Diego DI CARLO November 27, 2020

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Introduction



#### Sound

produced by sources



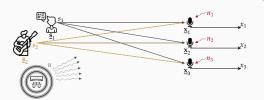


- produced by sources
- recorded by microphones

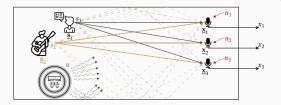




- produced by sources
- recorded by microphones
- corrupted by noise



- produced by sources
- recorded by microphones
- corrupted by noise
- propagates in the space



- produced by sources
- recorded by microphones
- corrupted by noise
- propagates in the space
- interacts with the room
  - $\hookrightarrow$  reverberation

## **Semantic** information



on nature and content

## **Semantic** information



on nature and content

## **Spatial** information



on position and geometry



## **Semantic** information



on nature and content

#### **Spatial** information



on position and geometry

#### Temporal information



on events activity



#### Semantic information



on nature and content

#### Spatial information



on position and geometry

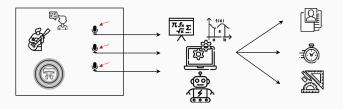
#### Temporal information



on events activity

## **Audio Scene Analysis**

Extraction and organization of all the information in the sound





#### Semantic information



on nature and content

#### Spatial information



on position and geometry

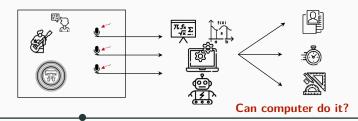
#### Temporal information

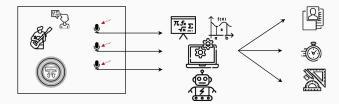


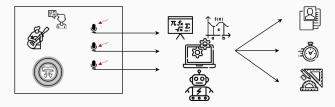
on events activity

## **Audio Scene Analysis**

Extraction and organization of all the information in the sound

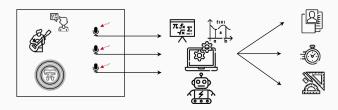






# **Signal Processing**

Mathematical models, frameworks and tools to tackle and solve such problems



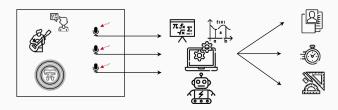
## **Signal Processing**

Mathematical models, frameworks and tools to tackle and solve such problems

#### Some (inverse) problems

- Speaker Identification
- Sound Source Separation (SSS)
- Speech Enhancement (SE)
- Automatic Speech Recognition (ASR)

- Voice Activity Detection
- Diarization
- RT<sub>60</sub> estimation
- Acoustic Channel Estimation
- Wall Absorption Estimation



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## Sound interacts with environment

```
it is reflected (specularly and diffusely)

+ it is diffracted
+ it is absorbed and transmitted
+ other physical interaction
```

#### **Acoustic Echoes**

- Elements of reverberation
- Standing out for time and strength
- Repetition of a sound but after
  - time ⇔ distance
  - same content



Room coloration

## **Everyday examples:**

Echo points Bat Typically sound propagation is

**Dolphins** 

■ ignored ⇒ simple processing

hut reverberation — noise

Echo-aware methods



## Thesis title:

Audio Scene Analysis



 $context\ and\ problems$ 



## Thesis title:

Audio Scene Analysis

Signal Processing



context and problems models and frameworks



#### Thesis title:

Audio Scene Analysis

↓ and problem Signal Processing



context and problems models and frameworks

Echo-aware ↓

better processing



#### Thesis title:

Audio Scene Analysis

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#### Thesis content

How to estimate them?

- Analytical method
- Learning-based method

#### How to use them?

- Source Separation
- Source Localization
- Speech Enhancement
- Room Geometry Estimation

#### Where to find them?

Echo-aware database for estimation and application

**Problem Statement** 

# Signal model

## Sound propagation process $\Leftrightarrow$ Source $\rightarrow$ Filter $\rightarrow$ Receiver model

$$\tilde{x}_i(t) = (\tilde{h}_i * \tilde{s})(t) + + \tilde{n}(t) \longrightarrow \text{noise term}$$
 noise term continuous-time convolution

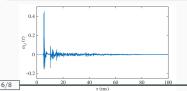
-3mm

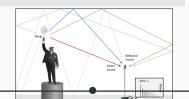


A continuous time

# Room Impulse Response (RIR)

- linear filtering effect of the sound
- acoustic response of a room to a (prefect) impulsive sound
- depends on spatial properties (room geometry, mic/src position)

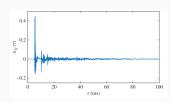




#### **Echoes in the RIR**

RIR model

$$\tilde{h}_i(t) = \frac{\tilde{h}_i^{\rm d}(t) + \tilde{h}_i^{\rm e}(t) + \tilde{h}_i^{\rm Irev}(t) + \varepsilon_i(t)}{$$



Echoes can be modeled as sum of Dirac's delta

$$\tilde{h}_i^{\text{echoes}} = \tilde{h}_i^{\text{d}}(t) + \tilde{h}_i^{\text{e}}(t) \approx \sum_{r=0}^R \alpha_i^{(r)} \delta(t - \tau_i^{(r)})$$

 $\textbf{Goal:} \ \text{estimated the} \ \tau_{i_{i,r}}$ 

## **Challenges:**

- $\alpha$  distortion (even if we know it  $\implies$  labeling)
- $\alpha \to \alpha(t)$  (sum of diracs  $\to$  sum of filters)
- $h_l$  reverberation is included in the noise term

# References i