

# Supplementary for “V2C: Visual Voice Cloning”

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We organise the supplementary materials as follows.

- In Section A, we provide a detailed analysis of the proposed V2C-Animation dataset, *e.g.*, word cloud, distribution of utterance length, *etc.*
- In Section B, we pose more visualised pitch tricks of the samples from the V2C-Animation dataset and the related datasets.
- In Section C, we exhibit more visual results of the mel-spectrogram derived from our V2C-Net with comparisons against baseline and ground truth.
- In Section D, we report the implementation details.
- In Section E, we depict more details of the vocoder.
- In Section F, we analyse the limits of MCD-DTW.

## A. More Analysis of V2C-Animation Dataset

### A.1. Word Cloud and Count

In Figure 1, we visualise the texts/subtitles of our V2C-Animation dataset as Venn-style word cloud [2], where the size of each word refers to the harmonic mean of its count.



Figure 1. Word cloud of the texts on our V2C-Animation dataset.

Besides, we also provide the top 30 words on our V2C-Animation dataset along with their counts in Figure 2. More

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results (top 100) are in the following:

(‘know’, 437), (‘oh’, 305), (‘right’, 255), (‘one’, 254), (‘now’, 250), (‘well’, 250), (‘go’, 233), (‘okay’, 217), (‘come’, 210), (‘want’, 201), (‘look’, 196), (‘got’, 181), (‘going’, 173), (‘think’, 167), (‘will’, 165), (‘thing’, 163), (‘gonna’, 163), (‘need’, 159), (‘see’, 155), (‘back’, 153), (‘never’, 151), (‘us’, 147), (‘time’, 141), (‘say’, 139), (‘hey’, 138), (‘mean’, 137), (‘let’, 137), (‘good’, 135), (‘yeah’, 131), (‘guy’, 128), (‘really’, 124), (‘make’, 124), (‘thank’, 124), (‘little’, 112), (‘way’, 108), (‘love’, 108), (‘ye’, 108), (‘find’, 104), (‘help’, 97), (‘tell’, 96), (‘wait’, 95), (‘take’, 93), (‘kid’, 92), (‘please’, 91), (‘sorry’, 88), (‘something’, 87), (‘great’, 87), (‘dad’, 87), (‘friend’, 84), (‘day’, 82), (‘game’, 80), (‘stop’, 75), (‘even’, 75), (‘Uh’, 74), (‘big’, 67), (‘work’, 66), (‘Ralph’, 66), (‘much’, 62), (‘give’, 62), (‘first’, 61), (‘everything’, 60), (‘new’, 59), (‘still’, 58), (‘life’, 58), (‘keep’, 58), (‘dragon’, 58), (‘family’, 57), (‘sure’, 56), (‘made’, 56), (‘talk’, 55), (‘world’, 53), (‘place’, 53), (‘heart’, 53), (‘every’, 53), (‘maybe’, 53), (‘stay’, 52), (‘wanna’, 51), (‘better’, 51), (‘people’, 50), (‘huh’, 50), (‘anything’, 50), (‘getting’, 49), (‘thought’, 48), (‘man’, 48), (‘mom’, 48), (‘listen’, 48), (‘guess’, 47), (‘fine’, 47), (‘around’, 47), (‘gotta’, 46), (‘believe’, 46), (‘two’, 45), (‘someone’, 45), (‘home’, 45), (‘call’, 45), (‘boy’, 45), (‘son’, 44), (‘put’, 43), (‘fix’, 43), (‘always’, 43)

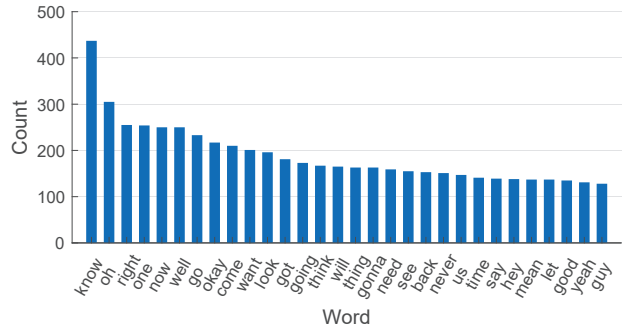


Figure 2. Top 30 words on V2C-Animation along with the counts.

## A.2. Distribution of Emotion Labels

Following the categories of FER-2013 [4] (a dataset for human facial expression recognition), we divide the collected video/audio clips into 8 types (*i.e.*, 0: angry, 1: disgust, 2: fear, 3: happy, 4: neutral, 5: sad, 6: surprise, and 7: others). The number and distribution of each emotion label can be found in Table 1 and Figure 3, respectively.

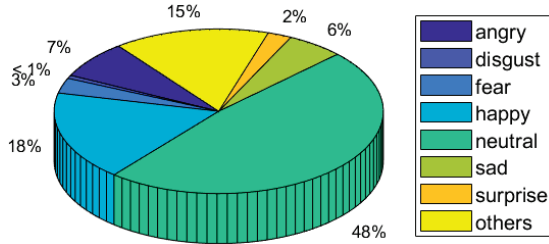


Figure 3. Distribution of emotion labels on V2C-Animation.

Emotion	angry	disgust	fear	happy
Count	756	64	305	1799
Emotion	neutral	sad	surprise	others
Count	4919	572	240	1562

Table 1. Counts of the emotion labels on V2V-Animation dataset.

## A.3. Distribution of Utterance Length

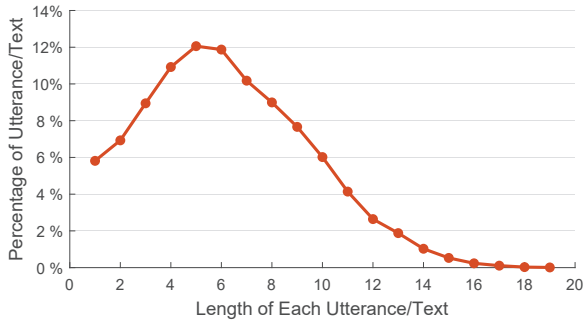


Figure 4. Distribution of utterance/text length.

Figure 4 exhibits the distribution of utterance/text length on V2C-Animation dataset, which shows that most utterance range from 3 to 8 words. Besides, we also list the number of utterance/text and their corresponding percentages in the following (format: *length, count, percentage*): (1, 594, 5.81%), (2, 708, 6.93%), (3, 914, 8.95%), (4, 1116, 10.92%), (5, 1232, 12.06%), (6, 1213, 11.87%), (7, 1040, 10.18%), (8, 919, 8.99%), (9, 783, 7.66%), (10, 615, 6.02%), (11, 423, 4.14%), (12, 270, 2.64%), (13, 192, 1.88%), (14, 105, 1.03%), (15, 54, 0.53%), (16, 24, 0.23%), (17, 11, 0.11%), (18, 3, 0.03%), (19, 1, 0.01%)

## A.4. More Examples of Subtitle and Video Clip

We show several examples of how to crop movies based on a corresponding subtitle file. Here, we use an SRT type subtitle file. Besides the subtitles/texts, the SRT file also contains starting and ending time-stamps to ensure the subtitles match with video and audio. The sequential number of subtitle (*e.g.*, No. 726 and No. 1340 in Figure 5) indicates the index of each video clip. Based on the SRT file, we cut movie into a series of video clips via FFmpeg toolkit [11] (an automatic audio and video processing toolkit).



Figure 5. Examples of how to cut a movie into a series of video clips according to subtitle files. Note that the subtitle files contain both starting and ending time-stamps for each video clip.

## A.5. Samples of Character’s Emotion

Figure 6 shows some samples of the reference videos on V2C-Animation dataset with their corresponding emotions.



Figure 6. Samples of the character’s emotion (*e.g.*, happy and sad) involved in the reference video. Here, we take *Elsa* (a character in movie *Frozen*) as an example.

## A.6. List of Animated Movies and Characters

As shown in Figure 7, we report all the names of our collected animated movies with their corresponding characters/speakers on the V2C-Animation dataset.

## B. V2C-Animation vs. Related Datasets

To compare the differences between the collected V2C-Animation dataset and several related datasets (*i.e.*, LJ Speech, LibriSpeech and LibriTTS), we visualise the pitch tricks of the samples from our dataset and others. Due to the varying lengths of audios, for a fair comparison, we cut two seconds of audio from each sample. As shown in Figure 8, the audio pitches from the existing datasets are more smooth and their ranges of frequency (Hz) are narrower than ours.

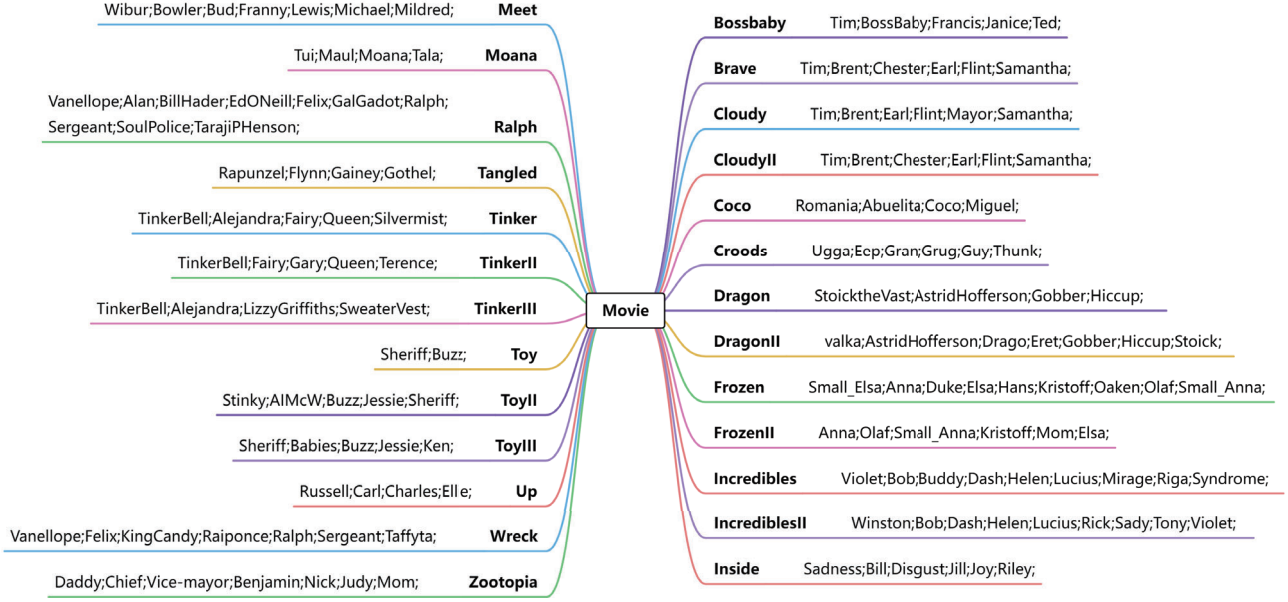


Figure 7. Movies with the corresponding speakers/characters on the V2C-Animation dataset.

### C. More Visual Results of Mel-spectrogram

We provide more visualised results of our V2C-Net with comparisons against the baseline method and ground truth. As shown in Figure 9, the mel-spectrograms generated by the proposed V2C-Net are more similar to the ground-truth ones. Note that the baseline method FastSpeech2 does not take the reference videos (*i.e.*, emotions) as inputs, which may lead to some misses of the prosody involving in the videos. The results further demonstrate the effect of the reference video when generating speech with rich emotions. Besides, the ranges of pitch for the mel-spectrograms are various due to the different emotions. For example, the pitch of the mel-spectrogram would be more drastic with the emotions “happy” or “sad”, while it would be more smooth if the emotion is “neutral”.

### D. Implementation Details

For the speaker encoder  $f_{spk}$ , we use the same architecture as [12], comprising three LSTM layers. The audio encoder maps a sequence of mel-spectrogram frames, derived from the reference audio, to a vector with a fixed dimension of 256. We optimise the model with a generalised end-to-end speaker verification loss, which ensure features from the same speaker are more similar than ones from different speakers. For the emotion encoder  $f_{emo}$ , we use a conventional I3D model [1], trained on our V2C-Animation dataset [1] with  $64 \times 10^3$  iterations and final output a vector with 1024 dimensions. For our synthesizer, we train the text encoder  $f_{txt}$  and the synthesizer in an end-to-end manner

with 16 batch size and  $2 \times 10^6$  iterations on our proposed V2C-Animation dataset. We train all models on a single GPU device (GeForce RTX 3090).

### E. Details of Vocoder

To synthesise the waveform of the speech from our generated mel-spectrogram, we use HiFi-GAN [7] as our vocoder. The HiFi-GAN model is based on Generative Adversarial Networks (GANs) [3], which consists of one generator and two discriminators, *i.e.*, a multi-period discriminator (MPD) and a multi-scale discriminator (MSD).

The generator of HiFi-GAN can be divided into two major modules: a transposed convolution (ConvTranspose) network and a multi-receptive field fusion (MRF) module. Specifically, we first upsample the mel-spectrogram by ConvTranspose, which seeks to take an alignment between the length of the output features and the temporal resolution of raw waveforms. Then, we feed the upsampled features into MRF module, which consists of multiple residual blocks [5], and take the sum of outputs from these blocks as our predicted waveform. Here, the residual blocks with different kernel sizes and dilation rates are used to ensure different receptive field.

For the two discriminators, the multi-period discriminator (MPD) contains several sub-discriminators, where each sub-discriminator handles a specific periodic part of the input audio. By contrast, the multi-scale discriminator (MSD) proposed in MelGAN [8], consisting of three sub-discriminators, tries to capture the consecutive patterns and long-term dependencies from input audio.

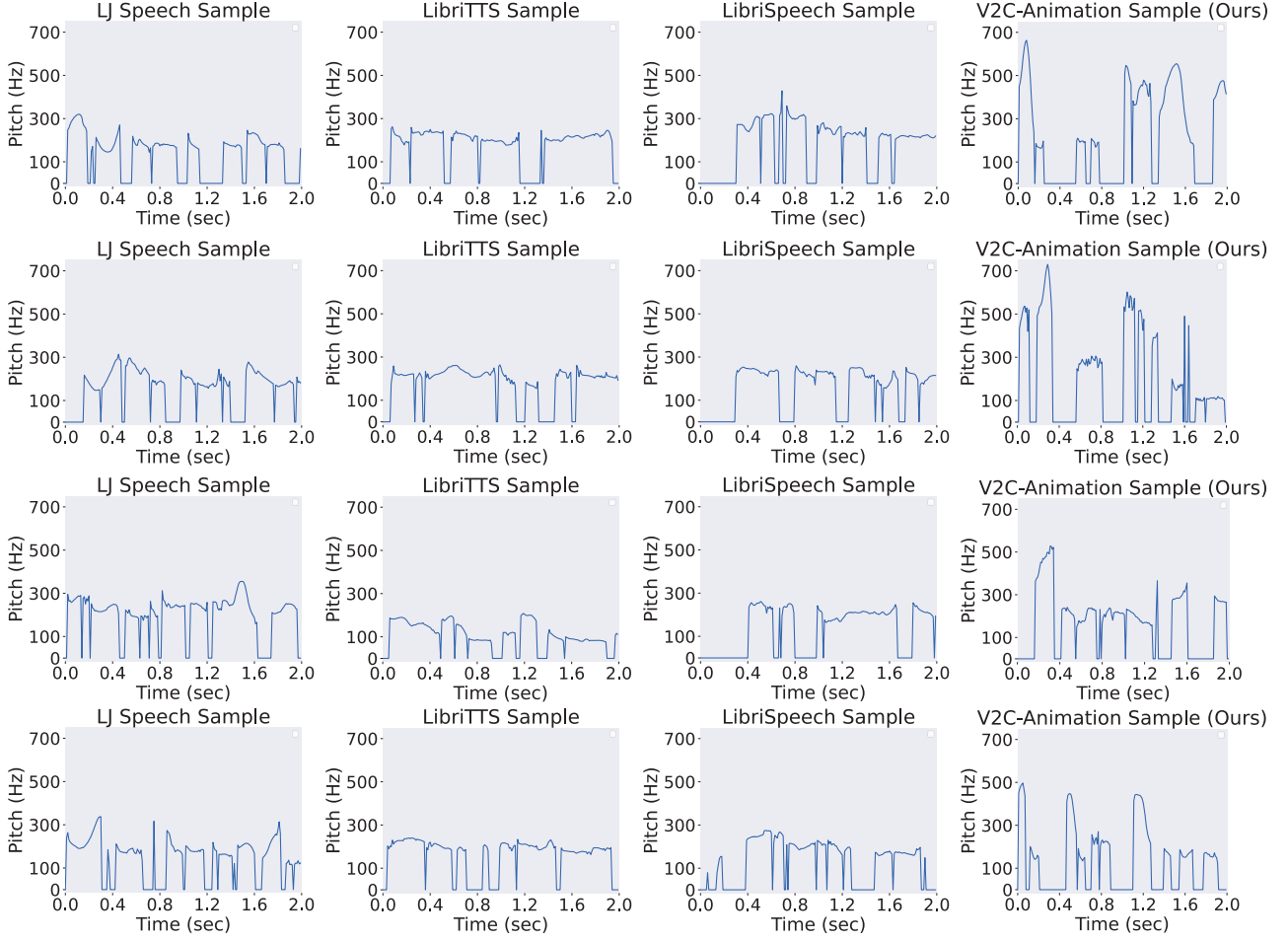


Figure 8. Visual comparison between our V2C-Animation dataset and the related datasets (*i.e.*, LJ Speech, LibriSpeech and LibriTTS). A pitch of 0 Hz refers to an unvoiced segment.

The generator and discriminators are trained adversarially, aiming to improve the training stability and the model performance. Specifically, the vocoder (*i.e.*, HiFi-GAN) is optimised via the objective function that contains an LSGAN-based loss [9], a mel-spectrogram loss [6], and a feature matching loss [8]. In practice, we use the vocoder (*i.e.*, HiFi-GAN) pretrained on the LibriSpeech dataset [10].

## F. Limitations of MCD-DTW

As mentioned in the main paper, “*MCD-DTW would achieve a better value as long as there is a match between two speeches. This is not reasonable as a better generated speech should have a similar length with the ground truth*”. To verify that, we take an example of the generated speech with its ground truth (GT) in Figure 10. The corresponding MCD-DTW and MCD-DTW-SL (ours) are 8.69 and 11.76, respectively. Smaller value of MCD-DTW demonstrates that it focuses more on the alignment within

pitch and energy while ignores whether the duration is reasonable or not. Our metric (MCD-DTW-SL) can reflect the alignment on both feature and length.

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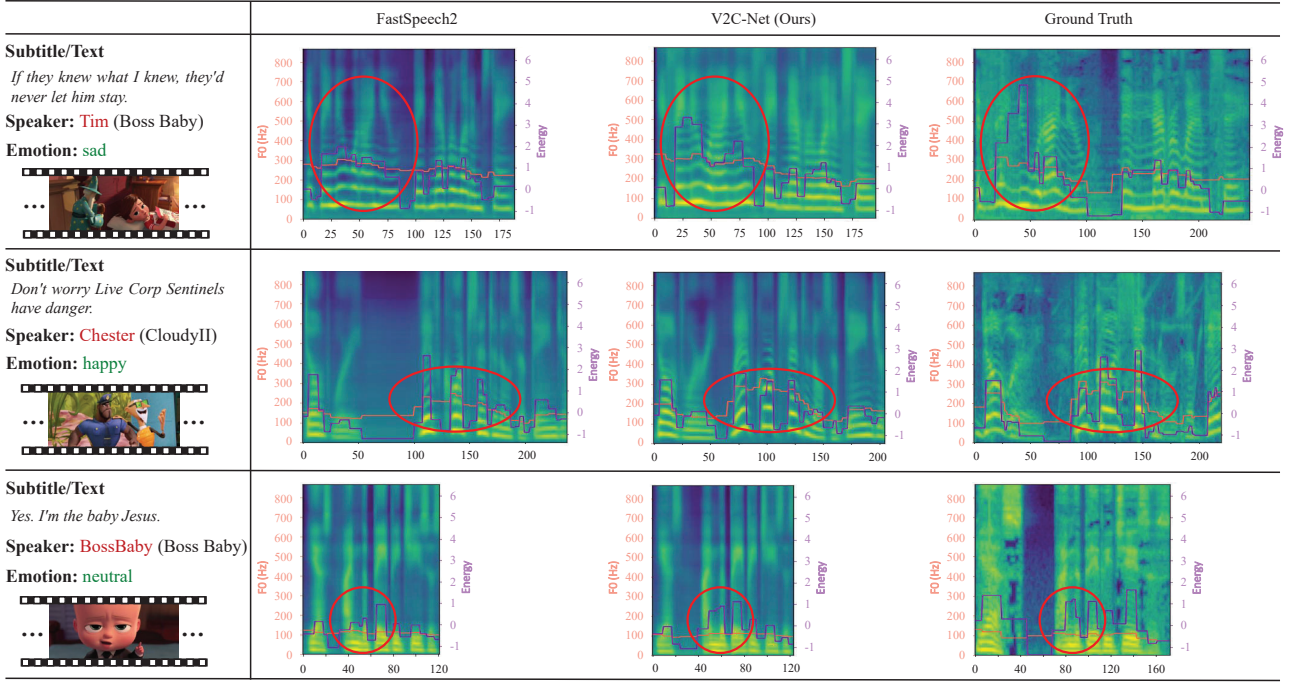


Figure 9. More visualised mel-spectrograms of generated and ground-truth audios. The orange curves are  $F_0$  contours, where  $F_0$  denotes the fundamental frequency of audio. The purple curves refer to energy (volume) of audio. Horizontal axis is the duration of audio. We highlight the main difference via red circle.

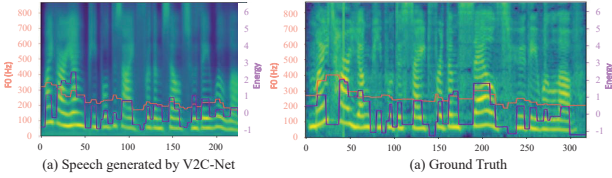


Figure 10. Similar shape of pitch and energy curves with different length between the synthesised speech and the ground-truth one.

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