ISMIR 2021 Paper Summary

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Abstract

A summary of music generation papers accepted by ISMIR 2021. (Publicly Available)

1 SURPRISENET: MELODY HARMONIZATION CONDI-TIONING ON USER-CONTROLLED SURPRISE CON-TOURS

SupriseNet Traits

- (1) relieves trade-off between coherence, which caused by the melody and surprisingness, which caused by the user-supplied condition.
- (2) the chords are correponded to the melody and follow the surprise contour.

Transition matrix $(a \in \mathbb{R}^{N \times N})$: encode the distribution of the chord at this time given the previous element.

 $p(\operatorname{chord}_t|\operatorname{chord}_{t-1})$

Surprisingness: equivalent to the definition of the information content in information theory. Surprisingness = $-\log p(\operatorname{chord}_t|\operatorname{chord}_{t-1})$.

The higher the surprise value at a certain time, the gerater the amount of information at that time.

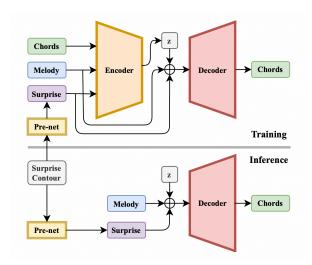


Figure 1: The structure of SURPRISENET.

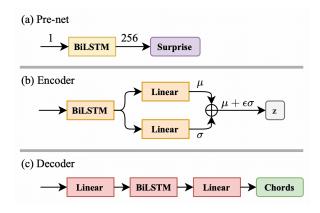


Figure 2: Three main components of SURPRISENET.

Pre-net: extend the feature from a scaler to a 256-dimensional vector.

CVAE: the features of the chord, melody, and extended surprise contour are concatenated and fed into the encoder to generate a latent code z subject to standard normal distribution.

Future work

- (1) the meaning of surprise: chord, melody?
- (2) the surprise contours, relationship between different surprise contours?
- (3) sample the surprise contours that generate melodious music?
- (4) surprisingness style classification.

2 CONTROLLABLE DEEP MELODY GENERATION VIA HIERARCHICAL MUSIC STRUCTURE REPRESENTA-TION

Problems

- (1) model larger scale music structure and multiple levels of repetition
- (2) controllability to create desired tempo, styles, and mood
- (3) scarcity of training data
- 1. MusicFrameworks
- (1) high-level representation: repeated sections and phrases
- (2) low-level representation: rhythm structure and melodic contour
- 2. Generate melodies from music frameworks
- 3. Achieve controllability by editing the music frameworks at any level

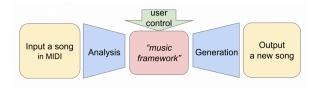


Figure 3: Architecture of MusicFrameworks.

Advantages of Music Frameworks A new song can be generated using the structure from song A, basic melody from song B, and basic rhythm form from song C.

Music Frameworks Analysis

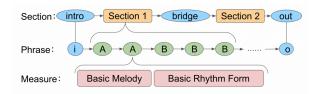


Figure 4: An example music framework.

- (1) section and phrase-level structure analysis results
- (2) basic melody and basic rhythm form within each phrase

Basic Melody: a sequence of half notes representing the most common pitch in each 2-beat segment of the original phrase

Basic rhythm form: consists of a per-measure descriptor with two components (a. pattern label; b. rhythmic complexity)

Generation Using Music Frameworks

- 1. user or the library provides the section and phrase structure.
- 2. generate a basic melody, generate rhythm using the basic rhythm form.
- 3. generate a new melody given the basic melody.

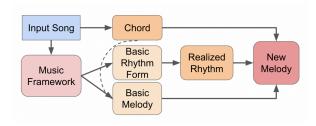


Figure 5: Generation process from music frameworks within each phrase.

Basic Melody Generation

input:
$$x_i = (pos_i, c_i, ...)$$

sampling with Dynamic Time Warping: melody contour rating function to estimate the contour similarity between two basic melodies.

Realized Rhythm Generation

rhythm pattern: r_i (256 possible rhythm patterns with a 2-beat length)

input: $x_i = (r_{i-1}, brf_i, pos_i)$

beam search

Realized Melody Generation

input: x_i includes the current note's duration, the basic melody pitch, the current chord (c_i) , three positional features (pos_i) .

Network Architecture

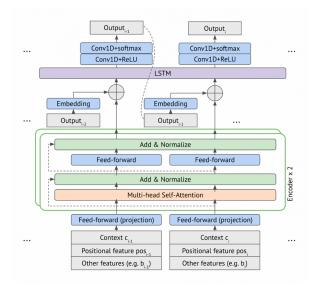


Figure 6: Transformer-LSTM architecture for melody, basic melody and rhythmic pattern generation.

Conclusion

- (1) The key idea: adopt an abstract representation music frameworks (long term repetitive structures; phrase-level basic melodies and basic rhythm forms)
- (2) Controllability: manipulation of music frameworks, can be edited and combined to guide compositions.

3 MINGUS: MELODIC IMPROVISATION NEURAL GEN-ERATOR USING SEQ2SEQ

MINGUS relies on two dedicated embedding models (pitch and duration) and exploits in prediction features such as chords (current and following), bass line, position inside the measure.

Data representation

P: [0-128], D: [0-12], C (Chord): $[0-128\times 4]$, NC (Next Chord): $[0-128\times 4]$, B: [0-128], BE: [0-3], O: [0-95]

Model Architecture

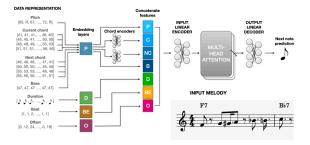


Figure 7: MINGUS model architecture and data representation.

MINGUS: two parallel transformer models with the same structure, predicting pitch and duration, composed by the sole encoder module with a forward mask and a pad mask.

The structure is the same for pitch and duration models.

Pitch model: D, C, B, BE, O Duration model: B, BE, O

4 SINTRA: LEARNING AN INSPIRATION MODEL FROM A SINGLE MULTI-TRACK MUSIC SEGMENT

SinTra

- (1) A novel pitch-group representation
- (2) A novel inspiration model
- (3) Three modules based on Transformer-XL are designed for each stage of multi-scale training to process multi-track music.

Data Representation

Pitch-group representation

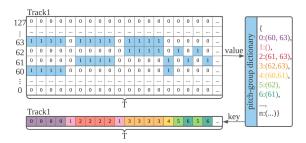


Figure 8: Illustration of the pitch-group representation.

Pyramid of Transformer-XL Model

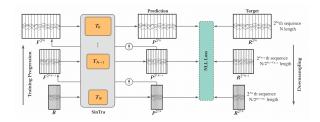


Figure 9: SinTra's multi-scale pipeline.

R: real music $\{R^{2^{x_0}},...,R^{2^{x_N}}\}$, $R^{2^{x_n}}$ is a downsampled version of $R^{2^{x_0}}$.

P: produced music P^{2^n} is the corresponding scale of $R^{2^{x_n}}$

The generation of a music sample starts at the coarsest scale and sequentially passes through all models up to the finest scale.

Network Structure

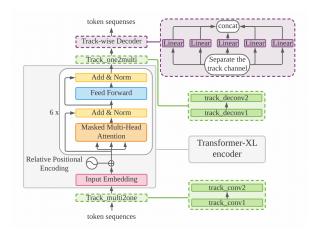


Figure 10: Network structure of each scale T_n .

Training Process

$$egin{aligned} 1st: P_{t+1}^4 &= Model_{1^{st}}(R_t^4) \ loss_{1^{st}} &= NLL(P_{t+1}^4, R_{t+1}^4) \ 2nd: F_{t+1}^8 &= Upsample(P_{t+1}^4) \ P_{t+1}^8 &= Model_{2^{nd}}(F_{t+1}^8) \ loss_{2^{nd}} &= NLL(P_{t+1}^8, R_{t+1}^8) \ 3rd: F_{t+1}^{16} &= Upsample(P_{t+1}^8) \ P_{t+1}^{16} &= Model_{3^{rd}}(F_{t+1}^{16}) \ loss_{3^{rd}} &= NLL(P_{t+1}^{16}, R_{t+1}^{16}) \end{aligned}$$

Figure 11: Training process.

5 COLLAGENET: FUSING ARBITRARY MELODY AND ACCOMPANIMENT INTO A COHERENT SONG

CollageNet: given a piece of melody and an irrelevant accompanient with the same length, fuse them into harmonic two-track music. Rhythm and pitch of the melody, and chord and texture of the accompaniment.

Model Architecture

We use the encoders of two VAEs to encode the melody and accompaniment to two latent vectors. melody: z_{mel} , $z_{mel} = z_p \oplus z_r^2$ (z_p : pitch vector, z_r : rhythm vector), EC²-VAE.

polyphonic accompaniment: z_{acc} , $z_{acc} = z_c \oplus z_t$ (z_c : chord vector, z_t : texture vector).

Feed z_{mel} and z_{acc} to the actor model G with transforms them into another latent vector pair \hat{z}_{mel} and \hat{z}_{acc} . The actor model G is trained under an adversarial framework together with a critic model D.

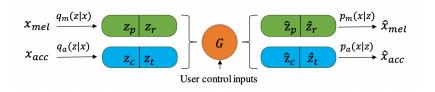


Figure 12: Inference diagram.

Training

- 1. pre-train the two VAEs for melodies and accompaniments.
- 2. train the G and D models in the latent space.

harmonic set (Ω_h) : $\{x_{mel}^{(i)}, x_{acc}^{(i)}\}_{i=1}^N$ disharmonic set (Ω_{dh}) : $\{x_{mel}^{(i)}, x_{acc}^{(j)}\}$ $(i \neq j)$

D model is trained to distinguish between positivie samples and negative samples.

positive samples

 $\{z_{mel}, z_{acc}\} \sim \Omega_h^z$ negative samples

- (1) latent vectors of the disharmonic pairs $\{z_{mel}, z_{acc}\} \sim \Omega_{dh}^z$
- (2) latent vectors samples from prior $\{z_{mel}, z_{acc}\} \sim p(z)$
- (3) latent vectors produced by the actor model $G(z_{mel}, z_{acc})$

G model is trained to transform disharmonic latent vector pairs into a more harmonic pair.

User Control

The input of the G model is extended with four scalars c_{mp} , c_{mr} , c_{ac} , $c_{at} \in [0, 1]$.

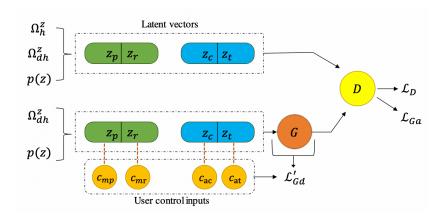


Figure 13: Training diagram.

6 DEEP MUSIC ANALOGY VIA LATENT REPRESENTATION DISENTANGLEMENT

Explicitly-constrained conditional variational autoencoder (EC²-VAE)

- (1) the disentanglement is explicitly coded.
- (2) the disentanglement does not sacrifice much of the reconstruction.
- (3) model is capable of making analogies in the inference phrase.

Analogy Algorithm

Any computational method capable of producing analogous versions of existing examples. strict analogy algorithm: requires not only learning the representations but also disentangling them. music representation learning: build a bi-directional mapping between the distributions of observation x and latent representation z.

Model Architecture

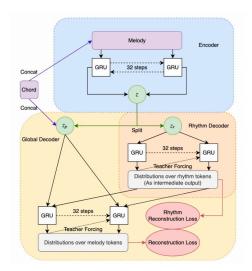


Figure 14: EC²-VAE model.

To disentangle the latent rhythm representation z_r from the overall z, encourage the intermediate

output of z_r to match the rhythm feature of the melody. z_p represents the other part of z, which is everything but rhythm.

SYMBOLIC MUSIC GENERATION WITH DIFFUSION **MODELS**

Denoising Diffusion Probabilistic Models (DDPMs)

forward process, reverse process.

Forward process:
$$q(x_t|x_{t-1}) = \mathcal{N}(x_t; \sqrt{1-\beta_t}x_{t-1}, \beta_t I)$$

 $\Leftrightarrow x_t = \sqrt{1-\beta_t}x_{t-1} + \beta_t Z_{t-1} (Z_t \sim \mathcal{N}(0, I))$
 $q(x_{1:N}|x_0) = \prod_{t=1}^N q(x_t|x_{t-1})$
Reverse process: $p_{\theta}(x_{t-1}; \mu_{\theta}(x_t, t), \sigma_{\theta}(x_t, t))$
 $p_{\theta}(x_{0:N}) = p(x_N) \prod_{t=1}^N p_{\theta}(x_{t-1}|x_t)$

$$q(x_{1:N}|x_0) = \prod_{t=1}^{N} q(x_t|x_{t-1})$$

$$p_{\theta}(x_{0:N}) = p(x_N) \prod_{t=1}^{N} p_{\theta}(x_{t-1}|x_t)$$

Loss function

The training objective: $\max_{\theta} p_{\theta}(x_0) = \int p_{\theta}(x_0, ..., x_N) dx_{1:N}$

$$L(\theta) = \mathbb{E}_{x_0, \epsilon, t} [|\epsilon - \epsilon_{\theta} (\sqrt{1 - \beta_t} x_0 + \sqrt{\beta} \epsilon, t)|^2]$$

Model

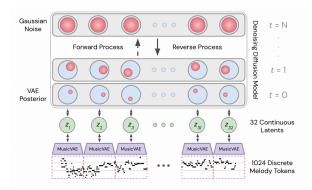


Figure 15: A diagram of the diffusion model.

MusicVAE embeddings

Transformer diffusion model: the network for $\epsilon_{\theta}(x_t, \sqrt{1-\beta_t})$ is a transformer.

Sampling

Algorithm 1 Sampling

$$\begin{split} x_T &\sim \mathbb{N}(0,I) \\ \textbf{for } t &= N,...,1 \ \textbf{do} \\ &\quad \epsilon \sim \mathbb{N}(0,I) \ \text{if } t > 1, \ \text{else } \epsilon = 0 \\ &\quad x_{t-1} = \frac{1}{\sqrt{1-\beta_t}} (x_t - \frac{\beta}{\sqrt{\beta}} \epsilon_\theta(x_t,t)) + \sigma_t z \\ \textbf{end for} \\ &\quad \textbf{return} \quad x_0 \end{split}$$

Infilling

s represents the partially occluded sample.

Algorithm 2 Infilling

```
Input: mask m, sample s, N steps, \beta_1, ..., \beta_N
x_N \sim \mathbb{N}(0, I)
for t = N, ..., 1 do
\epsilon_1, \epsilon_2 \sim \mathbb{N}(0, I) \text{ if } t > 1, \text{ else } \epsilon_1 = \epsilon_2 = 0
y = \sqrt{1 - \beta_t} + \sqrt{\beta_t} \epsilon_1, \text{ if } t > 1, \text{ else } s
x_{t-1} = \frac{1}{\sqrt{1-\beta_t}} (x_t - \frac{\beta}{\sqrt{\beta}} \epsilon_z \theta(x_t, t)) + \sigma_t \epsilon_2
x_{t-1} = x_{t-1} \odot (1 - m) + y \odot m
end for
return x_0
```