ORCHID: FLEXIBLE AND DATA-DEPENDENT CONVO-LUTION FOR SEQUENCE MODELING

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Abstract

In the rapidly evolving landscape of deep learning, the quest for models that balance expressivity with computational efficiency has never been more critical. Orchid is designed to address the quadratic computational complexity of attention models without sacrificing the model's ability to capture long-range dependencies. At the core of Orchid lies the data-adaptive convolution layers, which conditionally adjust their kernels based on input data using a conditioning neural network. This innovative approach enables the model to maintain scalability and efficiency for long sequence lengths. The adaptive nature of data-adaptive convolution kernel combined with the gating operations allows it to offer a highly expressive neural network. We rigorously evaluate Orchid across multiple domains, including language modeling and image classification, to showcase its generality and performance. Our experiments demonstrate that Orchid not only consistently outperforms traditional attention-based architectures in most scenarios but also extends the feasible sequence length beyond the constraints of dense attention layers. This achievement marks a significant milestone in the pursuit of more efficient and scalable deep learning models for sequence modeling.

1 INTRODUCTION

In the realm of modern deep neural networks, attention mechanisms have emerged as a gold standard, pivotal in domains such as natural language processing, image, and audio processing, and even complex fields like biology Vaswani et al. (2017); Dosovitskiy et al. (2020); Dwivedi & Bresson (2020). Despite their strong sequence analysis capabilities, these mechanisms face challenges, especially the computational complexity of Transformers, which scales quadratically with sequence length, creating significant hurdles in long-context tasks Dao et al. (2022); Chen et al. (2021a;b). In response to these computational challenges, the research community has explored alternatives to traditional dense attention layers. Methods like linear attention, sparse, and low-rank approximations of attention have been developed to reduce the computational complexity of attention layers in deep neural networks, enhancing scalability to larger sequences Child et al. (2019); Wang et al. (2020); Kitaev et al. (2020); Zhai et al. (2021); Schlag et al.. However, while these methods significantly reduce computational overhead, they often have lower expressiveness and performance.

Addressing the need for expressive, sub-quadratic, and hardware-efficient mixing operators is a formidable challenge. Recent studies have introduced sub-quadratic sequence mixing with long convolutions or state space models as potential solutions Gu et al. (2021); Romero et al. (2021); Mehta et al. (2022); Wang et al. (2022); Poli et al. (2023); Fu et al. (2023a;b). Orchid marks a significant advancement by offering an expressive, sub-quadratic primitive based on input-adaptive convolution, providing a robust alternative to the traditional Transformer paradigm and laying the foundation for further advancements in efficient modeling, opening new pathways to address the computational challenges in deep learning.



Figure 2.1: Orchid block. \odot and * denote element-wise multiplication and the convolution operator, respectively. The convolutions are implemented in the frequency domain using FFT. On the right two different conditioning networks, introduced in equations (2) and (3) as shift-invariant convolution kernels, are depicted.

2 PRELIMINARIES

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Self-Attention Mechanism: Given a length-*L* sequence of embeddings (of tokens) $x = (x_1, x_2, ..., x_L)$, self-attention generates a new sequence by computing a weighted sum of these embeddings.¹ It does this by linearly projecting x into three components: queries (Q), keys (K), and values (V), i.e., $Q = xW^Q$, $K = xW^K$, $V = xW^V$. Each head of self-attention can be expressed as a dense linear layer as follows:

$$y = \text{SelfAttention}(Q, K, V) = \text{SoftMax}\left(\frac{QK^T}{\sqrt{d_k}}\right)V = A(x)xW^V,$$

where the matrix A(x) is populated with the attention scores between each pair of tokens. This description of the attention layer highlights its notable benefits, including its capability to capture long-range dependencies efficiently, with a sublinear parameter count. The attention mechanism enables direct computation of interactions between any two positions in the input sequence, regardless of their distance, without a corresponding rise in parameter counts. Additionally, the attention layer implements a *data-adaptive* dense linear filter, effectively filtering the input while the filter weights are conditioned by a mapping of the data. However, these merits come at the expense of quadratic computational complexity and memory costs.

This motivates us to develop a scalable and efficient *data-adaptive convolution* mechanism, featuring an adaptive kernel that adjusts based on the input data. The kernel size of this convolution layer is as long as the input sequence length, allowing the capture of long-range dependencies across the input sequence while maintaining high scalability.

Linear Convolution: Discrete-time linear convolution is a fundamental operation in digital signal processing that computes the output as the weighted sum of the finite-length input x with shifted versions of the convolution kernel, h, also known as the impulse response of a linear time-invariant (LTI) system, formally as $\mathbf{y}[t] = (\mathbf{h} * \mathbf{x})[t] \triangleq \sum_{\ell=0}^{L-1} h[t-\ell]x[\ell]$. Circular convolution is defined as $\mathbf{y}[t \mod L] = (\mathbf{h} \circledast \mathbf{x})[t] \triangleq \sum_{\ell=0}^{L-1} h[t-\ell] x[\ell]$, which is equivalent to the linear convolution of two sequences when one is padded cyclically.

Fast convolution algorithm: One key advantage of convolution operators is that, according to the *convolution theorem*, they can be performed in the frequency domain, hence can be computed efficiently in $\mathcal{O}(L \log L)$ time using Fast Fourier Transform (FFT) algorithms. Generally speaking, convolution can be performed as $\hat{y} = \mathcal{F}^{-1}(\mathcal{F}(\hat{h}) \odot \mathcal{F}(\hat{x})) = T^{-1}(h_{\mathcal{F}} \odot T\hat{x})$, where T is the DFT matrix, \mathcal{F} denotes the discrete Fourier transformation, $\hat{x} = pad(x)$ denotes the zero-padded signal

3 Orchid

3.1 DATA-ADAPTIVE CONVOLUTION FILTER

We claim that making the kernel of convolution data dependent, renders the layer more expressive

$$y = h_{\theta}(x) * x = \mathrm{NN}_{\theta}(x) * x \tag{1}$$

We call the function $h_{\theta}(x) = NN_{\theta}(x)$, conditioning network parameterized by θ . Hence given an input, x, this operation defines how each token attends to the entire signal as a weighted sum whose weights are conditioned on the input itself.

In general, a discrete convolution is shift equivariance, that is, ignoring boundary (edge) effects, if the input data is spatially shifted, the output of the model shifts by the same amount. Boundaries doesn't affect this property in circular convolution, hence $\operatorname{shift}_m(y) = h \circledast \operatorname{shift}_m(x)$ where $\operatorname{shift}_m(x)[t] \triangleq x[t+m]$ (Bronstein et al., 2021). This property ensures that the operation's output is robust to shift of features within the input, thereby enhancing the model's generalization capabilities. This capability (which induces an inductive bias on the model) is at the core of the widespread success of convolution operations Thomas et al.. Therefore it is desirable to design conditioning network in the data-adaptive convolution operation it is sufficient to design filter kernel to be *shift invariant*. In the following, we present two methods for designing a shift-invariant conditioning network.

I) Suppressing the Phase of Frequency Components: A circular shift of a sequence u corresponds to multiplying its frequency components by a linear phase, *i.e.* $\mathcal{F}(\operatorname{shift}_m(u))[k] = u_{\mathcal{F}}[k] \cdot e^{-\frac{i2\pi}{L}km}$ Oppenheim (1999). Suppose g(x) is a shift-equivariant function (such as a depthwise Convld()), such that $g(\operatorname{shift}_m(x)) = \operatorname{shift}_m(g(x))$. The frequency components of g(x), when spatially shifted, can be expressed as: $\mathcal{F}(g(\operatorname{shift}_m(x))[k] = \mathcal{F}(g(x))[k] \cdot e^{-\frac{i2\pi}{L}km}$. Subsequently, by applying the absolute value (or the magnitude of complex numbers) non-linearity to the frequency components, we can achieve shift invariance. Defining $h_{\mathcal{F}}(x) = |\mathcal{F}(g(x))|$, it follows that $h_{\mathcal{F}}(\operatorname{shift}_m(x)) = h_{\mathcal{F}}(x)$, thereby satisfying shift invariance.

In our setup, we define g(x) as a 1D depth-wise linear convolution, denoted as Convld(x), with a short kernel length (typically 3-5) for each feature dimension, which is followed by a short convolution in the frequency domain. Consequently, the conditioning neural network is formulated as

$$h_{\theta}^{\mathcal{F}}(\boldsymbol{x}) = \operatorname{Convld}(|\mathcal{F}(\operatorname{Convld}(\boldsymbol{x}))|)$$
(2)

This architecture choice minimizes the number of parameters and reduces the computational burden that the conditioning network introduces to the overall model.

II) Using Cross-Correlation to Achieve Shift Invariance An alternative method to attain shift invariance involves computing the cross-correlation between two versions of a signal. Consider $k(\boldsymbol{x})$ and $q(\boldsymbol{x})$ as two shift-equivariant functions, satisfying: $k(\operatorname{shift}_m(\boldsymbol{x})) = \operatorname{shift}_m(k(\boldsymbol{x}))$ and $q(\operatorname{shift}_m(\boldsymbol{x})) = \operatorname{shift}_m(q(\boldsymbol{x}))$. Define $h(\boldsymbol{x})$ as the cross-correlation of $k(\boldsymbol{x})$ and $q(\boldsymbol{x})$, given by: $h(\boldsymbol{x})[t] = (k(\boldsymbol{x}) \star q(\boldsymbol{x}))[t] \triangleq \sum_{\ell=0}^{L-1} k(\boldsymbol{x})[\ell] \cdot q(\boldsymbol{x})[t+\ell \mod L]$. It can be demonstrated that $h(\boldsymbol{x})$ is shift invariant:

 $h(\mathrm{shift}_m(\boldsymbol{x})) = k(\mathrm{shift}_m(\boldsymbol{x})) \star q(\mathrm{shift}_m(\boldsymbol{x})) = \mathrm{shift}_m(k(\boldsymbol{x})) \star \mathrm{shift}_m(q(\boldsymbol{x})) = k(\boldsymbol{x}) \star q(\boldsymbol{x}) = h(\boldsymbol{x})$

Moreover, according to the convolution theorem, the cross-correlation can be efficiently computed in the frequency domain as $h_{\mathcal{F}}(\boldsymbol{x}) = \mathcal{F}(k(\boldsymbol{x}) \star q(\boldsymbol{x})) = k_{\mathcal{F}}^*(\boldsymbol{x}) \odot q_{\mathcal{F}}(\boldsymbol{x})$ where $k_{\mathcal{F}}^*$ denotes the complex conjugate of $k_{\mathcal{F}}$

Remark 3.1. By using the same function for both k and q, *i.e.* k(x) = q(x) = g(x), we derive $h_{\mathcal{F}}(x) = |g_{\mathcal{F}}(x)|^2$, implying that the cross-correlation-based approach generalizes approach (I).

In a similar manner, we leverage distinct 1D depth-wise short convolutions for k(x) and q(x) for both k(x) and q(x), followed by another convolution post cross-correlation in the frequency domain. As a result, the conditioning neural network is defined as

$$h_{\theta}^{\mathcal{F}}(\boldsymbol{x}) = \operatorname{Convld}\left(\mathcal{F}^{*}\left(\operatorname{Convld}(\boldsymbol{x})\right) \odot \sigma\left(\mathcal{F}\left(\operatorname{Convld}(\boldsymbol{x})\right)\right)\right). \tag{3}$$

The two conditioning functions (2), (3) are also illustrated in Figure 2.1. For operations involving long convolutions, we add a fixed (non data-adaptive) term which is implicitly parametrized using a positional embedding of time step (token index in the sequence) and a feed forward networks (FFNs) Romero et al. (2021); Poli et al. (2023) h(t) = FFN(PositionalEmbedding(t))). Subsequently, the final convolution kernel is $h = h(t) + h_{\theta}(\mathbf{x})$.

3.2 ORCHID BLOCK

In contrast to attention layers, convolution filters perform parameter sharing, meaning they utilize the same kernel weights across different positions within the input sequence. By integrating this data-adaptive convolution approach with element-wise multiplications, it's possible to achieve a location-dependent filtering scheme. Through element-wise multiplication, specific locations within the signal can be emphasized by assigning higher weights before applying the location-invariant convolution. The composition of a cascade of circulant and diagonal matrices has been demonstrated to serve as an efficient approximation for dense linear layers Moczulski et al. (2015); Cheng et al. (2015). The overall architecture of a block, incorporating our data-adaptive convolution block, is illustrated in Figure 2.1.

4 EXPERIMENTS

Our evaluation of Orchid focuses on three different Transformer-based models to assess its expressivity and generalization capabilities as an alternative to attention layers. Firstly, we conduct a set of experiments on a synthetic task to assess the **in-context learning ability** (Liang et al., 2022; Olsson et al., 2022) and scalability of the proposed model. It involves generating a value from a key given a string of key-value tuples from a random dictionary. For instance, given the input ([a, 1, b, e, 3, f], b), the model is expected to return e, the value associated with the key b. The results, illustrated in Figure 4.1 and Table 4.3, demonstrate that the Orchid model offers superior expressiveness and outperforms existing long convolution models in the associative recall task. Notably, in challenging scenarios with short sequence lengths of 128 and large vocabulary sizes, Orchid significantly improves the model's accuracy.

Subsequently, we evaluate the performance of the proposed architecture on **language modeling tasks**. Orchid is designed to integrate seamlessly with existing BERT-style language models, such as BERT Devlin et al. (2018), As the results outlined in Table 4.1 show, Orchid-BERT-base achieves 1.0 points in average GLUE score performance compared to the BERT-base on the GLUE benchmark with utilizing 30% fewer parameters. Similarly, Orchid-BERT-large outperforms the performance of BERT-large by 1.0 points with a 25% reduction in parameter counts.

Moreover, we extend our experiments to the Vision Transformer (ViT) architecture (Dosovitskiy et al., 2020) for **image classification tasks**, aiming to evaluate the model's generalizability across diverse domains. Our experiments are conducted on two widely recognized image datasets: CIFAR-10 and ImageNet-1K. The performance outcomes, as highlighted in Table 4.2 for both the CIFAR-10 and ImageNet-1K datasets, demonstrate that Orchid significantly outperforms baseline ViT-style models on both datasets. These results affirm the adaptability and effectiveness of the Orchid architecture beyond the realm of language modeling, showcasing its potential advantages in image processing tasks.

Model (size)	GLUE Score	Δ Params	Δ GLUE Score
BERT-base (110M) M2-BERT-base (80M) Orchid-BERT-base (77M)	79.6 79.9 80.6	-27.3% - 30.0%	+0.3 +1.0
BERT-large (340M) M2-BERT-large (260M) Orchid-BERT-large (254M)	82.1 82.2 82.7	-23.6% -25.3%	+0.1 +0.6

Table 4.1: Average GLUE Score of BERT-base and BERT-large (Devlin et al., 2018) in comparison to Orchid-BERT-base and Orchid-BERT-base, and M2-BERT-base and M2-BERT-large Dao et al. (2022). Baseline results are drawn from (Fu et al., 2023a).

Table 4.2: Performance comparison of Orchid with ViT-based models on ImageNet-1k and CIFAR-10 dataset. Baseline results are drawn from (Fu et al., 2023a).

Model (size)	Top-1 (%)	Top-5 (%)	Model (size)	Top-1 (%)
ImageNet-1k			CIFAR-10	
ViT-b (87M)	78.5	93.6	ViT (1.2M)	78.6
ViT-b + Monarch (33M)	78.9	94.2	ViT + Monarch (607K)	79.0
Hyena-ViT-b (88M)	78.5	93.6	Hyena-ViT (1.3M)	80.6
M2-ViT-b (45M)	79.5	94.5	M2-ViT (741K)	80.8
Orchid-ViT-b (48M)	80.2	94.9	Orchid-ViT (836K)	84.3



Figure 4.1: Performance of the associative recall task across different long implicit convolution models on various sequence lengths and vocabulary sizes (number of possible token values).

Table 4.3: This table shows the performance of in-context learning on the associative recall task with a vocabulary size of 20 and different sequence lengths. The results for the baseline models are drawn from Poli et al. (2023); Fu et al. (2023a). The symbol X indicates that the Transformer model failed to complete the task within a week or the model d oes not fit in memory.

Model	128	512	2K	8K	32K	128K
Transformer	100	100	100	100	×	×
Monarch-Mixer	-	98.7	99.4	99.4	99.4	99.4
Hyena	93	99	99.6	100	100	-
Orchid	99.2	99.8	100	100	100	100

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