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From interactive to adaptive mood-based music listening experiences in social or personal contexts

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Consumer cultures are increasingly shifting to cultures of participation supported by technology such as social computing. In the domain of interactive audio, listeners’ roles are revisited giving more importance to desire, context and the sense of control. We present new developments in our mood-driven music player Moodplay allowing both social (virtual jukebox) and personal usage. Moodplay relies on semantic computing technologies for musical mood using social tags and informative and aesthetic browsing visualisations. The prototype runs with a dataset of over 10,000 songs covering various genres and arousal and valence levels. Changes in the design of the system were made in response to a user evaluation including over 120 participants from 15 different sectors of work or education. The proposed client/server architecture integrates modular components powered by semantic web technologies and audio content feature extraction. This enables a shift from interactivity to adaptivity where recorded music content can be controlled in flexible and non linear ways.

1 Introduction

This work is concerned with the development of novel listening experiences of recorded music and is part of our larger project, Fusing Audio and Semantic Technologies for Intelligent Music Production and Consumption\(^1\). The music recording industry created the concept of fixation, to place a song into a tangible medium such as the Compact Disc [34]. Although this revolutionised accessibility to music, it positioned recorded music as a finished goods meant to be consumed passively from the creative perspective, which consequently settled a clear distinction between producers and consumers [7]. Rooted in such tradition, the first generation of digital music broadcast services left little room for interactions from end-users, i.e. listeners. We are nowadays experiencing a shift from consumer cultures to cultures of participation in different domains, from end-user software development [34] to music entertainment, with the introduction of multimedia systems responding to the desires and needs of the receivers and new interaction paradigms revisiting the roles and activities of receivers [13, 17, 39]. With regard to music listening such shift manifests itself by the development of adaptive music systems [8] defined here as systems enabling to select or alter music content according to user-related factors. Music recommender systems [33], which can be seen as a specific case of adaptive music systems, aim to present listeners with tracks matching their tastes and activities. Several methodologies for music recommendation have been proposed over time which may be categorised as follows, according to the sources of information they rely on: (i) collaborative filtering (relies on inter-user similitude of interests, see e.g. Last.fm\(^2\)), (ii) content-based (relies on inter-tracks similitudes, e.g. derived from audio [12]), (iii) user-based (relies on user personal factors, such as felt mood [15] or perceived mood [4]), (iv) context-aware (relies on the user situation, such as activity [11], location [18], motion [24], time [3]), and (v) a multimodal combination of information (see e.g. [26, 15, 38]).

In this paper we present advances made to our mood-driven and semantic web-based music player suited to both social, Moodplay [4], and personal, myMoodplay [2], usages. Amongst factors driving music recommendation, mood\(^3\) has received a growing interest in the past decade

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DOI: http://dx.doi.org/10.17743/jaes.2016.0042
\(^1\)http://www.semanticaudio.ac.uk/
\(^2\)http://www.last.fm
\(^3\)We will employ the words emotion and mood interchangeably as their distinctions are out of the scope of this article. We
Music psychology studies highlighted that music can actively be used for mood regulation [1] which provides a foundation to explain why music consumers express strong interest in being able to search and browse music by mood [20]. The communication of emotional meaning through music (perceived mood) differs from mood induction (felt mood), a difference which is often disregarded in music information retrieval research. A recommender system performing a one-to-one mapping between a user’s felt mood and the mood conveyed by a song (e.g. play happy tracks when one feels happy) would discard the mood regulation function of music (e.g. play party tracks to shift from calm to excited states). To address this issue, our system confers users the possibility to choose music based on a desired mood, which can or not correspond to their felt mood, depending on cases. We follow an iterative user-centred design approach [22] where the design is driven and refined by user-centred evaluation. The results from the survey conducted in [4] highlighted areas of improvements and new applications. We focus here on improving aspects such as personalisation, identification and mixing and present a new software architecture informed by user needs and expectations.

2 Related work

The design of the music player Moodplay (MP) was inspired by our participatory performance system Mood Conductor (MC) [13] which allows audiences to conduct performers in live improvisations using mood directions. In both MC and MP, the user-facing mood voting interface (see Figure 1) combines categorical and dimensional models of emotions by using a continuous two-dimensional space inspired by Russell’s circumplex model of affect

refer the reader to [23] for a discussion on the differences between emotion (“temporary and evanescent”) and mood (“relatively permanent and stable”).

![Figure 1. Example of user-defined mood trajectory in the myMoodplay player. Vertical and horizontal dimensions are arousal and valence, respectively.](image)

[27] (proved relevant for music by Thayer [36]) together with discrete emotion categories represented by mood tags (such as angry, haunting, sweet and fun). The two dimensions are arousal (A) or energy (how exciting/calming musical pieces are) and valence (V) or stress (how positive/negative musical pieces are). A color gradient is also used in the voting space according to color-emotion associations obtained from the [http://wefeelfine.org/](http://wefeelfine.org/) platform. We aimed to provide users with a self-explicit interface helping them to find desired moods easily which proved to be successful [21]. The first Moodplay prototype [4] can be seen as a virtual jukebox and targeted social music listening experiences where people collaboratively vote for the music to be played (see Figure 2). Visualisations and lighting effects provide informative and aesthetic feedback about individual mood votes and the mood selected by the system according to time and vote frequency considerations [13]. Time-varying individual mood vote representations are likewise a dynamic bubble chart where areas reflect vote frequency, whilst a moving trace similar to Dixon’s performance worm [10] is used to express current and past “elected” moods in the AV space. MC and MP are the first systems of their kind to the best of our knowledge.

The social jukebox presented in [25] presents similarities to Moodplay in terms of control (voting through a smartphone-friendly app) but music is selected in a more traditional way based on bibliographic (artist, title, etc.) rather than creative (mood) metadata. Following user evaluation (Section 4), we developed another prototype, myMoodplay (mMP), destined to be used on a personal level in a web browser environment [2]. As described in Section 3, MP and mMP rely on data-driven semantic mood space inferred from crowd sourced tag-to-song associations and onto which songs from a music collection are projected [28]. A similar approach is followed in Mood Cloud 2.0 [19] but the technique to obtain the semantic mood space is different; Moodplay uses dimensionality reduction techniques together with the affective circumplex transform (ACT) [28] to conform the semantic mood space to the AV space, whilst Mood Cloud 2.0 uses self-organising maps (SOMs). [37] proposed an audio content-based approach combined with two-dimensional visualisations of probabilistic mappings of features such as mood and genre. Although probabilistic mappings offer a promising way to counteract uncertainties of audio-based predictions, the generated browsing space required a learning curve for users as its dimensions are not explicit. [15] introduced an hybrid content- and context-based recommender system which supports reasoning about users’ desired emotion using a domain-specific semantic web ontology. Users’ desired emotions are inferred from context and user factors such as situation and taste, however user-based training of an emotion state transition model must be conducted for the system to operate. Some commercial proprietary services⁴ present similarities to our system but

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their methods are not disclosed. Music visualisation interfaces relying on features other than mood have also been proposed. For instance, [40] applies support vector regression to map high dimensional audio feature vectors to a two-dimensional tempo/timbre space.

![Image](image_url)

**Fig. 2.** Example of use of the Moodplay prototype in a social context (Digital Shoreditch 2015).

### 3 Semantic Computing of Musical Mood

Creating mood-based listening experiences requires the association of audio recordings with perceived moods. Various approaches are possible, ranging from using purposefully composed “mood” music also known as production music [28] and expert annotations of back catalogues, to automatic identification of mood by machine relying on acoustic features of recordings (see [6] for a review). While recent advances in content-based audio analysis allow for recognising the mood automatically, audio-based mood detection is challenging and far from being a solved problem [30]. We therefore developed methods that build mood models using social information.

The objective for this study was the construction of a well-balanced music dataset with good coverage of different genres and moods. To this end, we relied on the diverse music catalogue of I Like Music\(^5\) (ILM), a library consisting of over 3 million songs at the time of analysis. In order to obtain mood related tags, we first sampled over 1m songs from Last.fm using the approach introduced in [29]. Form the initial corpus, we found 218,032 tracks in the ILM database that matched one of the Last.fm tracks using string matching between artist names, song titles and track durations with penalties for spelling mistakes and other metadata discrepancies. We applied a two stage sampling method to fulfil several potentially conflicting criteria. We aimed to create a dataset that represents all broad genre categories following expert classification available from ILM as well as crowd sourced genre tags.

We first compute normalised Term Frequency-Inverse Document Frequency (TF-IDF) scores form the Last.fm tags, then create a three dimensional mood space by using non-metric Multi-Dimensional Scaling (MDS). To obtain a good coverage of this mood space, a Gaussian Mixture Model (GMM) with 5 components is fitted on the data. We then estimate the log-likelihood for each track under the fitted model. This likelihood is then used to sort songs in the respective genre and artist buckets before sampling. During iterative sampling, we first choose a genre category using a pseudo random process that favours genres with fewer artists associated with them. We then randomly pick an artist from this genre bucket and choose the least likely song from the selected artist under the GMM model fitted on the MDS space. The song is then removed from the pool. We also limit the number of tracks picked from any single artist or genre. This process maximises the number of unique artists in the collection, while ensures a balanced coverage of genres and moods.

The database, denoted ILM10K in the following, is then labelled with AV coordinates using the Affective Circumplex Transform (ACT) as in [31]. This transform uses MDS coordinates but also conforms the mood space to a reference configuration of terms. Such configuration can be extracted from Russell’s and Scherer’s studies [27, 32]. This is done via Procrustes transformation [14] which performs a linear mapping from one space to another while retaining the relative distances between objects in the original space.

The track metadata together with arousal-valence coordinates for tracks in the ILM10k dataset produced in the ACT experiments are stored in a triplestore and structured according to a light-weight OWL ontology. The ontology uses concepts from the Music Ontology\(^6\) for music production metadata and links these with tagging data structures described in the Modular Unified Tagging Ontology (MUTO)\(^7\). Each audio track entity in the triplestore is represented as a Track class from the Music Ontology and linked to a set of Last.FM tags by adding tagging properties to this class.

### 4 User-Centred Design

Our Mood Conductor and Moodplay systems have evolved and continue to evolve following a circular cycle involving prototyping, user evaluation, and inferred user and technical requirements. The Mood Conductor system is the result of a participatory design process involving performers with the objective of enabling creative interactions with audiences. Moodplay inherited from lessons learned with Mood Conductor [21], however its context of application is different as it deals with recorded music content rather than live performances. In [4] we conducted a user evaluation of the social Moodplay prototype with over 1200 participants from 15 different sectors of work or education across two public exhibitions held in London in May 2015. Photos and a short video showing interactions are provided at: [http://bit.ly/mp_photos](http://bit.ly/mp_photos) and [http://bit.ly/mp_video](http://bit.ly/mp_video). As in [25] we found that

\(^5\)http://www.illikemusic.com

\(^6\)http://musicontology.com/

\(^7\)http://muto.socialtagging.org/core/v1.html
the process of voting for music collectively created a rich source of interaction between participants as exemplified by the creation of different groups of users sharing the experience together, the generation of discussions within these groups (e.g. on musical tastes and music technology), playful competition and fun. About two-thirds of the participants found their experience with the system very satisfying. The most striking result was the wide range of applications suggested by the participants (see Figure 4) including over 100 different contexts of use (from e.g. party and home usage to marketing and therapy), some of which we had not envisioned. Amongst the different areas of improvement identified in [4] we focus in this paper on the following ones: (i) personalisation (e.g. personal usage with own local or cloud-based music collection, customisation through user preference and history, etc.), (ii) identification (users, tracks), and (iii) mixing (e.g. smoother and slower transitions, feature-based mixing). The next section describes the technical changes we implemented to provide improvements in these three areas. To be able to support different range of social or personal applications we opted for a decentralised model enabling a service-oriented architecture providing increased flexibility and accessibility. We also built on semantic web technologies following a linked data approach for content identification. In order to improve mixing aspects and produce seamless transitions between tracks as described in [9], we integrated automated sequencing technologies exploiting audio content-based features with AV mood coordinates, tag statistics and a set of identifiers for each track. The original software architecture includes user, audio, visualisation and lighting clients, and the server application used to communicate between the individual client components and process user interaction data (Figure 5). As an installation, the original system requires a location-specific setup, including a PA system, projector and lighting effects controller for creating the user experience. The new architecture (Figure 6) adds a JSON database storing content-based audio features to the data storage layer to improve mixing between tracks (see Section 5.2). The application logic is implemented in several independent modular units that can be located on one or multiple servers. The Web service API receives requests from client-facing components and information processing agents and connects to the data stores. The enhanced functionality to audio playback has introduced two more application logic software components, one to process content-based audio feature data and the other to efficiently stream audio fragments on-demand according to user interaction.

5 Architecture

The components from the Moodplay system architecture can be divided into a three-tiered structure by functionality: user-facing components, web application logic, and data storage. The original [4] and proposed architectures share the underlying data storage components including the ILM10K audio dataset and a separate triplestore with AV mood coordinates, tag statistics and a set of identifiers for each track. The original software architecture includes user, audio, visualisation and lighting clients, and the server application used to communicate between the individual client components and process user interaction data (Figure 5). As an installation, the original system requires a location-specific setup, including a PA system, projector and lighting effects controller for creating the user experience. The new architecture (Figure 6) adds a JSON database storing content-based audio features to the data storage layer to improve mixing between tracks (see Section 5.2). The application logic is implemented in several independent modular units that can be located on one or multiple servers. The Web service API receives requests from client-facing components and information processing agents and connects to the data stores. The enhanced functionality to audio playback has introduced two more application logic software components, one to process content-based audio feature data and the other to efficiently stream audio fragments on-demand according to user interaction.

5.1 Personalisation and identification

The user interface (UI) design and implementation depends on the particular context for which the architecture is deployed. In the original system the user client runs a Web-based application written in JavaScript supported by any modern Web browser. This design allows Moodplay to be used in a platform-independent manner and the interface serves solely as a control device for users to interact. This is in contrast to the decentralised architecture implemented for the web client version myMoodplay, in which different components directly involved in creating the user experience, are required to be incorporated into a common client application. In this case the user interface is presented in the same combined space as the visualisation functionality. This is achieved by layering an interactive surface on top of animated and background visualisations displaying user selections and playback indicators, to bet-
Fig. 4. Top 100 suggested applications for Moodplay [4].

Fig. 5. The architecture of the original Moodplay system.

Fig. 6. New architecture for the personal browser-based myMoodplay system. Dymo: dynamic music objects [35].

Fig. 7. myMoodplay browser-based UI with mood space visualisation.

5.2 Adaptive audio playback

The audio playback engine used in the original Moodplay system, implemented in SuperCollider\(^8\), simply performs crossfades between two successive tracks [4]. To offer advanced audio mixing capabilities we developed a new engine based on Semantic Web technologies and audio features. The engine uses the Dynamic Music Objects (dymos) linked data music format defined by the Mobile Audio Ontology [35]. The format consists of an abstract representation of the musical structure and its semantics\(^9\) which can be annotated with any type of audio features. Dymos can be queried and navigated in various ways and can include playback instructions that define how various controls and features map to predefined and custom musical parameters. The domain of such mappings can cover algorithmic controls, mobile sensor data, user interface controls, contextual information, web-sourced musical metadata, and analytical information extracted from the music itself. Dymos

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\(^8\)http://supercollider.github.io/

\(^9\)Based on the Common Hierarchical Abstract Representation for Music (CHARM) [16].
can be loaded and played back in any web environment simply by embedding the dedicated JavaScript library using the Bower package management system\(^\text{10}\). This library ensures that any necessary sensors or control units are allocated and it includes an audio scheduler that works for a great variety of use cases.

---

```
PREFIX mo: <http://purl.org/ontology/mo/>  
PREFIX mood: <http://isophonics.net/content/mood/>  

SELECT ?valence ?arousal ?mbid ?path  
WHERE  
  {  
    ?coords mood:valence ?valence ;  
    mood:arousal ?arousal ;  
    ?lfmid mood:coordinates ?coords ;  
    mo:available_as ?fileid ;  
    mo:musicbrainz_guid ?mbid .  
  }  
ORDER BY ?valence  
LIMIT 5
```

---

Listing 1. SPARQL query to retrieve the nearest 5 track identifiers to user specified x-y coordinates (query parameters @x and @y are substituted by floating point values when query is executed).

The audio client in the original Moodplay system connects to a SPARQL endpoint during system initialisation and retrieves mood coordinate data for all tracks in the dataset. However, this solution is not feasible for the on-demand data access model implemented in the browser-based version of the system. Instead, the nearest neighbor search is performed by sending a query to the SPARQL endpoint (example shown in Listing 1) that calculates the Manhattan distance between user selections and track coordinates in the mood space. The query determines a fixed number of nearest tracks along with their features from which the server generates dymos. These dymos are sent back to the playback engine which queries them locally to find which track will best follow the current track according to configurable transition rules (e.g. similar key, harmony, meter, tempo, etc). On the server, the audio feature data (extracted using the Vamp plugins framework with Sonic Annotator\(^\text{11}\)) is structured according to the Audio Feature Ontology\(^\text{12}\) and capture acoustic information that relate to perceptual attributes (e.g. rhythm, tonality, timbre, structural segmentation), often at a high temporal resolution. In dymos the features are summarised depending on the custom needs of the playback engine based on statistical methods. The use of Semantic Web technologies for the audio feature store and the dymos ensures that musical concepts and their relationships are properly defined throughout the process. At all times the playback engine is aware of the musical objects it deals with and it can rely on advanced querying methods of the Semantic Web to find optimal transition candidates and locations.

---

```
{  
  "@context": "http://tiny.cc/dymo-context",  
  "@id": "mix",  
  "@type": "Dymo",  
  "ct": "conjunction",  
  "parts": ["track1.json", "track2.json"],  
  "mappings": [{  
    "domainDims": [  
      {  
        "name": "Tempo", "@type": "Parameter",  
        "name": "duration", "@type": "Feature"},  
        "function": "[a, b] -> r(a/60*b);"},  
      "dymos": "[a, b] -> r(a/60*b);"},  
      "body": "return a[0]+a[1]-b[0]-b[1];"},  
      "dymos": "[mix]",  
      "range": "Tempo"},  
    "domainDims": [{  
      "name": "Fade", "@type": "Parameter",  
      "name": "afv:Tempo", "@type": "Feature"},  
        "function": "[a, b] -> r(a/60*b);"},  
      "body": "return b[0]+a*(b[1]-b[0]);"},  
      "dymos": "[mix]",  
      "range": "Tempo"},  
    "..."}
```

---

Listing 2. The main mix dymo used in myMoodplay.

Transitions between successive tracks can be achieved by interpolating in various musical dimensions. For the myMoodplay system we define a main mix dymo as a conjunction of two tracks that can be played back simultaneously (Listing 2). The dymo tracks are modeled as three-level hierarchical structures consisting of the main audio track, a bar-level and a beat-level. To create the transitions, we add custom higher-level parameters and define how they map to lower-level ones. For instance, the first mapping in Listing 2 defines the relationship between our custom parameter Tempo and the standard parameter TimeStretchRatio of all beat-level objects, also considering their individual duration feature. This parameter not only adjusts the overall tempo of the songs but also evens out the duration of the beats in order to ensure that the beats can be matched during playback. The second mapping defines how the Fade parameter controls an interpolation between the tempos of the two tracks (via the Tempo parameter just defined). In a similar fashion, we then define an interpolation between the tracks’ amplitudes also using low-pass filters. The interpolation mechanism can be controlled by mapping a Ramp control with a given duration to the Fade parameter. Whenever a new track is chosen by the system we replace the inaudible track of the mix dymo with the new one and trigger the Ramp.

The new adaptive playback engine also makes use of a specialised audio server that minimizes network traffic between components and responds more flexibly to user interactions. A dedicated service sends audio chunks rather than entire files upon request, based on the segmentation structure represented in the currently playing dymos. This is optimal for Moodplay due to the unpredictability and potential high frequency of user choices. The audio chunk server also paves the way for future developments by en-

\(^{10}\)http://bower.io/search/?q=dymo-core  
\(^{11}\)http://vamp-plugins.org/sonic-annotator/  
\(^{12}\)https://w3id.org/afo/onto/1.1#
5.3 Discussion

The modular flexible architecture of Moodplay has enabled experimenting with different solutions to user interaction. The two-dimensional interactive control interface can be supplemented or replaced by other means of mood estimation, including extracting valence and arousal features from biosignals using, for example, an electroencephalogram (EEG) to measure brain activity or electromyogram (EMG) for detecting muscle activity. An experimental interface for Moodplay using biosignals was designed at the Sonar Music Hack Day 2015, where brain, muscle and heart activity was converted to valence and arousal coordinates with the Neuroelctrics Enobio13 and BITalino14 biosignals systems and sent to the Moodplay server. Through mapping, the participants’ votes were registered and the mood trajectory calculated by the existing mechanism despite the nature of the sources and data input rates being significantly different from what the original touch surface interface affords. The new system can be used in various ways fulfilling different use cases. For instance, we can build a personalised browser-based version (myMoodplay) where a single user selects a mood or defines a mood trajectory and the music is played back locally in the client application (Figure 6). We can also build a web-based Moodplay that works in the same way as the original system, with multiple users collaboratively selecting moods and a centralized server generating the music which the clients then stream (in this case, the audio component shown as part of the client in Figure 6 is externalised to its own server).

6 Conclusions and future work

We presented a novel client/server architecture for our mood-based music player Moodplay which integrates modular components powered by semantic web technologies and audio content feature extraction. The proposed framework supports social (virtual jukebox) as well as personal applications and enables an adaptive control of music tracks responding to user-based (e.g. mood trajectory) and/or contextual factors (e.g. based on mobile phone sensor data). We plan to develop dynamic user personalisation techniques (e.g. using like/dislike) and explore other types of browsing interface using multimodal features (e.g. tempo and mood). With regard to audio playback, we plan to define different mixing instructions where for instance harmony or timbre are taken into account for even smoother transitions. Dynamic music objects could be used to create mashups on the fly of two or more simultaneous songs to be able to follow votes for multiple moods. We are also considering non-linear audio techniques which could transform the player into a creative tool, for instance by reorganising, compressing or expanding temporally pre-recorded content.

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8 REFERENCES


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