wav2vec

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base_100h.yaml

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一、fairseq安装

```
Traceback (most recent call last):
 1
2
      File "/home/zhuozhu.zz/tops/fairseq/bin/fairseq-hydra-train", line 33, i
    n <module>
        sys.exit(load_entry_point('fairseq', 'console_scripts', 'fairseq-hydra
3
    -train')())
      File "/home/zhuozhu.zz/tops/fairseq/bin/fairseq-hydra-train", line 25, i
4
    n importlib load entry point
         return next(matches).load()
5
      File "/home/zhuozhu.zz/.local/lib/python3.7/site-packages/importlib_meta
6
    data/__init__.py", line 166, in load
7
        module = import_module(match.group('module'))
8
      File "/usr/local/lib/python3.7/importlib/__init__.py", line 127, in impo
    rt module
9
         return bootstrap. gcd import(name[level:], package, level)
10
      File "<frozen importlib._bootstrap>", line 1006, in _gcd_import
      File "<frozen importlib._bootstrap>", line 983, in _find_and_load
11
      File "<frozen importlib._bootstrap>", line 967, in _find_and_load_unlock
12
13
      File "<frozen importlib._bootstrap>", line 677, in _load_unlocked
      File "<frozen importlib._bootstrap_external>", line 728, in exec_module
14
15
      File "<frozen importlib._bootstrap>", line 219, in _call_with_frames_rem
16
      File "/home/zhuozhu.zz/fairseq/fairseq_cli/hydra_train.py", line 15, in
    <module>
         from fairseg import distributed utils, metrics
17
      File "/home/zhuozhu.zz/fairseq/fairseq/__init__.py", line 21, in <module
18
    >
         from fairseq.logging import meters, metrics, progress_bar # noqa
19
      File "/home/zhuozhu.zz/fairseq/fairseq/logging/progress bar.py", line 31
20
    5, in <module>
21
         from torch.utils.tensorboard import SummaryWriter
22
      File "/usr/local/lib/python3.7/site-packages/torch/utils/tensorboard/ i
    nit__.py", line 4, in <module>
        LooseVersion = distutils.version.LooseVersion
23
24
    AttributeError: module 'distutils' has no attribute 'version'
25
26
    解决方案:
27
28
    $vim /usr/local/lib/python3.7/site-packages/torch/utils/tensorboard/ init
    ___. py
29
30
    import tensorboard
    from packaging.version import Version as LooseVersion
31
32
33
```

```
if not hasattr(tensorboard, '__version__') or LooseVersion(tensorboard.__v
34
    ersion__) < LooseVersion('1.15'):</pre>
         raise ImportError('TensorBoard logging requires TensorBoard version 1.
35
    15 or above')
36
37
        del LooseVersion
38
        del tensorboard
39
40
        from .writer import FileWriter, SummaryWriter # noga: F401
        from tensorboard.summary.writer.record_writer import RecordWriter # n
    oga: F401
```

```
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pip install --prefix=/home/zhuozhu.zz/tops/fairseq --editable ./

通过--prefix指定安装目录
```

```
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1 ValueError: numpy.ndarray size changed, may indicate binary incompatibilit y.

2 Expected 96 from C header, got 80 from PyObject

3 原因numpy 版本太低,解决方案 pip install numpy —upgrade
```

二、环境编译

base_100h.yaml

```
~
```

```
$cat examples/wav2vec/config/finetuning/base_100h.yaml
 1
 2
    # @package _group_
 3
    common:
 4
       fp16: false
 5
       log_format: json
 6
       log interval: 200
7
    checkpoint:
       no_epoch_checkpoints: true
8
9
       best_checkpoint_metric: wer
10
    task:
11
      _name: audio_pretraining
      data: /home/zhuozhu.zz/fairseq/manifest/finetune
12
13
       normalize: false
14
       labels: ltr
15
    dataset:
16
       num workers: 3
17
      max tokens: 800000
18
       skip_invalid_size_inputs_valid_test: true
19
       valid_subset: valid
20
    distributed training:
21
       ddp backend: legacy ddp
22
      distributed_world_size: 3
23
    criterion:
24
       name: ctc
25
       zero_infinity: true
26
    optimization:
27
      max update: 80000
28
      lr: [0.00003]
29
       sentence avg: true
       update_freq: [4]
30
    optimizer:
31
32
      name: adam
33
       adam_betas: (0.9,0.98)
34
       adam_eps: 1e-08
35
    lr scheduler:
36
      name: tri stage
37
       phase_ratio: [0.1, 0.4, 0.5]
38
       final_lr_scale: 0.05
39
    model:
40
       name: wav2vec ctc
      w2v_path: /home/zhuozhu.zz/fairseq/outputs/2021-02-25/10-14-58/checkpoin
41
    ts/checkpoint_last.pt
42
       apply mask: true
43
      mask_prob: 0.65
       mask_channel_prob: 0.5
44
```

```
mask_channel_length: 64
layerdrop: 0.1
activation_dropout: 0.1
feature_grad_mult: 0.0
freeze_finetune_updates: 0
```

boost 编译

```
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1 ./bootstrap.sh --prefix=/disk4/zhuozhu.zz/tops/boost-1.66 --with-python=pyt hon

2 ./b2 define=_GLIBCXX_USE_CXX11_ABI=1 install -j5
```

kenIm 编译

git diff CMakeLists.txt

```
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 1
    SET(CMAKE C COMPILER "/usr/local/bin/qcc")
2
    SET(CMAKE_CXX_COMPILER "/usr/local/bin/g++")
3
4
    set(CMAKE_CXX_FLAGS "${CMAKE_CXX_FLAGS} -std=c++11 -D_GLIBCXX_USE_CXX11_AB
    I=1")
5
    set(CMAKE_CXX_STANDARD 11)
6
7
    add_definitions(-D_GLIBCXX_USE_CXX11_ABI=1)
8
9
    SET(BOOST_INCLUDEDIR "/disk4/zhuozhu.zz/tops/boost-1.66/include")
    SET(BOOST_LIBRARYDIR "/disk4/zhuozhu.zz/tops/boost-1.66/lib")
10
11
12
     # We need boost
    find_package(Boost 1.66.0 REQUIRED COMPONENTS
13
```

setup.py

```
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1 ARGS = ['-g', '-DNDEBUG', '-DKENLM_MAX_ORDER='+max_order, '-std=c++11']
```

```
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1 diff --git a/bindings/python/setup.py b/bindings/python/setup.py
2 - cfg = "Debug" if self.debug else "Release"
3 +
4 + cfg = "Debug"
5 + #cfg = "Debug" if self.debug else "Release"
6 build_args = ["--config", cfg]
```

```
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1 diff --git a/cmake/FindMKL.cmake b/cmake/FindMKL.cmake
2 - SET(_found_gomp true)
3 + SET(_found_gomp false)
4 FOREACH(_lib_name ${OpenMP_CXX_LIB_NAMES})
```

三、学习资料

VQ-WAV2VEC: SELF-SUPERVISED LEARNING OF DISCRETE SPEECH REPRESENTATIONS https://arxiv.org/pdf/1910.05453.pdf

wav2vec 2.0: A Framework for Self-Supervised Learning of Speech Representations https://arxiv.org/pdf/2006.11477.pdf

UniSpeech: Unified Speech Representation Learning with Labeled and Unlabeled Data https://arxiv.org/pdf/2101.07597.pdf

Pushing the Limits of Semi-Supervised Learning for Automatic Speech Recognition https://arxiv.org/pdf/2010.10504.pdf

Self-training and Pre-training are Complementary for Speech Recognition (Xu et al., 2020) https://arxiv.org/pdf/2010.11430.pdf

fairseq 文档

https://fairseq.readthedocs.io/en/latest/data.html

wav2vec Librispeech

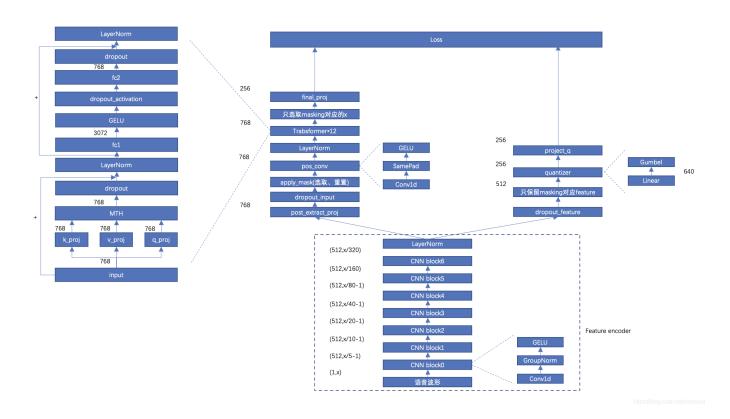
- 1 \$sh run_librispeech_predict.sh
- INFO:__main__:Namespace(all_gather_list_size=16384, autoregressive=False, 2 azureml_logging=False, batch_size=None, batch_size_valid=None, beam=5, bea m_size_token=100, beam_threshold=25.0, best_checkpoint_metric='loss', bf16 =False, bpe=None, broadcast_buffers=False, bucket_cap_mb=25, checkpoint_sh ard_count=1, checkpoint_suffix='', constraints=None, cpu=False, criterion ='ctc', curriculum=0, data='/disk4/zhuozhu.zz/librispeech_raw/LibriSpeech/ librispeech_wav2vec_test', data_buffer_size=10, dataset_impl=None, ddp_bac kend='pytorch_ddp', decoding_format=None, device_id=0, disable_validation= False, distributed_backend='nccl', distributed_init_method=None, distribut ed_no_spawn=False, distributed_port=-1, distributed_rank=0, distributed_wo rld_size=1, diverse_beam_groups=-1, diverse_beam_strength=0.5, diversity_r ate=-1.0, dump_emissions=None, dump_features=None, empty_cache_freq=0, ena ble_padding=False, eos=2, eval_wer=False, eval_wer_post_process='letter', eval_wer_tokenizer=None, fast_stat_sync=False, find_unused_parameters=Fals e, finetune_from_model=None, fix_batches_to_gpus=False, fixed_validation_s eed=None, force_anneal=None, fp16=False, fp16_init_scale=128, fp16_no_flat ten_grads=False, fp16_scale_tolerance=0.0, fp16_scale_window=None, gen_sub set='/disk4/zhuozhu.zz/librispeech_raw/LibriSpeech/librispeech_wav2vec_tes t/train', heartbeat_timeout=-1, iter_decode_eos_penalty=0.0, iter_decode_f orce_max_iter=False, iter_decode_max_iter=10, iter_decode_with_beam=1, ite r_decode_with_external_reranker=False, keep_best_checkpoints=-1, keep_inte rval_updates=-1, keep_last_epochs=-1, kenlm_model='/disk4/zhuozhu.zz/libri speech_raw/LibriSpeech/lm.bin', kspmodel=None, labels='ltr', lenpen=1, lex icon='/disk4/zhuozhu.zz/librispeech_raw/LibriSpeech/lexicon.txt', lm_path= None, lm_weight=2.0, load_checkpoint_on_all_dp_ranks=False, load_emissions =None, localsgd_frequency=3, log_format=None, log_interval=100, lr_schedul er='fixed', lr_shrink=0.1, match_source_len=False, max_len_a=0, max_len_b= 200, max_sample_size=None, max_tokens=4000000, max_tokens_valid=4000000, m aximize_best_checkpoint_metric=False, memory_efficient_bf16=False, memory_ efficient_fp16=False, min_len=1, min_loss_scale=0.0001, min_sample_size=No ne, model_overrides='{}', model_parallel_size=1, nbest=1, no_beamable_mm=F alse, no_early_stop=False, no_epoch_checkpoints=False, no_last_checkpoints =False, no_progress_bar=False, no_repeat_ngram_size=0, no_save=False, no_s ave_optimizer_state=False, no_seed_provided=False, normalize=False, nprocs _per_node=1, num_shards=1, num_workers=1, optimizer=None, optimizer_overri des='{}', pad=1, path='/disk4/zhuozhu.zz/wav2vec_vox_960h_pl.pt', patience =-1, pipeline_balance=None, pipeline_checkpoint='never', pipeline_chunks= 0, pipeline_decoder_balance=None, pipeline_decoder_devices=None, pipeline_ devices=None, pipeline_encoder_balance=None, pipeline_encoder_devices=Non e, pipeline_model_parallel=False, post_process='letter', prefix_size=0, pr int_alignment=None, print_step=False, profile=False, quantization_config_p ath=None, quiet=False, replace_unk=None, required_batch_size_multiple=8, r equired_seq_len_multiple=1, reset_dataloader=False, reset_logging=False, r eset_lr_scheduler=False, reset_meters=False, reset_optimizer=False, restor

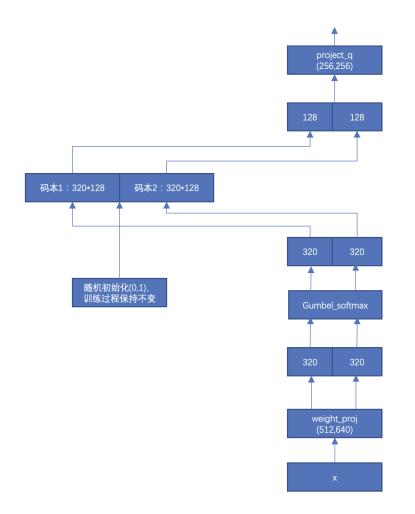
e_file='checkpoint_last.pt', results_path='train', retain_dropout=False, r etain_dropout_modules=None, retain_iter_history=False, rnnt_decoding_type ='greedy', rnnt_len_penalty=-0.5, sacrebleu=False, sample_rate=16000, samp ling=False, sampling_topk=-1, sampling_topp=-1.0, save_dir='checkpoints', save_interval=1, save_interval_updates=0, score_reference=False, scoring ='bleu', seed=1, shard_id=0, sil_weight=0.0, skip_invalid_size_inputs_vali d_test=False, slowmo_algorithm='LocalSGD', slowmo_momentum=None, suppress_ crashes=False, task='audio_pretraining', temperature=1.0, tensorboard_logd ir=None, threshold_loss_scale=None, tokenizer=None, tpu=False, train_subse t='train', unit_lm=False, unk=3, unk_weight=-inf, unkpen=0, unnormalized=F alse, user_dir=None, valid_subset='valid', validate_after_updates=0, valid ate_interval=1, validate_interval_updates=0, w2l_decoder='kenlm', wandb_pr oject=None, warmup_updates=0, wer_args=None, wer_kenlm_model=None, wer_lex icon=None, wer_lm_weight=2.0, wer_word_score=-1.0, wfstlm=None, word_score =-1.0, zero_infinity=False, zero_sharding='none') 3 INFO:__main__:| decoding with criterion ctc 4 dict_path /disk4/zhuozhu.zz/librispeech_raw/LibriSpeech/librispeech_wav2ve c_test/dict.ltr.txt 5 target_dictionary <fairseq.data.dictionary.Dictionary object at 0x7f579881</pre> 6 INFO:__main__:| loading model(s) from /disk4/zhuozhu.zz/wav2vec_vox_960h_p l.pt 7 INFO:fairseq.data.audio.raw_audio_dataset:loaded 2489, skipped 0 samples tgt_dict ['<s>', '<pad>', '</s>', '<unk>', '|', 'E', 'T', 'A', '0', 'N', 'I', 'H', 'S', 'R', 'D', 'L', 'U', 'M', 'W', 'C', 'F', 'G', 'Y', 'P', 'B', 'V', 'K', "'", 'X', 'J', 'Q', 'Z'] 9 tgt dict {'<s>': 0, '<pad>': 1, '</s>': 2, '<unk>': 3, '|': 4, 'E': 5, 'T': 6, 'A': 7, '0': 8, 'N': 9, 'I': 10, 'H': 11, 'S': 12, 'R': 13, 'D': 1 4, 'L': 15, 'U': 16, 'M': 17, 'W': 18, 'C': 19, 'F': 20, 'G': 21, 'Y': 2 2, 'P': 23, 'B': 24, 'V': 25, 'K': 26, "'": 27, 'X': 28, 'J': 29, 'Q': 3 0, 'Z': 31} 10 INFO:__main__:| /disk4/zhuozhu.zz/librispeech_raw/LibriSpeech/librispeech_ wav2vec_test /disk4/zhuozhu.zz/librispeech_raw/LibriSpeech/librispeech_wav 2vec_test/train 2489 examples 11 <fairseq.data.add_target_dataset.AddTargetDataset object at 0x7f57954a1c18</pre> 12 use W2lKenLMDecoder 13 [flashlight] load LM start. 14 [flashlight] load LM finish. 15 load vocab success. 16 token HAMMER, lmIdx 11030 17 token TO, lmIdx 11707 18 token HANDLE, lmIdx 5552 19 token CONTRAST, lmIdx 10751 20 token TREMENDOUSLY, lmIdx 13803 21 token ARGUING, lmIdx 12788 22 token THORLEIF, lmIdx 4043 23 token COVERING, lmIdx 8529

token WAVES,lmIdx 8381
token EXERTED,lmIdx 14503
token PLACARD,lmIdx 3869
INFO:__main__:WER: 1.4511028381569993
INFO:__main__:| Processed 2489 sentences (273750 tokens) in 3445.1s (0.72s entences/s, 79.46 tokens/s)
INFO:__main__:| Generate /disk4/zhuozhu.zz/librispeech_raw/LibriSpeech/lib

https://blog.csdn.net/xmdxcsj/article/details/115787729

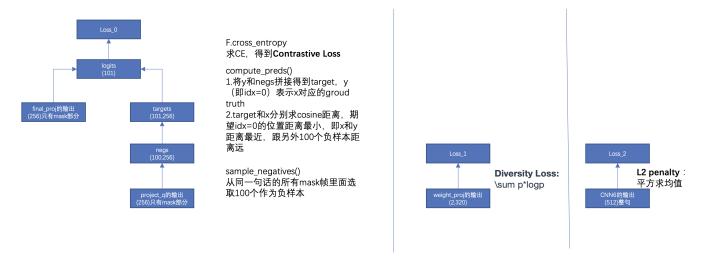
rispeech_wav2vec_test/train with beam=5





两个向量中为1的idx为i和j, 从码本1提取第i行,从码 本2提取第j行,然后拼接

320维的向量,最大值对 应的idx设为1,其他为0



loss部分