

### Unified Stochastic Reverberation Modeling

#### **Roland Badeau**

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September 6, 2018

26th European Signal Processing Conference (EUSIPCO)

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### **副選擇的** Why this research work?

Applications of reverberation models:

- Dereverberation (Belhomme et al., 2017),
- Source separation (Leglaive et al., 2018),
- Source localization, denoising, audio inpainting...

A. Belhomme, R. Badeau, Y. Grenier, and E. Humbert. Amplitude and phase dereverberation of harmonic signals. In *Proc. of IEEE WASPAA*, New Paltz, New York, USA, October 2017

S. Leglaive, R. Badeau, and G. Richard. Student's t source and mixing models for multichannel audio source separation. *IEEE Trans. Audio, Speech, Language Process.*, 26(5):1–15, May 2018

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# **副選択的** Why this research work?

- Applications of reverberation models:
  - Dereverberation (Belhomme et al., 2017),
  - Source separation (Leglaive et al., 2018),
  - Source localization, denoising, audio inpainting...
- Existing stochastic models of late reverberation:
  - Time domain (Schroeder, 1962; Moorer, 1979)
  - Frequency domain (Schroeder, 1962)
  - Space-frequency domain (Cook et al., 1955)
  - Time-frequency domain (Polack, 1988)

A. Belhomme, R. Badeau, Y. Grenier, and E. Humbert. Amplitude and phase dereverberation of harmonic signals. In *Proc. of IEEE WASPAA*, New Paltz, New York, USA, October 2017

S. Leglaive, R. Badeau, and G. Richard. Student's t source and mixing models for multichannel audio source separation. *IEEE Trans. Audio, Speech, Language Process.*, 26(5):1–15, May 2018







- II Properties of reverberation
- III Review of reverberation models
- IV Definition of the new stochastic model
- V Statistical properties of the model
- VI Experimental validation
- VII Conclusion





### Part II

### **Properties of reverberation**

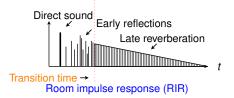
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# **新發展的**Time-frequency profile of reverberation

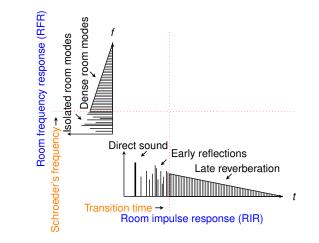




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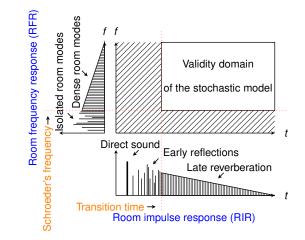


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# **新發展的**Time-frequency profile of reverberation



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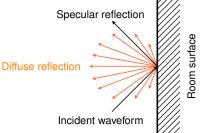
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# Space domain: diffuse sound field

*Diffusion*: reflections on the room surfaces are not

 specular (mirror-like), but rather scattered in various directions

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#### T.J. Schultz. Diffusion in reverberation rooms. Journal of Sound and Vibration, 16(1):17 – 28, 1971





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### Space domain: diffuse sound field Diffusion: reflections on the room surfaces are not specular (mirror-like), but rather scattered in various directions

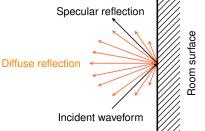
The acoustic field can be approximated as diffuse (Schultz, 1971)

- inside the time-frequency validity domain of the stochastic model
- if source/sensors are at least a half-wavelength away from walls



# Specular reflections on the room surfaces are not specular (mirror like) but Diffuse reflection

 specular (mirror-like), but rather scattered in various directions



- The acoustic field can be approximated as diffuse (Schultz, 1971)
  - inside the time-frequency validity domain of the stochastic model
  - if source/sensors are at least a half-wavelength away from walls
- After many reflections, the acoustic field is *uniform* and *isotropic*

T.J. Schultz. Diffusion in reverberation rooms. Journal of Sound and Vibration, 16(1):17 – 28, 1971





Measured RIRs from C4DM database (169 RIRs, Fs=96 kHz)

- Octagon room: 8 walls 7.5m length and domed ceiling 21m height
- 13 x 13 sensor positions distributed on a uniform square grid
- Space sampling of the omnidirectional microphone grid: D = 1m
- Reverberation time: RT60  $\approx$  2s

R. Stewart and M. Sandler. Database of omnidirectional and b-format room impulse responses. In *IEEE ICASSP*, pages 165–168, Center for Digital Music (C4DM), QMUL, London, March 2010

Emmanuel Vincent and Douglas R. Campbell. Roomsimove. GNU Public License, 2008. http://homepages.loria.fr/evincent/software/Roomsimove.zip



### **新聞 Experiments**

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Measured RIRs from C4DM database (169 RIRs, Fs=96 kHz)

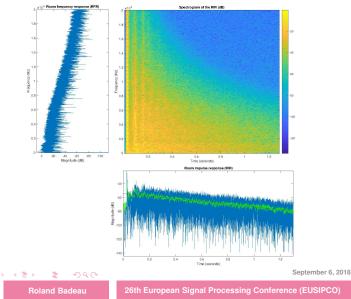
- Octagon room: 8 walls 7.5m length and domed ceiling 21m height
- 13 x 13 sensor positions distributed on a uniform square grid
- Space sampling of the omnidirectional microphone grid: D = 1m
- Reverberation time: RT60  $\approx$  2s
- Synthetic RIRs from Roomsimove toolbox (400 RIRs, Fs=16 kHz)
  - Shoebox room: 4 x 5 x 2.5 m<sup>3</sup>
  - Random source and sensor positions, random sensor orientations
  - Distance between the omnidirectional microphones: *D* = 20cm
  - Reverberation time: RT60  $\approx$  0.1s

R. Stewart and M. Sandler. Database of omnidirectional and b-format room impulse responses. In *IEEE ICASSP*, pages 165–168, Center for Digital Music (C4DM), QMUL, London, March 2010

Emmanuel Vincent and Douglas R. Campbell. Roomsimove. GNU Public License, 2008. http://homepages.loria.fr/evincent/software/Roomsimove.zip



# Ime-frequency profile (C4DM database)



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### Part III

### **Review of reverberation models**

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#### Schroeder (1962) and Moorer (1979): the RIR at microphone i is

$$h_i(t) = b_i(t)e^{-\alpha t}\mathbf{1}_{t\geq 0}$$

Manfred R. Schroeder. Frequency-correlation functions of frequency responses in rooms. *The Journal of the Acoustical Society of America*, 34(12):1819–1823, 1962

James A. Moorer. About this reverberation business. Computer Music Journal, 3(2):13-28, 1979





#### Schroeder (1962) and Moorer (1979): the RIR at microphone i is

$$h_i(t) = \mathbf{b}_i(t)\mathbf{e}^{-\alpha t}\mathbf{1}_{t\geq 0}$$

**b\_i(t)** is a centered white Gaussian process

Manfred R. Schroeder. Frequency-correlation functions of frequency responses in rooms. *The Journal of the Acoustical Society of America*, 34(12):1819–1823, 1962

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Schroeder (1962) and Moorer (1979): the RIR at microphone *i* is

$$h_i(t) = b_i(t)e^{-\alpha t}\mathbf{1}_{t\geq 0}$$

 $b_i(t)$  is a centered white Gaussian process

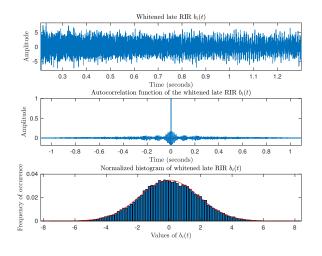
•  $\alpha > 0$  is related to the reverberation time:  $RT_{60} = \frac{3 \ln(10)}{\alpha}$ 

Manfred R. Schroeder. Frequency-correlation functions of frequency responses in rooms. *The Journal of the Acoustical Society of America*, 34(12):1819–1823, 1962

James A. Moorer. About this reverberation business. Computer Music Journal, 3(2):13-28, 1979



# Validation of time model (C4DM database)



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#### The RFR is the Fourier transform of the RIR:

$$\mathcal{F}_{h_i}(f) = \int_{t \in \mathbb{R}} h_i(t) e^{-2i\pi f t} dt$$

Manfred R. Schroeder. Frequency-correlation functions of frequency responses in rooms. *The Journal of the Acoustical Society of America*, 34(12):1819–1823, 1962





The RFR is the Fourier transform of the RIR:

$$\mathcal{F}_{h_i}(f) = \int_{t \in \mathbb{R}} h_i(t) e^{-2i\pi f t} dt$$

Schroeder (1962):  $\mathcal{F}_{h_i}(f)$  is a stationary random process

Manfred R. Schroeder. Frequency-correlation functions of frequency responses in rooms. *The Journal of the Acoustical Society of America*, 34(12):1819–1823, 1962





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$$\mathcal{F}_{h_i}(f) = \int_{t \in \mathbb{R}} h_i(t) e^{-2\imath \pi f t} dt$$

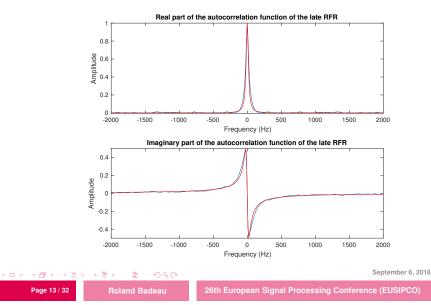
- Schroeder (1962):  $\mathcal{F}_{h_i}(f)$  is a stationary random process
- Complex autocorrelation function of  $\mathcal{F}_{h_i}(f)$ :

$$\operatorname{corr}\left[\mathcal{F}_{h_i}(f_1), \, \mathcal{F}_{h_i}(f_2)\right] = \frac{1}{1 + \imath \pi \frac{f_1 - f_2}{\alpha}}$$

Manfred R. Schroeder. Frequency-correlation functions of frequency responses in rooms. *The Journal of the Acoustical Society of America*, 34(12):1819–1823, 1962



# Validation of spectral model (Roomsimove)



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Correlation at frequency *f* between sensors (Cook et al., 1955):  $\begin{bmatrix} T & (D) & T & (D] \\ (2\pi fD) \end{bmatrix}$ 

corr 
$$\left[\mathcal{F}_{h_1}(f), \mathcal{F}_{h_2}(f)\right] = \operatorname{sinc}\left(\frac{2\pi f D}{c}\right)$$

D is the distance between microphones

c is the speed of sound in the air (≈ 343 m/s)

R. K. Cook, R. V. Waterhouse, R. D. Berendt, S. Edelman, and M. C. Thompson Jr. Measurement of correlation coefficients in reverberant sound fields. *The Journal of the Acoustical Society of America*, 27(6):1072–1077, 1955



## Space-frequency domain

Correlation at frequency *f* between sensors (Cook et al., 1955):

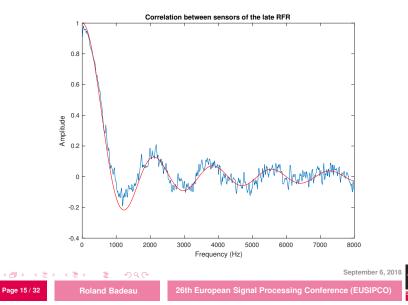
corr 
$$\left[\mathcal{F}_{h_1}(f), \mathcal{F}_{h_2}(f)\right] = \operatorname{sinc}\left(\frac{2\pi f D}{c}\right)$$

- D is the distance between microphones
- *c* is the speed of sound in the air ( $\approx$  343 m/s)
- Assumptions:
  - Plane waves (far field)
  - Isotropic incident waves (diffuse acoustic field)

R. K. Cook, R. V. Waterhouse, R. D. Berendt, S. Edelman, and M. C. Thompson Jr. Measurement of correlation coefficients in reverberant sound fields. *The Journal of the Acoustical Society of America*, 27(6):1072–1077, 1955

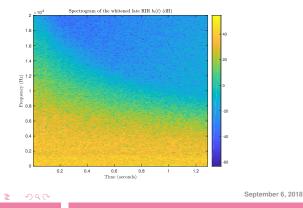


# Validation of space model (Roomsimove)



### **新资额** Time-frequency domain

- Moorer (1979): the RIR at microphone *i* is  $h_i(t) = b_i(t)e^{-\alpha t}\mathbf{1}_{t\geq 0}$ where  $b_i(t)$  is a centered white Gaussian process
- Spectrogram of  $b_i(t)$  (C4DM database):





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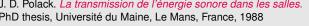
• Moorer (1979): the RIR at microphone *i* is  $h_i(t) = b_i(t)e^{-\alpha t}\mathbf{1}_{t>0}$ 

J. D. Polack. La transmission de l'énergie sonore dans les salles. PhD thesis, Université du Maine, Le Mans, France, 1988

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- Moorer (1979): the RIR at microphone *i* is  $h_i(t) = b_i(t)e^{-\alpha t}\mathbf{1}_{t>0}$
- **Polack** (1988):  $b_i(t)$  is a centered stationary Gaussian process, whose power spectral density (PSD) B(f) has slow variations

J. D. Polack. La transmission de l'énergie sonore dans les salles. PhD thesis, Université du Maine, Le Mans, France, 1988

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### **副選擇的** Time-frequency domain

• Moorer (1979): the RIR at microphone *i* is  $h_i(t) = b_i(t)e^{-\alpha t}\mathbf{1}_{t\geq 0}$ 

Polack (1988):  $b_i(t)$  is a centered *stationary* Gaussian process, whose power spectral density (PSD) B(t) has slow variations

Polack (1988): the Wigner distribution of the RIR is

$$\mathcal{W}_{h_i,h_i}(t,f) = B(f)e^{-2\alpha t}\mathbf{1}_{t\geq 0}.$$

J. D. Polack. *La transmission de l'énergie sonore dans les salles.* PhD thesis, Université du Maine, Le Mans, France, 1988



### **新聞 Time-frequency domain**

• Moorer (1979): the RIR at microphone *i* is  $h_i(t) = b_i(t)e^{-\alpha t}\mathbf{1}_{t\geq 0}$ 

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Polack (1988): the Wigner distribution of the RIR is

$$\mathcal{W}_{h_i,h_i}(t,f) = B(f)e^{-2\alpha t}\mathbf{1}_{t\geq 0}.$$

• Wigner distribution of two  $2^{nd}$  order random processes  $\psi_1$ ,  $\psi_2$ :

$$\mathcal{W}_{\psi_1,\psi_2}(t,f) = \int_{\mathbb{R}} \operatorname{cov}[\psi_1(t+rac{u}{2}),\psi_2(t-rac{u}{2})]e^{-2\imath\pi f u} du$$

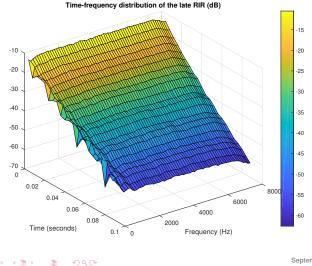
J. D. Polack. *La transmission de l'énergie sonore dans les salles.* PhD thesis, Université du Maine, Le Mans, France, 1988

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# Validation of Polack's model (Roomsimove)



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### Part IV

### Definition of the new stochastic model

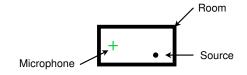
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# Source image principle (shoebox room)











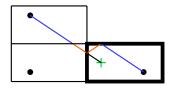
# Source image principle (2 reflections)





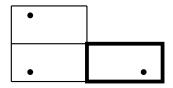














# Source image principle (infinite space)

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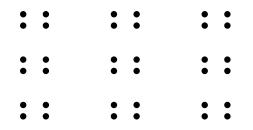
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# Source image principle (no wall)







Source image principle: specular reflections in a shoebox room ⇒ spatially uniform distribution of source images

Jean-Dominique Polack. Playing billiards in the concert hall: The mathematical foundations of geometrical room acoustics. *Applied Acoustics*, 38(2):235 – 244, 1993

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- Source image principle: specular reflections in a shoebox room ⇒ spatially uniform distribution of source images
- Proposed stochastic model: source images are spatially distributed according to a uniform Poisson distribution:

Jean-Dominique Polack. Playing billiards in the concert hall: The mathematical foundations of geometrical room acoustics. *Applied Acoustics*, 38(2):235 – 244, 1993

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- Source image principle: specular reflections in a shoebox room ⇒ spatially uniform distribution of source images
- Proposed stochastic model: source images are spatially distributed according to a *uniform Poisson distribution*:
  - for any volume  $V \subset \mathbb{R}^3$ ,  $N(V) \sim \mathcal{P}(\lambda|V|)$  with  $\lambda > 0$

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  - for any volume  $V \subset \mathbb{R}^3$ ,  $N(V) \sim \mathcal{P}(\lambda|V|)$  with  $\lambda > 0$
  - independent of the microphone position and true source position

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  - independent of the microphone position and true source position
  - independent of the room geometry (Polack, 1993)

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Jean-Dominique Polack. Playing billiards in the concert hall: The mathematical foundations of geometrical room acoustics. *Applied Acoustics*, 38(2):235 – 244, 1993

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holds even more in a *diffuse* (spatially uniform) acoustic field

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holds even more in a *diffuse* (spatially uniform) acoustic field

Assumption: microphone and source images are omnidirectional

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- Source image principle: specular reflections in a shoebox room ⇒ spatially uniform distribution of source images
- Proposed stochastic model: source images are spatially distributed according to a *uniform Poisson distribution*:
  - for any volume  $V \subset \mathbb{R}^3$ ,  $N(V) \sim \mathcal{P}(\lambda|V|)$  with  $\lambda > 0$
  - independent of the microphone position and true source position
  - independent of the room geometry (Polack, 1993)
  - holds even more in a diffuse (spatially uniform) acoustic field
- Assumption: microphone and source images are *omnidirectional* 
  - The attenuation of sound waves is exponential w.r.t. the distance, isotropic and independent of frequency

Jean-Dominique Polack. Playing billiards in the concert hall: The mathematical foundations of geometrical room acoustics. *Applied Acoustics*, 38(2):235 – 244, 1993



#### **副發展聞**Unified stochastic reverberation model

For a set of sensors at positions  $\{\mathbf{x}_i\}_i \in \mathbb{R}^3$ , the RIRs are

$$h_i(t) = \int_{\boldsymbol{x} \in \mathbb{R}^3} h(t, \|\boldsymbol{x} - \boldsymbol{x}_i\|_2) e^{-\frac{lpha}{c} \|\boldsymbol{x} - \boldsymbol{x}_i\|_2} dN(\boldsymbol{x})$$

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#### Washington and the stochastic reverberation model Section 2018 Section 201

For a set of sensors at positions  $\{\boldsymbol{x}_i\}_i \in \mathbb{R}^3$ , the RIRs are

$$h_i(t) = \int_{\boldsymbol{x} \in \mathbb{R}^3} h(t, \|\boldsymbol{x} - \boldsymbol{x}_i\|_2) e^{-\frac{\alpha}{c} \|\boldsymbol{x} - \boldsymbol{x}_i\|_2} dN(\boldsymbol{x})$$

•  $\mathbf{X} \in \mathbb{R}^3$  is a possible source image position





#### We unified stochastic reverberation model

For a set of sensors at positions  $\{\boldsymbol{x}_i\}_i \in \mathbb{R}^3$ , the RIRs are

$$h_i(t) = \int_{\boldsymbol{x} \in \mathbb{R}^3} h(t, \|\boldsymbol{x} - \boldsymbol{x}_i\|_2) e^{-\frac{\alpha}{c} \|\boldsymbol{x} - \boldsymbol{x}_i\|_2} dN(\boldsymbol{x})$$

- $\textbf{\textit{x}} \in \mathbb{R}^3$  is a possible source image position
- $dN(\mathbf{x}) \sim \mathcal{P}(\lambda d\mathbf{x})$  are independent Poisson increments



#### 

For a set of sensors at positions  $\{\boldsymbol{x}_i\}_i \in \mathbb{R}^3$ , the RIRs are

$$h_i(t) = \int_{{\pmb{x}} \in \mathbb{R}^3} h(t, \|{\pmb{x}} - {\pmb{x}}_i\|_2) e^{-\frac{lpha}{c} \|{\pmb{x}} - {\pmb{x}}_i\|_2} dN({\pmb{x}})$$

- $\mathbf{X} \in \mathbb{R}^3$  is a possible source image position
- $dN(\mathbf{x}) \sim \mathcal{P}(\lambda d\mathbf{x})$  are independent Poisson increments
- $\alpha > 0$  is the attenuation coefficient (in Hz)



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For a set of sensors at positions  $\{\boldsymbol{x}_i\}_i \in \mathbb{R}^3$ , the RIRs are

$$h_i(t) = \int_{\boldsymbol{x} \in \mathbb{R}^3} h(t, \|\boldsymbol{x} - \boldsymbol{x}_i\|_2) e^{-\frac{\alpha}{c} \|\boldsymbol{x} - \boldsymbol{x}_i\|_2} dN(\boldsymbol{x})$$

- $\mathbf{X} \in \mathbb{R}^3$  is a possible source image position
- $dN(\mathbf{x}) \sim \mathcal{P}(\lambda d\mathbf{x})$  are independent Poisson increments
- $\alpha$  > 0 is the attenuation coefficient (in Hz)
- c > 0 is the speed of sound in the air ( $\approx 343$  m/s)



#### We will Unified stochastic reverberation model

For a set of sensors at positions  $\{\boldsymbol{x}_i\}_i \in \mathbb{R}^3$ , the RIRs are

$$h_i(t) = \int_{\boldsymbol{x} \in \mathbb{R}^3} h(t, \|\boldsymbol{x} - \boldsymbol{x}_i\|_2) e^{-\frac{lpha}{c} \|\boldsymbol{x} - \boldsymbol{x}_i\|_2} dN(\boldsymbol{x})$$

- $\mathbf{x} \in \mathbb{R}^3$  is a possible source image position
- $dN(\mathbf{x}) \sim \mathcal{P}(\lambda d\mathbf{x})$  are independent Poisson increments
- α > 0 is the attenuation coefficient (in Hz)
- c > 0 is the speed of sound in the air ( $\approx$  343 m/s)
- h(t, r) is a coherent sum of monochromatic spherical waves:

$$h(t,r) = \int_{f \in \mathbb{R}} A(f) \, \frac{e^{2i\pi f\left(t-\frac{f}{c}\right)}}{r} df$$

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#### We unified stochastic reverberation model

For a set of sensors at positions  $\{\boldsymbol{x}_i\}_i \in \mathbb{R}^3$ , the RIRs are

$$h_i(t) = \int_{x \in \mathbb{R}^3} h(t, \|x - x_i\|_2) e^{-\frac{\alpha}{c} \|x - x_i\|_2} dN(x)$$

- $\mathbf{x} \in \mathbb{R}^3$  is a possible source image position
- $dN(\mathbf{x}) \sim \mathcal{P}(\lambda d\mathbf{x})$  are independent Poisson increments
- α > 0 is the attenuation coefficient (in Hz)
- c > 0 is the speed of sound in the air ( $\approx 343$  m/s)
- *h*(*t*, *r*) is a coherent sum of *monochromatic spherical waves*:

$$h(t,r) = \int_{f\in\mathbb{R}} A(f) \, \frac{e^{2i\pi f\left(t-\frac{r}{c}\right)}}{r} df$$

• We get  $h_i(t) = e^{-\alpha(t-T)} b_i(t)$ ,  $b_i(t) = \int_{\mathbf{x} \in \mathbb{R}^3} \frac{g\left(t-T-\frac{\|\mathbf{x}-\mathbf{x}_i\|_2}{c}\right)}{\|\mathbf{x}-\mathbf{x}_i\|_2} dN(\mathbf{x})$ with  $g(t) \in L^2([-T, T])$  satisfying technical conditions





#### Part V

#### Statistical properties of the model

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Asymptotic normality: with  $h_i(t) = e^{-\alpha(t-T)} b_i(t)$ , when  $t \to +\infty$ ,  $b(t) = [b_i(t), b_j(t)]^{\top}$  converges to a stationary Gaussian process



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#### 送 聞 Space-time domain

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## Space-time domain

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## Space-time domain

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- When  $t \to +\infty$ , if  $b_i(t)$  is white, we retrieve (Moorer, 1979)
- Between two sensors:  $\forall t \ge 2T + \frac{D}{c}$ ,  $\boldsymbol{b}(t) = [b_i(t), b_j(t)]^\top$  is a centered WSS process, of cross-PSD

$$B_{i,j}(f) = B(f)\operatorname{sinc}(\frac{2\pi fD}{c})$$
 (new)



## Space-frequency domain

Between two sensors:  $\forall f_1, f_2 \in \mathbb{R}$ ,

$$\operatorname{corr}[\mathcal{F}_{h_{i}}(f_{1}), \mathcal{F}_{h_{j}}(f_{2})] = \frac{e^{-\frac{\alpha D}{c} - 2\iota\pi(f_{1} - f_{2})(T + \frac{D}{c}c)\operatorname{sinc}(\frac{\pi(f_{1} + f_{2})D}{c})}{1 + \iota\pi\frac{f_{1} - f_{2}}{\alpha}} \quad (\text{new})$$



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At one sensor (i = j, D = 0) with  $b_i(t)$  white (T = 0):

$$\operatorname{corr}[\mathcal{F}_{h_i}(f_1), \mathcal{F}_{h_i}(f_2]) = \frac{1}{1 + i\pi \frac{f_1 - f_2}{\alpha}} \quad (\text{Schroeder}, 1962)$$

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#### **警務部間** Space-frequency domain

Between two sensors:  $\forall f_1, f_2 \in \mathbb{R}$ ,

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At one frequency ( $f_1 = f_2 = f$ ), with no attenuation ( $\alpha = 0$ ):

 $\operatorname{corr}[\mathcal{F}_{h_i}(f), \mathcal{F}_{h_j}(f)] = \operatorname{sinc}(\frac{2\pi f D}{c})$  (Cook et al., 1955)

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## Space-time-frequency domain

Between two sensors:  $\forall f \in \mathbb{R}, \forall t \geq 2T + \frac{D}{2c}$ ,

$$\mathcal{W}_{h_i,h_j}(t,f) = B(f)e^{-2\alpha(t-T)}\operatorname{sinc}(\frac{2\pi fD}{c})$$
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#### **警務部間** Space-time-frequency domain

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$$\mathcal{W}_{h_i,h_i}(t,f) = B(f) e^{-2\alpha(t-T)}$$
 (Polack, 1988)

Time-frequency correlation:  $\forall f \in \mathbb{R}, \forall t \ge 2T + \frac{D}{2c}$ ,

$$rac{\mathcal{W}_{h_i,h_j}(t,f)}{\mathcal{W}_{h_i,h_j}(t,f)} = \operatorname{sinc}(rac{2\pi f D}{c})$$
 (new)

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#### Part VI

#### **Experimental validation**

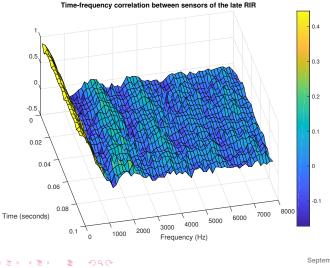
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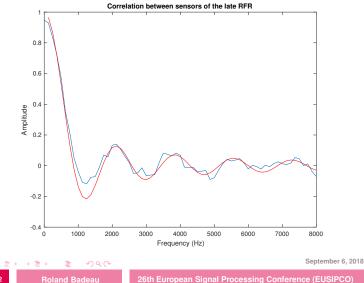
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Image: 0

# 記题 Trojection on the frequency axis



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#### Part VII

#### Conclusion

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- Summary
  - New reverberation model that unifies and generalizes known results







- Summary
  - New reverberation model that unifies and generalizes known results
  - Also applicable before the transition time:







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    - the Poisson distribution makes  $h_i(t)$  impulsive in early reverberation







#### Summary

- New reverberation model that unifies and generalizes known results
- Also applicable before the transition time:
  - the Poisson distribution makes  $h_i(t)$  impulsive in early reverberation
- Perspectives







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    - Directional sources, directional microphones
    - Non-perfectly diffuse acoustic fields
    - Frequency-dependent attenuation  $\alpha$



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  - Acoustics
    - Directional sources, directional microphones
    - Non-perfectly diffuse acoustic fields
    - Frequency-dependent attenuation  $\alpha$
  - Signal processing
    - Fast algorithm to estimate the model in discrete time
    - Applications: source separation, dereverberation,...











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Polack (1988): the RIR at microphone i is

$$h_i(t) = b_i(t)e^{-\alpha t}\mathbf{1}_{t\geq 0}$$

where  $b_i(t)$  is a centered *stationary* Gaussian process, whose PSD B(f) has slow variations

J. D. Polack. *La transmission de l'énergie sonore dans les salles.* PhD thesis, Université du Maine, Le Mans, France, 1988





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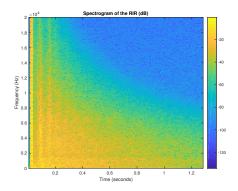
Then the Wigner distribution of the RIR is

$$\mathcal{W}_{h_i,h_i}(t,f) = B(f)e^{-2\alpha t}\mathbf{1}_{t\geq 0}.$$

J. D. Polack. *La transmission de l'énergie sonore dans les salles.* PhD thesis, Université du Maine, Le Mans, France, 1988



## Work in progress (C4DM database)



Polack (1988): the attenuation actually depends on the frequency:

$$\mathcal{W}_{h_i,h_i}(t,f)=B(f)e^{-2lpha(f)t}\mathbf{1}_{t\geq 0}$$

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