Speech Recognition using Spiking Neural Networks on GPU

Swapnil Ahlawat

ID No. 2018A7PS0178G

Department of Computer Science and Information Systems f20180178@goa.bits-pilani.ac.in

Neelay Shah

ID No. 2018A8PS0400G

Department of Electrical and Electronics Engineering f20180400@goa.bits-pilani.ac.in

Under the supervision of

Dr. Basabdatta Sen Bhattacharya Department of Computer Science and Information Systems

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Abstract

Presently, artificial neural networks (ANNs) are the mainstream modelling technique for automatic speech recognition [1]. A conventional ANN features a deep multi-layer architecture which makes use of multiple matrix computations for processing data. The rapid progress in the integration of voice interfaces has been viable on account of the remarkable performance of the ASR systems using ANNs for acoustic modeling

The performance gains however, come with immense computational requirements often due to the time-synchronous processing of input audio signals. These models are computationally intensive and memory inefficient to operate as compared to the biological brains. Alternatively, event-driven models such as spiking neural networks (SNNs) inspired by the human brain have attracted ever-growing attention in recent years. Unlike ANNs, asynchronous and event-driven information processing of SNNs resembles the computing paradigm that observed in the human brains, whereby the energy consumption matches the activity levels of sensory stimuli.

The aim of this project is to implement and simulate biologically plausible SNNs on GPU compute for a speech recognition task. We attempt to train an SNN capable of classifying speech signals into one of 10 digits (0-9). We use a supervised Spike-timing-dependent plasticity (STDP) learning rule to train network to differentiate between speech signals belonging to different categories of digits.

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1 Introduction

Spiking neural networks are computational models that simulate neural behavior in biological systems. Like biological networks, they are driven by discrete spike trains whose instantaneous frequency encodes data. These spikes, by either polarisation or depolarisation, will alter the membrane potential of neurons. Dynamics of the neuron are decided by a set of differential equations that relate membrane voltage with conductance and synaptic current. If a neuron's membrane potential exceeds a certain threshold value, it spikes, passing the integrated information to the next neuron. The influence of a synapse on the membrane voltage of the next neuron can be of two types: excitatory and inhibitory. Excitatory synapses depolarise the neuron, pushing it towards the threshold voltage while inhibitory synapses polarise, thereby reducing the likelihood of a spike. Inhibitory neurons are generally useful to regulate spiking frequency and increase more information contained per spike of the postsynaptic neuron.

There are various neuron models that mimic fairly accurately, the dynamics of biological neurons. The Hodgkin-Huxley model is a conductance-based model that uses a set of non-linear differential equations to explain the electrical characteristics of neurons. While it is a powerful model which is biophysically meaningful, it is very complex and computationally expensive to simulate. Hence, simple spiking neuron models are favoured for applications. A simple model which is widely used is Integrate-and-Fire model. Integrate-and-Fire model considers the neuron like an R-C circuit with a current I(t) flowing through the circuit. This current can be simulated either by external current or presynaptic input spikes. option recently has been the Izhikevich neuron model since it is both accurate and computationally efficient for simulation purposes We focus on the Izhikevich neuron model in the next section.

2 Izhikevich neuron model

The Izhikevich's Neuron model is a simple neuron model which combines the biological plausi- bility, accuracy, scalability and flexibility of the Hodgkin-Huxley model and the computational inexpensiveness of the Integrate and Fire Neuron model.

$$\frac{dv(t)}{dt} = 0.04v^{2}(t) + 5v(t) + 140u(t) + Ipsc(t) + Idc$$

$$\frac{du(t)}{dt} = a(bv(t) - u(t))$$
If $v(t) > V peak$, then $v(t) \leftarrow c$; $u(t) \leftarrow u(t) + d$

where a, b, c, d are parameters that affect the dynamics of the model and can be chosen to obtain different kinds of spiking behavior. $\mathbf{u}(t)$ is the recovery variable $\mathbf{v}(t)$ is the membrane voltage. Ipsc is the post-synaptic current and Idc is the DC bias current stimulus

3 Spike Time Dependent Plasticity

Spike timing dependent plasticity (STDP) is a biological process that adjusts the strength of connections between neurons in a network. The process adjusts the connection strengths based on the relative timing of a particular neuron's output and input action potentials (or spikes). An increase in the strength of the connection is called as Long-Term Potentiation(LTP) whereas a decrease is called Long-Term Depression(LTD)

3.1 Hebbian and Anti-hebbian STDP

In the case of Hebbian STDP, if the postsynaptic spike is generated immediately after receiving the presynaptic spike, the presynaptic spike has a causal role in the output neuron firing. The synaptic weight is thus increased (LTP). Conversely, if a postsynaptic spike occurs before the presynaptic spike, the strength is reduced (LTD), as seen in the equation below

$$\Delta w_{ji} = \begin{cases} Ae^{\frac{(-|t_j - t_i|)}{\tau +}}, & t_j - t_i > 0, A > 0. \\ Be^{\frac{(-|t_j - t_i|)}{\tau -}}, & t_j - t_i < 0, B < 0. \end{cases}$$
(1)

In the above, the first case (Case 1) covers LTP and the second case (Case 2) covers LTD. Both cases are decaying exponentials that decay with the distance between and pre and postsynaptic spikes. A>0 and B<0 scale the amplitude of the exponential, and $\tau+$ and τ are the respective time constants

Anti-hebbian STDP is simply the reverse of Hebbian STDP

$$\Delta w_{ji} = \begin{cases} Be^{\frac{(-|t_j - t_i|)}{\tau_-}}, & t_j - t_i > 0, B < 0. \\ Ae^{\frac{(-|t_j - t_i|)}{\tau_+}}, & t_j - t_i < 0, A > 0.. \end{cases}$$
 (2)

4 PyGeNN - A neuronal network simulation framework

GPU-enhanced Neuronal Networks (GeNN) is a software package to enable neuronal network simulations on NVIDIA GPUs by code generation. It is a cross-platform C++ library for generating optimized CUDA code for GPU accelerated spiking neural network simulations.

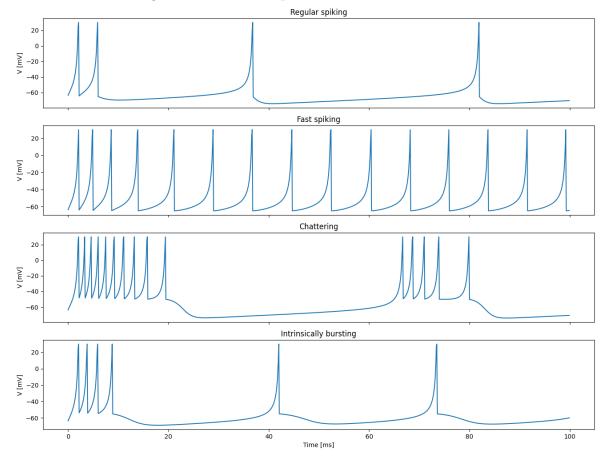
Python interface to GeNN (PyGeNN) is an abstraction on top of GeNN that allows users to access all GeNN features from Python

Given below is an example code snippet highlighting how an Izhikevich neuron can be simulated in 4 regimes for 100 ms using PyGeNN -

```
import numpy as np
2 import matplotlib.pyplot as plt
3 from pygenn.genn_model import GeNNModel
5 # Create a single-precision GeNN model
6 model = GeNNModel("float", "pygenn")
8 # Set simulation timestep to 0.1ms
9 \text{ model.dT} = 0.1
# Initialise IzhikevichVariable parameters
izk_init = {"V": -65.0,
              "U": -20.0,
              "a": [0.02,
                               0.1,
                                       0.02,
                                                0.02],
14
              "b": [0.2,
                               0.2,
                                       0.2,
                                                0.2],
              "c": [-65.0,
                               -65.0,
                                       -50.0,
16
              "d": [8.0,
                               2.0,
                                       2.0,
19 # Add neuron populations and current source to model
pop = model.add_neuron_population("Neurons", 4, "IzhikevichVariable",
                                       {}, izk_init)
model.add_current_source("CurrentSource", "DC", "Neurons",
                           {"amp": 10.0}, {})
25 # Build and load model
26 model.build()
27 model.load()
29 # Create a numpy view to efficiently access the membrane
30 voltage from Python
voltage_view = pop.vars["V"].view
33 # Simulate
34 v = None
35 while model.t < 100.0:
      model.step_time()
     model.pull_state_from_device("Neurons")
      v = np.copy(voltage_view) if v is None
          else np.vstack((v, voltage_view))
```

```
40
41 # Create plot
42 figure, axes = plt.subplots(4, sharex=True)
43
44 # Plot voltages
45 for i, t in enumerate(["RS", "FS", "CH", "IB"]):
46    axes[i].set_title(t)
47    axes[i].set_ylabel("V [mV]")
48    axes[i].plot(np.arange(0.0, 200.0, 0.1), v[:,i])
49 axes[-1].set_xlabel("Time [ms]")
```

Figure 1: Membrane potential vs Simulation Time



5 Current Work

We aim to replicate the results of A spiking network that learns to extract spike signatures from speech signals [2]. The method described in the paper uses a compact, non-recurrent SNN consisting of Izhikevich neurons equipped with STDP for a spoken digit recognition task.

We are making use of the Free Spoken Digit Dataset (FSDD) - an open speech dataset consisting of recordings of spoken digits in English by 6 native English speakers.

Described hereon are the different steps we have completed towards trying to replicate the previous speech recognition work -

5.1 Observing behaviour of Izhikevich neurons with custom parameters

PyGeNN allows users the flexibility of simulating Izhikevich neurons with custom parameters. We use the values of parameters mentioned in [2] in order to have a faithful comparison when trying to replicate results.

We simulated a single Izhikevich neuron for a range of current values and for a time duration of 1000 ms with a timestep of 1 ms to determine the current sweep for which the neurons display desired spiking behaviour (Figure 2). We observed the frequency of spikes for the neuron for the range of current values. Simultaneously, we also recorded the membrane potential of the neuron to keep track of the model dynamics.

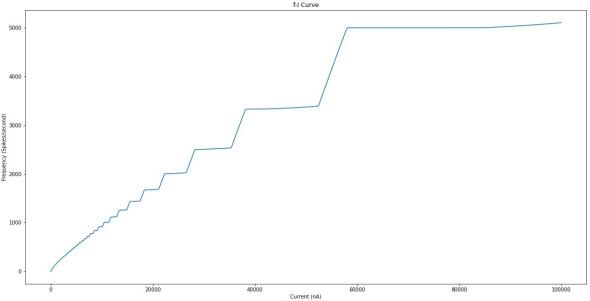


Figure 2: Spiking frequency vs Current

We observed that the neuron starts spiking at an input current of 52 nA, and the spiking frequency saturates at a current value of 59000 nA. We checked the membrane potential over time for the neuron to confirm whether 59000 nA was a suitable upper

bound of input current. However, We noticed irregularties in the action potentials of the neuron for a current value of greater than 52000 nA. Hence, we decided to use 52000 nA as the upper bound of input current to be fed to the Izhikevich neurons during the speech recognition task, with 52 nA being the lower bound.

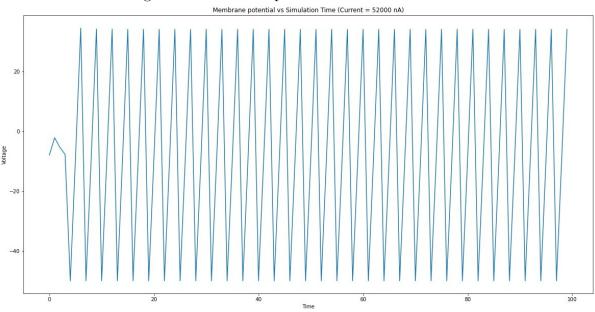


Figure 3: Membrane potential vs Simulation time

5.2 Feature extraction from speech dataset

We follow the same procedure as [2] for extracting representative feature vectors to be used as input to the SNN from raw speech signals. The speech signal is divided into small overlapping time sections called speech frames. A fixed number of frames, N, is used for for each spoken digit. The length, L, of a spoken digit varies from 500 to 1000 ms, and was divided into N=40 frames with 50% overlap to support a frame length of 10-50 ms. The 50% overlap captures the temporal characteristics of the changing spectrum of the speech signal.

After framing, a small feature vector based on the frame's frequency spectrum is extracted. The spectrum values are calculated for all the frames temporally to represent the speech signal spectrogram. A frame encompassing an R Hz frequency range can be divided into M frequency bands. We use R=4000 Hz and M=5 to produce a feature vector of length N*M=200 for each spoken digit. The number of filter banks (M=5) is small enough to create a minimal SNN.

5.3 Network Architecture

We use Izhikevich neurons in pur spiking network. The speech signal is divided into 40 input vectors and each input vector has 5 features. Each of these are connected to one neuron. Therefore, the input layer has 200 neurons (denoted by y unit layer).

The output layer has 10 neurons (denoted by z unit layer). Each unit corresponds to one of the ten spoken digit categories (class labels). The y units are fully connected to the z units, i.e., there is no hidden layer.

Finally, there is a supervision signal (called a teacher) that monitors the z units in order to determine the form of the STDP used in training. The teacher determines which z units undergo Hebbian versus anti-Hebbian STDP. During training, whenever a z unit emits a spike, it undergoes some form of STDP. If the z unit represents the target category, then it undergoes Hebbian STDP. Otherwise, it undergoes anti-Hebbian STDP The teaching signal is only used for the training phase.

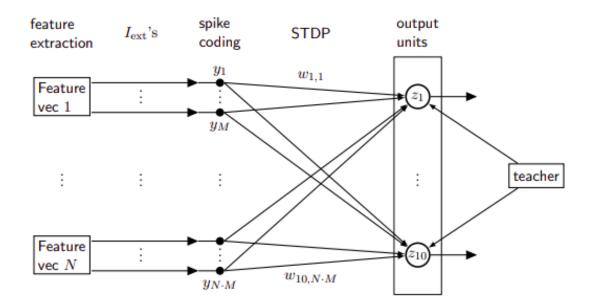


Figure 4: Network Architecture

5.4 Feeding data into input neurons

We employ an input layer of 200 Izhikevich neurons in our network; one neuron for each element in the feature vector of a speech signal. The values of the features are used as constant DC current sources for the network simulation. We normalize the feature values between the lower and upper bounds determined previously.

We supply one sample speech signal as input and simulate the network for a period of 100 ms. We observe the behaviour of the 200 input neurons in the form of a raster plot. (Figure 5)

5.5 Implementing STDP in PyGeNN

We implemented both hebbian and anti-hebbian forms of STDP in PyGeNN for a spiking neural network of 2 neuron layers. We use the utility of defining custom learning rules

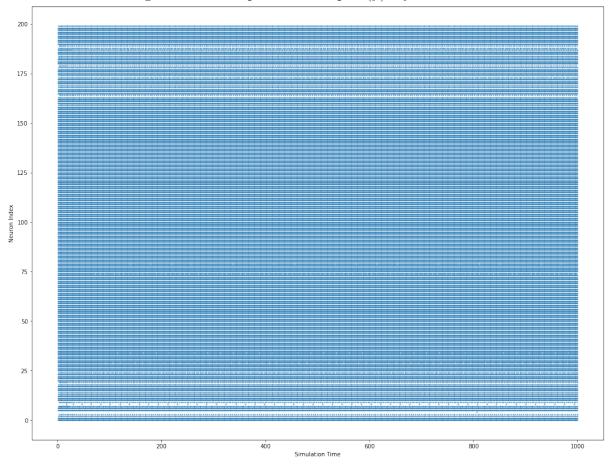


Figure 5: Raster plot of the input (y) layer neurons

PyGeNN provides to implement both the STDP forms. We initialize the synaptic weights between neurons uniformly between 0.1 and 1.0 and limit them to the same upper and lower bounds during the training process as well. We employ the same STDP parameters (in equations (1) and (2)) as [2]. We track the synaptic weights, spike times and membrane potentials of all the neurons in the network using state variables.

5.6 Teacher supervision in PyGeNN

To facilitate the supervision signal, we make use of an 'index' parameter in the network. Every z unit (output neuron) has the aforementioned 'index' parameter which corresponds to what digit that particular z unit represents. For example, the first z unit would have an index of 0 because it corresponds to the digit 0. We also employ a variable called 'label' which denotes which digit category a speech signal belongs to. The 'label' variable is updated every time a new speech signal is being processed.

Every time a z unit spikes, its index is compared to the label of the speech signal currently being processed. If the index matches the label, the 'correct' neuron for that particular signal has spiked and it undergoes hebbian SDTP. If the index of a neuron doesn't match the label, it undergoes anti-hebbian STDP.

5.7 Network training

We simulated the network for 1 speech sample using teacher supervision as described above. We plotted and observed the behaviour of the neurons in the output (z) layer. Figure 6 shows the membrane potential over time of one z unit. Regular spiking behaviour is observed.

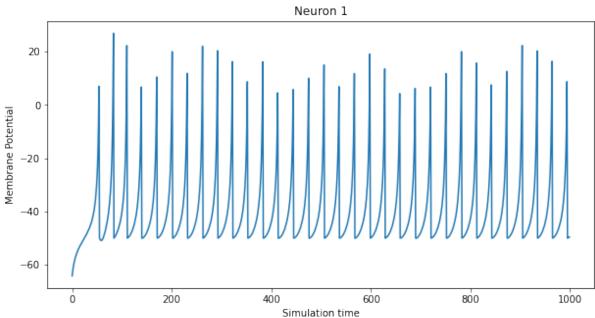
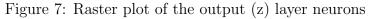


Figure 6: Membrane potential of neuron 1 in the output layer (1st z unit)



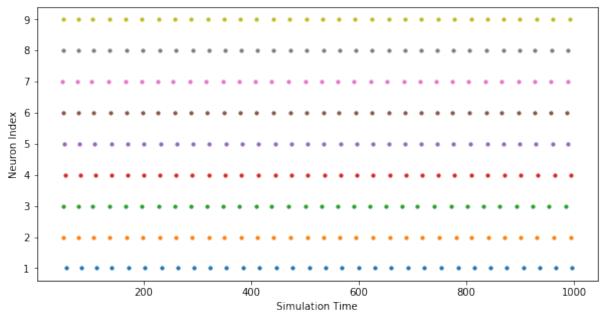


Figure 7 displays a raster plot of the output layer neurons. Since the network hasn't been trained yet with enough training samples to distinguish between speech signals belonging to different categories, all the output layer neurons display similar spiking behaviour. After training, we expect the spiking behaviour of the neurons to be representative of the category of the speech signal being processed.

6 Conclusion and Future Work

During the course of the project, we were able to learn how to simulate neuronal networks on GPU compute using PyGeNN. We understood how spiking neurons process information, how learning takes place in spiking neural networks and what are the ways by which we can decode what an SNN has learnt.

We were able to develop a neuronal network for speech recognition in PyGeNN with an aim of replicating the results in [2]. We tested the working of the network we developed on 1 training sample from a spoken digit speech dataset.

We plan to carry out the following steps here onwards -

- Train the network for the entire training portion of the speech dataset.
- Obtain spike signatures for a test portion of speech signals in the dataset and calculate evaluation metrics to judge performance.

Appendix

PyGeNN code for training the network using supervised STDP -

```
1 # Imports
3 import pandas as pd
4 import numpy as np
5 from pygenn import genn_model, genn_wrapper
6 from pygenn.genn_model import (
     create_custom_neuron_class,
      create_custom_current_source_class,
      create_custom_weight_update_class,
      GeNNModel,
10
11 )
12 import argparse
13 from os.path import exists, join
14 import os
17 # Command line arguments
19 parser = argparse.ArgumentParser("Script to train model the SNN using
     supervised STDP")
parser.add_argument(
     "--datafile",
     type=str,
      required=True,
      help="Path to the .npy file containing the speech data",
26 )
27 parser.add_argument(
      "--outdir",
      type=str,
      default="output",
      help="Name of folder where all the ouput files (membrane potentials
      etc) should be stored. Folder doesn't need to exist beforehand. (
     default = output)",
32 )
parser.add_argument(
     "--n_samples",
34
      type=int,
      default=1,
      help="Number of samples in the dataset for which the network should
      be simulated. (default=1)",
38 )
args = parser.parse_args()
42 # Define custom neuron class
44 izk_neuron = create_custom_neuron_class(
     "izk_neuron",
     param_names=["a", "b", "c", "d", "C", "k"],
    var_name_types=[("V", "scalar"), ("U", "scalar")],
```

```
sim_code="""
                       (V) += (0.5/\$(C))*(\$(k)*(\$(V) + 60.0)*(\$(V) + 40.0)-\$(U)+\$(Isyn)
                       //at two times for numerical stability
                       (V) + (0.5/\$(C)) * (\$(k) * (\$(V) + 60.0) * (\$(V) + 40.0) - \$(U) + \$(V) + (0.5/\$(C)) * (V) + (V
                       (U) += (a)*((b)*((V) + 60.0)-(U))*DT;
                      """,
52
             reset_code="""
                      (V) = (c);
54
                       (U) += (d);
                       0.00
56
              threshold_condition_code="$(V) > 30",
58 )
59
60
_{61} # Set the parameters and initial values of variables for the IZK neuron
              (whose class has been defined above)
63 izk_params = {"a": 0.03, "b": -2.0, "c": -50.0, "d": 100.0, "k": 0.7, "
           C": 100.0}
65 izk_var_init = {
             "V": -60.0,
66
             "U": 0.0,
68 }
70
71 # Define the weight update (supervised STDP learning) rule
73 # An additional called parameter called 'index' is created. Each neuron
             in the ouput layer corresponds to a digit from 0-9. This digit a
            neuron in the output layer corresponds to is stored in 'index'.
            Therefore, each post neuron has an index
_{75} # An additional variable called 'label' is created. 'label' is the
            category (digit:0-9) that the speech signal currently processed
            belongs to. 'label' is updated every time a new speech signal is
            being processed
77 # 'sim_code' is called when a pre-synaptic spike occurs
79 # 'learn_post_code' is called whena post-synaptic spike occurs
81 # Whether the correct post neuron corresponding to a particular speech
            signal being processed has spiked is determined by comparing the '
            label' of the speech signal to the 'index' of the post neruon
83 # Logic in 'sim_code': when a pre-synaptic spike occurs -
84 #
                                                                                                                                                              if
            the the correct post neuron for the speech signal currently being
            processed fires , the synaptic strength is reduced
            learning case 2)
85 #
                                                                                                                                                              if
           the the incorrect post neuron for the speech signal currently being
```

```
processed fires , the synaptic strength is reduced (Anti-hebbian
      learning case 2)
87 # Logic in 'learn_post_code': when a post-synaptic spike occurs -
      the the correct post neuron for the speech signal currently being
      processed fires , the synaptic strength is increased
                                                               (Hebbian
      learning case 1)
                                                                          if
      the the incorrect post neuron for the speech signal currently being
      processed fires , the synaptic strength is increased (Anti-hebbian
      learning case 1)
90
  supervised_stdp = create_custom_weight_update_class(
       "supervised_stdp",
92
       param_names=["tauMinus", "tauPlus", "A", "B", "gMax", "gMin", "
      index"],
       var_name_types=[("g", "scalar"), ("label", "scalar")],
       sim_code="""
95
           $(addToInSyn, $(g));
           scalar dt = $(t) - $(sT_post);
97
           if(dt > 0) {
               if($(index) == $(label)){
99
                   scalar timing = exp(-dt / $(tauMinus));
                   scalar newWeight = $(g) - ($(B) * timing);
                   $(g) = fmax($(gMin), newWeight);
                                }
103
104
               else{
                   scalar timing = exp(-dt / $(tauPlus));
                   scalar newWeight = $(g) + ($(A) * timing);
108
                   $(g) = fmin($(gMax), newWeight);
               }
109
           }
110
           . . .
112
       learn_post_code="""
113
           scalar dt = $(t) - $(sT_pre);
114
           if (dt > 0) {
               if($(index) == $(label)){
                   scalar timing = exp(-dt / $(tauPlus));
117
                   scalar newWeight = $(g) + ($(A) * timing);
118
                   $(g) = fmin($(gMax), newWeight);
119
                                }
               else{
                   scalar timing = exp(-dt / $(tauMinus));
123
                   scalar newWeight = $(g) - ($(B) * timing);
124
                   $(g) = fmax($(gMin), newWeight);
126
           }
127
           0.00
128
       is_pre_spike_time_required=True,
       is_post_spike_time_required=True,
130
131
```

```
# Set the initial values of variables of the weight update rule
135
# Initialize weights uniformly between 0 and 1
138 stdp_var_init = {
      "g": genn_model.init_var("Uniform", {"min": 0.1, "max": 1.0}),
      "label": 0.0,
141 } # Initialize label to 0
143
144 # Define GeNN model
146 model = genn_model.GeNNModel("float", "speech_recognition")
147
148
# Add neuron populations
151 num_inp_neurons = 200
152 num_output_neurons = 10
inp_layer = model.add_neuron_population(
      "input_layer", num_inp_neurons, izk_neuron, izk_params,
      izk_var_init
157
  neuron_layers = [inp_layer]
160 for i in range (10):
      neuron_layers.append(
161
162
           model.add_neuron_population(
               "output_neuron_" + str(i), 1, izk_neuron, izk_params,
163
      izk_var_init
           )
164
       )
165
166
167
168 # Create synaptic connections
170 # Each synapse group contains synapse connections from all 200 input
      neurons to 1 particular neuron in the output layer. 10 such synapse
      groups are created, one for every neuron in the output layer
172 # Every synapse group contains (belonging to specific output neuron)
      contains the 'index' of that neuron
174 \text{ syn_io} = []
  for i in range(num_output_neurons):
       syn_io.append(
           model.add_synapse_population(
177
               "synapse_input_output_" + str(i),
178
               "DENSE_INDIVIDUALG",
               genn_wrapper.NO_DELAY,
180
               inp_layer,
181
```

```
neuron_layers[i + 1],
                supervised_stdp,
183
184
                    "tauMinus": 20.0,
185
                    "tauPlus": 20.0,
186
                    "A": 0.1,
187
                    "B": 0.1,
188
                    "gMax": 1.0,
189
                    "gMin": 0.1,
190
                    "index": float(i),
191
                },
192
                stdp_var_init,
193
                {},
194
                {},
195
                "DeltaCurr",
196
                {},
                {},
198
           )
200
201
202
203 # Define current source
204
205 current_source = create_custom_current_source_class(
      "current_source",
       var_name_types=[("magnitude", "scalar")],
       injection_code="$(injectCurrent, $(magnitude));",
209 )
210
212 # Create current input
214 current_input = model.add_current_source(
      "input_current", current_source, inp_layer, {}, {"magnitude": 0.0}
216 )
217
219 # Set simulation parameters
221 timesteps_per_sample = (
      1000.0 # No. of timesteps one speech signal in the dataset is
      presented for
223
224 resolution = 1
225
227 # Build and load model
229 model.dT = resolution
230 model.build()
model.load()
232
234 # Load data
```

```
dataset = np.load(args.datafile, allow_pickle=True).item()
237 data = dataset["data"]
238 labels = dataset["labels"]
241 # Initialize data structures for variables and parameters we want to
242
243 layer_spikes = [(np.empty(0), np.empty(0)) for _ in enumerate(
      neuron_layers)]
244 layer_voltages = [l.vars["V"].view for l in neuron_layers]
245 current_input_magnitude = current_input.vars["magnitude"].view[:]
246 neuron_labels = [
       syn_io[neuron].vars["label"].view for neuron in range(
      num_output_neurons)
248
249
input_voltage_view = inp_layer.vars["V"].view[:]
251 input_voltage = None
252 output_voltage = {}
253 output_voltage_view = {}
for index, output_neuron in enumerate(neuron_layers[1:]):
       output_voltage[index] = None
       output_voltage_view[index] = output_neuron.vars["V"].view
257
259 synaptic_weights = {}
260 synaptic_weight_views = {}
261 for index, synapse in enumerate(syn_io):
       synaptic_weights[index] = None
       synaptic_weight_views[index] = synapse.vars["g"].view[:]
263
265
266 # Simulate
268 num_simulation_samples = args.n_samples
  while model.t < timesteps_per_sample * num_simulation_samples:</pre>
270
271
       timestep_in_example = model.t % timesteps_per_sample
272
       sample = int(model.t // timesteps_per_sample)
      if timestep_in_example == 0: # If a new sample is starting to be
275
      processed
276
           label = labels[sample]
277
           print(f"Processing sample {sample}: {label}")
278
279
           current_input_magnitude[:] = data[
               sample
281
              # Update the current input for the new sample
           model.push_var_to_device("input_current", "magnitude")
283
           for 1, v in zip(neuron_layers, layer_voltages):
285
```

```
v[:] = -65.0 # Manually 'reset' voltage
287
               1.push_var_to_device("V")
288
           for index, synapse in enumerate(syn_io):
200
               neuron_labels[index] = float(label)
291
               synapse.push_var_to_device("label")
                                                      # Update the 'label'
292
      for the new sample
293
       model.step_time()
                          # Simulate a timestep
294
205
       # record input neurons membrane potential
296
297
       model.pull_state_from_device("input_layer")
298
       input_voltage = (
299
           np.copy(input_voltage_view)
300
           if input_voltage is None
           else np.vstack((input_voltage, input_voltage_view))
302
304
       # record output neurons membrane potential
306
       for index, output_neuron in enumerate(neuron_layers[1:]):
           model.pull_state_from_device("output_neuron_" + str(index))
308
           output_voltage[index] = (
               np.copy(output_voltage_view[index])
310
               if output_voltage[index] is None
               else np.hstack((output_voltage[index], output_voltage_view[
312
      index]))
           )
313
       # record synaptic weights
315
316
       for synapse_index, synapse in enumerate(syn_io):
317
           synapse.get_var_values("g")
318
           synaptic_weights[synapse_index] = (
310
               np.copy(synaptic_weight_views[synapse_index])
320
               if synaptic_weights[synapse_index] is None
321
322
               else np.vstack(
                    (synaptic_weights[synapse_index], synaptic_weight_views
323
      [synapse_index])
               )
324
           )
325
       # record spikes
327
       for i, l in enumerate(neuron_layers):
329
           model.pull_current_spikes_from_device(l.name)
           spike_times = np.ones_like(l.current_spikes) * model.t
331
           layer_spikes[i] = (
               np.hstack((layer_spikes[i][0], l.current_spikes)),
333
               np.hstack((layer_spikes[i][1], spike_times)),
           )
335
338 # Create ouput directory
```

References

- [1] J. Wu, E. Yılmaz, M. Zhang, H. Li, and K. C. Tan, "Deep spiking neural networks for large vocabulary automatic speech recognition," *Frontiers in Neuroscience*, vol. 14, p. 199, 2020.
- [2] A. Tavanaei and A. S. Maida, "A spiking network that learns to extract spike signatures from speech signals," *Neurocomputing*, vol. 240, p. 191–199, May 2017.