# "Speech resynthesis from discrete disentangled self-supervised representations" online resource

[1] A. Polyak et al., "Speech resynthesis from discrete disentangled self-supervised representations," arXiv preprint arXiv:2104.00355, Jul. 2021.

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## Outline

- Action item
  - 找 disentangled 相關研究:使用 [1] 所提供的 open online resource
- Status report
  - □ 實作步驟分為以下四點:
    - 1. Environment setup

此專案預設使用 GPU 執行,目前執行方案二

- 方案一:使用 AC922 (GPU) 執行,但在 PPC64el 架構下,Anaconda 不提供作者要求所使用套件 torch=1.8.0 的版本,需自行從 <u>source code</u> 建構 (附錄 p.4)
- 方案二:將程式中所使用到 GPU 的指令更改為 CPU 執行,並架設 Google Cloud Platform (GCP) 中免費版的 Virtual Machine (VM) (附錄 p.5)

#### 2. Dataset preprocess

● 作者提供的連結為 flac 檔,需自行將 flac 轉換為 wav 檔,才能繼續 audio file downsampling

#### 3. Network training

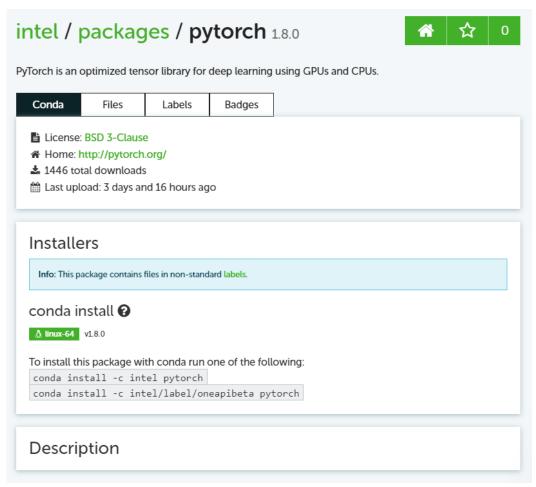
- A. F0 Quantizer Model (訓練中)
- B. Resynthesis Model,需完成 A 才能繼續訓練

#### 4. Generate Audio file

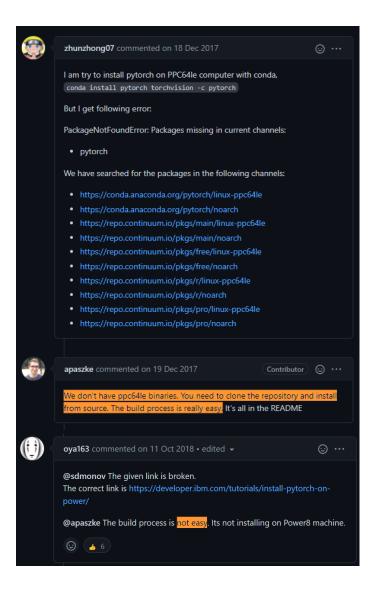
- □ 後續規劃
  - 1. (使用 VM) 等待 F0 Quantizer Model 訓練完接著訓練 Resynthesis Model
  - 2. (使用本機) 將目前 F0 Quantizer Model 訓練到一半的 checkpoint 拿去訓練 Resynthesis Model,之後與 VM 完整訓練完結果比較
  - 3. 再次嘗試使用 AC922 進行實作 (在 Environment setup 中,從 source code 建構所需環境)

## 附錄

## Pytorch don't have ppc64le binaries



Pytorch :: Anaconda.org



## Using GCP's VM

```
Epoch: 37
Steps : 25020, Gen Loss Total : 0.054, s/b : 0.648
Steps : 25025, Gen Loss Total : 0.061, s/b : 0.489
Steps : 25030, Gen Loss Total : 0.063, s/b : 0.512
Steps : 25035, Gen Loss Total : 0.062, s/b : 0.429
Steps: 25040, Gen Loss Total: 0.036, s/b: 0.473
Steps : 25045, Gen Loss Total : 0.040, s/b : 0.462
Steps : 25050, Gen Loss Total : 0.030, s/b : 0.495
Steps : 25055, Gen Loss Total : 0.049, s/b : 0.470
Steps : 25060, Gen Loss Total : 0.048, s/b : 0.471
Steps : 25065, Gen Loss Total : 0.044, s/b : 0.497
Steps: 25070, Gen Loss Total: 0.054, s/b: 0.464
Steps: 25075, Gen Loss Total: 0.047, s/b: 0.528
Steps: 25080, Gen Loss Total: 0.055, s/b: 0.484
Steps: 25085, Gen Loss Total: 0.032, s/b: 0.470
Steps : 25090, Gen Loss Total : 0.044, s/b : 0.470
Steps : 25095, Gen Loss Total : 0.040, s/b : 0.476
Steps : 25100, Gen Loss Total : 0.044, s/b : 0.451
Steps : 25105, Gen Loss Total : 0.057, s/b : 0.443
Steps : 25110, Gen Loss Total : 0.039, s/b : 0.449
Steps : 25115, Gen Loss Total : 0.061, s/b : 0.481
Steps : 25120, Gen Loss Total : 0.039, s/b : 0.415
Steps : 25125, Gen Loss Total : 0.042, s/b : 0.417
Steps : 25130, Gen Loss Total : 0.029, s/b : 0.486
Steps : 25135, Gen Loss Total : 0.038, s/b : 0.398
/home/fq6ts15test/.local/lib/python3.8/site-packages/scipy/signal/signaltools.p
v:1531: UserWarning: kernel size exceeds volume extent: the volume will be zero
-padded.
 warnings.warn('kernel size exceeds volume extent: the volume will be '
Steps : 25140, Gen Loss Total : 0.038, s/b : 0.489
Steps : 25145, Gen Loss Total : 0.034, s/b : 0.428
Steps : 25150, Gen Loss Total : 0.034, s/b : 0.415
Steps: 25155, Gen Loss Total: 0.036, s/b: 0.507
Steps : 25160, Gen Loss Total : 0.038, s/b : 0.389
Steps: 25165, Gen Loss Total: 0.033, s/b: 0.460
Steps : 25170, Gen Loss Total : 0.046, s/b : 0.510
Steps : 25175, Gen Loss Total : 0.049, s/b : 0.435
                    2:bash 3:bash 4:unzip LJSpeech- 5:trai
```

目前執行到 Steps: 25175 最終 Steps: 40000

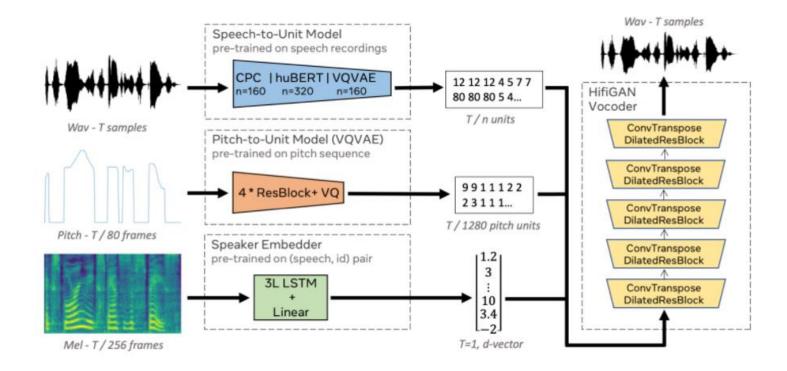
訓練完 F0 Quantizer Model 會在下方資料夾 (lj\_f0\_vq) 看到 g 00040000 的 checkpoint 檔

```
fg6ts15test@ubuntu-focal-2:/mnt/disks/nvmessd/speech-resynthesis/data$ c d .. fg6ts15test@ubuntu-focal-2:/mnt/disks/nvmessd/speech-resynthesis$ cd c checkpoints/ configs/ fg6ts15test@ubuntu-focal-2:/mnt/disks/nvmessd/speech-resynthesis$ cd c checkpoints/ configs/ fg6ts15test@ubuntu-focal-2:/mnt/disks/nvmessd/speech-resynthesis$ cd che ckpoints fg6ts15test@ubuntu-focal-2:/mnt/disks/nvmessd/speech-resynthesis/checkpo ints$ ls lj_f0_vq lj_vqvae fg6ts15test@ubuntu-focal-2:/mnt/disks/nvmessd/speech-resynthesis/checkpo ints$ [] fg6ts15test@ubuntu-focal-2:/mnt/disks/nvmessd/speech-resynthesis/checkpo ints$ [] fg6ts15test@ubuntu-focal-2:/mnt/disks/nvmessd/speech-resynthesis/checkpo ints/li_f0_vq$ le config.json g_00010000 g_00020000 logs
```

並使用 g\_00040000 checkpoint 繼續訓練第二個 Resynthesis Model

## Paper Demo Link

Speech Resynthesis from Discrete Disentangled Self-Supervised Representations



HiFi-GAN: High-Fidelity Denoising and Dereverberation Based on Speech Deep Features in Adversarial Networks