

**A readily-interpretable  
fully-convolutional  
autoencoder-like  
algorithm for  
unlabelled waveform  
analysis**

*23<sup>rd</sup> October 2020*

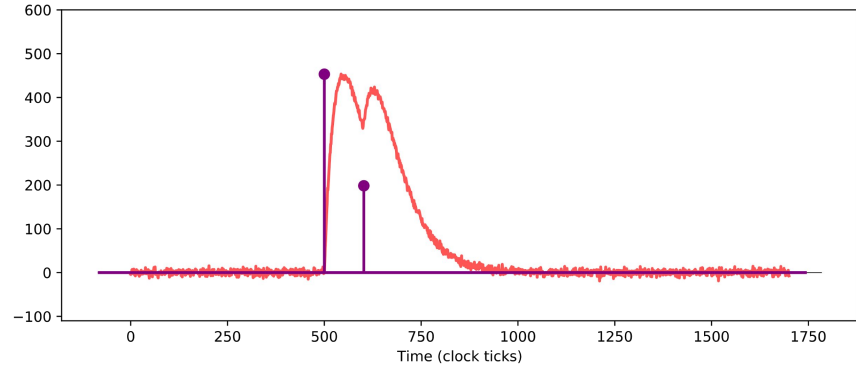
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Chris Wright



# Waveform analysis

Goals of pulse reconstruction:

- What energy, when, what type?



Methods:

- Simple sum or maxima  $\Rightarrow$  Poor resolution
- Filter noise:  $\Rightarrow$  better but sub-optimal resolution, and parameters to tune
- Template fitting  $\Rightarrow$  optimal resolution but need a template

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**Learn the optimal template  
without labelling the data ?**

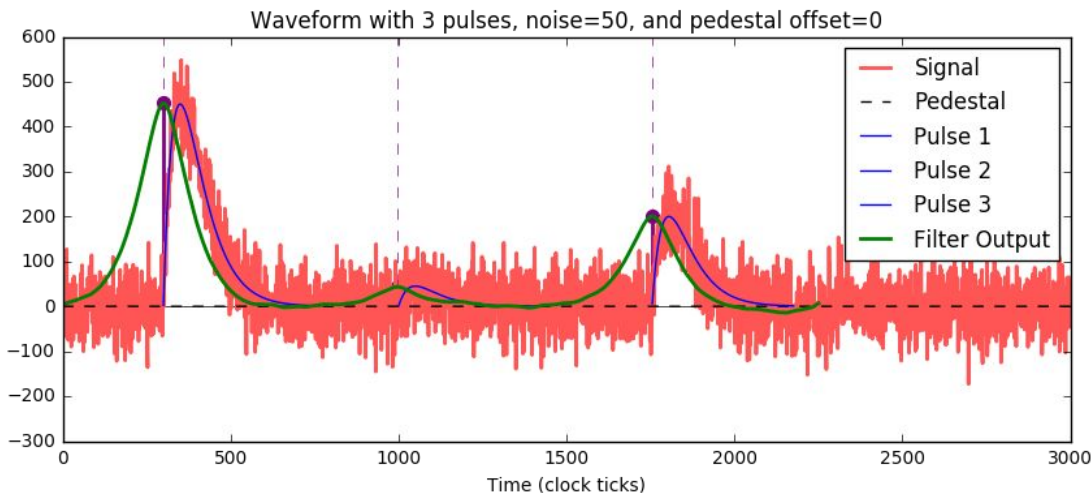
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# The matched filter

If you know the expected pulse shape:

- Convolution  $\Rightarrow$  Best possible signal to noise ratio for any linear filter
- Maxima indicate:
  - When pulse happens
  - What energy it has

**Convolutional layers in CNNs are matched filters**

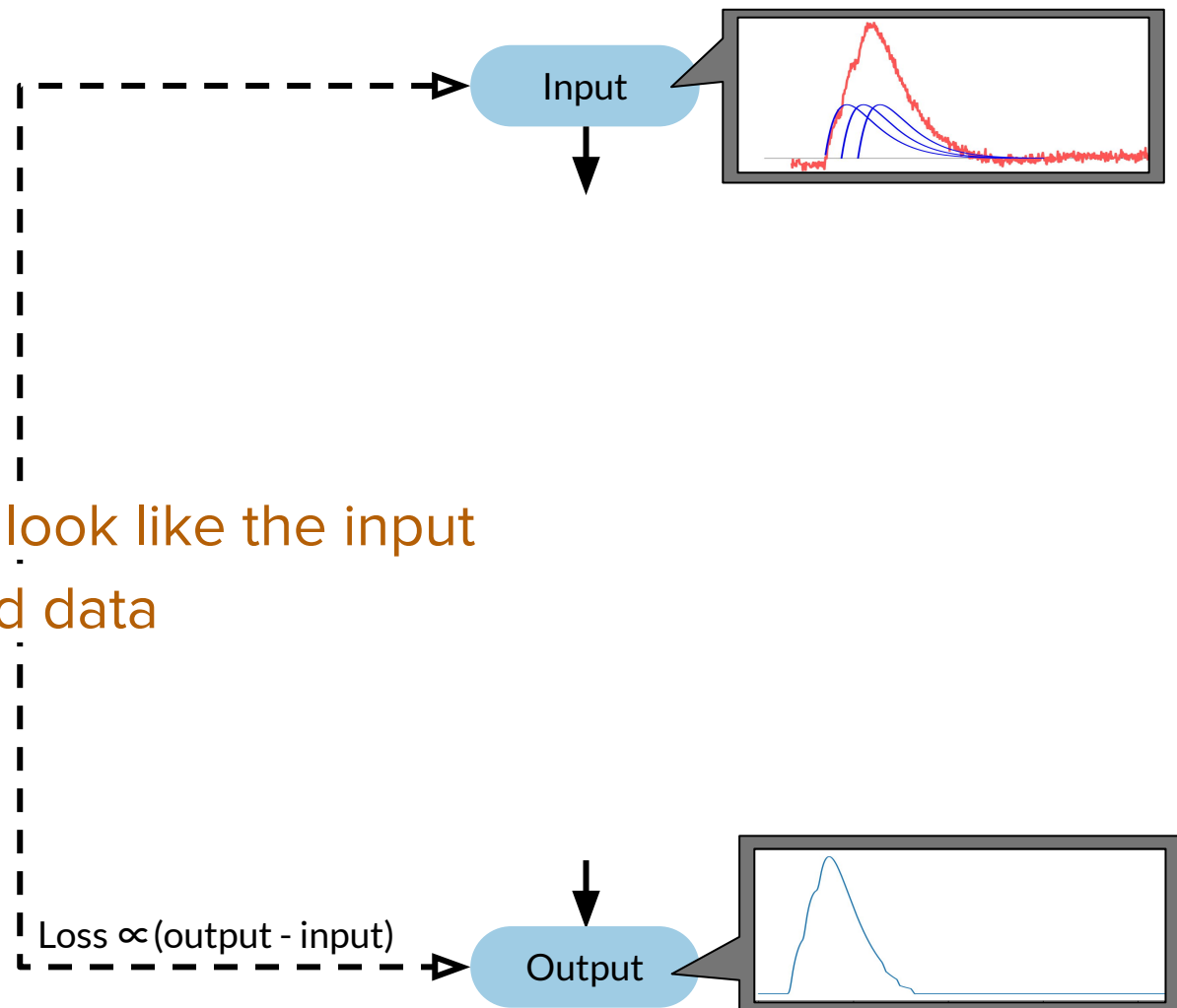


# Our model

Autoencoder-like:

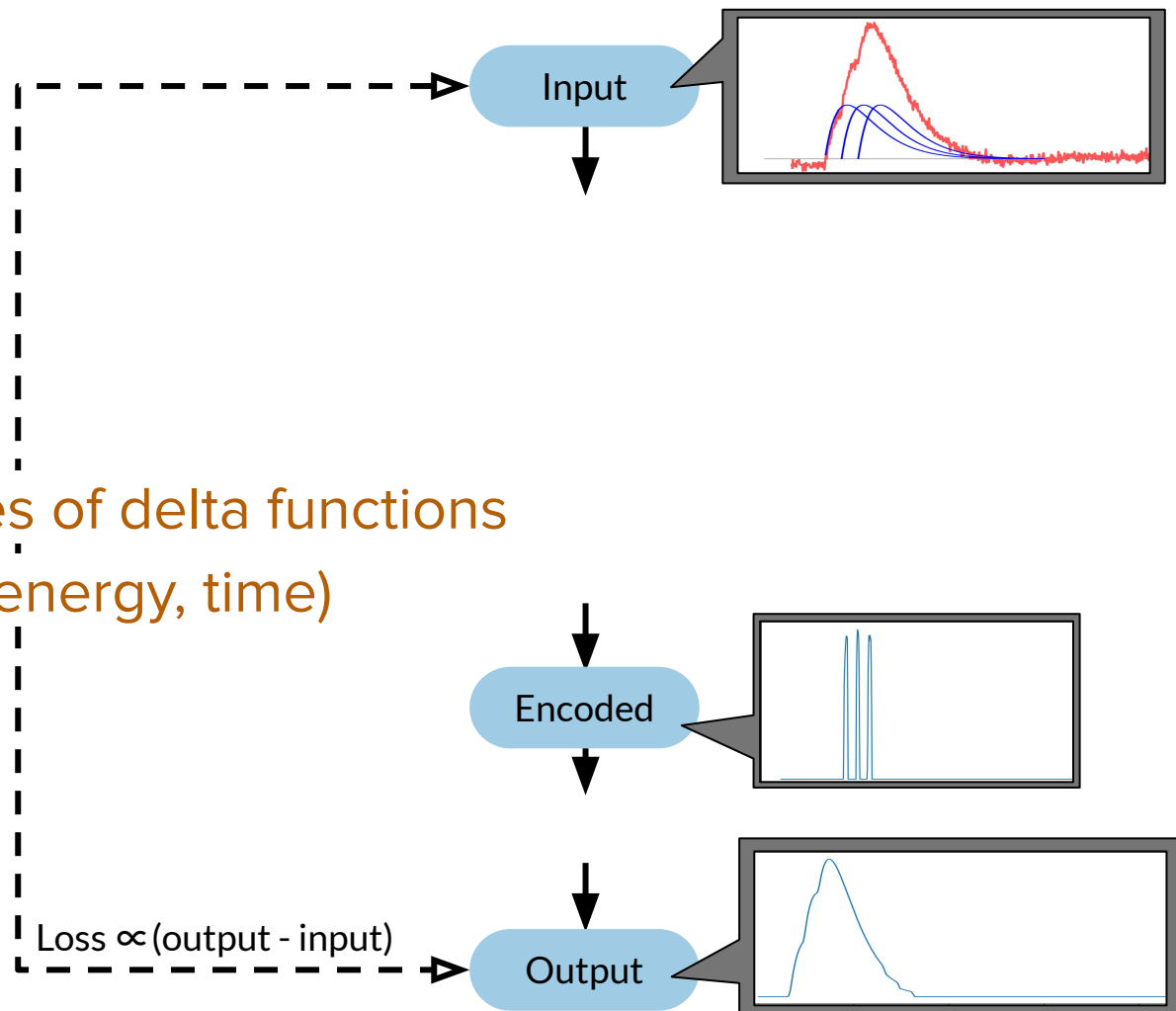
⇒ want the output to look like the input

⇒ works on unlabeled data

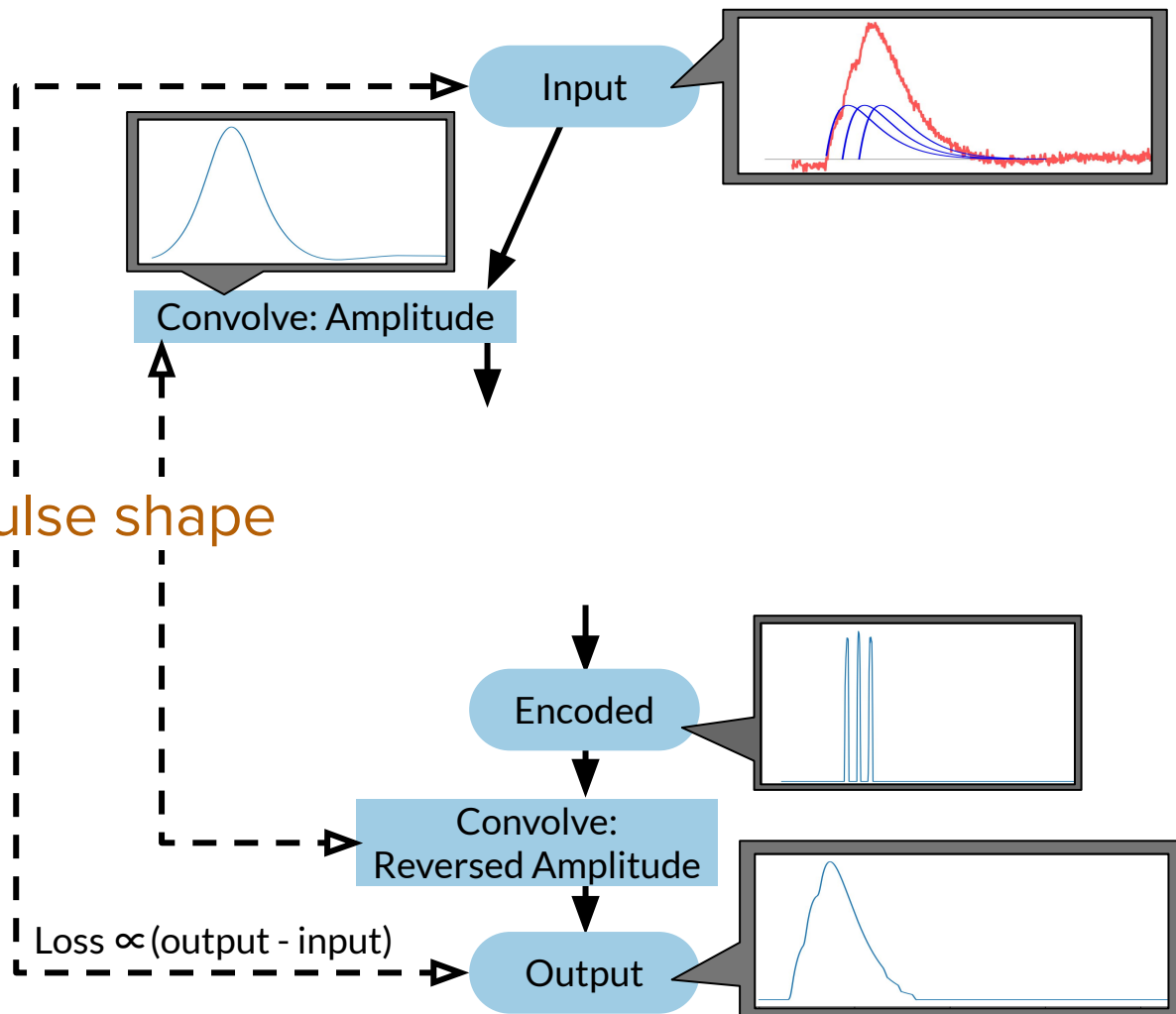


# Our model

Encoded space: series of delta functions  
⇒ Pulse parameters (energy, time)



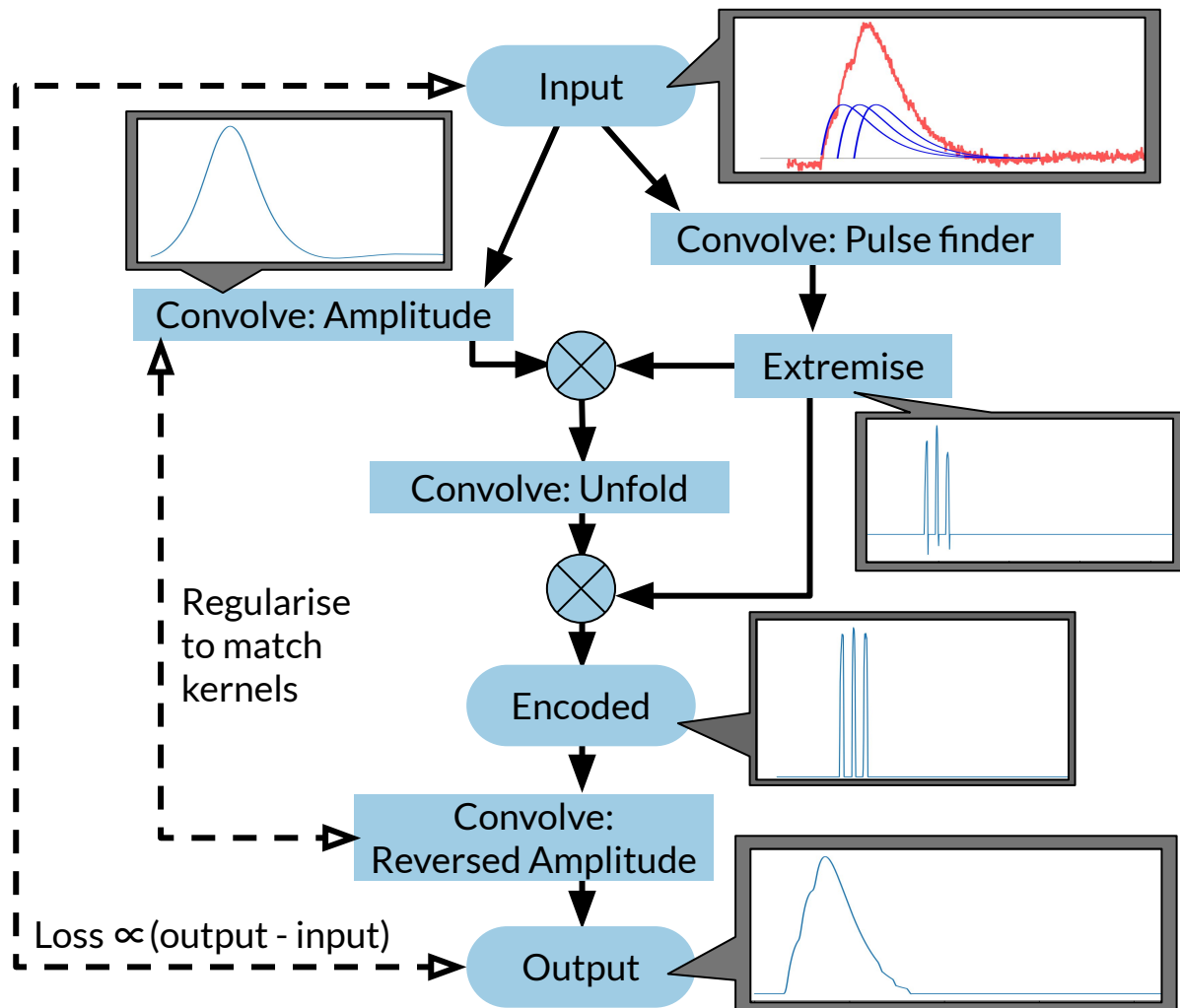
# Our model



Convolve with learnt pulse shape

# Our model

Detect pulses and disentangle contributions to observed amplitudes

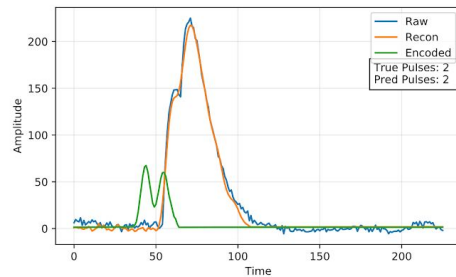
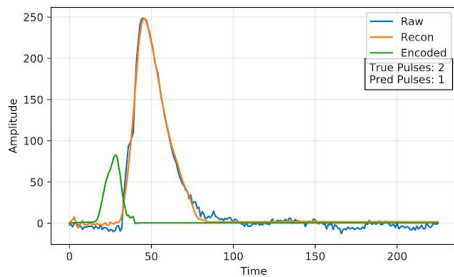
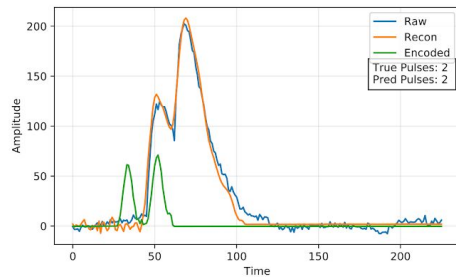
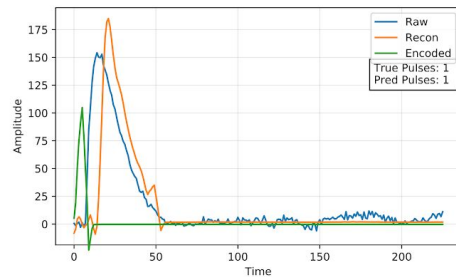
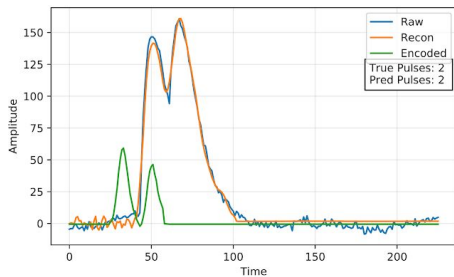
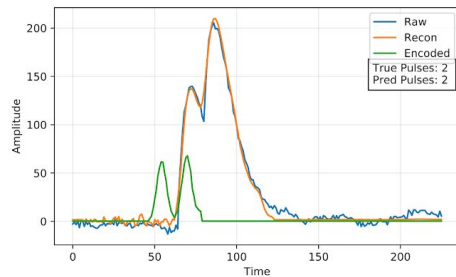
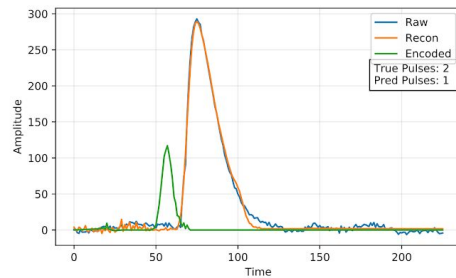
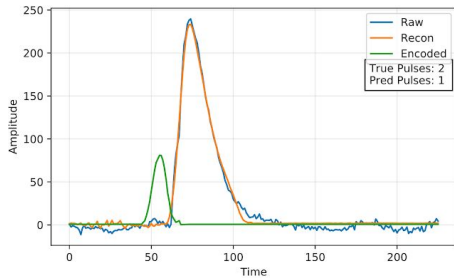
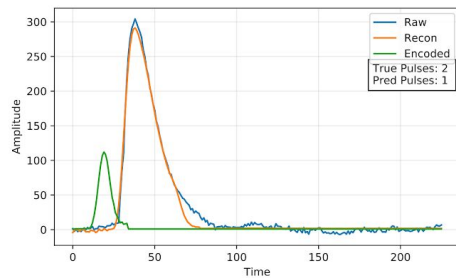




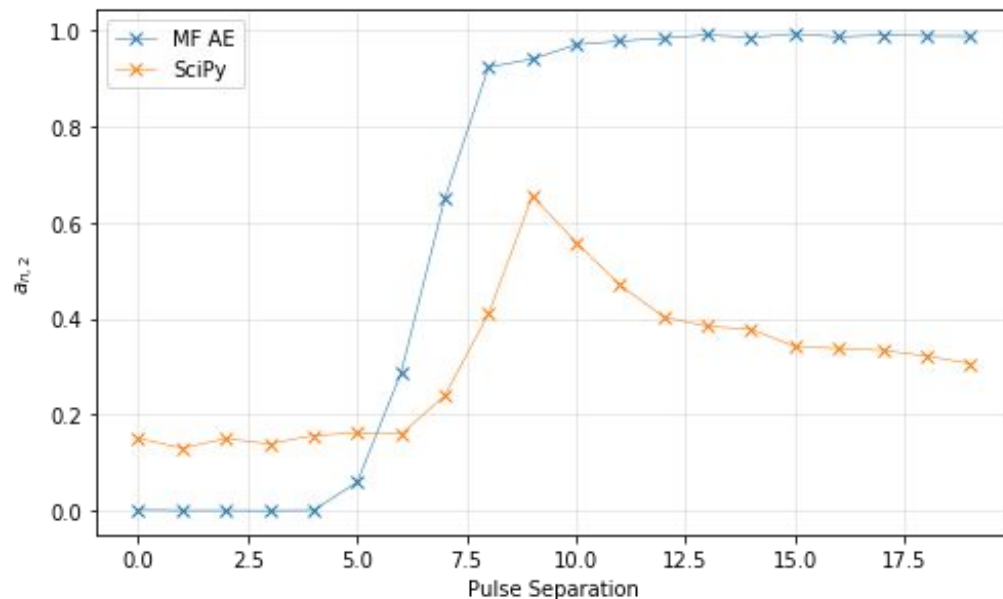
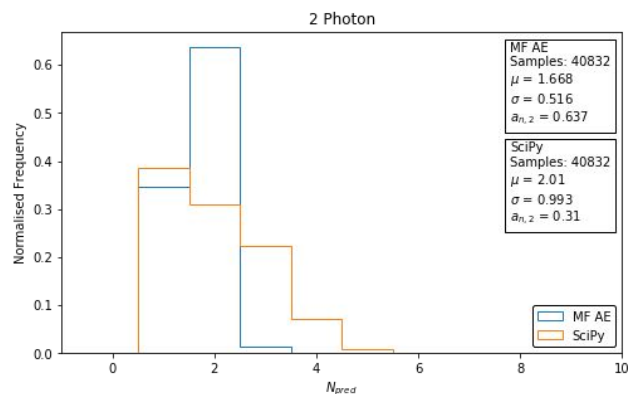
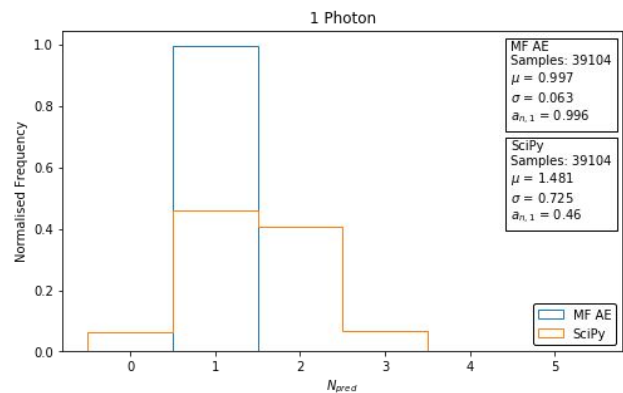
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**Does it work?**

# Examples on toy data



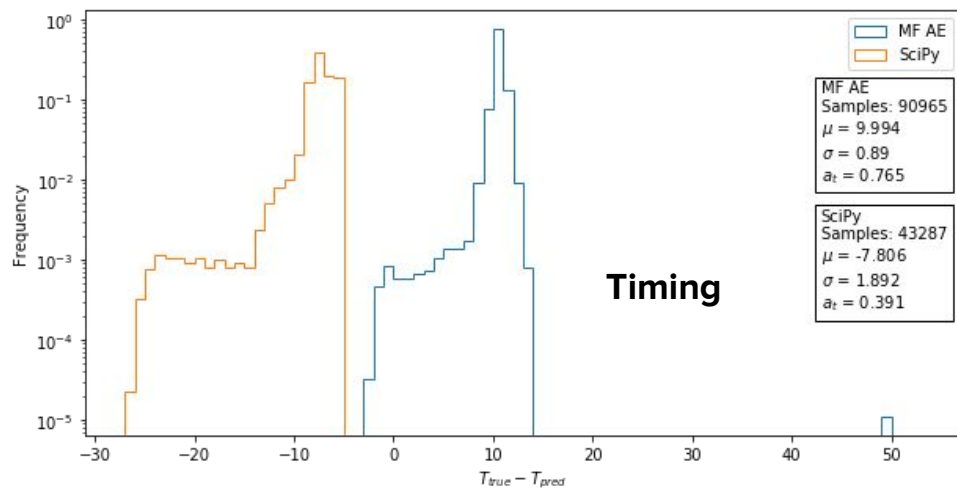
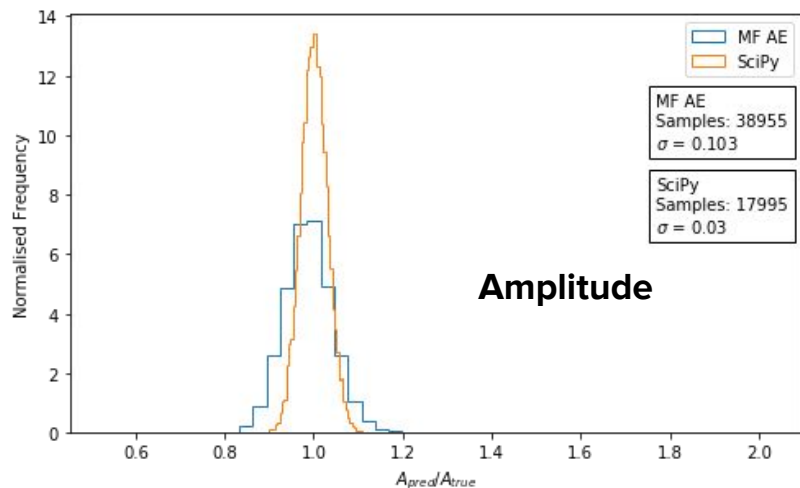
# Handling overlapping pulses



**MF AE = Our model**

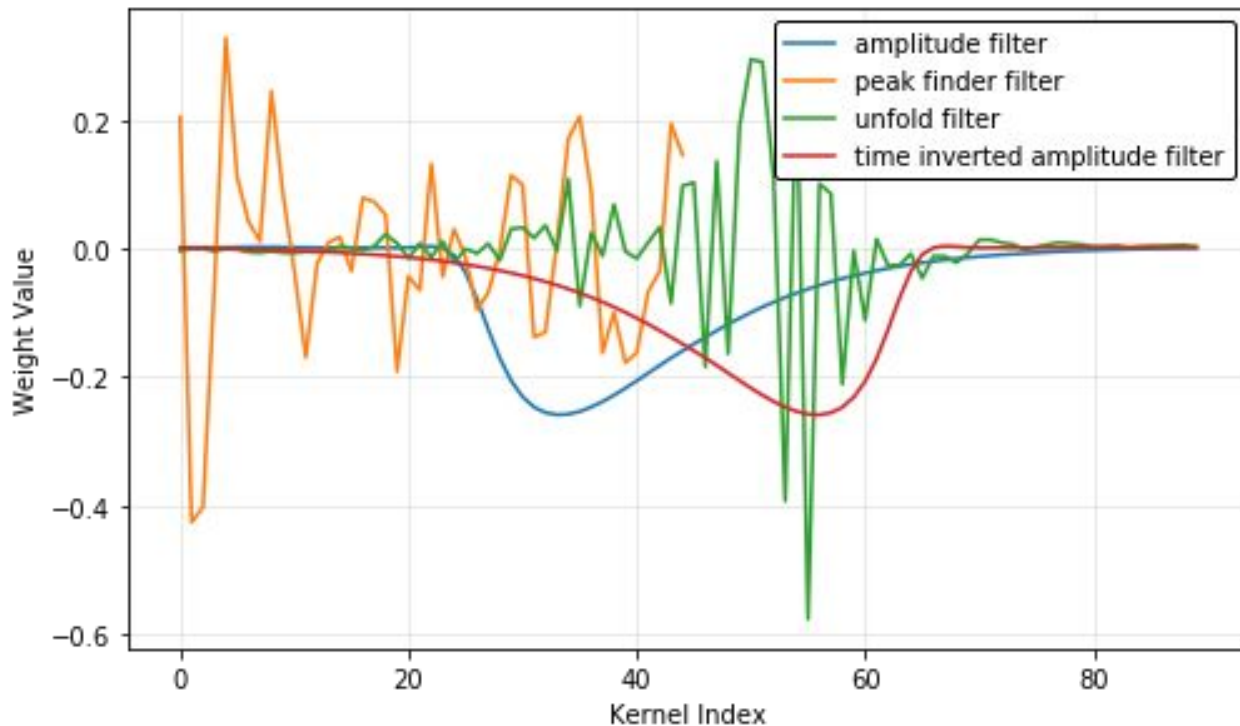
**SciPy = SciPy's peak finder**

# Parameter accuracy



	std. dev. amplitude	std. dev. timing
Our model	0.10	0.89
Scipy	0.03	1.89

# Can we understand what it's learnt?



- **Amplitude:** Pulse shape at the right hand edge. Zero-mean ringing ?
- **Reversed Amplitude:** matching input
- **Pulse finder:** second derivative of rising edge of pulse
- **Unfold layer:** Looks roughly like I'd expect but shifted to edge rather than centred

# What now and next

## Now:

- Improve stability and sparsity
- Test against more realistic scenarios
  - real data
  - real algorithms for comparison
- Finalise built-in pulse classification (see backups)

## Next:

- Extend to target multiple types of pulse shapes (layers with multiple kernels)
- Multiple simultaneous waveforms (e.g. output of many PMTs)



**Thank You**

# Analytical fit

Define chi-square for goodness of fit

$$\chi^2 = \sum (S_i - a_{\text{fit}} T_i)^2$$

Minimal chi-square value when

$$\frac{d\chi^2}{da} = -\sum_i T_i (S_i - a T_i) = 0$$

So the least-squares fitted amplitude is given (semi-analytically) by:

$$a = \frac{\sum_i T_i S_i}{\sum_j T_j^2}$$

Which is just a weighted sum over samples in the waveform



# Pulse separation

## Can invert the output of the matched filter and disentangle the contributions of each pulse

A two-pulse signal would be formed by:

$$S_i = aT_{i-k} + bT_{i-m} + \epsilon_i$$

So the matched filter output becomes:

$$F_j = \sum T_i S_{i-j} = aT_{k-j}^2 + bT_{m-j}^2 + E_j$$

Where for simplicity we define:

$$T_k^2 = \sum_i T_i T_{i-k} \quad (\text{the auto-correlation function})$$

The response of the filter at the two pulse times is given by

$$\begin{pmatrix} F_k \\ F_m \end{pmatrix} = \begin{pmatrix} T_0^2 & T_{k-m}^2 \\ T_{k-m}^2 & T_0^2 \end{pmatrix} \begin{pmatrix} a \\ b \end{pmatrix}$$

Now invert and solve for  $a$  and  $b$

# Pulse Classification

- The Matched Filter gives the best signal-to-noise ratio
  - If the template used is the same as the underlying signal pulse shape
- Return to the chi-square, which is a measure of the noise:

$$\chi^2 = \sum_i (S_i - a_{\text{fit}} T_i)^2$$

So minimising the chi-square leads to the matched filter output

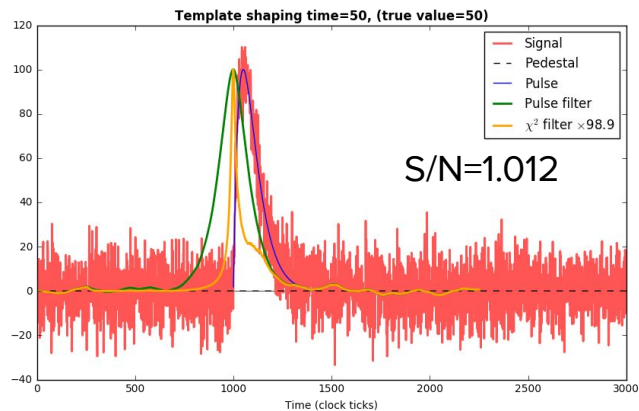
$$a = \frac{\sum_i T_i y_i}{\sum_j T_j^2} \implies F_j = \sum_i T'_i S_{i-j}$$

Expand the chi-square formula, and replace 'a' with the matched filter output

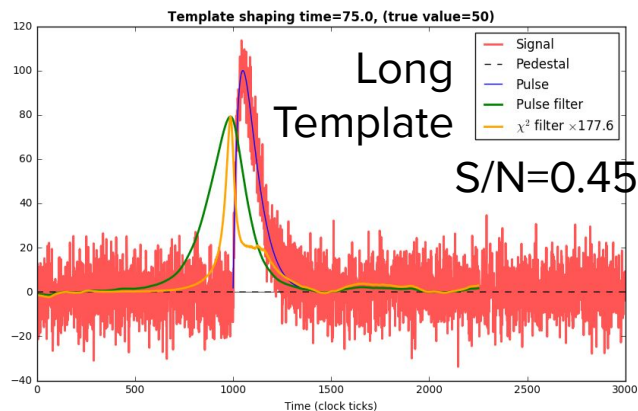
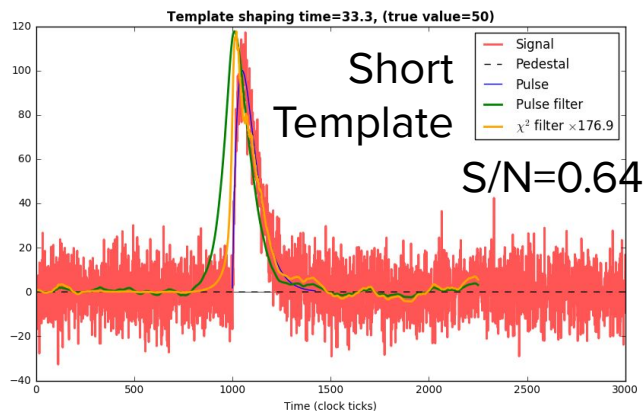
And finally define a signal-to-noise filter

$$C_i = \frac{1}{N-1} \left( \sum_{j=0}^N y_{j-i}^2 - F_i^2 \sum_{j=0}^N T_j^2 \right) \longrightarrow \text{SN}_i = \frac{F_i (N-1)}{\sum_{j=0}^N y_{j-i}^2 - F_i^2 T_0^2}$$

# Pulse classification demo



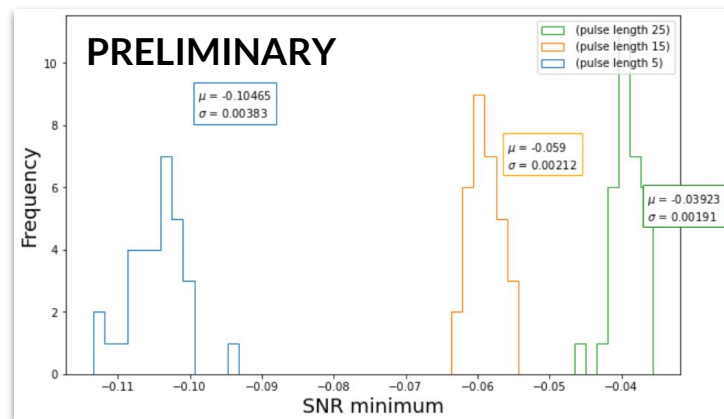
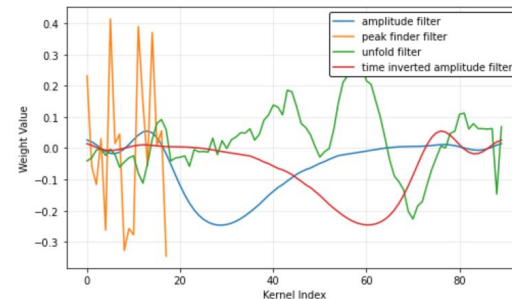
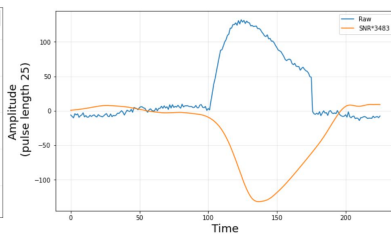
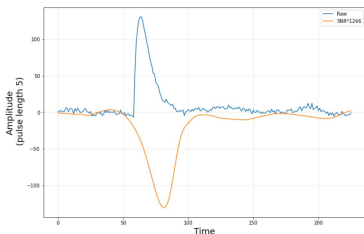
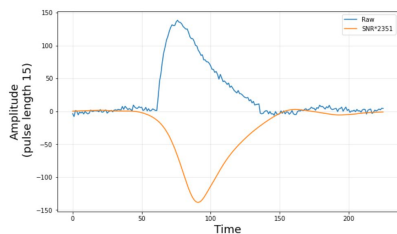
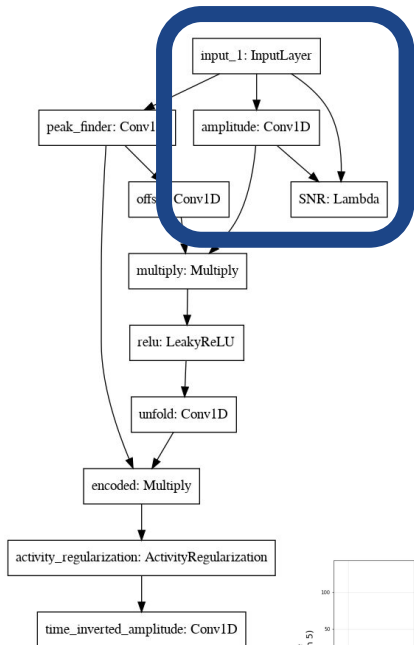
Correct  
Template



# SNR activation in the model

Train on pulses with shaping time = 15

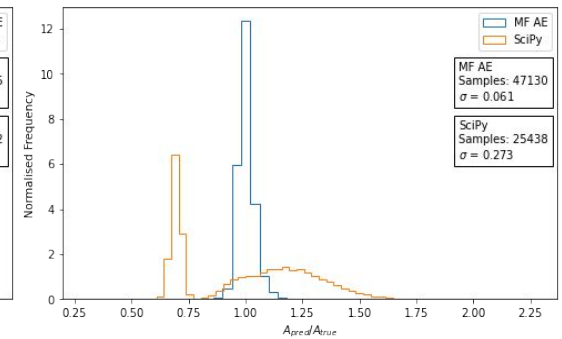
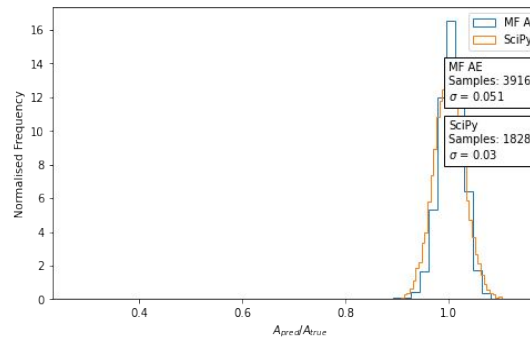
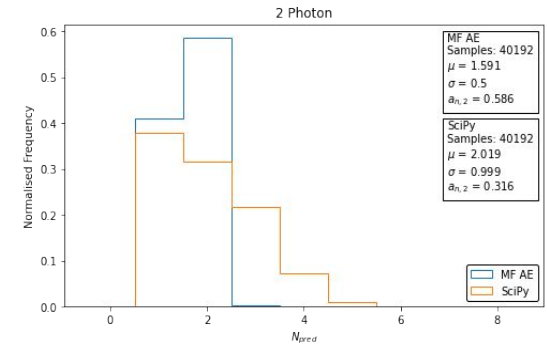
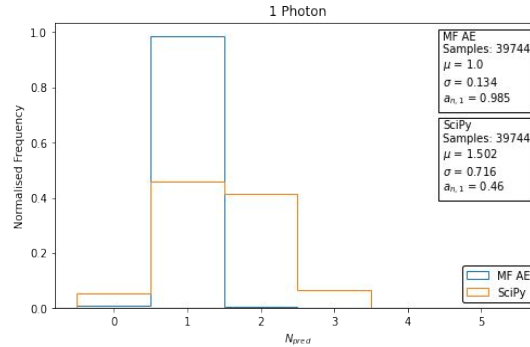
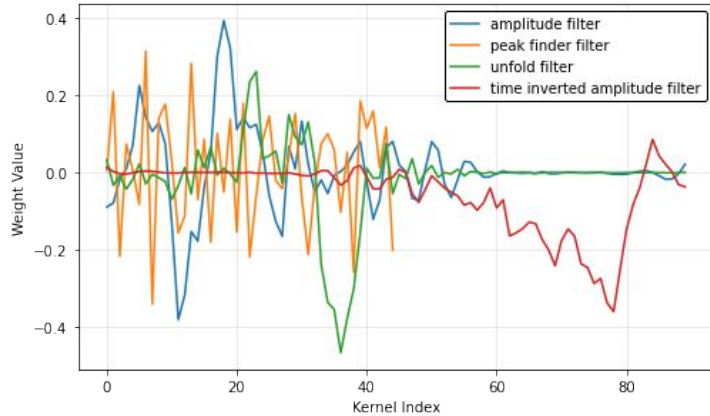
How does the SNR filter respond to different shaping times?



Issue in our input pulse normalisation: don't have the same "signal strength" for pulses with different shape..

⇒ "Does SNR activation of model trained on different pulse shapes correctly identify if a pulse matches its training set?"

# Interpretability vs performance



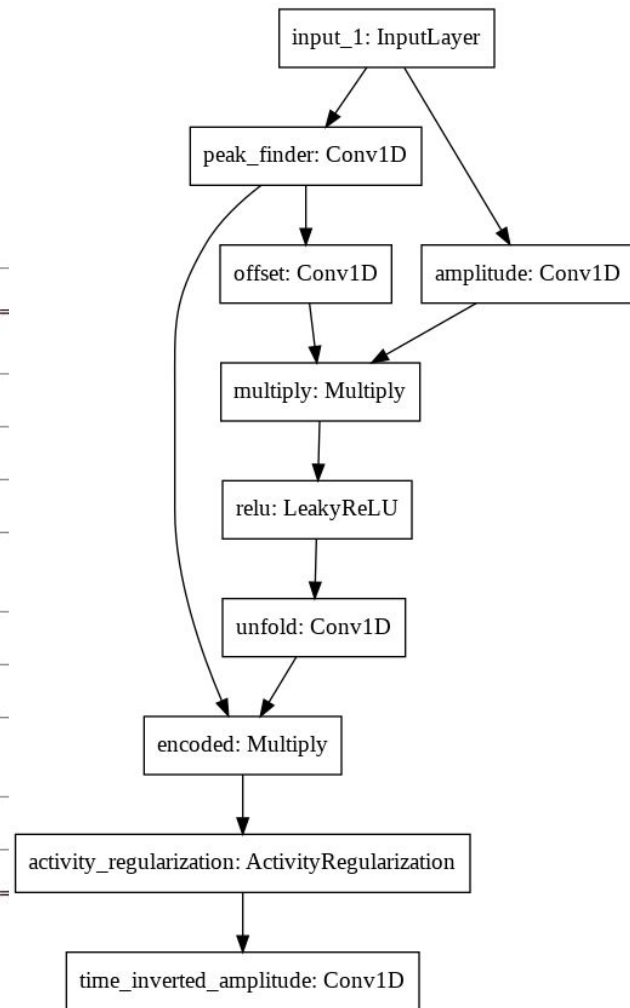
Relaxing desire for obvious interpretation can improve performance

# Which makes for a heavily constrained model

Layer (type)	Output Shape	Param #	Connected to
input_1 (InputLayer)	[(None, None, 1)]	0	
peak_finder (Conv1D)	(None, None, 1)	19	input_1[0][0]
amplitude (Conv1D)	(None, None, 1)	91	input_1[0][0]
offset (Conv1D)	(None, None, 1)	91	peak_finder[0][0]
multiply (Multiply)	(None, None, 1)	0	amplitude[0][0] offset[0][0]
relu (LeakyReLU)	(None, None, 1)	0	multiply[0][0]
unfold (Conv1D)	(None, None, 1)	91	relu[0][0]
encoded (Multiply)	(None, None, 1)	0	peak_finder[0][0] unfold[0][0]
activity_regularization (ActivityRegularization)	(None, None, 1)	0	encoded[0][0]
time_inverted_amplitude (Conv1D)	(None, None, 1)	91	activity_regularization[0][0]

Total params: 383  
 Trainable params: 292  
 Non-trainable params: 91

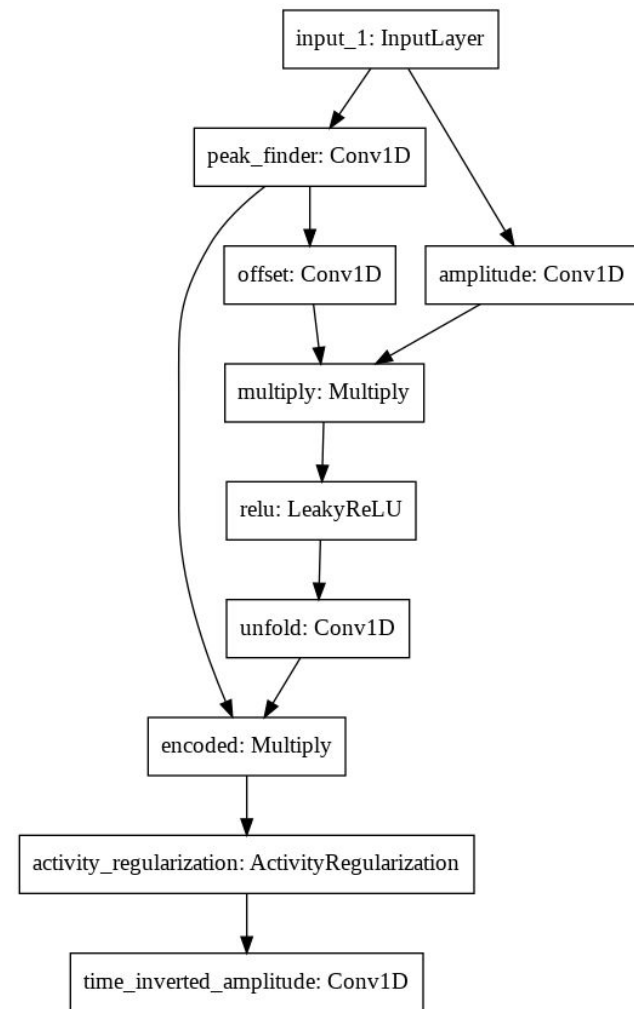
<300 trainable weights



# Optimising the performance

There are many hyperparameters to be tuned:

- Regularisation strengths:
  - L1 on the encoded space activation  
→ Makes it “sparse”
  - L2 on the derivative of the amplitude layer weights → make this smooth
  - Mirror symmetry between input / output amplitude layers
- Architecture:
  - Lengths of the layers (3 params)
  - Activation functions for “extremising” the pulse finding layer
  - Relative alignment of filter outputs (2 params)
- Learning parameters:
  - Batch sizes, learning rate, early stopping, etc

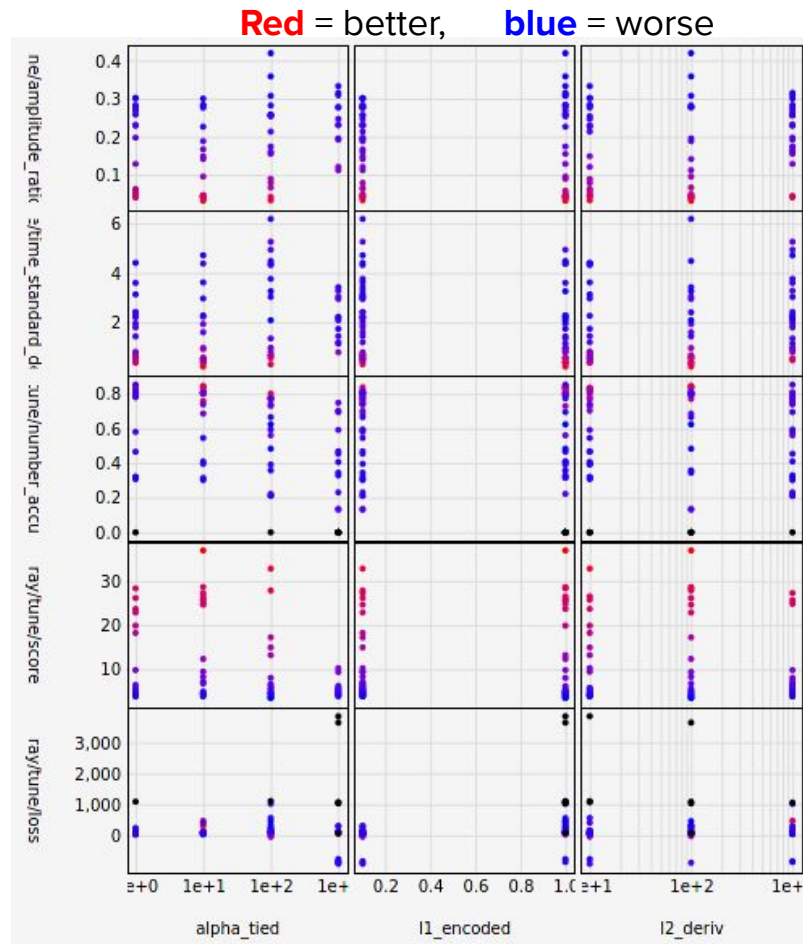


# Optimising the performance

## Some preliminary conclusions:

- No single parameter studied strongly correlated to performance
  - Need to study learning parameters
- Stability is difficult:
  - For some points, model often collapses to predict nothing (vanishing gradients...?)

Different performance metrics





# Generating input data

## **Toy detector sim** for development:

- Input = impulse delta functions + white noise
  - Final waveform =  
input \* convolve w. pulse shape + white noise
- Pulse shape:  $T_i = x_i A e^{(1 - \frac{x_i}{\tau})}$
- Shaping time, tau = 8 samples

## **Underlying waveform truth** (not for training):

- 1 or 2 pulses per waveform
- Randomised pulse times

