

TEN HAND PIANO



CATÓLICA
UNIVERSIDADE CATÓLICA PORTUGUESA | PORTO
Escola das Artes



PSOs @ CASA DA MÚSICA



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2000/2006 – MTG, UPF in Barcelona, Spain

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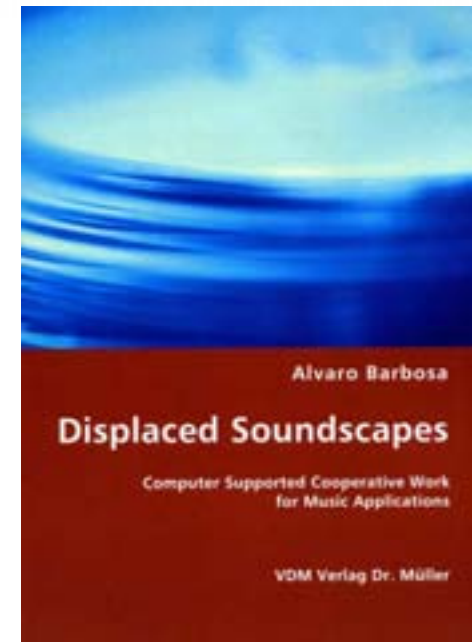
Displaced Soundscapes



Displaced Soundscapes is a metaphorical description of the way lively generated sounds can be perceived over the Internet.

(Book) Barbosa, A. 2008. “Displaced Soundscapes”
VDM Verlag (ISBN: 978-3-8364-7154-1)

- Extensive Survey (up to 2006)
- Research on Latency and Internet Acoustics
- Models for Networked Music Systems
- Implementation “Public Sound Objects”





Latency and Network Acoustics

For the Human ear to perceive two simultaneous sounds, they should not be displaced in time over **20ms** (Hirsh, 1959)

The perception of two different sounds performed simultaneously is strongly dependent on:

- sound characteristics (ADSR, Timbre, Pitch or Loudness)
- music style (rhythmic, melodic,...)
- complementary feedback modalities (visual or physical stimuli)
- musicians background and experience

Jitter and delay asymmetry introduce further disruption in Network Communication.

Furthermore Latency is not a technological condition that can be overcome in the near future. Consider mobile or satellite technology or a communication setup that spans worldwide.



Moscow – Santiago (fiber-optic link)
Bidirectional Latency= **94,3 ms**

EPT<20ms
(Schuett 2002, Lago 2004)

EPT<25ms
(Alexander Carôt 2006)



Minimizing Latency's Disruption

Fast Network Connectivity

Large bandwidth, Smart Routing, Fast protocols, Fast ADC/DAC, ...

Latency Adaptive Tempo (LAT)

Tolerance to Latency increases in lower musical Tempo (BPMs). This inverse proportion principle results in a possible application which consists of a simple software function for network acoustic communication systems, that dynamically adapts the Musical Tempo (typically a referenced by a metronome sound) to the maximum value tolerated by the least “latency-tolerant” musician of an ensemble.

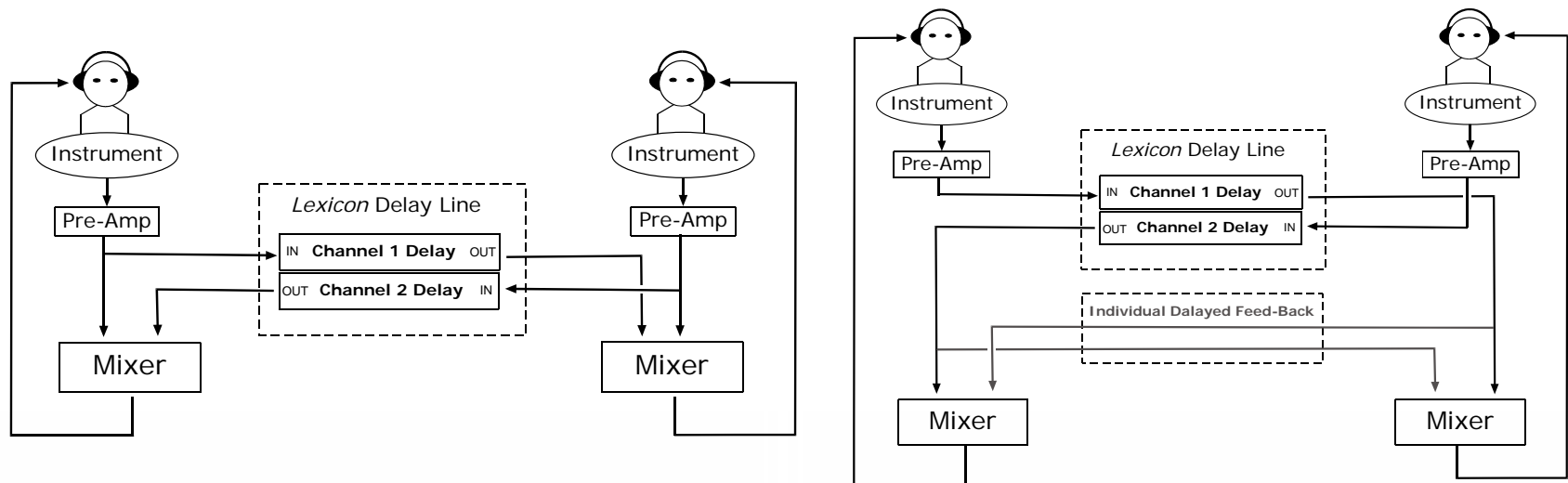
This dynamic adaptation is based on real-time latency measurement between peers.

LAT allows musicians to rehearse music as fast as their Network connectivity speed allows them to.

Individual Delayed Feed-Back (IDF)

A specific Feed-Back Topology that can enhance individual latency-tolerance.

If instead of having ensembles in which each musician receives direct acoustic feedback from their own instrument mixed with delayed feedback from the other performers, **every musician listens to his acoustic feed-back delayed, together and in sync with the others.**





Accepting Net-Delay

- Music is often (if not always) “custom designed” to the media where it will be broadcasted.

Whether it is about adapting a performance technique to the acoustics of a specific Church (like the XI century Venetian polychoral music style) or to condition the duration of popular music songs based on the audio storage capacity of early recording technologies.

- An approach to face such a scenario is accepting net-delay as the natural acoustics of the Internet
- To create music that makes sense in the internet media

Video @ Atau Tanaka's Interview, 2001

<http://crossfade.walkerart.org/>



The Shared Nature of the Internet

- Going beyond the paradigm in which we connect two or more remote spaces for a performative session by creating a **shared space** in a computer network, inside which users can achieve a certain degree of **immersion and flexibility in their behaviour**
- Users can freely join or leave this space, choosing to participate or just to listen to an on-going acoustic piece
- It is not oriented towards a time limited event scenario



Shared Sonic Environments

Auracle

Neuhaus and Freeman@Gtech

Co-Audicle for CHUCK

Wang @Princeton

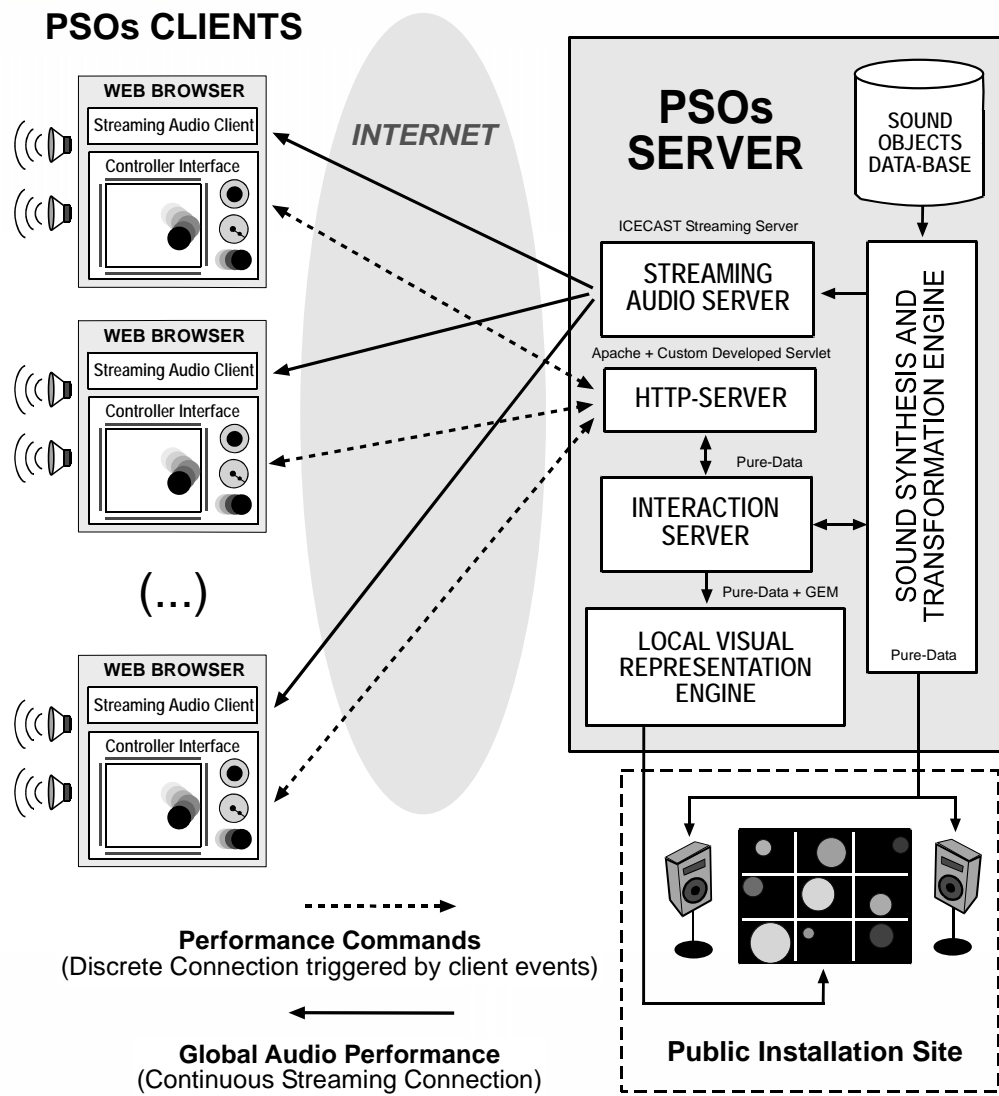
Public Sound Objects

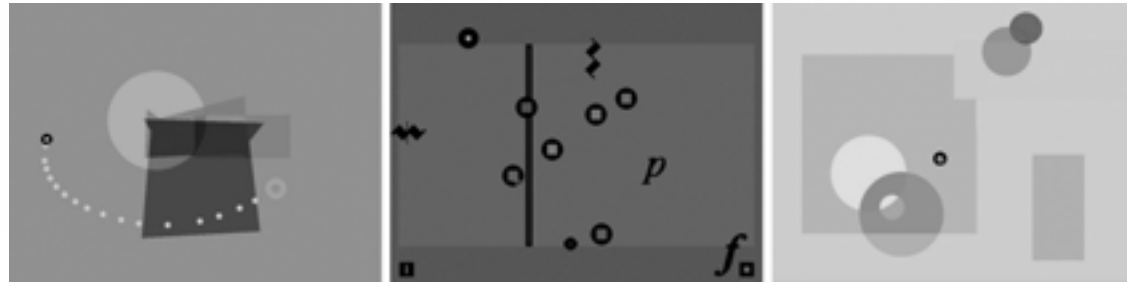
(Barbosa@MTG)

Original PSOs – 2004/2006, MTG (BCN)

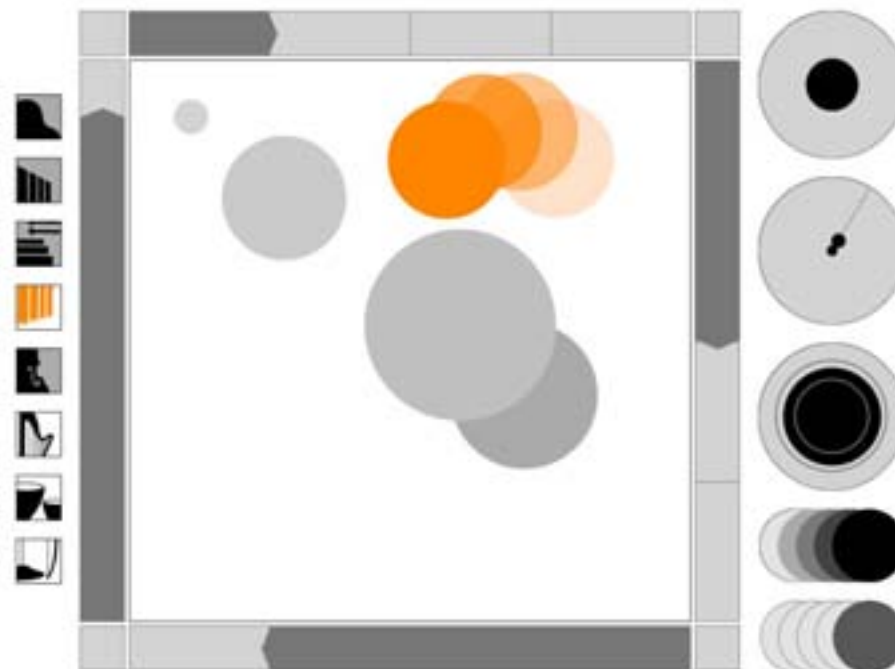


Public Sound Objects





Inspiration in **Small Fish (Fujihata and Furukawa 1999)**
A Behavior Driven Interface





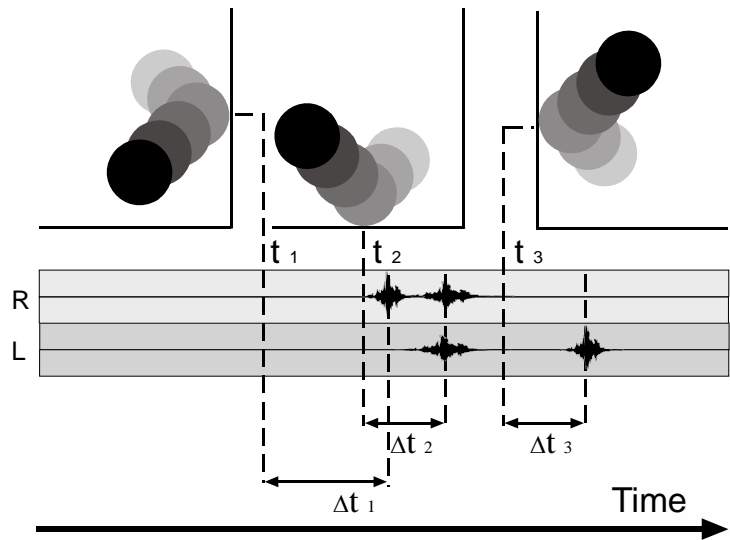
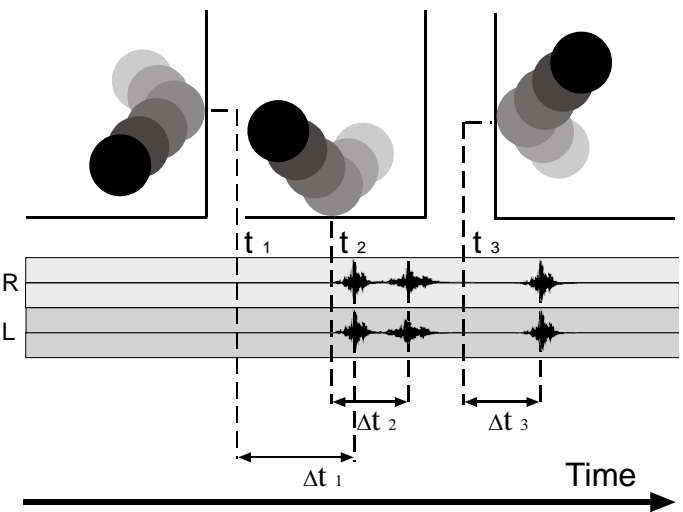
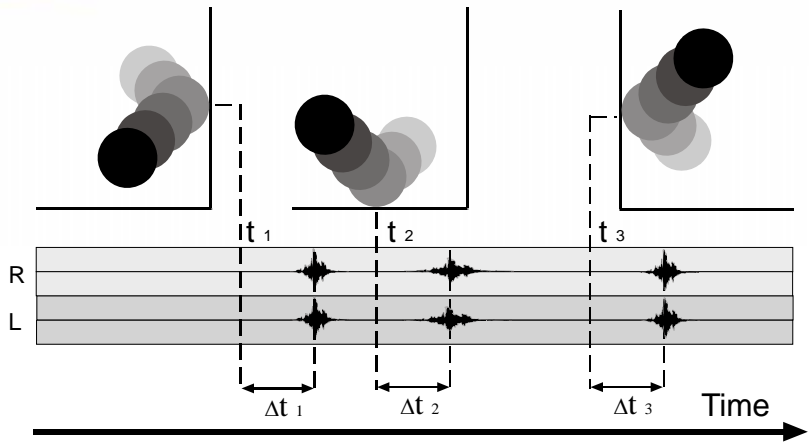
PSOs Interface

It is based on Pitch, Dynamics, Tempo and Timbre.

It tolerates Latency because:

- The resulting soundscape is musical. Yet, it does not require a strict synchronization between sound events
- Behavior driven interactive Interface (loose coupling interaction)
- Adaptive speed and dynamics (to network conditions)
- Individual Delayed Feed-Back (same piece to everyone)
- Coherent with visual cues (spatialization of sound helps)

Public Sound Objects: Panorama Cues



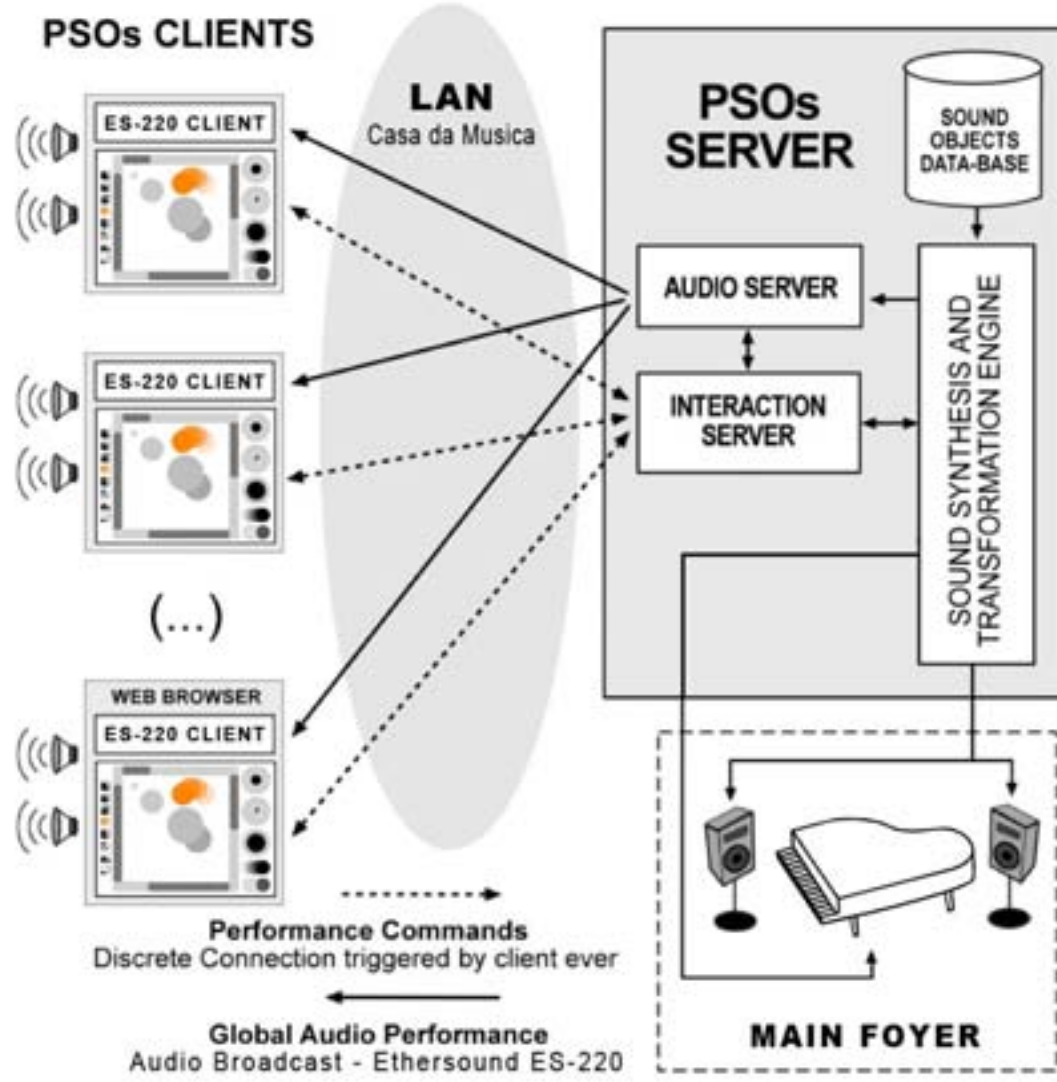
Local Network of PSOs – CITAR (Porto)

commissioned by “*Casa da Musica*”

http://en.wikipedia.org/wiki/Casa_da_musica



Public Sound Objects: 2007





EtherSound product range

Simple and cost-effective audio distribution
over standard Ethernet with extremely low latency

EtherSound allows to easily and economically create audio networks with extremely low latency using standard Ethernet IEEE 802.3x compliant cabling and components. Digigram's EtherSound devices exchange up to 64 channels of 24-bit digital audio at 48 kHz in each direction, plus control and monitoring information, via a single Ethernet cable with only 100 microseconds latency.



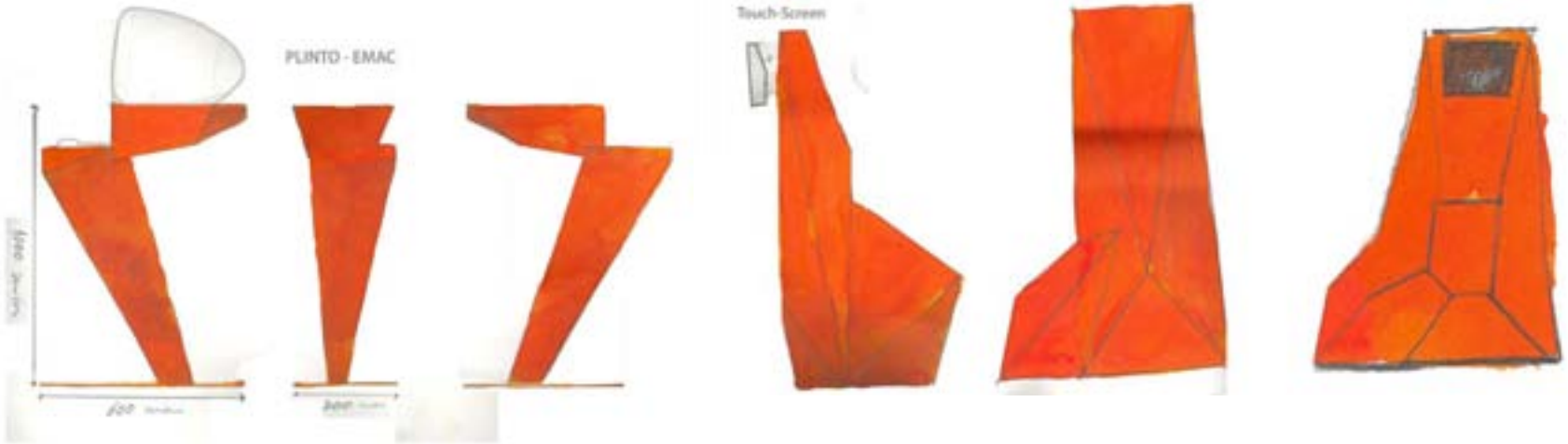
Ether
ES
Sound



**EtherSound
ES220/ES220-L**

Configuration		
Size	1 U 1/3 19" rack: 42 x 146.5 x 210 mm optional 19" rack mount bracket for up to three units	
Local audio channel selection	Manually by internal rotary switches	
Inputs/Outputs		
Audio	2 balanced analog mono line inputs, 2 servo-balanced analog mono line outputs ES220: inputs AND outputs ES220-L: inputs OR outputs (jumper selectable)	
Input/output impedance	22.2 k Ω / <100 Ω	
Maximum input level	+22 dBu or +10 dBV (jumper selectable)	
Maximum output level	+22 dBu (software adjustable)	
Analog output gain	from -72 dBu to 0 dB (software adjustable)	
Connectivity		
Analog/digital audio	15-pin D-Sub	
EtherSound	2 EtherCon female RJ45 compatible (connections "IN"/"OUT")	
GPIO	4 inputs and 4 outputs on 15-pin D-Sub	
Audio specifications		
Sampling frequency	48 kHz or 44.1 kHz (Primary Master: 48 kHz only)	
A/D and D/A converter resolution	24 bits	
Frequency response at 48 kHz	20 Hz – 20 kHz: \pm 0.2 dB	
Dynamic range –60 dBfs with Fs=48 kHz (20 Hz/20 kHz, unweighted)	>102 dB	
Distortion and noise (THD+N) at 1 kHz (–1 dBfs with Fs=48 kHz)	<–95 dB (0.0018%)	
Phase difference between channels: 20 Hz/20 kHz	0.5°/2°	
Crosstalk:	Inputs:	Outputs:
at 1 kHz	<–116 dB	120 dB
at 15 kHz (–1 dBfs with Fs=48 kHz)	<–92 dB	105 dB

Public Sound Objects: 2007



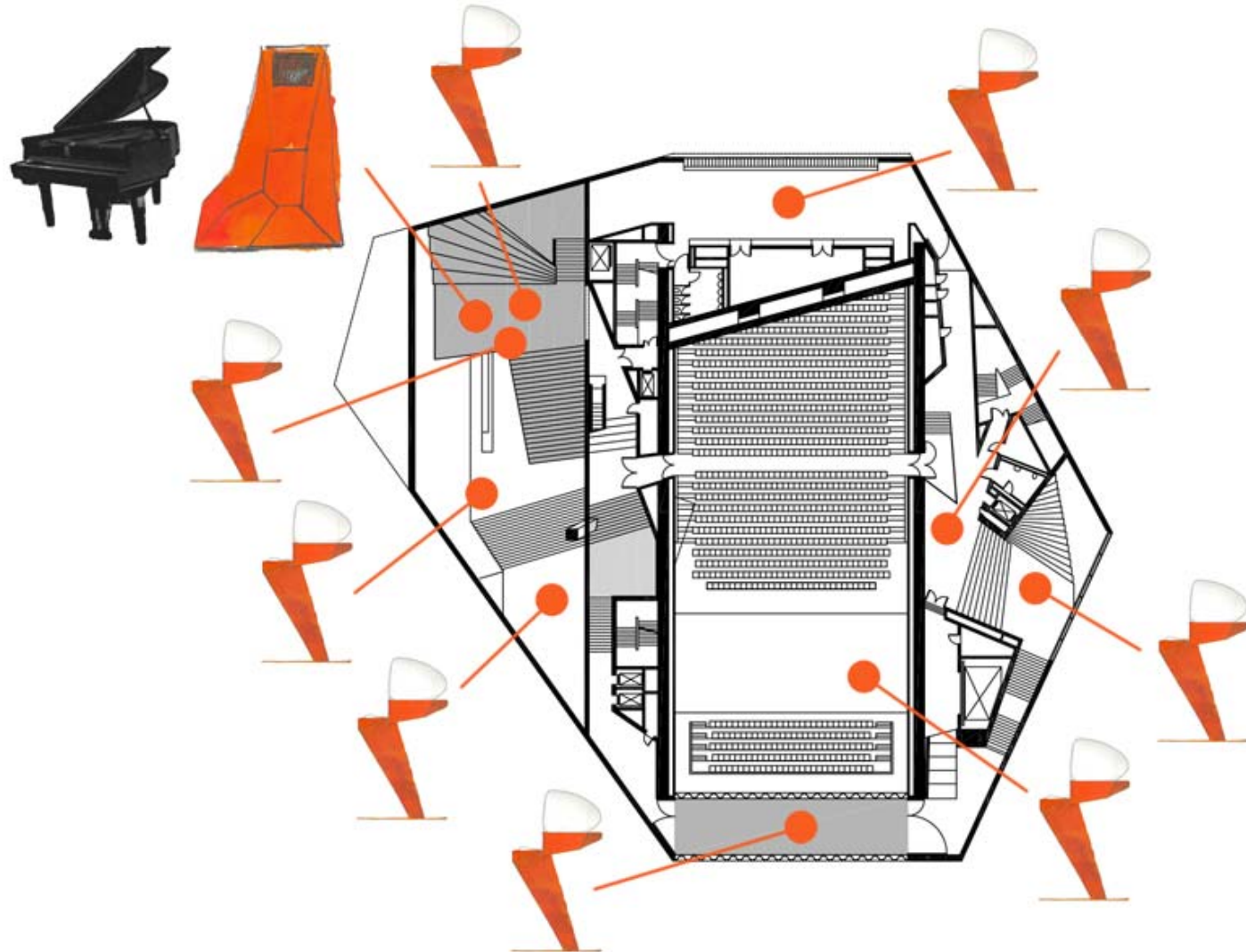


Client



Server

Public Sound Objects: 2007





CASA DA MÚSICA



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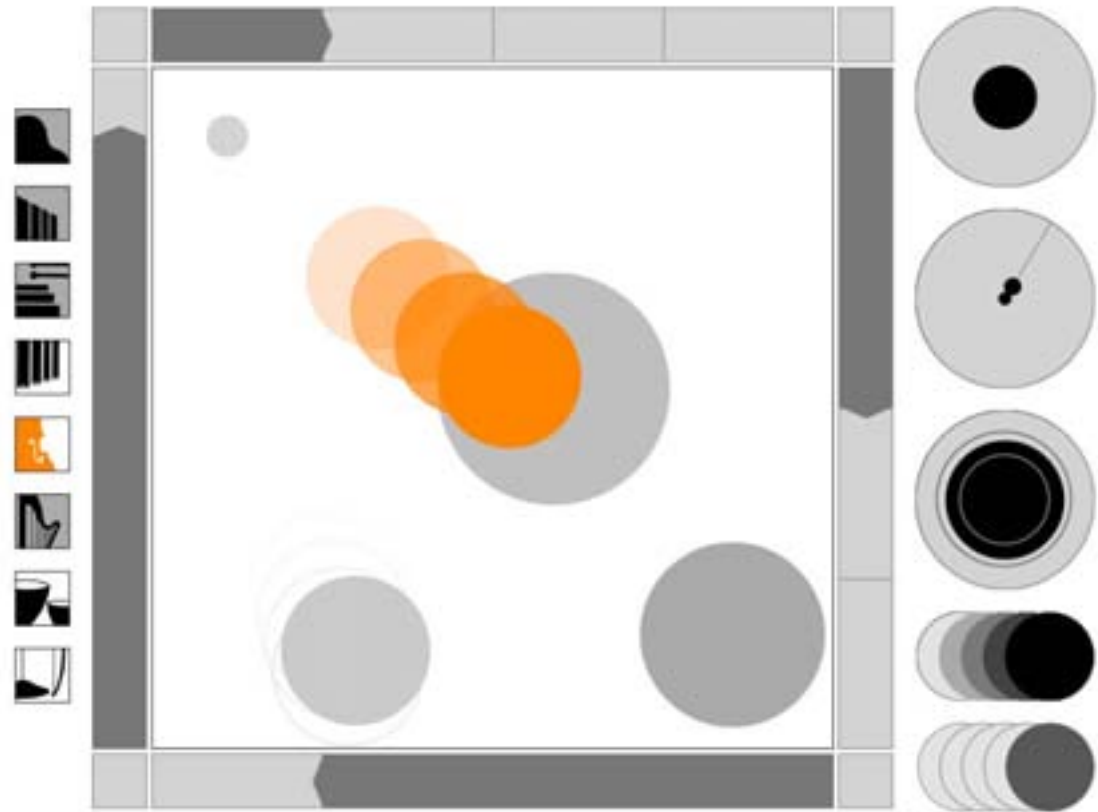
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Thank You!!!!

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<http://www.abarbosa.org/pso/>