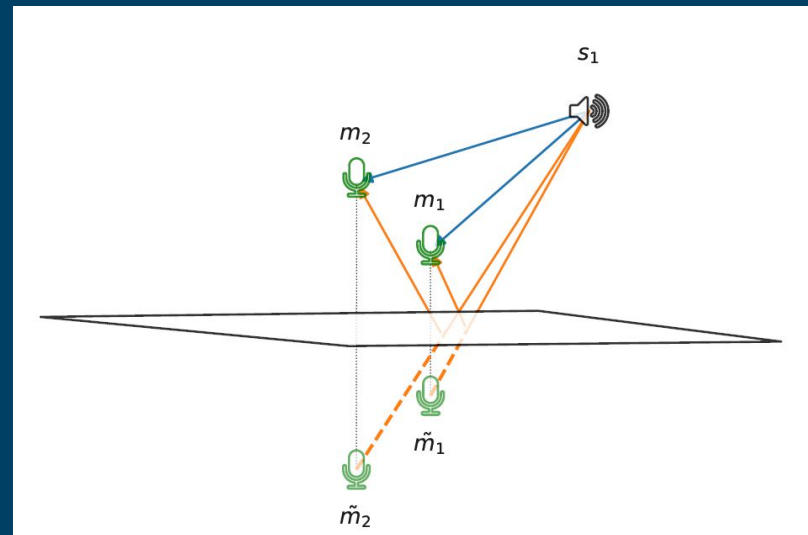
Early echoes are assumed perfectly known in this work

MIRAGE

*Microphones array
augmentation with echoes*

Accepted at ICASSP 2019

[Diego Di Carlo, Antoine Deleforge, Nancy Bertin]

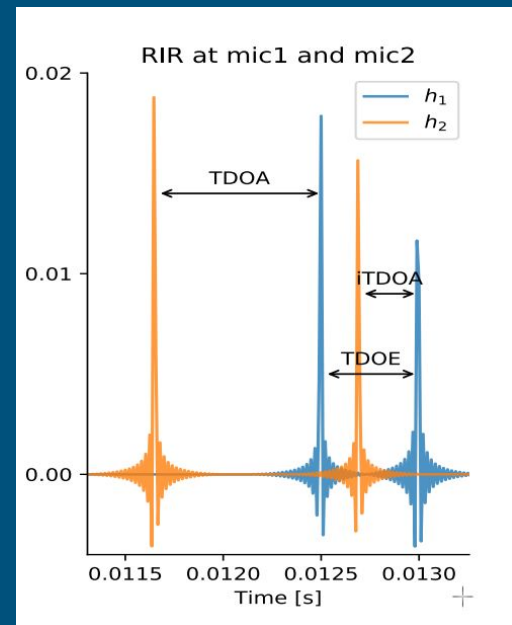
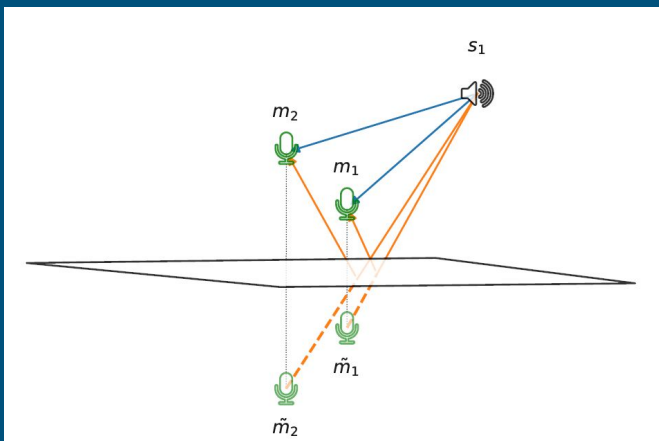


The signal received at m_1 can be seen as the **sum** of anechoic signals received at m_1 and an **image microphone \tilde{m}_1**

Image source -> Image microphones

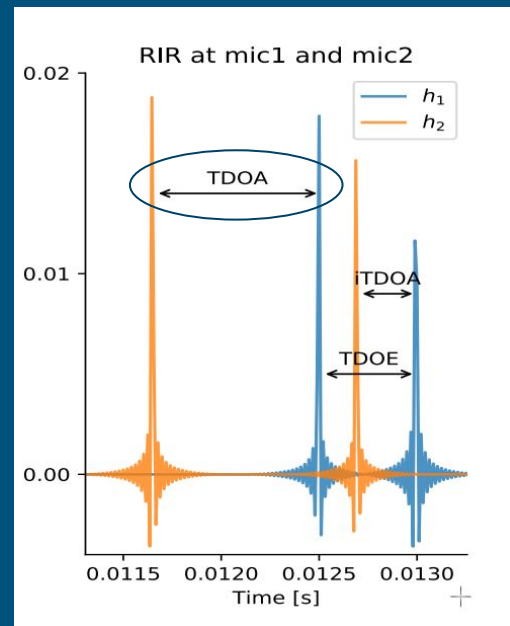
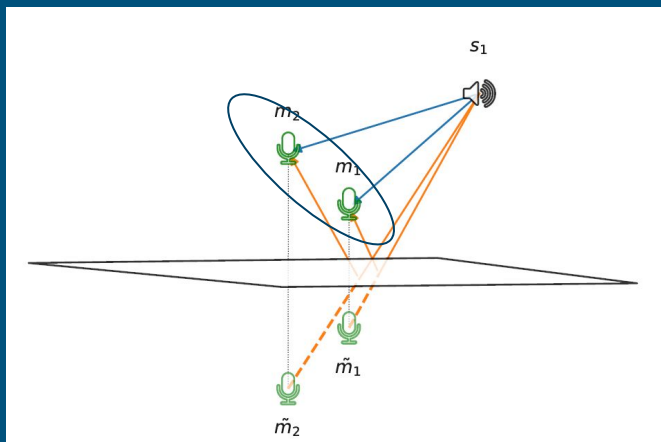
Sound Source Localization *with a little help from echoes*

- More microphones... better audio signal processing!
- How to « access » image microphones?



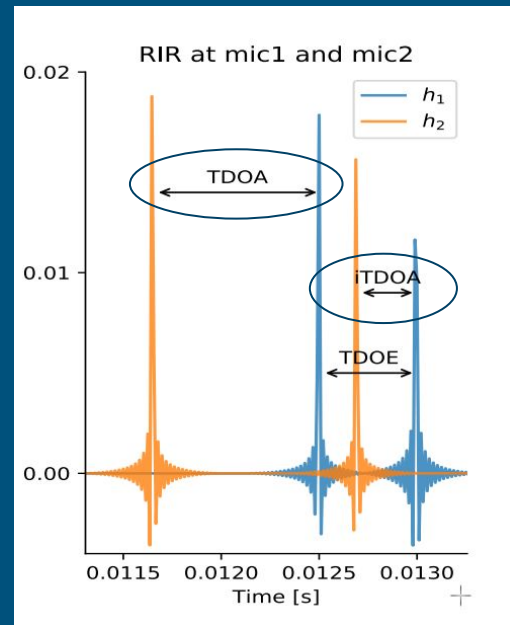
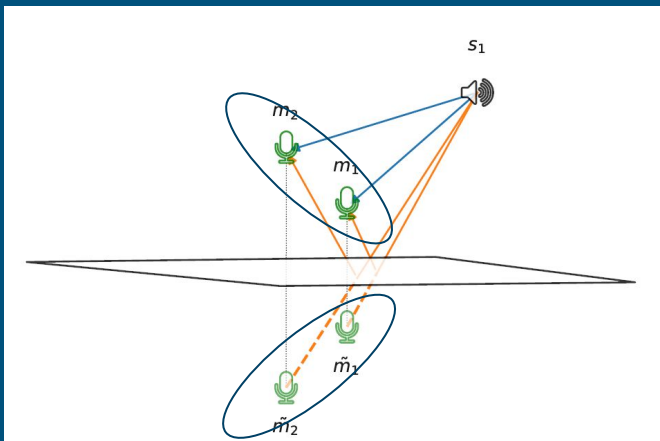
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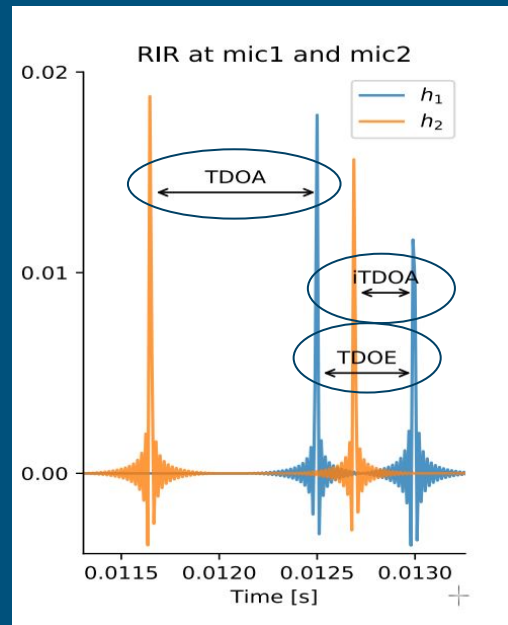
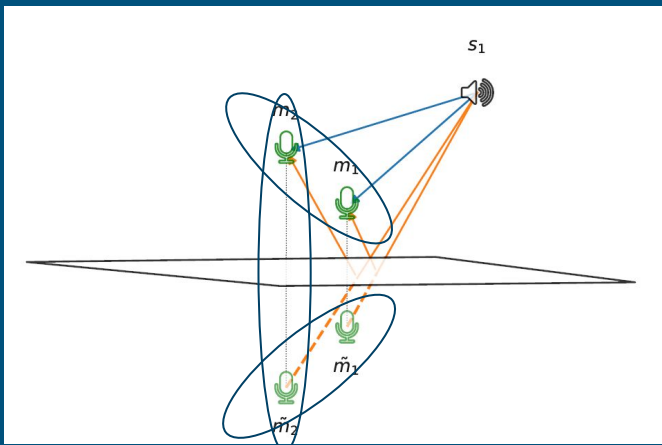
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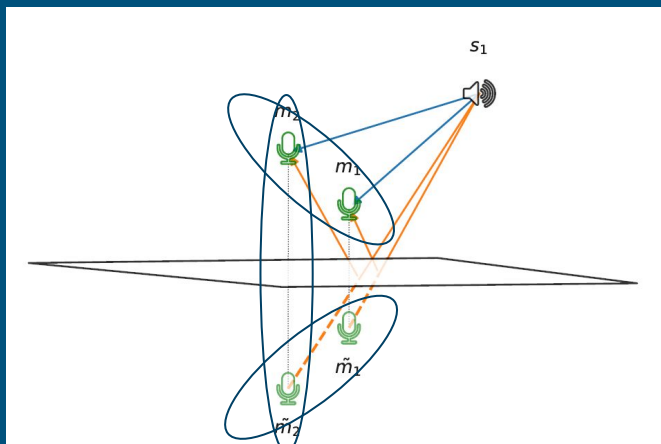
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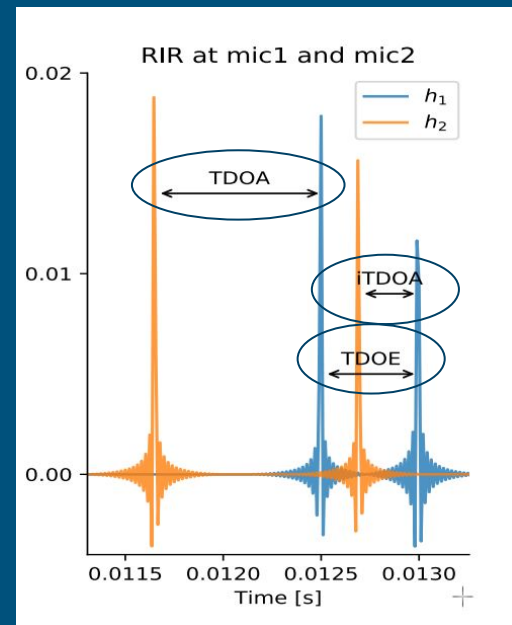


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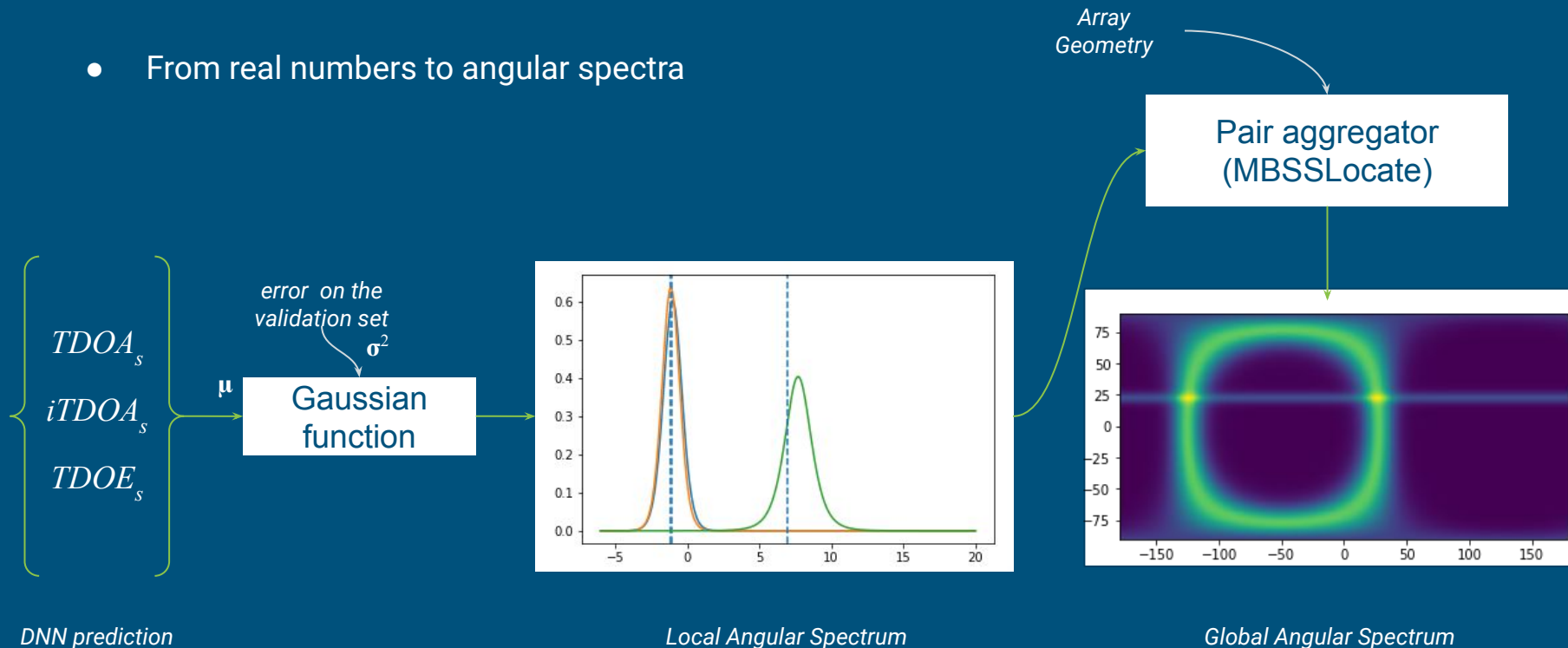


- Each pair in the **augmented array** is associated to impulse response characteristics



Sound Source Localization *with a little help from echoes*

- From real numbers to angular spectra

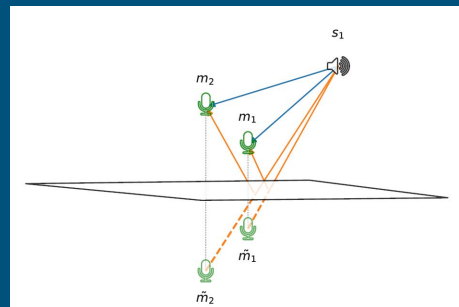


Sound Source Localization *with a little help from echoes*

- Aggregating (with MBSSLocate) time differences of arrival from multiple microphone pairs enables **2D sound source localization**
- The microphone and surface positions are assumed known
- Promising «**impossible localization**» results using clean signals and white noise sources
- Future work:
 - Aggregating multiple pairs
 - Test on real data
 - Perfect symmetries breaks the model

Results on test set [ICASSP19]

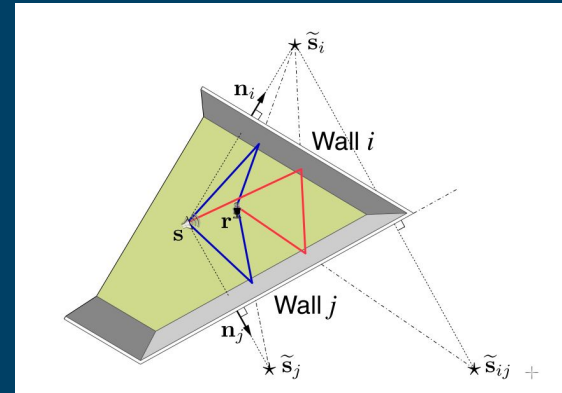
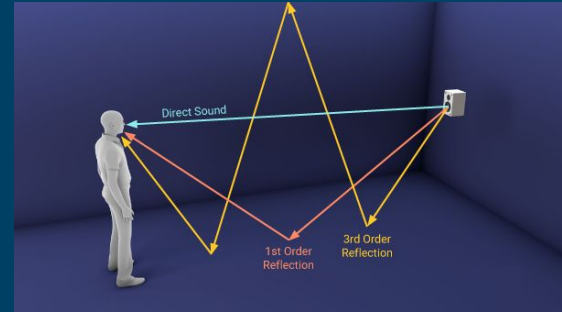
DoA	Input	ACCURACY < 10°		ACCURACY < 20°	
		θ	ϕ	θ	ϕ
MIRAGE	wn	4.5 (59)	3.9 (71)	6.8 (79)	5.9 (88)
MIRAGE	wn+n	4.4 (18)	5.5 (26)	9.4 (35)	11.1 (66)
MIRAGE	sp	4.6 (45)	4.8 (59)	8.1 (71)	7.2 (83)
MIRAGE	sp+n	5.2 (17)	5.9 (12)	10.7 (38)	12.3 (43)



ROOM GEOMETRY

Plan for the Third Year

[Crocco2016,Dockmanic2013, Antonacci2010, Tervo2010]



How to know them?



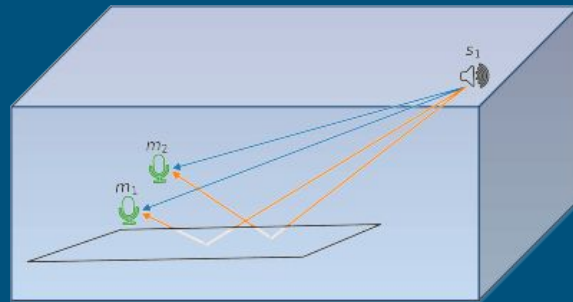
In all the presented works, the echoes are assumed known.

How to estimate them?

Blind Echo Estimation

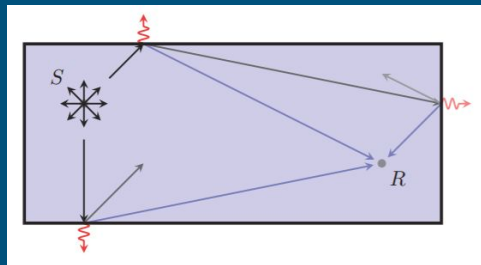
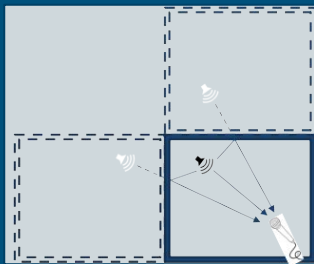
An easy - yet common - scenario: the **pic-nic**

- One Source
- Two microphones
- Random shoe-box rooms
- Nearest surface is the most reflective



10'000 Auditory pic-nic scenes generated using [Schimmel et al. 2009] software

Specular
reflection

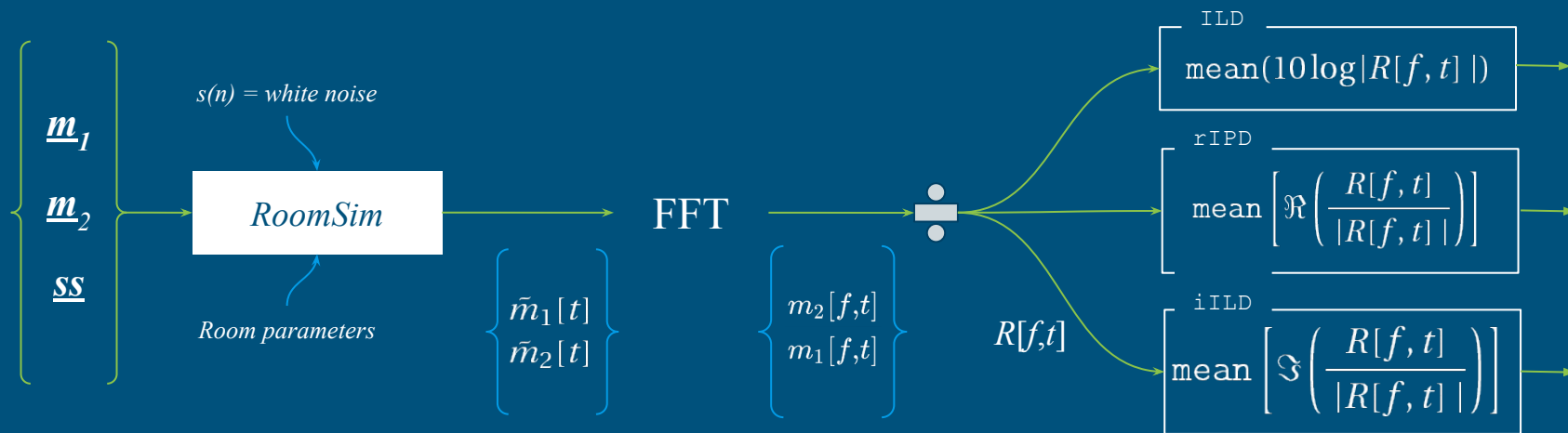


Scattering

Blind Echo Estimation

Why only two microphones?

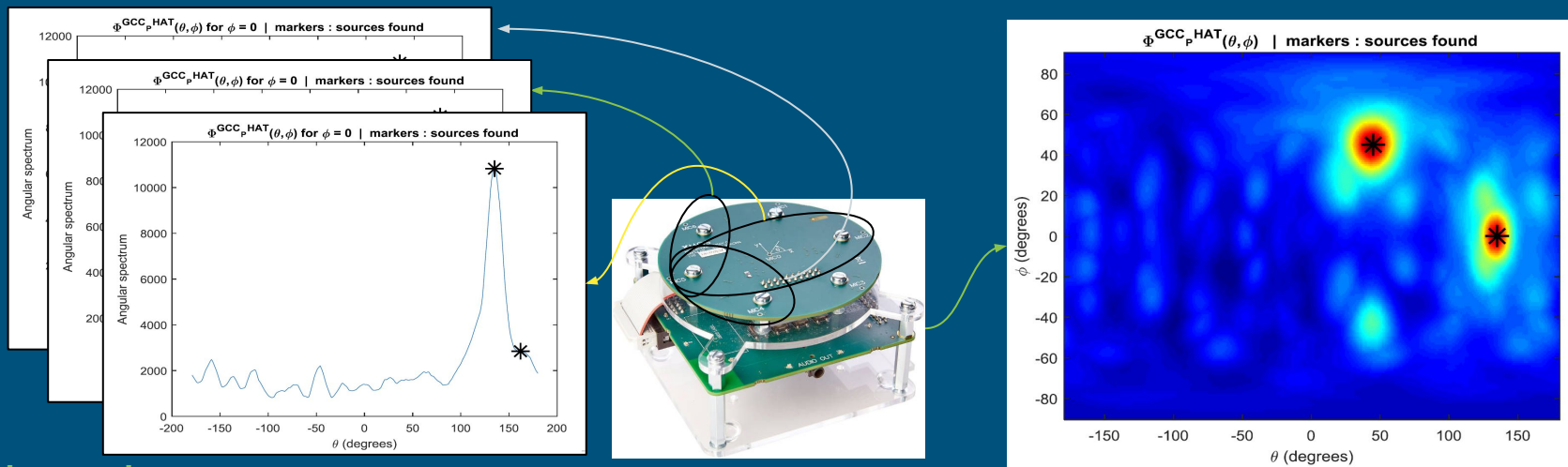
- The **relative transfer function** can be computed



Blind Echo Estimation

Why only two microphones?

- The contribution of multiple microphone pairs can be aggregated together
 - If the geometry of the microphone array is known a priori [MBSSLocate, DiBiase et al 2001]



Blind Echo Estimation

Why only two microphones?

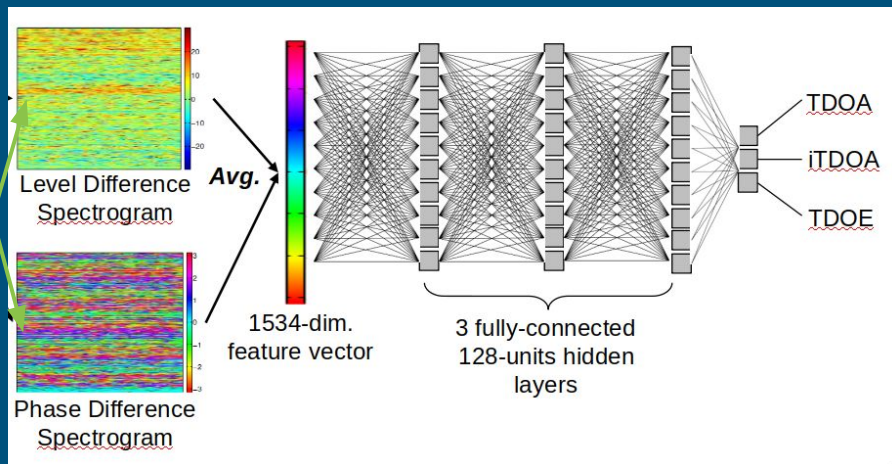
- The **relative transfer function** can be computed

$$\begin{cases} \tilde{m}_1[t] = h_1[t] * \tilde{s}[t] \\ \tilde{m}_2[t] = h_2[t] * \tilde{s}[t] \end{cases} \Rightarrow R[f, t] = \frac{m_2[f, t]}{m_1[f, t]} = \frac{h_2[f]s[f, t]}{h_1[f]s[f, t]} = \frac{h_2[f]}{h_1[f]}$$

- Ideally it removes the dependency from the source signal
- If there is no noise **and** filter shorter than the fft window

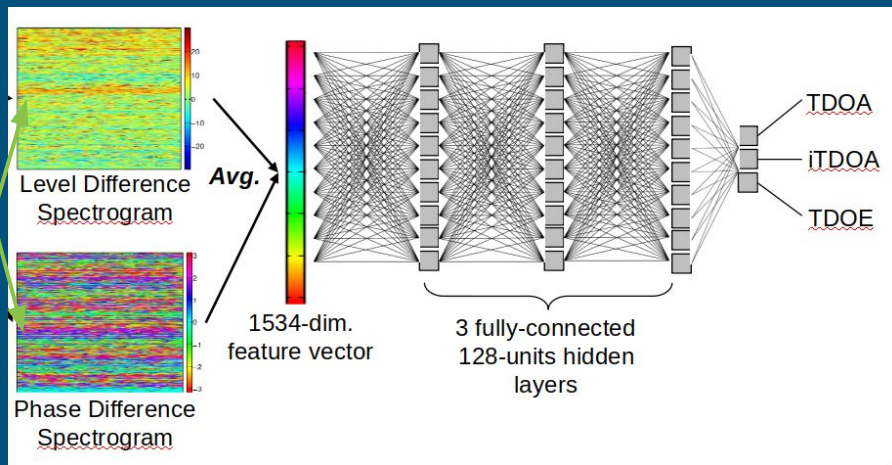
Blind Echo Estimation - Learning approach

Deep Neural Network Learning



Blind Echo Estimation - Learning approach

Deep Neural Network Learning



Results on test set

[Submitted to ICASSP19]

		nRMSE		
	Input	TDOA	iTDOA	TDOE
MIRAGE	wn	0.18	0.28	0.25
MIRAGE	wn+n	0.68	0.69	0.89
MIRAGE	sp	0.31	0.34	0.56
MIRAGE	sp+n	0.99	0.98	1.48
GCC-PHAT	wn	0.21	-	-
GCC-PHAT	wn+n	0.68	-	-
GCC-PHAT	sp	0.32	-	-
GCC-PHAT	sp+n	1.38	-	-

Also tried with a Gaussian Locally-Linear Mapping (GLLiM) \Rightarrow It failed

BRAIRE

*Blind and constrained acoustic room
impulse response estimation*

Collaboration with Clement Elvira:

- *An application for the theory of Super Resolution / Continuous Dictionary*
- *Threat as off-grid spike-retrieval*



The Hunt Continues...

Current research

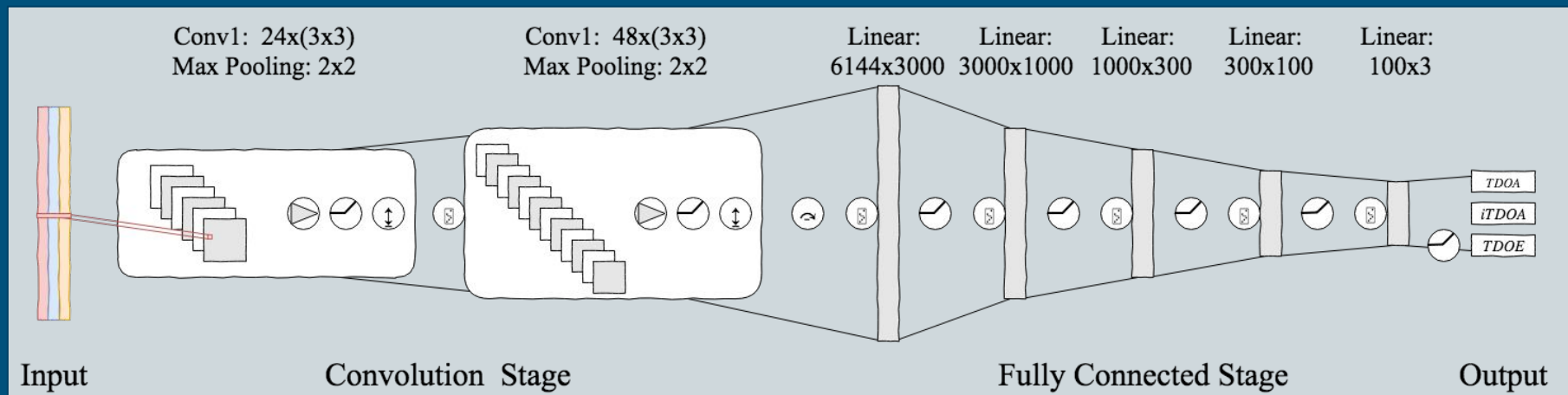


1. State of the art DNN architectures
2. Ad-hoc loss functions for uncertainty
3. Extensions to microphone arrays
4. Test on real world data
 - a. *Honda Haru*



Echo hunting continues...

- What's next? What's now?
 - **State of The Art DNN** architecture: CNN [Chakrabarty et al 2017, Nguyen et al. 2018]



Echo hunting continues...

- What's next? What's now?
 - **State of The Art DNN** architecture: CNN
 - **Gaussian and Student-T likelihood** Loss Function for estimating both TDOA, iTDOA and TDOE and their uncertainties

$$\mathcal{L}(\theta) = \frac{1}{3} \sum_{n=1}^N |\tau_{a,n} - \tau_{\tilde{a},n}|^2 + |\tau_{i,n} - \tau_{\tilde{i},n}|^2 + |\tau_{e,n} - \tau_{\tilde{e},n}|^2$$

Echo hunting continues...

- What's next? What's now?
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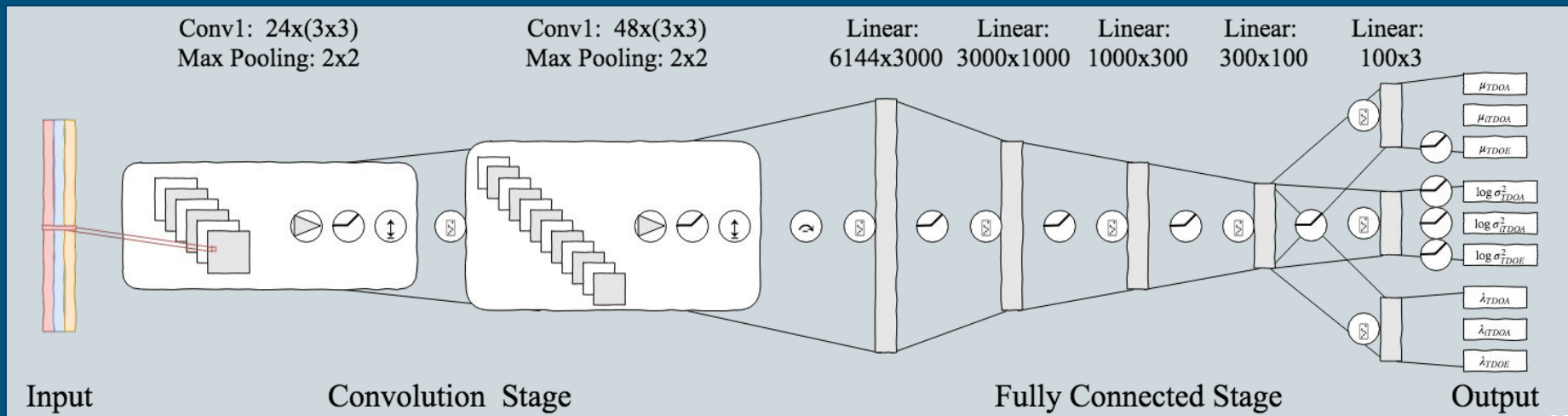
$$\mathcal{L}(\theta) = \frac{1}{3} \sum_{n=1}^N |\tau_{a,n} - \tau_{\tilde{a},n}|^2 + |\tau_{i,n} - \tau_{\tilde{i},n}|^2 + |\tau_{e,n} - \tau_{\tilde{e},n}|^2$$

$$p(\tau_k | X; \theta) \sim \mathcal{N}(\mu_{\tau_k}(x_n; \theta), \sigma_{\tau_k}^2(x_n; \theta)) \quad k = a, i, e$$

$$\mathcal{L}(\theta) = \sum_{n=1}^N \log \sigma_{\tau_a}^2(x_n) + \frac{|\tau_a - \mu_{\tau_a}(x_n)|^2}{\sigma_{\tau_a}^2(x_n)} + \dots$$

Echo hunting continues...

- What's next? What's now?
 - State of The Art DNN architecture: CNN
 - Gaussian and Student-T likelihood Loss Function for estimating both TDOA, iTDOA and TDOE and their uncertainties



Echo hunting continues...

- What's next? What's now?
 - State of The Art DNN architecture: CNN
 - Gaussian and Student-T likelihood Loss Function

Results on test set
[Accepted at ICASSP19]

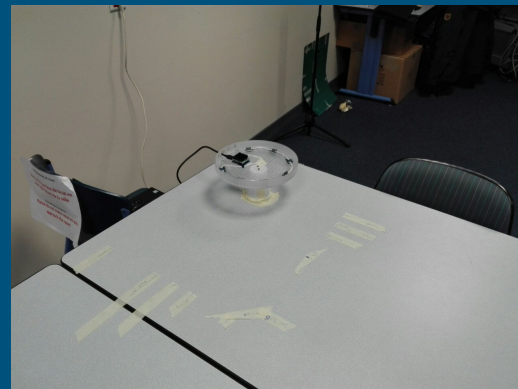
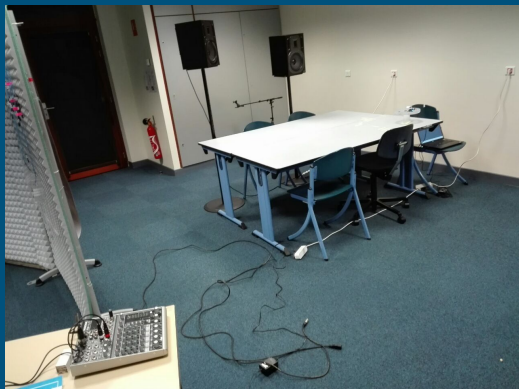
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Current Results on test set
with noise

distr	snr	phase	test_signal	tdoa	itdoa	tdoe1
gaussian	0	Test	noise	0.103131	0.110806	0.248462
gaussian	15	Test	noise	0.102640	0.110342	0.280237
gaussian	30	Test	noise	0.101439	0.108265	0.323202
none	0	Test	noise	0.137354	0.145333	0.209920
none	15	Test	noise	0.192951	0.196020	0.284383
none	30	Test	noise	0.148980	0.151179	0.222592
student	0	Test	noise	0.099268	0.107615	0.237591
student	15	Test	noise	0.110567	0.111748	0.310297
student	30	Test	noise	0.106170	0.113793	0.294742

Echo hunting continues...

- What's next? What's now?
 - State of The Art DNN architecture: CNN
 - Gaussian and Student-T likelihood Loss Function
 - Test on real data and microphone array (HONDA HARU array)





HONDA HARU

An application for MIRAGE

Submitted to HONDA on March 2019

Phase II: passed
Phase III: next year

Research project funded by HONDA





SPCUP2019: DREGON

*Drone ego noise reduction
for sound source localization*

[ICASSP 2019]

IEEE Signal Processing Cup 2019

*Search and Rescue via Drone-Embedded Sound
Source Localization*

Student Competition

Organizing committee:





MIRA for DOLBY

*Music Interference
Reduction Algorithm*

Submitted to DOLBY on December 2018
Project Accepted

*Algorithm for microphone leakage removal on
full-length audio recordings*

No Echo model at All : Phase is considered Random

Tasks:

- Soundcheck dataset
 - Learning with sound-check
 - Removal on the actual song
- Gorillaz concert
 - Remove PA from Audience
- Football match
 - Commentator enhancement
 - Ball enhancement



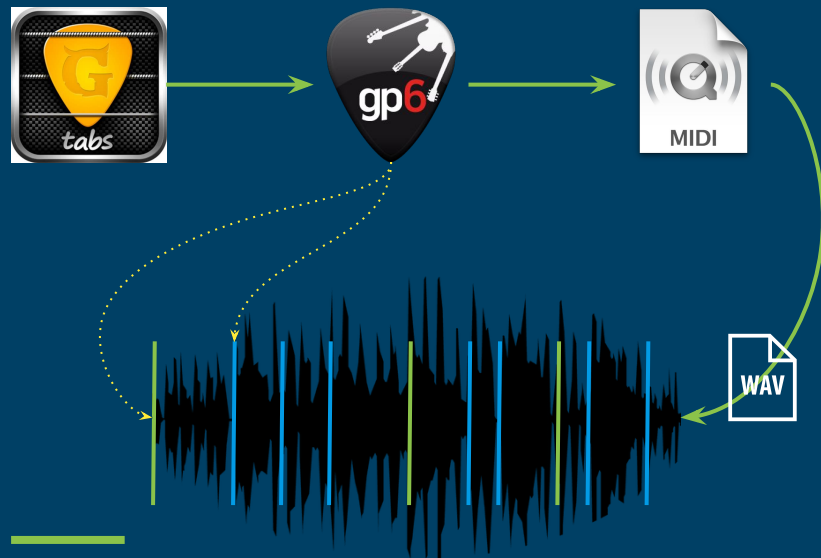
BEATLESS

*downbeat Detection by Learning on
Synthetic Sources*

Collaboration with Magdalena Fuentes
Telecom ParisTech/CentraleSupelec

*Virtually-Supervised
Learning-based*

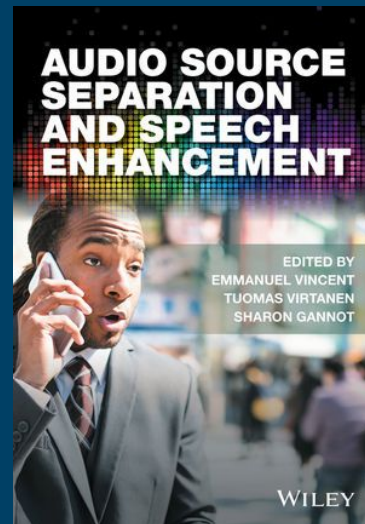
Downbeat detection and micro-timing





Prof. GANNOT

*Visiting Bar'Ilan University
November 2019 - February 2020*





RAPPLE

Rap APP battLE



THANK YOU

Need for some answers?

