



NATURAL BORN INTERACTIVE



Long Project with Audiogaming

Additive Synthesis with Inverse Fourier Transform for Non-Stationary Signals

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Introduction

- The company

- Objective

- Context of the Project

- Work Environment and Project Management

Method Overview

- Additive Synthesis (Time Domain)

- Windowing

- Peak Detection

- Result

The additive synthesis

- General approach: The time domain

- General approach: The frequency domain

- General approach: The frames

Stationary Signals

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- ▶ Localization: Toulouse, Paris
- ▶ Activity: Audio plug-in (VSTs and RTAS)
- ▶ Main customers: Film and Video Game Industry (Sony, Ubisoft)
- ▶ 10 employees

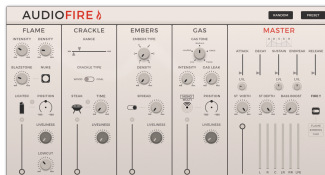


Figure: Audiofire: audio plug-in that recreates fire sound



- We are continuing the Audiogaming long project from 2015 (Emilie Abia, Lili Zheng, Quentin Biache)

Objective : Synthesizing sounds from their spectrum with a FFT^{-1}

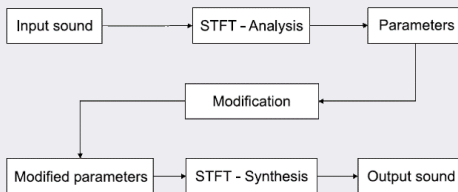


Figure: General approach for modifying a sound in the spectral domain

- We have to implement a new method of additive synthesis \Rightarrow computationally very fast



- ▶ 6 weeks only \Rightarrow Focus on the synthesis method only.

Given codes in Python and Matlab from the 2015 project :

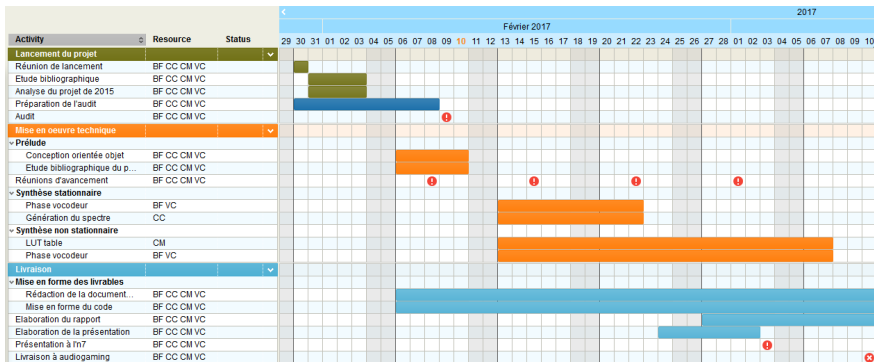
- ▶ Python : Analysis estimator of sinus parameters and sinus generation with those parameters (only stationary)
 - ▶ Matlab : Some reasearch on the Non-stationary synthesis with the LUT of lobes
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- ▶ We made our own OOP codes in Python
 - ▶ We have taken the analysis estimator code to test our final synthesis



Figure: *PyCharm* as Python IDE , *Slack* to communicate, *GitHub* to stock the codes and have a versionning, *Freedcamp* to plan the project events

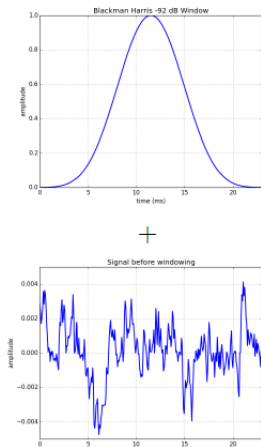
Introduction

Project Management : Gantt Chart



Method Overview : Additive synthesis

Additive Synthesis



Windowing step :

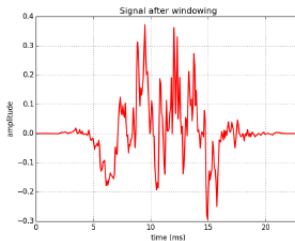
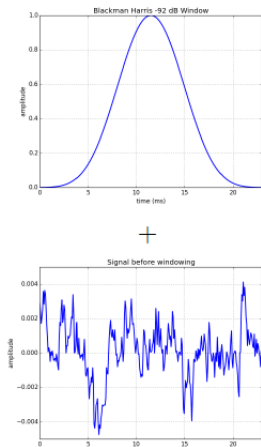


Figure: *Windowing step*

Method Overview : Analysis

Windowing



Windowing step :

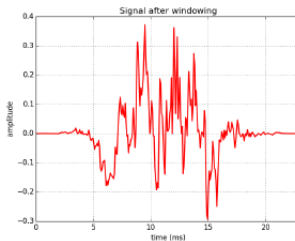


Figure: *Windowing step*

Method Overview : Analysis

Peak Detection



Peak detection and extraction of parameters by STPT (particular Short Time Fourier Transform):

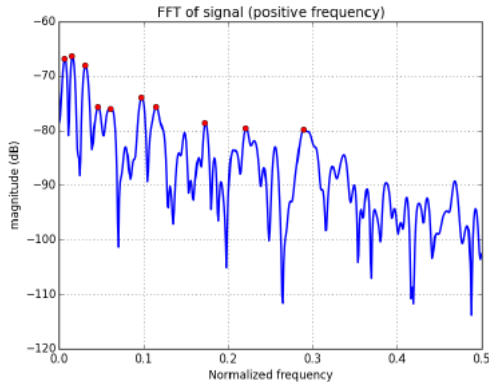


Figure: *Peak detection*



Additive synthesis according to the parameters from the analysis:

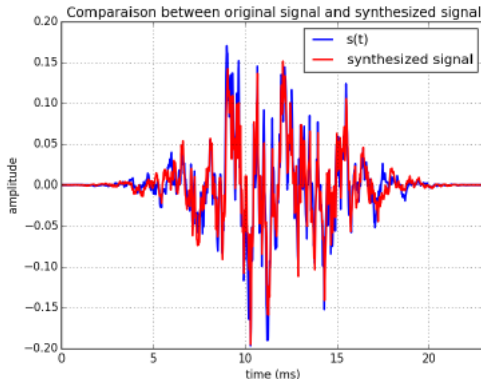


Figure: *Synthesized signal vs Original signal*

The additive synthesis

General approach: The time domain



The sound signal is represented as a sum of N sinusoids:

$$x(t) = \sum_{n=1}^N a_n \sin(2\pi f_n t + \phi_n)$$

- ▶ Very costly to implement
- ▶ Impossible to compute in real-time

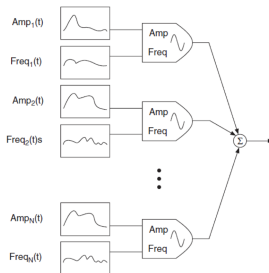


Figure: *The additive synthesis*

The additive synthesis

General approach: The frequency domain



We generate the sinusoids in frequency domain in order to reduce the computation time :

- ▶ Window the signal to maximize the energy in the main lobe
- ▶ We only keep the main lobe for each sine (9 points)
- ▶ We assume that the parameters (amplitude, frequency, phase) are already given by the analysis

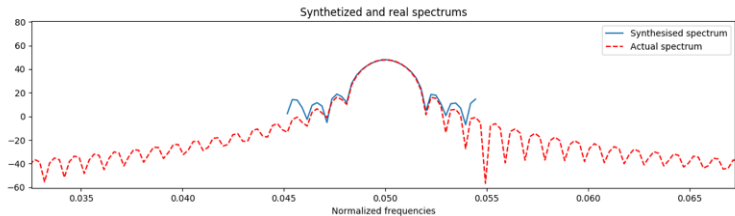


Figure: *Windowed sine lobe*

The additive synthesis

The frames



The sound signal is a frame-by-frame signal:

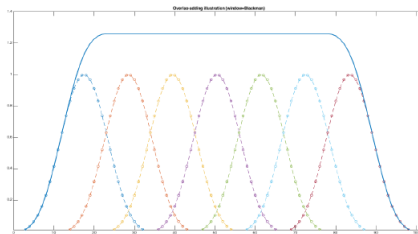


Figure: *Sum of small size Hanning windows*

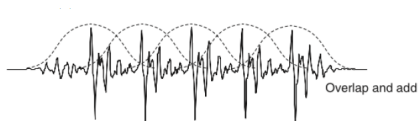


Figure: *Overlap and add*

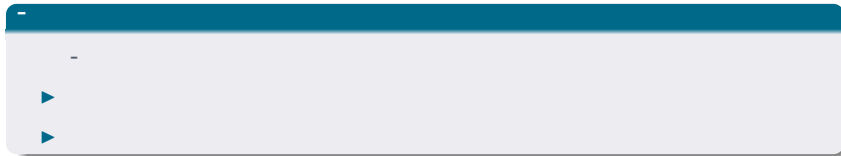




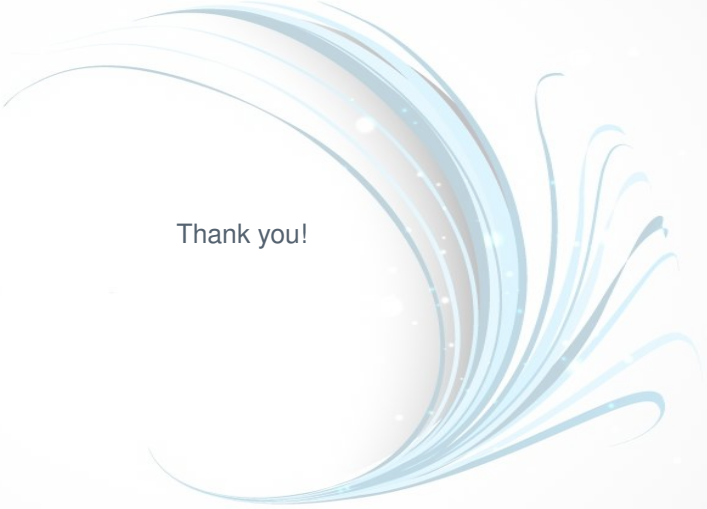
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Thank you!