# Semantic browsing of sound databases without keywords

**Grégoire Lafay**[1], **Nicolas Misdariis**[2], **Mathieu Lagrange**[1], **Mathias Rossignol**[2]
(gregoire.lafay@irccyn.ec-nantes.fr)

*IRCCyN, Ecole Centrale de Nantes* [1]  *IRCAM, STMS, UPMC* [2]

In this paper, we study the relevance of a semantic organization of sounds to ease the browsing of a sound database. For such a task, semantic access to data is traditionally implemented by a keyword selection process. However, various limitations of written language, such as word polysemy, ambiguities, or translation issues, may bias the browsing process.

We present and study the efficiency of two sound presentation strategies that organize sounds spatially so as to reflect an underlying semantic hierarchy. For the sake of comparison, we also consider a display whose spatial organization is based on acoustic cues. Those three displays are evaluated in terms of search speed in a crowdsourcing experiment. Results demonstrate the usefulness of using an implicit semantic organization to display sounds, both in terms of search speed and of efficiency of learning.

Audio content management and display, Semantic sound data mining

## 0 Introduction

With the growing capability of recording and storage devices, the problem of indexing large databases of audio has recently been the object of much attention [22]. Most of that effort is dedicated to the automatic inference of indexing metadata from the actual audio recording [23, 21]; in contrast, the ability to browse such databases in an effective manner has been less considered.

Most media assets management are based on keyword-driven queries. The user enters a word which best characterizes the desired item, and the interface presents him with items related to this word. The effectiveness of this principle is primarily based on the typological structure and nomenclature of the database. However, for databases and more specifically for databases of sounds, several issues arise:

1. Sounds, as many others things, can be described in many ways. Sound may be designated by their sources (a car door), as well as by the action of those sources (the slamming of a car door) or their environments (slamming a car door in a garage) [8, 13, 2]. Designing an effective keyword-based search system requires an accurate description of each sound, which has to be tunable to the sound representation of each user.
2. Pre-defined verbal descriptions of the sounds made available to the users may potentially bias their browsing and final selection.
3. Localization of the query interface is made difficult as the translation of some words referring to qualitative aspects of the sound, such as its ambiance, is notoriously ambiguous and subject to cultural specificities.
4. Unless considerable time and resources are invested into developing a multilingual interface, any system based on verbal descriptions can only be used with reduced performance by non-native speakers of the chosen language.

To circumvent those issues, not relying on keywords is desirable. However, conveying semantics remains a necessity for ease of browsing, which raises the question of what alternate means, other than written language, can be used to that end.

We thus consider in this paper three means of displaying sounds without relying on any textual representation, all based on a spatial organization of sounds. The first one, considered as a reference baseline, does not rely on any semantic information and positions sounds according to their acoustical properties (time averaged spectral features). The second and third displays exhibit a spatial organization based on a predefined hierarchical semantic organization; those last two displays differs by how semantics are used.

Their effectiveness is studied through a search-based task whose aim is to find a given target sound by browsing the database using the display under evaluation.

The paper is organized as follows: in Section 1, previous work on the topic of sound database browsing is reviewed. Then, the corpus used in this study is described in Section 2. The three displays under evaluation are next described in Section 3, and the crowdsourcing test used to compare them is presented in Section 4. Finally, the outcomes of this experiment are discussed in Section **??**.

## 1 Previous work

Most of the research effort in the sound design and Music Information Retrieval (MIR) communities is focused on acoustical based indexing and browsing [20, 19, 7, 16]. Typically, the items (sound effects or pieces of music) are modeled by processing the digital audio waveform through some signal processing pipeline in order to get a compact description of each item [5] with sometime additional textual features. Then, statistical projections or embedding techniques, like Principal Components Analysis (PCA) [12], Multi Dimensional Scaling (MDS) [17, 4], Self Organizing Maps (SOM) [14, 15, 9] and the like are used to project the items in a two or three dimensional space while preserving as much as possible the distances among them. One of the advantages of such an approach is its ability to scale to very large databases [18] as it does not need any kind of manual annotation and allows the user to search by similarity efficiently according to acoustical properties.

However, acoustical models are inherently subject to observation noise and biases. Selecting the most relevant features to achieve the correct projection of the data may only be performed by an expert user, taking into account the specificities of the data. If done *a priori* by an expert or by some above cited dimensionality reduction technique, the induced bias can strongly limit the users' ability to access what they search for. In that respect, semantic tags, if available, have the advantage of implicitly structuring the similarity space, thus possibly easing the browsing process even if the actual tags are not – as in this study – explicitly exposed to the user.

## 2 Datasets

The datasets considered in this study are respectively composed of 149 urban environmental sound events and 137 musical sounds produced by musical instruments.

The former has a semantic organization as follows: a sound is characterized by a tag describing the physical source of the sound (*man-yelling*, *car-passing*). Sounds are then hierarchically grouped into classes according to their tags (*car > car-passing*; *car > car-starting*). Those classes are in turn packed into classes until high level classes describing broader concepts are reached (*traffic > car > car-passing*). The sound dataset is thus organized into a hierarchical structure of semantic classes as described in Figure 1. Strictly speaking, the sound
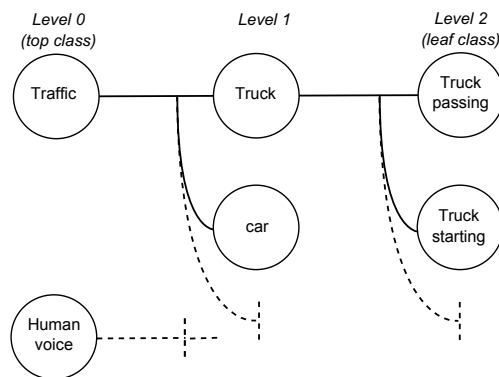


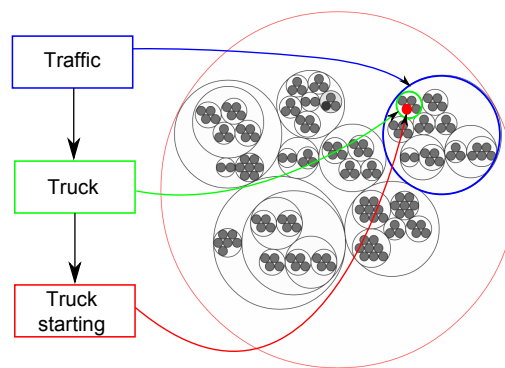Fig. 1. Semantic hierarchical structure of the dataset of urban environmental sounds.



Fig. 2. Semantic displays are based on the semantic hierarchical structure of the dataset.

samples of the dataset are the leaf semantic levels. All the other classes are represented by a *prototype sound* that best characterizes the sounds belonging to the class.

In order for the semantic hierarchical structure to be perceptually valid, the *tags* that describe the classes are chosen from sound categories presented in studies addressing environmental auditory scenes perception [13, 2, 6]. In cognitive psychology, sound categories may be regarded as intermediaries between collective sound representations and individual sensory experiences [6]. It is our belief that using such category names to build the hierarchical structure makes it perceptually motivated, and can thus help the users to efficiently browse the dataset.

Gregoire : Similarly, the latter also has a semantic organization: high and mid level classes categorize musical instruments according to the means by which sound is produced, while leaf level classes are related to specific instruments (*string > plucked-string > viola*; *free-aerophone > brassy > trumpet*)

## 3 Displays

The aim of the displays described in this section is to allow the user to efficiently browse the above presented dataset of sounds without any written textual help. In all displays, each sound is graphically represented by a filled
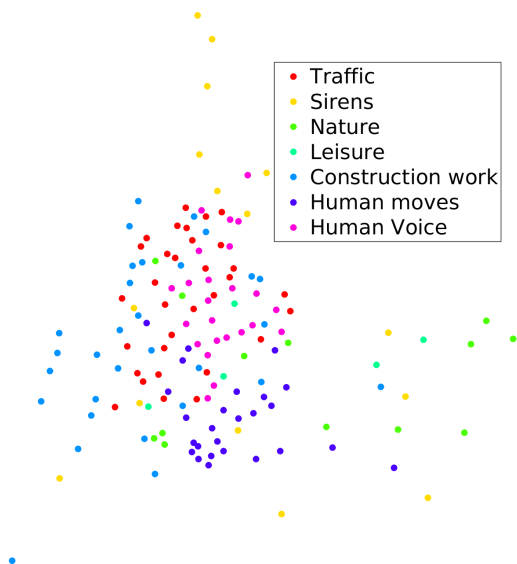
Fig. 3.   Acoustical Display for the urban sound dataset (AD). Colors and legend are removed during the experiment.



Fig. 4.   Acoustical Display for the musical sound dataset (AD). Colors and legend are removed during the experiment.

circle which, once clicked, plays the actual sound. The spatial organization of circles is specific to each display.

As a reference, an Acoustic Display (AD) provides a spatial representation based on acoustic descriptors. Each sound is described by time-averaged Mel-Frequency Cepstrum Coefficients (MFCCs) computed with standard parameter settings (13 lowest quefrency coefficients are kept out of 40). The Euclidean distance between time averaged features is then computed for each sound pair. A non metric multidimensional scaling (MDS) with Kruskal's normalized stress [11] is then employed to project the data into two dimensions according to this distance, see Figures 3 4. The latter exhibit a clear distinction between percussive (left) and sustained sounds (right).

Among the several alternative that exists, MFCCs features are considered as they are widely used in the audio processing community. The MDS is selected as it has lower sensitivity to outliers than the PCA and have the advantage of not relying on any parameter setting unlike the SOM approach.

Two semantically oriented displays are then proposed, called Full Semantic Display (FSD) and Progressive Semantic Display (PSD)Both display consider the hierarchical structure of the dataset to organize sounds. Each sound class is represented by a filled circle. Circles are then packed together according to the hierarchical semantic organization of the dataset, as shown in Figure 2. Thus, subclasses belonging to the same class are close to each others. Circle packing functions of the D3.js (Data-Driven Documents) javascript library [1] are used to distribute the sound classes in the space. Depending on the
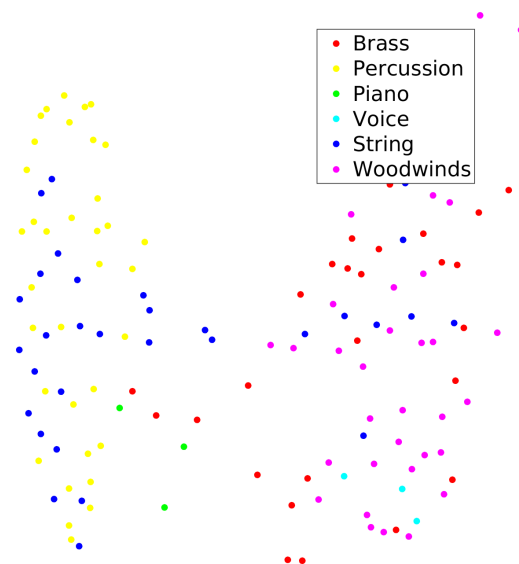
display, the user can either access each leaf class directly, or has to click through intermediate levels of the hierarchy.

More precisely, with the FSD display, users can directly visualize the whole hierarchy, down to the leaf classes, while with the PSD display, users only have access, at first, to the intermediate semantic levels of the hierarchy. Upon first using PSD, they observe circles representing the top classes of the semantic hierarchical structure of the dataset. When users click on a circle, they hear the sound prototype of the class and the subclasses are progressively revealed, represented by white circles contained in the circle that has just been clicked. The same action repeats itself until the user reaches the leaf classes of the hierarchy. The leaf classes are represented with small gray circles, indicating that there is no subclass to uncover. Thus the PSD has a constrained exploration system. Each time a sub-circle is automatically revealed, its sound prototype is played. Users may stop the discovery process by clicking on an other circle. In this display, the leaf classes are distributed in the same manner as FSD. Thus, the spatial configuration of the unfolded version of PSD, which may be obtained after discovering all the classes and subclasses, is equivalent to the one of FSD.

The latter interface is introduced in order to study if a progressive top-down display of the hierarchical structure helps the user explore and understand the dataset hierarchy.

## 4 Experiments

### 4.1 Objective

During this experiment, the three displays are compared using two corpora: the urban sound corpus and the

musicla sound corpus. By comparing AD and xSD (the two semantic displays), the goal is to evaluate the relative efficiencies of semantic based and acoustic based spatial configurations. By comparing FSD and PSD, we study the impact of an enforced hierarchical exploration of the corpus by the user. The goal is to check if constraining the user to first browse the high levels of the hierarchy helps him to grasp and memorize the spatial configuration and the organization of the sound classes.

## 4.2 Experimental protocol

We evaluate and compare those three displays with a crowdsourcing web-based experiment [1] for which we adopt a between subject approach, *i.e.* a subject can only test one display. In this experiment, subjects are asked to successively retrieve 13 target sounds among the whole dataset. The target sounds are selected such that there are at least two target sounds in each top-level class of the semantic hierarchical structure of the dataset. To reduce order effects, target sounds are presented to each subject in random order.

First, the subject clicks a *Play target sound* button to listen to the target sound. They may then replay the target sound as many times as they like, even while browsing. A timer starts when the subject clicks on a circle for the first time. When the target sound is found, the subject puts an end to the search by clicking on the *Click if found* button. This action 1) stops the timer and 2) loads a new target sound. If the subject designates and incorrect sound, an error message appears, and the experiment goes on with the same target sound.

During the experiment, two visual cues are shown to the subject:

- *The search state*: which can be "pause" if the subject is not currently looking for a target sound (at the beginning of the experiment, or between two target sounds) or "in progress" if they are.
- *Remaining target sounds*: the number of target sounds that remain to be found.

The experiment ends when all the target sounds have been found. One shall notice that the PSD display is not reinitialized at each change of the target sound. Thus, when a circle is unfolded, it remains visible during the whole experiment.

## 4.3 Data Collection

Four types of data are collected during the experiment:

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[1]The tests are available at `http://soundthings.org/research/speedSoundFinding/index.html` and `http://soundthings.org/research/speedSoundFindingMusic/index.html`, each of the 3 displays is also available by appending a digit from 1 to 3 to *index* in the above cited url.

- the total duration of the entire experiment. It includes breaks between two target sound searches and it is called the *absolute duration*.
- the duration of each search. The sum of the 13 search durations, which is the *absolute duration* minus the break times between two target sound searches, is called the *total search duration*.
- the name of each sound which has been heard.
- the time at which each sound has been heard.

## 4.4 Apparatus

A crowdsourcing approach is adopted. The experiment is designed to be performed using the *Chrome* web browser. The link to the experiment has been sent to three mailing lists, namely *Music-Ir, Auditory* and an internal IRCAM mailing list. Subjects are only allowed to perform the experiment once, and on one interface only. Data were automatically collected server-side at the end of the experiments. Subjects were asked to use headphones. All sounds of the dataset are normalized to the same root mean square (RMS) level.

## 4.5 Participants

60 subjects have completed the experiment for the urban sound dataset, 20 for each interface. Gregoire : 43 subjects have completed the experiment for the urban sound dataset: 15 for the PSD interface, 13 for the FSD interface and 15 for the AD interface.

## 4.6 Outliers Detection

Outlier detection is an important step of any crowdsourcing experiment as experimenters do not control the environment in which the subjects perform the experiment [10, 3]. A commonly used method to detect outliers in human-computer interaction studies is to consider as outlier an observation which deviates of at least $\pm 2$ standard deviation from the average [10]. However, as this latter method is not robust to the presence of isolated extreme observations (as it is often the case for crowdsourcing experiment), we follow the method proposed by [10] and use the Inter-Quartile Range ($IQR$). With this approach, an observation is considered to be an outlier if its value is more than $3 * IQR$ higher than the third quartile or more than $3 * IQR$ less than the first quartile. This methods is applied in this study to the four following criteria :

1. the *absolute duration*: to detect subjects who took abnormally long breaks between searches.
2. the *total search duration*: to detect subjects who spent an abnormally long time to complete the experiment.
3. number of heard sounds: to detect subjects who had to hear an abnormally high number of sounds to complete the experiment.
4. the maximum number of times a target sound had to be heard before being found: to detect subjects who had difficulty to recognize the target sounds.

## 4.7 Statistical analysis

Gregoire : We use one-way ANOVA to test for statistical significance. Post hoc analyses are done using two samples *t* tests with Bonferroni correction. All statistical analyses are performed at the 5% significance level using native functions of the MATLAB computing environment.

## 5 Results on Urban Sounds

### 5.1 Outliers

Using the *IQR* method, 5 subjects are detected as outliers for the urban experiment, and removed from the analysis. 2 of those subjects used PSD and 3 subjects AD. 3 subjects are detected by considering the *absolute duration*. They spent respectively 50 minutes, 5 hours and 17.5 hours to complete the experiment. 1 subject is detected by considering the *total search duration* (46 minutes), and an other by observing the *total number of heard sounds* (1800 heard sounds, roughly 12 times the total size of the corpus). And lastly 2 subjects are detected by observing the maximum number of times they had to hear a target sound before finding it (4 and 8 times).

### 5.2 Efficiency

To study the relative efficiency of the three displays, three metrics are considered:

- the *total search duration*
- the number of heard sounds
- the number of unique sounds heard. By "unique" we mean that, if a same sound is heard 10 times during the 13 searches of the experiment, it counts only for one.

The first two metrics quantify the notion of efficiency by considering the time and the number of clicks needed to achieve the task, *ie.* reach the target. The goal for those values is to be as low as possible. The third data allows us to measure the selectivity of the interfaces. A low number of unique heard sounds indicates that subjects understood the spatial organisation of the dataset, and used this knowledge to improve their searches. On the contrary, a high number of heard sounds without duplication suggests that the subject did not understand the way sounds are organized in space, and tended to play all the sounds at each search. The maximum number of unique heard sounds is the corpus size: 149 sounds.

The averages and standard deviations for the three metrics are presented in Figure 5. The type of display has a significant effect on the the *total search duration* ($F[2,52] = 3.63; p < 0.05$), the number of heard sounds ($F[2,52] = 5.65; p < 0.01$) and the number of unique heard sounds ($F[2,52] = 8.65; p < 0.01$).

Post hoc analysis on the total search time (Figure 5.a) shows no significant differences between the three interfaces (PSD-FSD: $p = 0.09$; PSD-AD: $p = 1$; FSD-AD: $p = 0.07$), although the FSD interface achieves the best results.
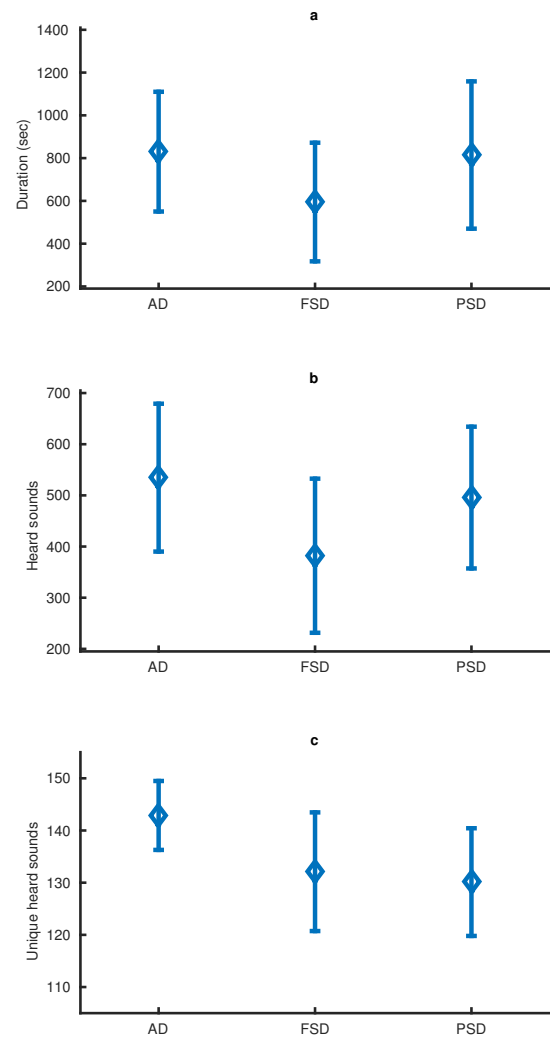


Fig. 5. Average and standard deviation for (a) the total search durations, (b) the number of heard sounds and (c) the number of unique heard sounds.

Concerning the numbers of heard sounds (Figure 5.b), FSD significantly outperforms the AD interface (FSD-AD: $p < 0.01$), whereas PSD and FSD as well as PSD and AD show similar outcomes (PSD-FSD: $p = 0.06$; PSD-AD: $p = 1$).

Figure 5.c shows the results for the number of unique heard sounds. This time the results of AD are significantly lower than those of both PSD and FSD (PSD-AD: $p < 0.01$; FSD-AD: $p < 0.01$). For AD, 75% of the subjects heard more than 140 sounds, and 25% heard at least 148 sounds, that is almost the entire database. Considering PSD, 75% of subjects heard less than 134 sounds, versus 143 for FSD. This time PSD and FSD perform equivalently (PD-FD: $p = 1$).

According to those results, a spatial display following a hierarchical organization of the dataset based on semantic values (PSD and FSD) is more efficient than an organization based on acoustic descriptors (AD) as it allows the users to quickly apprehend the dataset range and thus to specify their search. Furthermore the FSD interface allows the users to retrieve the 13 target sounds
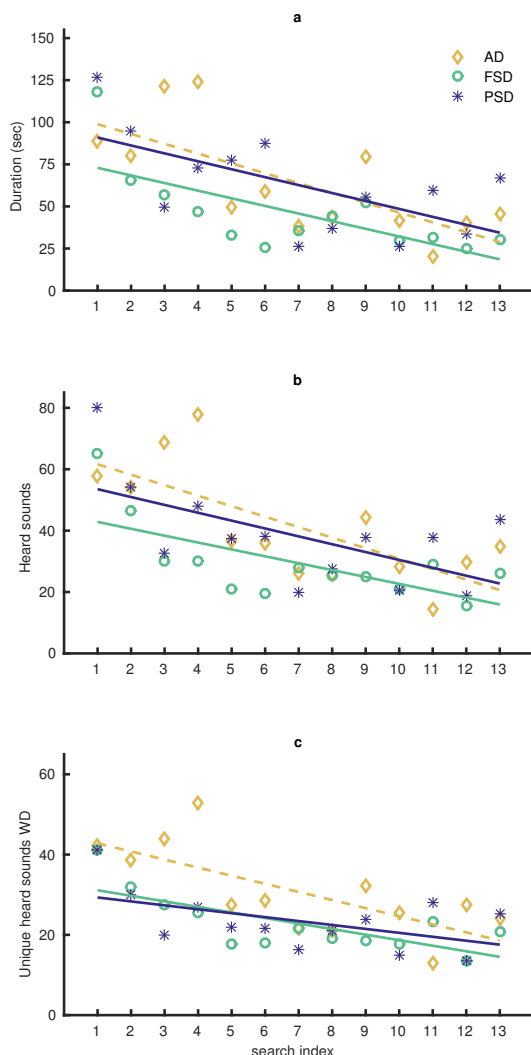
Fig. 6. evolution of (a) the average search durations, (b) the average numbers of heard sounds, and (c) the average numbers of unique heard sounds at each target sound search. Lines are linear regression fits.

Fig. 7. Average and standard deviation for (a) the total search durations, (b) the number of heard sounds and (c) the number of unique heard sounds.

by listening to a smaller amount of sounds, than the AD. However, those two effects are significantly compromised when users have to parse the entire hierarchy to reach the first target sound, as in the PSD display. It seems that enforcing an explicit exploration of the hierarchy disturbs or confuses the user instead of allowing him to learn the semantically motivated spatial organization of the classes.

## 5.3 Learning

We now study if and how users progressively acquire knowledge about the spatial organization of the classes. To do that, the evolution of the above described metrics with respect to the search indexes is considered. Three sets of collected data are used:

- the duration of each target sound search,
- the number of heard sounds for each target sound search,
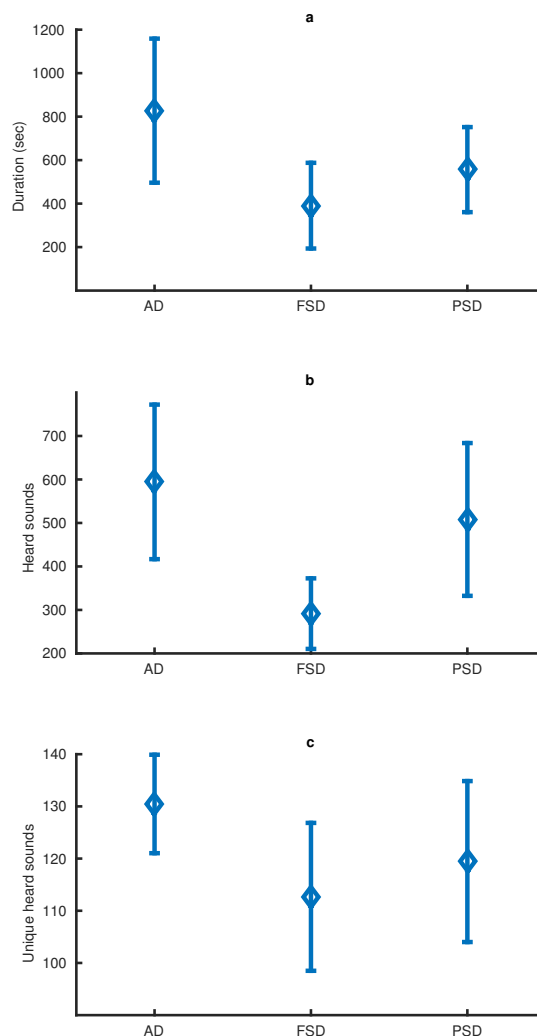- the number of unique heard sounds for each target sound search.

Figure 6.a shows the evolution of the average duration of each target sound search observed for AD, FSD and PSD. As shown by a linear regression of the data, there is an overall increase of efficiency with respect to the index of the search. For the duration and the number of heard sounds (Figure 6.b), the AD and PSD perform equivalently, whereas the FSD exhibit better performance. In the case of the number of unique heard sounds (Figure 6.c), the two semantic displays are equivalent. This latter result suggest that for both semantic displays, users have parsed the same range of the dataset, although users of FSD have done so by listening to fewer sounds and more quickly.

## 6 Results on Musical Sounds
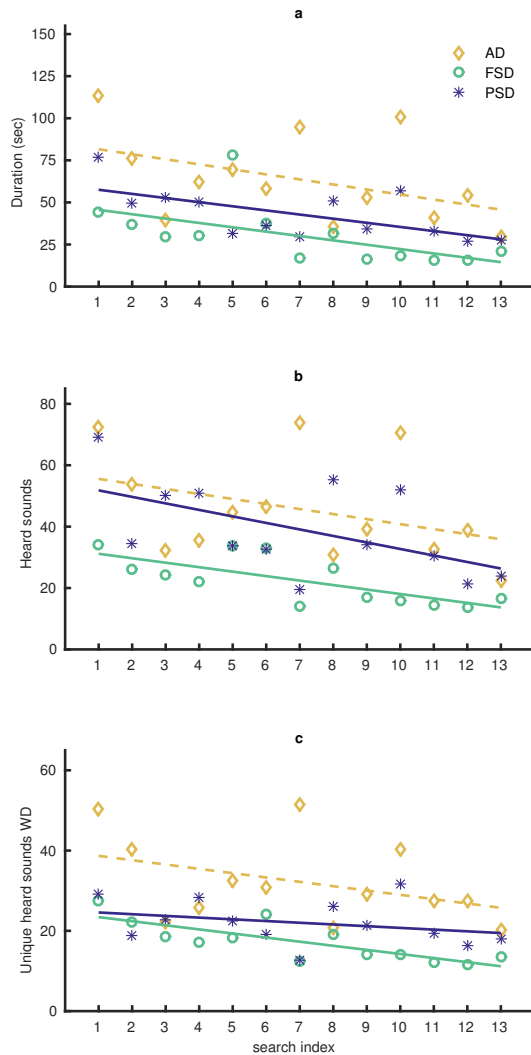
Gregoire : merge fig. 5 and 7

Fig. 8. evolution of (a) the average search durations, (b) the average numbers of heard sounds, and (c) the average numbers of unique heard sounds at each target sound search. Lines are linear regression fits.

## 6.1 Outliers

Gregoire : 2 subjects are detected as outliers for the music experiment and removed from the analysis. 1 subject used PSD and 1 subject used FSD. The first outlier is detected by considering the *absolute duration* (5 hours to complete the experiment) and the second by looking at the *absolute duration* (1 hour and 30 minutes) as well as the *total search duration* (1 hour and 15 minutes) and the total *number of heard sounds* (901).

## 6.2 Efficiency

Gregoire : The averages and standard deviations for the three metrics are presented in Figure 7. The type of display has a significant effect on the the *total search duration* ($F[2,38] = 10.25; p < 0.01$), the number of heard sounds ($F[2,38] = 13.15; p < 0.01$) and the number of unique heard sounds ($F[2,38] = 6.39; p < 0.01$).

Gregoire : Considering the *total search duration* (Figure 7.a), post hoc analyses shows that PSD and FSD

perform better than the AD interfaces (PSD-AD: $p < 0.05$; FSD-AD: $p < 0.01$;). No significant difference is found between PSD and FSD (PSD-FSD: $p = 0.32$).

Gregoire : Figure 7.b shows the results for the numbers of heard sounds. This time AD and PSD exhibit similar results (PSD-AD: $p < 0.43$), while being both outperformed by FSD (FSD-PSD: $p < 0.01$; FSD-AD: $p < 0.01$).

Gregoire : Considering the numbers of unique heard sounds (Figure 7.c), FSD significantly outperforms the AD interface (FSD-AD: $p < 0.01$), while both PSD *vs.* FSD and PSD *vs.* AD show similar outcomes (PSD-FSD: $p = 0.6$; PSD-AD: $p = 0.9$).

Gregoire : This results confirms the trends observed for the urban sounds (see Section 5):

- A semantic organization of the dataset significantly reduces the number of heard sounds needed to find a target sound.
- Forcing the users to parse the entire hierarchical organization of the dataset seems to diminish the interface efficiency by increasing

Gregoire : It has to be noted that this time a semantic organization of the dataset (PSD and FSD) significantly reduces the search durations compared to an acoustic organization (AD).

## 6.3 Learning

Gregoire : Figure 8 shows the evolution of the metrics computed at each target sound. Results for the music dataset are similar to those observed for the urban dataset. An overall increase of efficiency with respect to the index of the search is observed for the tree interfaces. For the duration and the number of heard sounds (Figure 8.a and .b) the FSD interface exhibits better performance. Considering the number of unique heard sounds (Figure 8.c), the learning of the dataset range seems to be quicker for the FSD interface than for the PSD interface, thus confirming that forcing the users to parse the hierarchy disturbs their researches.

## 7 Conclusion

In this paper, two displays allowing users to explore a semantically organized sound dataset without written textual help are presented. The interfaces distribute sounds represented by circles on a 2D space, following an underlying hierarchical semantic organization. Those two semantic displays are assessed and compared to a third display in which the spatial configuration depends upon acoustic features. The tests consist in data retrieval tasks. The Full Semantic Display (FSD), that allows users to directly visualize the leaf classes of the semantic hierarchical structure, proves to be the most effective interface for the task.

Two main conclusions may be derived from this experiment. First, a spatial configuration based on

semantic features is more effective to retrieve target sounds than a spatial configuration based on acoustic features. Second, imposing the exploration of the semantic hierarchical structure by constraining its visibility does not help the user understand and learn the spatial configuration of the semantic class, but instead disturbs the navigation.

In the dataset considered in this study, there are as many leaf classes as sounds in the dataset, which would be unrealistic for large datasets, but the organization can easily be adapted by considering the leaf classes as *collections* of semantically similar sound samples. Thus, two sounds of *male-yelling* would be grouped into a single leaf class with the tag *male-yelling*. The leaf class would then also have a *prototype sound* being the most representative item of the different *male-yelling* sounds belonging to the leaf class.

## 8 Acknowledgements

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**THE AUTHORS**