

VAE	To	Oc	Sc	R1	R2	A1	A2	A3	A4
$\beta$	1	.95	.14	.64	.64	.28	.23	.28	.24
AR	1	.95	.34	.09	.09	.02	.02	.02	.02
I	1	.99	.36	.73	.77	.50	.52	.54	.53
S2	1	.94	.12	.65	.70	.29	.23	.34	.25

**Table 1.** Latent Density Ratio (LDR) for four dMelodies VAE models. Attributes are Tonic (To), Octave (Oc), Scale (Sc), Rhythm Bar (R) and Arp Chord (A).

verse from a position  $z$  in the latent space, along the corresponding dimension  $z_d$  between a minimum and maximum value over a number of steps. This is repeated over a small batch of points. Three measures are defined by [7] based on the work of [26]: *Consistency* measures how constant the attribute is across the batch for the same value of  $z_d$ , *Restrictiveness* measures how constant other attributes are when  $z_d$  changes, and *Linearity* measures how linear the attribute is with respect to the latent dimension  $z_d$ . An *attribute change matrix*  $A(d, n)$  is proposed by [11] which computes the net change in the  $n^{th}$  attribute while traversing regularised dimension  $d$ . When an attribute is well controlled  $A$  should be high on the diagonal, and low elsewhere, and these relationships can be inspected by visualising the matrix. Another measure of controllability is the correlation between the interpolated value  $z_d$  and the resulting attribute [14], which measures linearity but not restrictiveness.

Another important factor is the *Latent Density Ratio* (or *LDR*), the proportion of a batch of random latent samples that decode with valid attributes [11]. As shown in Table 1, errors in dMelodies occur most frequently with the *Scale*, *Rhythm Bar* and *Arp Chord* attributes. If notes are generated that are outside the three defined scales (major, minor, blues) the *Scale* attribute is invalid. If a bar does not have exactly 6 note onsets the *Rhythm Bar* attribute does not match one of the 28 codes, and is invalid. If the chord notes are not consistently ascending or descending, the *Arp Chord* attribute is invalid. A traversal function should aim to be more accurate than these base-line LDR rates.

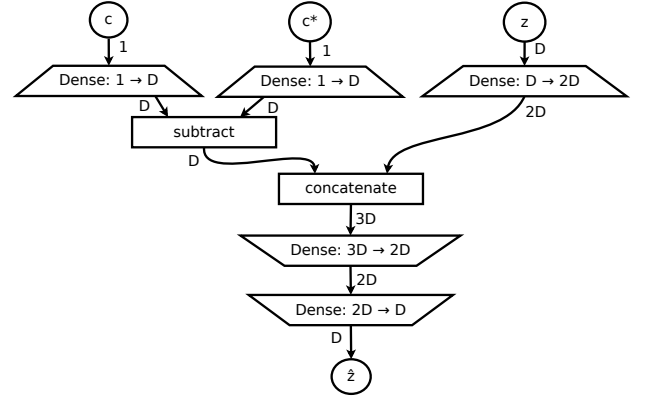
### 3. METHOD

#### 3.1 Neural Latent Traversal

This section describes SeNT-Gen as applied to musical VAE models. Further details are available for a GAN-based image synthesis application [27].

Let  $x$  be a sample of the data space  $X = \mathbb{R}^N$ , a piano-roll encoding of a melody. Let  $z$  be a sample of the latent space  $Z = \mathbb{R}^D$  of the VAE model. Let  $c = [c_1, c_2, \dots, c_K]$  be the vector of  $K$  semantic attributes of  $x$ , and  $R_1(\cdot), \dots, R_K(\cdot)$  be  $K$  functions that compute the value of attribute  $k$  of  $x$  in the normalised range  $[0, 1]$ :  $c_k = R_k(x)$ . Let  $\mathcal{F}(\cdot)$  be the encoder, and  $\mathcal{G}(\cdot)$  be the decoder of the VAE, so that  $\mathcal{G}(\mathcal{F}(x)) \approx x$ . We use  $x^*$  for the *target value* of  $x$ , and  $\hat{x}$  for its *predicted* or *achieved* value.

Given a sample  $x$ , we aim to find the modified  $x^*$  such that the value of attribute  $k$  is changed from  $c_k$  to  $c_k^*$  while all other attributes remain unchanged:  $c_i^* = c_i, i \neq k$ .



**Figure 2.** Implementation of SeNT-Gen neural traversal function  $\mathcal{T}_k(z, c_k, c_k^*)$ . Input is the latent code  $z$ , and the desired attribute change from  $c_k$  to  $c_k^*$ . Each Dense layer includes ReLU activation, except for the Tanh activation on the final layer. Output is the latent code  $\hat{z}$  which approximates the solution  $z^*$ .

This adjustment will be done in the latent space. Given the latent encoding  $z = \mathcal{F}(x)$ , we seek a modified  $z^*$  that generates  $x^* = \mathcal{G}(z^*)$ , with the desired attribute value  $c_k^* = R_k(\mathcal{G}(z^*))$ .

SeNT-Gen implements contextual traversal of the latent space for each of the attributes. The traversal function  $\mathcal{T}_k$  predicts the new value  $\hat{z}$  given the old value  $z$ , and the old and new values of the attribute  $c_k$  and  $c_k^*$ :

$$\hat{z} = \mathcal{T}_k(z, c_k, c_k^*)$$

The traversal function  $\mathcal{T}_k$  is implemented by training the neural network shown in Figure 2 which outputs the value  $\hat{z}$ , an approximate solution for  $z^*$ .

Three constraints are imposed during training. The first constraint is to minimise the perceptual loss by ensuring that the changed attribute value  $\hat{c}_k = R_k(\mathcal{G}(\hat{z}))$  is close to the target  $c_k^*$ :

$$\mathcal{L}_c = \mathbb{E}_{z \sim P(Z)} [\|R_k(\mathcal{G}(\hat{z})) - c_k^*\|_2^2] \quad (1)$$

The term  $\|R_k(\mathcal{G}(\hat{z})) - c_k^*\|_2$  represents the deviation between  $\hat{c}_k$  and the target  $c_k^*$ , and is in the range  $[0, 1]$ . When  $R_k$  is invalid for  $\mathcal{G}(\hat{z})$  a deviation of 1 is used instead.

The other two constraints are required to align the attributes of the traversal function with the relevant dimensions of the latent space. In a disentangled latent space each semantic attribute corresponds to one latent dimension, but in general there will likely be a small number of relevant dimensions that are strongly related to each attribute. For *relevant* dimensions  $r$ ,  $\hat{z}_r$  should be close to the correct solution  $z_r^*$ . For the other *irrelevant* dimensions  $i$ ,  $\hat{z}_i$  should be close to the original location  $z_i$ .

Relevance is expressed using the vector  $\rho_k \in \mathbb{R}^D$ , which is 1 for relevant dimensions and 0 otherwise. Under supervised training, this relation will be known and for a consistent numbering of attributes and dimensions  $\rho_k[i] = 1$  for  $i = k$ , and 0 for  $i \neq k$ . Otherwise, for unsupervised models  $\rho_k$  can be calculated using mutual

information as described below. The two remaining loss functions are thus:

$$\mathcal{L}_I = \mathbb{E}_{z, z^* \sim P(Z)} [\|\rho_k^T(\hat{z} - z^*)\|_2^2] \quad (2)$$

$$\mathcal{L}_{-I} = \mathbb{E}_{z, z^* \sim P(Z)} [\|(1 - \rho_k)^T(\hat{z} - z)\|_2^2] \quad (3)$$

Equation 2 pulls  $\hat{z}$  towards  $z^*$  for relevant dimensions, and Equation 3 pulls  $\hat{z}$  towards  $z$  for irrelevant dimensions.

When  $\rho_k$  is not known a-priori it can be calculated using the mutual information (MI) between the dimensions  $d \in \{1, \dots, D\}$  of  $z$  and the  $c_k$  values:

$$\rho_k = \text{threshold}(\text{softmax}(\text{MI}(z_d, c_k)), \gamma) \quad (4)$$

Where  $\gamma$  is a hyper-parameter controlling the number of latent dimensions related to each semantic attribute. Equation 4 selects  $\gamma$  dimensions of  $z$  that have the most mutual information with each attribute  $c_k$ .

### 3.2 Latent Traversal Training

The traversal function  $\mathcal{T}_k$  is trained on similar pairs of latent vectors  $(z, z^*)$  which differ only in attribute  $c_k$ . Such pairs are good examples of attribute modifications. We sample pairs of the data space  $(x, x^*)$ , and compute their attributes  $(c_k, c_k^*)$ , where  $c_k = R_k(x)$ , and  $c_k^* = R_k(x^*)$ . We use the encoder to compute their latent representation  $(z, z^*)$ , where  $z = \mathcal{F}(x)$  and  $z^* = \mathcal{F}(x^*)$ . We discard pairs where there is a significant difference in attributes other than  $k$ . A training pair is considered valid if:

$$\sum_{j \neq k} \|c_j - c_j^*\|_2 \leq \epsilon, \quad \text{for } j \in \{1, \dots, K\} \quad (5)$$

Where  $\epsilon \geq 0$  is a slack parameter. One implementation is to enumerate all pairs from a set of candidate samples, and adjust  $\epsilon$  to be as small as possible while also yielding enough similar pairs.

After generating the training pairs, we train the traversal model  $\mathcal{T}_k$  by minimising the loss:

$$\mathcal{L}_k \triangleq \lambda_1 \mathcal{L}_I + \lambda_2 \mathcal{L}_{-I} + \lambda_3 \mathcal{L}_c \quad (6)$$

where  $\lambda_1$ ,  $\lambda_2$ , and  $\lambda_3$  are hyper-parameters that balance perceptual loss versus semantic relevance.

## 4. EXPERIMENTS

We train each of the four dMelodies learning models ( $\beta$ -VAE, AR-VAE, I-VAE and S2-VAE) for three different hyperparameter settings and 3 different random seeds, a total of 36 models. Input is a two-bar melody, with a latent space of  $D = 32$  dimensions. For  $\beta$ -VAE we vary  $\beta \in \{0.2, 1, 4\}$ , and for the other models we vary the regularisation strength  $\Gamma \in \{0.1, 1, 10\}$  for  $\beta = 0.2$ . The default settings use 1016k melodies (75%) for training the original models, which leaves 338k melodies (25%) as candidates for training and testing the SeNT traversal function  $\mathcal{T}_k$ . Setting  $\epsilon = 0$  yields between 42k and 157k pairs (see Equation 5) and from these we allocate 80% for

training and 20% for testing. Since dMelodies is generated by combinatorial expansion, attributes with the most states (*Tonic*, *Rhythm Bar*) have the most pairs, while those with the fewest states (*Arp Chord*) have the least. Latent codes  $z$  are normalised to the range  $[-1, 1]$ , and attributes  $c_k$  to  $[0, 1]$ . Neural traversal is trained for  $\gamma = 3$  (see Equation 4),  $\lambda_1 = \lambda_2 = \lambda_3 = 1$  (Equation 6).

### 4.1 Performance Metrics

Performance of the traversal function  $\mathcal{T}_k$  is evaluated by measuring accuracy on a set of attribute changes. A set of testing pairs  $(z, z^*)$  is generated using the same method that generated the training pairs (see Equation 5). These define a starting state  $z$  and an attribute change from  $c_k$  to  $c_k^*$  that leaves other attributes unchanged. Using the traversal function we predict the new latent code  $\hat{z} = \mathcal{T}_k(z, c_k, c_k^*)$ , and compute its attributes  $\hat{c}_i = R_i(\mathcal{G}(\hat{z}))$  for all  $i$ . This approach ensures that only feasible attribute changes are tested, and that these have been unseen during training.

For a targeted change of an attribute, the important measures are: (1) How far is the edited attribute value  $\hat{c}_k$  from the desired target  $c_k^*$ ? and (2) How far are the unedited attributes  $\hat{c}_u$  from the original values  $c_u$ ? The *target deviation*  $\Delta_k$  for target attribute  $k$  is defined as:

$$\Delta_k = |\hat{c}_k - c_k^*| \quad (7)$$

where  $\hat{c}_k = R_k(\mathcal{G}(\hat{z}))$ . The *non-target deviation*  $\nabla_u$  for non-target attribute  $u$  is defined as:

$$\nabla_u = |\hat{c}_u - c_u| \quad (8)$$

When  $\Delta_k$  is low, the traversal achieves the desired attribute  $k$  value. When  $\nabla_u$  is low, the traversal avoids unintended changes to other attributes  $u$ . When combined, these measures are similar to the attribute change matrix of [11], except that  $\Delta_k$  is with respect to specific target values, rather than arbitrary interpolation points.

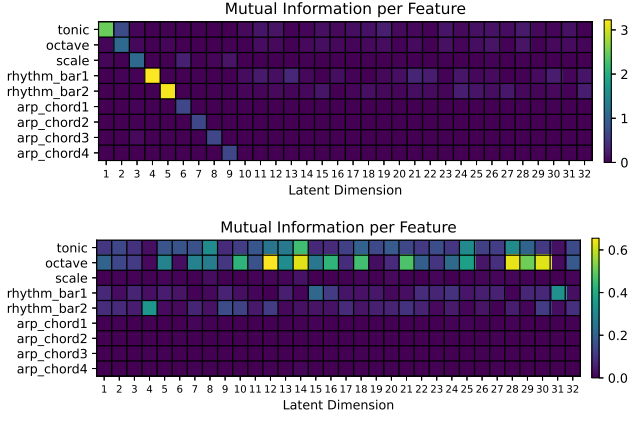
Occasionally errors occur in the generated melodies, due to the decoder quality and the proximity of samples to holes in the latent space. Ideally  $\mathcal{T}_k$  should avoid these holes, and to measure this we define the *Target Density Ratio* (or TDR) to be the proportion of the decoded target  $\hat{z}$  values with valid attributes. We aim for TDR to be substantially higher than the underlying LDR (defined in 2.1).

## 5. RESULTS

For brevity we summarise the 36 models by selecting the best performing hyper-parameters, and aggregate over the three random seeds. For supervised models these are the settings with the most regularisation  $\Gamma = 10$ , and for  $\beta$ -VAE, the median  $\beta = 1$ . Source code is available online with further results and technical details.<sup>1</sup>

Figure 3 shows Mutual Information between semantic attributes (vertical axis) and latent dimensions (horizontal axis). The supervised S2-VAE (top) shows good disentanglement, with each attribute uniquely related to one

<sup>1</sup> <https://github.com/stewartgreenhill/sentgen>



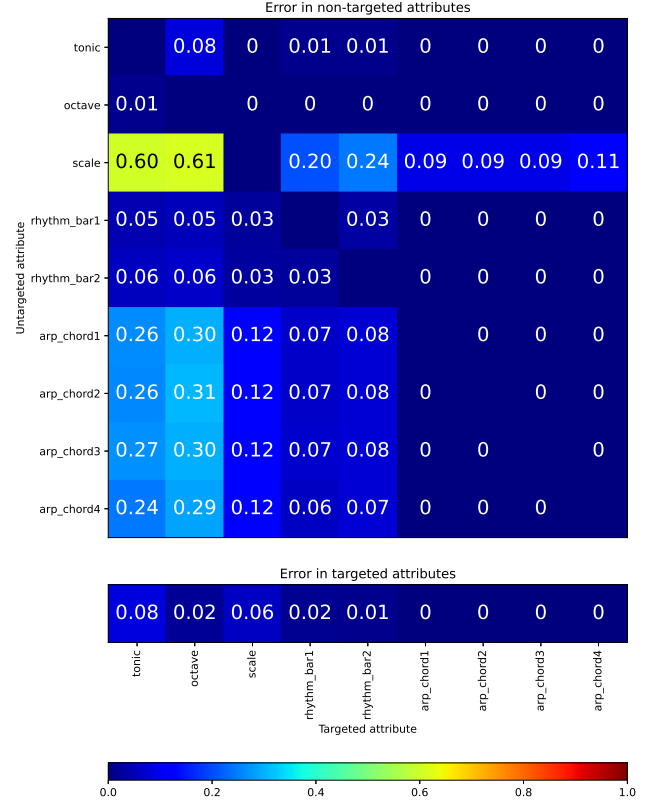
**Figure 3.** Mutual Information between semantic attributes and dimensions of the latent space for supervised S2-VAE (top), and unsupervised  $\beta$ -VAE (bottom).

of the first 9 latent dimensions. The strongest relationships are for the attributes with the most states: *Tonic* and *Rhythm Bar*. The unsupervised  $\beta$ -VAE (bottom) shows some strong but less well separated relationships, and some very weak relationships: *Scale*, *Arp Chord*. Mutual Information is important in SeNT-Gen for determining  $\rho_k$  which aligns semantic attributes to the latent space (Equations 2–4).

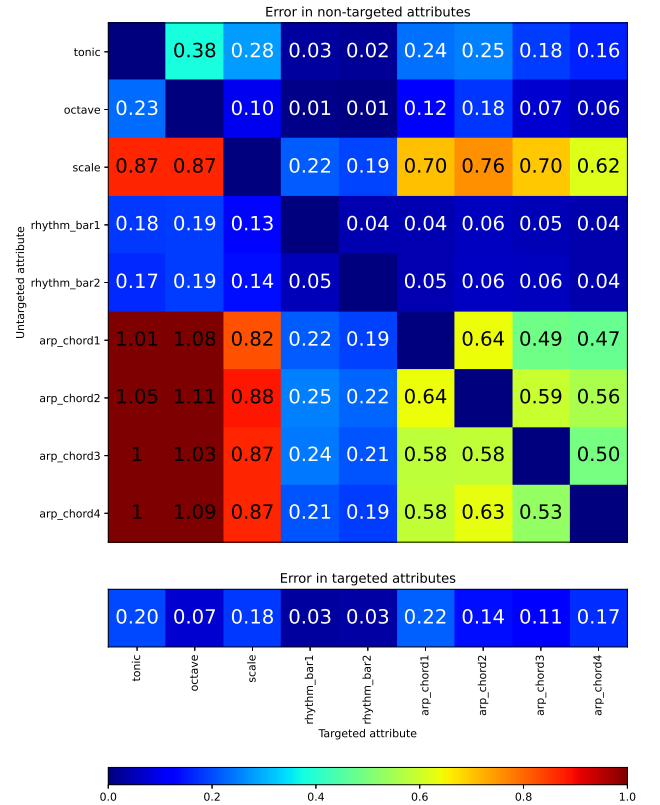
Figure 4 shows SeNT-Gen traversal accuracy for the S2-VAE. The bottom chart shows mean target deviation  $\Delta_k$  for target attributes  $k$ . Smaller deviations are better. For most attributes the deviation is 2% or less, and the worst performance is 8% for attribute *Tonic*. The top chart shows the mean non-target deviation  $\nabla_k$ , with non-target attributes on the vertical axis, and target attributes on the horizontal axis. The *Scale* attribute is most influenced by changes to other attributes, particularly *Tonic*, and *Octave* where the influence approaches 60%. This makes sense since these attributes are both changing the overall pitch of the melody, and it only requires one transposition “error” amongst the 12 melody notes to cause the scale to be invalid or altered. To a lesser extent the *Arp Chord* attributes are also susceptible for the same reason. Changes to *Arp Chord* attributes are the most accurate, with no deviation in the target or non-target attributes other than *Scale*.

Figure 5 shows SeNT-Gen traversal accuracy for the  $\beta$ -VAE which is the worst performing model. Rhythm attributes show a good target deviation of 3%, with the pitch based attribute deviations ranging from 7 to 22%. Non-target errors are lowest for changes to *Rhythm Bar 1* & 2, but are generally much higher than for the S2-VAE. Here *Scale* and *Arp Chord* attributes are most influenced by changes to other attributes. This is expected since these attributes are only weakly related to dimensions of the latent space (see Figure 3) so are essentially invisible to the traversal constraints.

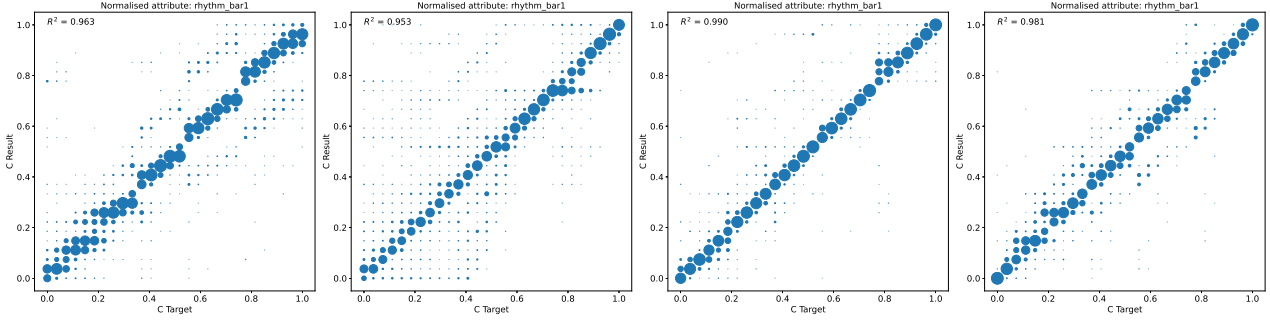
Another way to evaluate traversal accuracy is to look at correlation  $R^2$  between target and achieved attribute values. Figure 6 shows the target  $c_k^*$  value (horizontal) versus the achieved  $\hat{c}_k$  (vertical) for rhythm\_bar1, the attribute



**Figure 4.** Accuracy of S2-VAE traversal, showing *target deviation*  $\Delta$  (bottom), and *non-target deviation*  $\nabla$  (top). Target attributes are on the horizontal axis, and non-targets on the vertical.



**Figure 5.** Accuracy of  $\beta$ -VAE traversal, showing *target deviation*  $\Delta$  (bottom), and *non-target deviation*  $\nabla$  (top). Target attributes are on the horizontal axis, and non-targets on the vertical.



**Figure 6.** Correlation  $R^2$  between normalised target  $c_k^*$  (horizontal axis) and result  $\hat{c}_k$  (vertical axis) for  $k = \text{rhythm\_bar1}$ . Dot area is proportional to number of samples. Models are (left to right)  $\beta$ -VAE, AR-VAE, I-VAE, and S2-VAE.

VAE	To	Oc	Sc	R1	R2	A1	A2	A3	A4
$\beta$	.09	.78	.19	.96	.97	.11	.45	.56	.32
AR	.07	0	-1.3	.95	.97	.98	.99	1	1
I	.26	.34	.98	.99	.97	1	1	1	1
S2	.77	.93	.78	.98	.99	1	1	1	1

**Table 2.** Correlation  $R^2$  between  $\hat{c}_k$  versus  $c_k^*$  for four dMelodies VAE models. Attributes are Tonic (To), Octave (Oc), Scale (Sc), Rhythm Bar (R) and Arp Chord (A).

VAE	To	Oc	Sc	R1	R2	A1	A2	A3	A4
$\beta$	1	1	.30	.92	.92	.57	.52	.64	.62
AR	1	1	.49	.76	.82	.72	.73	.73	.73
I	1	1	.92	.95	.91	.90	.91	.90	.92
S2	1	.99	.65	.93	.93	.91	.90	.91	.91

**Table 3.** Target Density Ratio (TDR) for four dMelodies VAE models. See Table 1 for Latent Density Ratio (LDR).

with the most states. The frequency and location of errors can be visualised by inspecting the off-diagonal elements. Table 2 shows the  $R^2$  values for each of the attributes over four dMelodies VAE models, excluding samples which yield invalid results. As expected, this measure shows high correlations where target deviation  $\Delta$  is low. The three supervised models show strong control over the *Rhythm Bar* and *Arp Chord* attributes with weaker control over *Tonic*, *Octave* and *Scale*.

Table 3 shows the Target Density Ratio (TDR), the proportion of the decoded targets  $\hat{x} = \mathcal{G}(\hat{z})$  that have valid attributes, thus avoiding potential holes in the latent space. The best performer from this perspective is the I-VAE which exhibits good TDR for all attributes. All models score better than the corresponding latent density ratio (LDR, Table 1). Notably, the AR-VAE shows good control over *Arp Chord* with TDR=0.73 despite scoring very low LDR=0.02 for these attributes.

## 6. CONCLUSION

This paper presents a novel algorithm to control musical attributes of deep generative models. The SeNT-Gen method implements a neural traversal function  $\mathcal{T}_k(z, c_k, c_k^*)$  that predicts the latent position  $z^*$  required to change attribute  $k$  of  $z$  from  $c_k$  to  $c_k^*$ . Previous works in this field focus on disentanglement but do not implement traversal functions

and instead assess controllability via latent space interpolation which does not allow the specification of a particular target value  $c_k^*$  for the controlled attribute.

The SeNT-Gen method is demonstrated using the dMelodies data set and various VAE models. Performance is strongest for highly regularised models. The best performance was obtained using the S2-VAE model, which shows strong control over most attributes, and few side-effects except for *Scale*. Some attributes that depend on note pitch (*Scale* and *Arp Chord*) are significantly less stable when adjusting overall pitch via *Tonic* or *Octave*, and also show low LDR scores. The definition of these attributes may be fragile to small changes in the melody notes. The Target Density Ratio (TDR) measures the robustness of the method to holes in the latent space, aggregating factors related to the latent space regularisation, the traversal algorithm, and the fidelity of the decoder. TDR for I-VAE and S2-VAE is good for most attributes, and AR-VAE shows very strong improvement over a poor baseline LDR.

Although each SeNT-Gen traversal function controls only one attribute, more complex operations involving multiple attributes could be performed using a sequence of separate attribute changes. Future improvements to the algorithm could include better support for categorical variables, as well as mixtures different variable types. Categorical variables would normally be one-hot encoded, but alternative distance metrics might be required, for example Jaccard distance where Euclidean distance is currently used (Equations 1 and 5).

While demonstrated here using VAE models, SeNT-Gen can also be used in other latent space models such as Generative Adversarial Networks (GAN). There is an underlying assumption that the relationship between semantic attributes and the latent dimensions will be fairly sparse. This is normally true for supervised models, but can also apply in other models too. In any case the hyper-parameter  $\gamma$  should be chosen to reflect this. An extensive study of the other hyper-parameters  $\lambda_1$ ,  $\lambda_2$ , and  $\lambda_3$  was outside the scope of this work, but fine-tuning these values through meta-optimisation may improve the overall performance.

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# MUSIC REPRESENTATION LEARNING BASED ON EDITORIAL METADATA FROM DISCOGS

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## ABSTRACT

This paper revisits the idea of music representation learning supervised by editorial metadata, contributing to the state of the art in two ways. First, we exploit the public editorial metadata available on Discogs, an extensive community-maintained music database containing information about artists, releases, and record labels. Second, we use a contrastive learning setup based on COLA, different from previous systems based on triplet loss. We train models targeting several associations derived from the metadata and experiment with stacked combinations of learned representations, evaluating them on standard music classification tasks. Additionally, we consider learning all the associations jointly in a multi-task setup. We show that it is possible to improve the performance of current self-supervised models by using inexpensive metadata commonly available in music collections, producing representations comparable to those learned on classification setups. We find that the resulting representations based on editorial metadata outperform a system trained with music style tags available in the same large-scale dataset, which motivates further research using this type of supervision. Additionally, we give insights on how to preprocess Discogs metadata to build training objectives and provide public pre-trained models.

## 1. INTRODUCTION

Developing robust representations is crucial to improving the performance of existing supervised Music Information Retrieval (MIR) tasks on datasets with few annotations. While conventional approaches are based on classification [1–3], other directions include self-supervised strategies [4–9] including leveraging generative models [10], supervision by editorial metadata [11–15], playlist co-occurrences [16], co-listen statistics [15], natural language text [17], or combinations of them [10, 14–16]. While self-supervised systems are narrowing the performance gap

with the supervised approaches, the latter seem to be reaching a performance ceiling in the academic research, partially due to the tremendous difficulty to collect larger annotated datasets.

Using editorial metadata as a source of supervision was already considered in other domains, for example, in scientific publication classification [18] or film recommendation [19]. Likewise, music is naturally rich in metadata. For example, physical formats typically contain detailed editorial information on their covers, digital audio files support containers such as ID3 for this purpose, and most streaming platforms offer album or artist-level browsability. Most music digital service providers routinely require such metadata from content uploaders, and therefore it is often available for music collections in the industry by default. Furthermore, editorial metadata does not suffer from the subjectivity problems common for music tags and tends to be more consistent. Because of these reasons, such metadata allows the creation of potentially larger and less noisy datasets than tag annotations.

Park *et al.* showed that certain types of editorial metadata (artist name) could be used to learn music representations using the triplet loss [11]. Motivated by their study and following the recent success of unsupervised representation learning approaches such as SimCLR [9, 20], we are interested whether modifying such systems to operate with similarity relations based on editorial metadata improves the learned features.

Our proposal differs from previous works on metadata-based music representation learning in two aspects. First, we use editorial metadata from Discogs,<sup>1</sup> a website containing an extensive metadata database including the artists, year, country, record label, and genres/styles of each release. Discogs publishes monthly dumps of their data under the Creative Commons CC0 license that we use to label 3.3 million tracks from our in-house data collection, allowing us to extract conclusions about systems trained on large datasets. Second, we adopt a contrastive approach already performant in a self-supervised setup instead of siamese networks. Specifically, we choose COLA [21] for its simplicity, operating directly on spectral representations, and requiring no data augmentation, which makes it efficient and suitable for large-data regimes. COLA works by maximizing the bilinear similar-



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<sup>1</sup> <https://www.discogs.com/search>



ity between a pair of mel-spectrogram patches (anchor and positive) cropped from the same audio clip while minimizing it for the rest of the patches in the batch. We propose to modify this self-supervised approach by constructing the anchor/positive pairs according to the different types of metadata considered (i.e., same track, release, artist, and record label). Additionally, we explore whether different metadata associations generate complementary information by combining the embeddings produced by their respective models and creating multi-task systems jointly optimized to learn them.

To summarize, we propose investigating the usability of metadata as an inexpensive source of supervision to train contrastive feature extractors for music classification. On top of the original self-supervised COLA approach, we experiment with associations based on editorial metadata to construct the positive pairs. We report results on public benchmarking datasets to facilitate comparison with the SOTA and provide publicly available pre-trained embedding and downstream models.

## 2. RELATED WORK

This paper combines three ideas: supervision based on editorial metadata, usage of the Discogs database, and the recent advances in contrastive learning for MIR. In this section, we review relevant works on these topics.

### 2.1 Metadata-based music representation learning

Park *et al.* shows that the artist name can be used as training target, with the resulting features being close in performance to those obtained from systems trained on crowd-sourced tags [11]. As the artist information is too sparse to be learned efficiently in a classification setup, they approached it as a metric learning task by creating triplets with anchor and positive samples belonging to the same artist and a negative sample belonging to a different one. In a posterior study, they extend the approach to learning track-, album-, artist-level associations, and a combination of all, finding that the latter two provide the best representation [13].

Kim *et al.* propose dealing with the high dimensionality of artist vectors by summarizing them into Latent Dirichlet Allocation topics [22] to create targets suitable for a traditional classification setup [12]. Other works exploit editorial metadata (release year) among other learning sources in multi-task setups [14].

### 2.2 Discogs in MIR research

The Discogs database has already been used for research. Bogdanov *et al.* propose using it in the context of music recommendation [23], MIR, and computational musicology [24]. The former study proposes a recommendation system based on the similarity between artist representations in the form of a tag cloud of the associated genre, style, record label, and release year and country metadata. The latter illustrates the potential uses of the database on various cultural analysis examples including the evolution

of physical distribution formats, genre and style trends, and their co-occurrences. Similarly, some studies analyze music artist collaboration networks [25–28].

In the context of music genre classification, the AcousticBrainz Genre dataset contains mappings across different music genre taxonomies, including the one from Discogs [29]. Hennequin *et al.* use the genre and style labels from Discogs for genre tag disambiguation [30].

### 2.3 Contrastive Learning in MIR

The MIR community has lately adopted contrastive learning approaches, both supervised and self-supervised. Regarding supervised methods, Favory *et al.* use a contrastive objective to align embedded representations of tags and audio [31, 32]. In a subsequent study, the system is enriched with playlists-track interactions as an additional modality to align [16].

On the self-supervised side, several contrastive systems from the computer vision domain have been adapted and evaluated for music-related tasks. CLMR [5] adapts SimCLR [20] to operate on waveforms using musically-meaningful augmentations. BYOL-A [6] relies on the BYOL [33] framework and adapts it to the audio domain by proposing specific augmentations and evaluating it on several audio tasks, including instrument classification. S3T [8] combines the MoCo framework [34] with Swin Transformers [35] to learn music classification features. Wang *et al.* [9] modify SimCLR by using a normalization-free SlowFast [36] backbone and improve the performance in several audio tasks, including music-auto-tagging. PEMR [7] deals with the lack of temporal resolution of existing systems by learning to mask important or irrelevant parts of the mel-spectrogram to produce self-augmented positive/negative samples. COLA [21] is a simple contrastive model for audio that takes random mel-spectrogram patches of the same clip instead of augmented versions of the same patch as anchor/positive pairs. It relies on the bilinear similarity to compute a distance matrix between all the anchors and positives and uses the categorical cross-entropy loss to maximize the similarity between correspondent anchor/positive pairs while using the rest of the positives in the batch as negatives.

We follow a straightforward implementation of COLA, but instead of relying on the same audio clip (purely self-supervised), we form the anchor/positive pairs according to relationships from the metadata. The complete pipeline is represented in Figure 1 (left). This is, to the best of our knowledge, the first usage of this framework in the context of MIR.

## 3. METHODOLOGY

We are interested in assessing the representation power of features derived from associations of commonly available editorial metadata. This goal is motivated by previous works that already showed the usability of the track, artist, and album similarities as supervision targets [13]. As these works already studied the influence of the dataset size and