Multimodal music information processing and retrieval: survey and future challenges

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Abstract—Towards improving the performance in various music information processing tasks, recent studies exploit different modalities able to capture diverse aspects of music. Such modalities include audio recordings, symbolic music scores, mid-level representations, motion and gestural data, video recordings, editorial or cultural tags, lyrics and album cover arts. This paper critically reviews the various approaches adopted in Music Information Processing and Retrieval, and highlights how multimodal algorithms can help Music Computing applications. First, we categorize the related literature based on the application they address. Subsequently, we analyze existing information fusion approaches, and we conclude with the set of challenges that Music Information Retrieval and Sound and Music Computing research communities should focus on in the next years.

Index Terms—Multimodal music processing, music information retrieval, music description systems, information fusion

I. INTRODUCTION

Beginning with the oldest evidence of music notation, music has been described in several different forms [1]. Such descriptions have been used by computational systems for facilitating music information computing tasks. Interestingly, when observing the history of music, one can see how the various descriptive forms have gradually emerged with a strict dependence both on technology advancements and changes in music practices.

Initially, no written description systems for music existed besides text. Between the 6th-7th cen., Isidore of Seville, Archbishop and theologian, wrote that no melody could be written. Indeed, the first systems to memorize music were based solely on lyrics and only later some signs over the words appeared. Such notation, called neumatic, evolved in more complex forms, which differed from region to region. Due to the need of more powerful tools to express music features, new notation systems, called pitch specific, took place, such as the alphabetic and the staff-based notations. In particular, the system introduced by Guido d’Arezzo (10th-11th cen.) was particularly successful and similar conventions spread all over Europe. Music notation was now able to represent text, pitches and durations at the same time. During the following centuries, other types of symbols were introduced addressing directly the performer towards peculiar colors, or sentiment(s). At the crossing of the 16th and 17th cen., Opera was born in Italy, after a long tradition of plays, including Greek drama, medieval entertainers and renaissance popular plays (both liturgic and profane) [2]. The tremendous success of the Opera in Italy and then in the rest of Europe, determined a fundamental way to connect music and visual arts for the future centuries. A turning point in the history of music description systems was the invention of the phonograph cylinder by Thomas Edison in 1877 and the disc phonograph diffused by Emile Berliner ten years later [3]. In the same years, Edison and the Lumière brothers invented the first devices to record video [4]. Since then, a number of technologies were born paving the way for new music description systems. With the invention of computers and the beginning of the digital era, the elaboration of sound signals highlighted the need for more abstract information characterizing audio recordings. Thus, researchers started proposing mid-level representations [5], with reference to symbolic and physical levels [6]. Nowadays, the availability of vast, easily accessible quantities of data, along with appropriate modern computational technologies, encourages the collection of various types of meta-data, which can be either cultural or editorial [7].

From a cognitive point of view, the connecting, almost evolutionary, element between the above-mentioned representations is that each one relates to a different abstraction level. Psychology, indeed, is almost unanimous in identifying an abstraction process in our music cognition [8]: we can recognize music played on different instruments, with different timings, intensity changes, various metronome markings, tonalities, tunings, background noises and so on. The different descriptions of music developed in different era or contexts, can be seen as an answer to the necessity of representing new modalities – such as the visual one – or new unrevealed abstraction levels – such as the audio recordings and the mid-symbolic levels, or the pitch specific notation compared to the neumatic one.

Aside from these historical and cognitive considerations, it is a fact that in the last two decades researchers have obtained better results through multimodal approaches in respect to single-modalities approaches [9], [10]. As Minsky said [11]:

To solve really hard problems, we’ll have to use several different representations.

We argue that music processing tasks can benefit profoundly from multimodal approaches, and that a greater focus is needed by the research community in creating such a syn-
ergistic framework. A fundamental step would be the study and design of suitable algorithms through which different modalities can collaborate. Then, a particular effort should be devoted in developing the needed technologies. In fact, given the course of history summarized above, we could expect that in the future, new disparate music representations will be born.

In this paper, we review the existing literature about Music Information Retrieval techniques which exploit multiple descriptions of music to the end of multimodal fusion \[12\]. The paper is organized as follows: in section II, we give some basic definition and discuss previous reviews on similar topics to explain the categorization and the taxonomy we used. Sections III to VII describe the different tasks faced with multimodal approaches, the various features extracted, the preprocessing steps and the fusion approaches adopted in literature; in section VIII we express our idea about how the multimodal paradigm can be enhanced.

II. DEFINITIONS, TAXONOMY AND PREVIOUS REVIEWS

We have found no univocal definition of modality. In the music computing literature, authors use the word multimodal in two main contexts:

- in computational psychology, where modality refers to a human sensory channel;
- in music information retrieval, where modality usually refers to a source of music information;

Since we are focusing on music information retrieval methods, to the purpose of the present paper, with modality we mean a specific way to digitize music information. Different modalities are obtained through different transducers, in different places or times, and/or belong to different media. Examples of modalities that may be associated to a single piece of music include audio, lyrics, symbolic scores, album covers, and so on.

Having defined what we mean by modality, we define multimodal music information processing as an MIR \[13\] approach which takes as input multiple modalities of the same piece of music. All the papers which we are going to discuss show methods which take as input various music representations. Conversely, we are not considering those approaches which exploit features derived through different methods from the same modality: an example is pitch, rhythmic and timbral features, when they are all derived from the audio \[14\]. Similarly, we are not considering approaches which process multiple representations of the same modality: an example is spectrograms (treated as 2D images) and traditional time-domain acoustic features \[15\], which are both derived from the audio. Moreover, we do not focus on general multimodal sound processing: the idea which moves our effort is that music is characterized by the organization of sounds in time; thus, we are interested in exploiting this organization, which is not available in general sound processing.

One previous review on multimodal music processing was written in 2012 \[12\]. However, that work was more focused on a few case studies rather than on an extensive survey. The authors recognized a distinction between “the effort of characterizing the relationships between the different modalities”, which they name cross-modal processing, and “the problem of efficiently combining the information conveyed by the different modalities”, named multimodal fusion. To our analysis, this distinction is useful if with cross-modal processing we mean the end-user systems which offer an augmented listening experience by providing the user with additional information. If this is the case, we are primarily interested in multimodal fusion; nevertheless, some synchronization algorithms, which are classified as cross-modal processing by the previous authors \[12\], are used as pre-processing steps in other works. Because of this ambiguous distinction, we base our classification on the performed task rather than on the processing stage – see section III.

Almost all authors dealing with multimodal information fusion talk about two approaches: early fusion and late fusion. Figure 1 shows the main difference between the two approaches: in early fusion, data is used “as is” in one single processing algorithm which fuse the data representation, while in late fusion data from each modality is first processed with specific algorithms and then all the output are merged, so that it is the output to be fused and not the data. Because of this, early fusion is also called feature-level fusion, and late fusion is also called decision-level fusion, even if features extraction and decision algorithms are not the only approaches for multimodal processing. Some reviews \[9\] also talk about hybrid fusion for multimedia analysis, but we have found no example in the music domain.

Finally, we have found useful to introduce a new diagram to represent the data flow in retrieval systems (see fig. 2). Indeed, in most of these systems, one modality is used to query a database for retrieving another modality; in such cases, no fusion exists, but just a data conversion and a similarity computation.

An exhaustive and continuously updated table, which summarizes all the works reviewed in this paper, is available online\[1\].

III. MULTIMODAL MUSIC PROCESSING TASKS

To date, several tasks have been experimented in multimodal approaches. We found two possible categorizations for the application level:

- less vs more studied tasks: some tasks have been extensively studied with a multimodal approach, such as audio-to-score alignment, score-informed source separation, music segmentation, emotion or mood recognition; other tasks, instead, have been little explored and are worth of more attention.
- macro-task based categorization: we identified 4 different macro-tasks, that are a partial re-elaboration of a previous effort \[13\]: classification of music, synchronization of different representations, similarity computation

\[1\] Link: https://frama.link/multimodal-MIR
The difference modalities between the query and the retrieved system. The multimodal approach, here, can exist either in explicitly expressed by the user or implicitly deduced by the documents from a collection through a query, which can be obtained by merging and processing multiple modalities. Possibly the most studied task within this family is score-informed source separation [17], in which symbolic music data and audio recordings of a musical ensemble are used to create different documents or in the query itself. A very common example of explicit queries for retrieving another modality is query-by-humming or query-by-example, in which the query is represented by an audio recording and the system retrieves the correct song; this task is usually performed with two main approaches: by using a collection of recordings in a single-modality fashion, or by exploiting multimodality with a collection of symbolic data [20], [21]. An example of implicit query systems, instead, are recommender systems and playlist generators, where the user is usually not aware of which specific parameters are used for the recommendations; most of the recent research in this field tries to exploit multimodal approaches – also called hybrid – involving metadata, user context, audio features [22], [23]. An emerging field in the retrieval context is the so-called multimodal queries, where the user can explicitly create a query by using different parameters for different modalities [15], [24]. Following this line of thought, some researchers devised and studied novel tasks in the context of multimodal music retrieval. Some example are: a system for retrieving music score images through audio queries [19]; an algorithm to retrieve the cover of a given song [25]; systems to retrieve audio recordings through symbolic queries [26], [27]; an approach to query a music video database with audio queries [28].

C. Classification

The classification process consists in taking as input a music document and returning one or more labels. A popular multimodal classification task is the mood or emotion recognition [29], while an emerging one is genre classification [30]–[37]. Both these two tasks can take advantage of audio recordings, lyrics, cover arts and meta-tags. Additionally, emotion recognition can exploit EEG data, while for genre classification one can use music video and generic text such as critic reviews. Usually, just one modality is considered in addition to audio recordings, but an interesting work [37] tries to exploit more than two modalities. Other multimodal classification tasks found in the literature are:

- artist identification, through lyrics and audio fusion [38];
- derivative works classification of youtube video through audio, video, titles and authors [39];
- instrument classification by exploiting audio recordings and performance video [40], [41];
- tonic identification, that is: given an audio recording and the note level, find the tonic [42];
- expressive musical description, which consists in associating a musical annotation to an audio recording by extracting musical features with the help of symbolic level [43].

D. Time-dependent representation

With time-dependent representation, we mean the creation of a time-dependent description of the music data, created by merging and processing multiple modalities. Possibly the most studied task within this family is score-informed source separation [17], in which symbolic music data and audio recordings of a musical ensemble are used to create different

![Diagram showing the flow of information in early-fusion and late-fusion. Early fusion process takes as input the output of the pre-processing of the various modalities, while the late fusion takes as input the output of specific processing for each modality. Hybrid fusion, instead, uses the output of both early and late fusion.](image)
Figure 2. Multimodal retrieval: usually, the query and the collection contain different modalities, so that the diagram should be collapsed to the highlighted elements; however a more general case is possible [19], in which both the query and the collection contain multiple modalities.

Figure 3. The tasks identified in literature, divided in 4 macro-tasks and plotted along a less - more studied axis. Tasks for which only one paper has been found appear at the left-side (less studied); at the rightmost side are tasks for which extensive surveys are already available; the other tasks are placed in the remaining space proportionally to the number of corresponding papers found in literature. All references to these tasks can be found in the discussion and in the online spreadsheet – see footnote[1]. Note that labels refer to the multimodal approach at hand and not to generic MIR tasks – e.g. genre classification task is intended to be performed with a multimodal approach and thus it has been less studied than emotion or mood classification in the context of multimodal approaches.

audio recordings for each different instrument. A number of researchers have also tried to use audio and video recordings of a music performance or of a dancer to extract beat tracking information [44]–[48]. An emerging task is piano tutoring, which consists in the tracking of errors in a piano performance: to this end, the audio recording, the knowledge about the instrument timbre and the symbolic score can be exploited [49]–[55]. Less studied tasks are:
• music segmentation, in which audio and video, lyrics or note level can be exploited to identify the music piece structure [56]–[58];
• spatial transcription, that is the inference, starting from audio and video, of the note level of songs for fretted instruments, so that the resulting score includes the annotation of fingering [59], [60];
• onset detection through audio and performer video [61] or rhythmic structure knowledge;
• chords labeling, by comparing multiple audio recordings of the same work [62];
• source association, that is the detection of which player is active time by time by exploiting audio, video and music scores [63], [64];
• multi-pitch estimation, that is the transcription of parts being played simultaneously, with the help of performance video to detect play-nonplay activity of the various instruments [65].

IV. DATA PRE-PROCESSING

Data pre-processing is the elaboration of data to the end of transforming their representation to a more suitable format for the subsequent steps. We have identified a number of possible non-exclusive types of pre-processing:
• Synchronization: the synchronization process described in section III-A is sometime used as pre-processing step
to align multiple modalities; thus, the pre-processing itself can be multimodal. For example, in piano tutoring and score-informed source separation, an audio-to-score alignment is performed; audio-to-audio synchronization is a fundamental pre-processing step in tasks requiring comparison of multiple recordings of the same piece [62]; audio-to-score alignment is also used in several previously cited works [26], [27], [43], [58].

- **Feature extraction:** usually, music representations are not used as they are, but a number of features are extracted – see section V.
- **Other pre-processing steps include:**
  - conversion from one modality to the other, such as in query-by-humming – which includes a conversion from audio to the symbolic level – or in audio-to-score alignment where symbolic scores can be converted to audio through a synthesis process.
  - feature selection through Linear Discriminant Analysis (LDA) [28] or ReliefF [43].
  - normalization of the extracted features [48].
  - source-separation in lyrics-to-audio alignment and source association [63], [64].
  - chord labeling on audio only [62].
  - multi-pitch estimation on audio only [65].
  - video-based hand tracking [59].
  - tf-idf-based statistics – see section V-C – adapted for audio [38].

Finally, we think that a step worthy of a particular attention is the conversion to a common space of the extracted features, to make them comparable. We will talk about this step in section VI. The accompanying online table (see footnote 1) contains a short description of the pre-processing pipeline adopted in each cited paper.

V. FEATURE EXTRACTION IN MULTIMODAL APPROACHES

Various types of features can be extracted from each modality. In this section, we provide a general description for audio, video, textual and symbolic score features.

A. Audio features

This section is mainly written with reference to a previous review [66]. Audio features can be broadly subdivided in physical and perceptual.

1) Physical features: these can be computed in various domains, such as time, frequency or wavelet. Time-domain features can be computed directly on the digitally recorded audio signal and include zero-crossing rate, amplitude, rhythm and power-based features, such as the volume, the MPEG-7 temporal centroid or the beat histogram. Frequency-domain features are the richest category; they are usually computed through a Short-Time Fourier Transform (STFT) or an autoregression analysis and can be subdivided in: autoregression-based, STFT-based and brightness, tonality, chroma or spectrum shape related. Features in the Wavelet-domain are computed after a Wavelet transform, which has the advantage of being able to represent discontinuous, finite, non-periodic or non-stationary functions. Image-domain features are computed through a graphic elaboration of the spectrogram, that is a matrix that can be represented as a one-channel image computed with the STFT; often, spectrogram is used as input for a convolutional neural network (CNN), which is trained to compute ad-hoc features, which lack straightforward interpretation.

2) Perceptual features: these try to integrate human sound perception processing in the feature extraction stage or in the elaboration of physical audio features. Most of them aim at mapping certain measurements to a perceptual-based scale an/or metrics. For example, Mel Frequency Cepstral Coefficients (MFCC) are derived by mapping the Fourier transform to a Mel-scale, thus improving the coherence with human perception. Perceptual wavelet packets [67] employ a perceptually motivated critical-band based analysis to characterize each component of the spectrum using wavelet packets. Loudness is computed from the Fourier transform with the aim of providing a psychophysically motivated measure of the intensity of a sound.

B. Video and image features

This section is mainly written with reference to a previous work [48]. Video features used in the music domain are very similar to visual features used in general purpose video analysis. Image features can be based on the color space (RGB or HSV), on edges detection, on the texture – such as the LBP –, or on the moment of a region. In video, motion detection is also possible and can be performed with background detection and subtraction, frame difference and optical flow. Object tracking has been also used to detect hand movements, for example in piano-tutoring applications. Object tracking can happen by exploiting the difference between frames of the detected object contours, by using deviations frame-to-frame of whole regions or generic features. In video, one can also detect shots, for example by analyzing the variation of the color histograms in the video frames, using the Kullbach-Leibler distance [68] or other metrics.

In genre and mood related analysis, other features can also be exploited [69]. The use of tempo is essential to express emotions in video clips, and can be analyzed through features related to motion and length of video shots. Another relevant factor is lighting, that can be measured through brightness-based features. Colors have an affective meaning too, and color features are consequently useful for genre or emotion recognition.

Finally, images can also be used as they are as input of CNNs.

C. Text features

This section is written with reference to a previous review [70]. The most common text representations are based on tf-idf. In this context, tf(d, t) is the term frequency and is computed as the number of occurrences of a term t in a document d. Instead, idf(d, t) is a short for inverse document frequency and is needed to integrate the discrimination power
of the term \( t \) for the document \( d \), considering the whole collection; it is related to the inverse ratio between the number of documents containing \( t \) at least once and the total number of documents in the considered collection:

\[
\text{idf} = \frac{\text{docs in collection}}{\text{docs containing } t}
\]

Usually, \( \text{tf-idf} \) takes the following form:

\[
\text{tf-idf}(d, t) = \text{tf}(d, t) \times \log[\text{idf}(d, t)]
\]

Features based on \( \text{tf-idf} \) are often used in Bag-of-Words (BoW) models, where each document is represented as a list of words, without taking care of the cardinality and order of words. In order to make BoW and \( \text{tf-idf} \) models effective, a few preliminary steps are usually performed, such as punctuation and stop-words removal and stemming. More sophisticated methods are also available, allowing for topic- or semantics-based analysis, such as Latent Dirichlet Allocation (LDA), Latent Semantic Analysis (LSA), Explicit Semantic Analysis (ESA) [71] and CNN feature extraction.

For lyrics analysis, other types of features can be extracted, like rhymes or positional features. Finally, when the available text is limited, one can extend it with a semantic approach consisting in knowledge boosting [37].

D. Symbolic score features

Symbolic music scores have been rarely used in feature extraction approaches. Most of the papers which deal with symbolic scores use MIDI-derived representations, such as the pianoroll [17] or inter-onset intervals (IOD) [58]. To the end of audio-symbolic comparison, one can compute chromograms, that are also computable from the audio modality alone. However a number of representation exist and have been tested in Music Information Retrieval applications, such as pitch histograms, Generalized Pitch Interval Representation (GPIR), Spiral Array, Rico-Íñesta trees, Pinto graphs, Orin-Rodà graphs and others. A brief review of the music symbolic level representations is provided in a previous work [72].

VI. CONVERSION TO COMMON SPACE

The conversion of the extracted features to a common space is often a mandatory step in early fusion approaches. Nevertheless, almost no authors emphasize this aspect. Thus, we think that greater attention should be posed on this step of the pre-processing pipeline.

The conversion to a common space consists in the mapping of the features coming from different modalities to a new space where they are comparable. This can be needed in single-modality approaches too, when the features refer to very different characteristics of the signal. Indeed, many papers describe techniques which include a mapping of the features to a common space, both in the pre-processing and in the processing stages, but no particular attention to the conversion itself. Common methods include:

- **normalization**, that is the most basic approach;
- **conversion from one modality to another**, so that features can be computed in the same units;
- **machine learning algorithms** such as CNNs or SVMs: SVMs compute the best parameters for a kernel function that is used to transform the data into a space where they are more easily separable; CNNs, instead, can be trained to represent each input modality in a space such that the last network layers can use as input the concatenation of these representations;
- **dimensionality reduction** algorithms, which usually search for a new space where data samples are representable with a fewer number of dimensions without losing the ability to separate them; examples are Principal Component Analysis (PCA) and Linear Discriminant Analysis (LDA).

It must be said that some types of features are suitable for multimodal fusion without any conversion step. For example, chroma features can be computed from both the audio recordings and the symbolic scores and thus can be compared with no additional processing.

A possible categorization of the conversion to common space methods is between coordinated and joint: in the former type, the mapping function takes as input a unimodal representation, while in the latter type it takes as input a multimodal representation [10]. In other words, coordinated conversion learns to map each modality to a new space trying to minimize the distance between the various descriptions of the same object, while joint conversion learns the best mapping function which uses all the modalities and optimizes the subsequent steps – e.g. SVM.

VII. INFORMATION FUSION APPROACHES

Two major information fusion approaches exist: early fusion and late fusion – see fig. [1]. Some authors also report a hybrid approach [9], which consists in fusing information both in a early and late fashion and in adding a further step to fuse the output of the two approaches. Nevertheless, we did not find any existing application to the music domain. Before discussing in detail the two approaches, we recall that no fusion is usually needed in similarity tasks, but just a comparison of the various modalities and, thus, a conversion to a common space. The accompanying online table (see footnote [1]) contains a short description of the fusion approach used in all the cited papers. To our understanding the main difference between early and late fusion is about their efficiency and ease of development; however authors disagree about which one is the more effective.

A. Early fusion

Early fusion consists in the fusion of the features of all the modalities, using them as input in one single processing algorithm. Although the development of such techniques is more straightforward, they need a more careful treatment because the features extracted from various modalities are not always directly comparable.

To the end of synchronization, the most used approach exploits Dynamic Time Warping algorithms (DTW) [73]. DTW is a well-known technique based on a similarity matrix
between two sorted sets of points, for example two time-sequences. By using a dynamic programming algorithm, one can exploit the similarity matrix to find the best path which connects the first point in one modality to the last point in the same modality and which satisfies certain conditions. This path will indicate the corresponding points between the two modalities. Other common methods for synchronization purposes are Hidden Markov Models (HMMs) [18], [58] where hidden states represent points in one modality and observations represent points in a second modality; this is particularly effective for real-time alignment or generic sequence fusion such as in time-dependent descriptions.

Aside HMMs, many additional machine learning approaches are used to perform early fusion: Support Vector Machines (SVMs), Gaussian Mixture Models (GMMs), Convolutional Neural Networks (CNNs) and Particle Filters are the most used techniques.

Another interesting method is Non-negative Matrix Factorization (NMF), through which audio and symbolic scores can be exploited to the end of precise performance transcription, as in score-informed source separation and piano tutoring applications [17]. In NMF, a matrix $A$ is decomposed in two components $C$ and $B$, so that $A = B \times C$. If $A$ is a spectrogram and $B$ a template matrix dependent on the instrumentation, then we can think to $C$ as a pianoroll matrix – see fig. 4. Consequently, one can use an optimization algorithm to minimize the function $f(B, C) = A - B \times C$, by initializing $C$ with a symbolic score; at the end of the optimization, $C$ will be a precise transcription of the performance contained in $A$.

Finally, feature fusion can also happen at the feature selection stage [38], [43].

B. Late fusion

Unlike early fusion, late fusion is the fusion of the output of various ad-hoc algorithms, one for each modality. It is also called decision-level fusion, even if a decision process is not mandatory. The main advantage of late fusion is that it allows for a more adjustable processing of each modality. However, it is usually more demanding in terms of development costs.

In classification and time-dependent description tasks, the most used types of late fusion are rule-based. Rules can include voting procedures [62], [64], linear and weighted combinations [41], [75], maximum and minimum operations [41], [75]. Many authors have developed sophisticated algorithms to execute this step, such as in beat tracking, piano tutoring

and structural segmentation [57], multi-pitch estimation [65] and tonic identification [42].

In synchronization tasks, instead, no late-fusion approach is possible, since the task consists in creating associations between points in different modalities and, thus, the process must take as input all the modalities, eventually in some common representation.

VIII. FUTURE DIRECTIONS

In this paper, we have analyzed the literature on multimodal music information processing and retrieval. Based on our study, we propose the following concluding remarks.

First of all, we note the unavailability of datasets of suitable size. This issue is usually addressed with various methods such as co-learning approaches [10], which has the side-effect of impoverishing the goodness of the developed algorithms. Although a few datasets have been recently created [57], [76]–[78], a great effort should still be carried out in this direction. Indeed, existing multimodal music datasets are usually characterized by limited sizes and only rarely include a wide range of modalities. However an exhaustive list of the available datasets is out of the scope of this paper. We argue that this limit is due to two main reasons: first, the precise alignment of various modalities is a hard computational task and should be controlled by human supervision; second, no largely adopted standard exists for multimodal music representation. About the first point, more effort should be devoted to the development of algorithms for the alignment of various sequences. The representation of the intrinsic music multimodality, instead, is faced by the IEEE 1599 standard and the Music Encoding Initiative; moreover, the W3C group is currently working on a new standard with the purpose of enriching MusicXML with multimodal information. The course of history described insertion and the rapid technology advancements of our times suggest that new representation modalities could be needed in the future and that multimodal representation standards should also focus on this challenge.

Another challenge that multimodal music researchers should face in the next years is the exploration of various techniques already used in multimodal processing of multimedia data, that have not been tested in the musical domain. According to previous surveys [9], [10], multimodal methods never applied to the music domain include: the hybrid approach, the Dempster-Shafer theory, Kalman filters, the maximum entropy model, Multiple Kernel Learning and Graphical Models. Moreover, we have found only one paper in which the information fusion happens during the feature extraction itself [58] and not afterwards. This approach should be explored more deeply.

Finally, we suggest that the conversion to common space should be more rigorously addressed. To this end, transfer learning technologies could be explored towards forming a

\[\text{IEEE 1599 website: http://ieee1599.lim.dii.unimi.it/}\]

\[\text{W3C music notation group website: https://www.w3.org/community/music-notation/}\]
synergistic feature space able to meaningfully represent multiple modalities [79]. [80]. Such a direction may include the use of an existing feature space characterizing a specific modality, or the creation of a new one where multiple modalities are represented. Such a space could satisfy several desired properties, such as sparseness, reduced dimensionality, and so on.

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