Jukebox Empty

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[]: !nvidia-smi
[]: !python --version
[]: |pip install --upgrade git+https://github.com/craftmine1000/jukebox-saveopt.git
[]: import jukebox
     import torch as t
     import librosa
     import os
     from IPython.display import Audio
     from jukebox.make_models import make_vqvae, make_prior, MODELS, make_model
     from jukebox.hparams import Hyperparams, setup_hparams
     from jukebox.sample import sample_single_window, _sample, \
                                sample_partial_window, upsample, \
                                load prompts
     from jukebox.utils.dist_utils import setup_dist_from_mpi
     from jukebox.utils.torch_utils import empty_cache
     try:
         if device is not None:
            pass
     except NameError:
         rank, local_rank, device = setup_dist_from_mpi()
[]: model = "5b_lyrics" # or "1b_lyrics"
     hps = Hyperparams()
     hps.sr = 44100
     hps.n_samples = 3 if model=='5b_lyrics' else 8
     hps.name = 'Jukebox Samples Mozart noLyrics'
     chunk_size = 16 if model=="5b_lyrics" else 32
     max_batch_size = 3 if model=="5b_lyrics" else 16
     hps.levels = 3
     hps.hop_fraction = [.5, .5, .125]
     vqvae, *priors = MODELS[model]
     vqvae = make_vqvae(setup_hparams(vqvae, dict(sample_length = 1048576)), device)
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top_prior = make_prior(setup_hparams(priors[-1], dict()), vqvae, device)
[]: sample_length_in_seconds = 30
                                            # Full length of musical sample to_
      →generate - we find songs in the 1 to 4 minute
                                            # range work well, with generation time_
      ⇔proportional to sample length.
                                            # This total length affects how quickly_
      → the model
                                            # progresses through lyrics (model also⊔
      \rightarrow generates differently
                                             # depending on if it thinks it's in the
      →beginning, middle, or end of sample)
     hps.sample_length = (int(sample_length_in_seconds*hps.sr)//top_prior.
      →raw_to_tokens)*top_prior.raw_to_tokens
     assert hps.sample_length >= top_prior.n_ctx*top_prior.raw_to_tokens, f'Please_
      ⇒choose a larger sampling rate'
[]: metas = [dict(artist = "wolfgang amadeus mozart",
                 genre = "classical",
                 total_length = hps.sample_length,
                 offset = 0,
                 lvrics = """
                 0.00
                 ),
               ] * hps.n_samples
     labels = [None, None, top_prior.labeller.get_batch_labels(metas, 'cuda')]
[]: sampling_temperature = .98
     lower_batch_size = 16
     max_batch_size = 3 if model == "5b_lyrics" else 16
     lower_level_chunk_size = 32
     chunk_size = 16 if model == "5b_lyrics" else 32
     sampling_kwargs = [dict(temp=.99, fp16=True, max_batch_size=lower_batch_size,
                             chunk size=lower level chunk size),
                         dict(temp=0.99, fp16=True, max_batch_size=lower_batch_size,
                              chunk_size=lower_level_chunk_size),
                         dict(temp=sampling_temperature, fp16=True,
                              max_batch_size=max_batch_size, chunk_size=chunk_size)]
[]: zs = [t.zeros(hps.n_samples,0,dtype=t.long, device='cuda') for _ in_
     →range(len(priors))]
     zs = _sample(zs, labels, sampling_kwargs, [None, None, top_prior], [2], hps)
[]: Audio(f'{hps.name}/level_2/item_0.wav')
```