UberShuffle: Faster Distributed Learning via Erasure Coded Data Shuffling

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Abstract

Modern large-scale learning algorithms are deployed on hundreds of distributed compute instances, each computing gradient updates on a subset of the training data. It has been empirically observed that these algorithms can offer higher statistical performance when the training data is *shuffled* once every few epochs. However, *data shuffling* is often avoided due to its heavy communication costs. Recently, erasure coding ideas have been proposed to minimize the communication cost of shuffling, but have been only successful in theory. In this work, we implement *UberShuffle*, a new erasure coded shuffling system and compare its performance against various shuffling algorithms in distributed learning setups. We observe that our shuffling framework for machine learning can achieve significant speed-ups compared to the state of the art. In some cases, the data shuffling time is reduced by about 50%, and the training time is reduced by about 30%.

1 Introduction

Distributed machine learning systems are becoming increasingly popular due to their promise of high scalability and substantial speedups gains. In a prototypical distributed learning setup, each worker computes gradient updates on parts of the dataset, and these updates are periodically synchronized at a master node (*i.e.*, parameter server). Recent works show that if the training data set is reshuffled across the worker nodes, this can lead to superior convergence performance [26, 33]. In practice, however, data shuffling incurs a huge communication cost and many practitioners avoid it.

One of the first coded shuffling algorithms for distributed machine learning was proposed in [17] to leverage the local caches of the worker nodes and curtail the communication cost of the shuffling process. Based on a novel coding technique, the master node broadcasts linear combinations of data points, which are carefully designed such that every worker can successfully decode its allocated batch. The authors show that such coded algorithm can –in theory– reduce the communication overhead by a factor of $\Theta(n)$, where the number of workers is n. However, the practical efficacy of the coded shuffling algorithm has not been demonstrated yet. Indeed, the theoretical guarantees of [16, 17] hold only when 1) the number of data points is approaching infinity, and when 2) there exists a perfect broadcast channel between the master node and the workers. The goal of this work is to exhibit that erasure coded algorithms for data shuffling can indeed lead to significant performance gains in practice.

In this work, we present a new and implementable coded shuffling algorithm, called UberShuffle, based on the original work of [17]. We implement a distributed machine learning system that can run generic distributed machine learning algorithms combined with any shuffling procedure. Using

¹In [17], the authors show that one can achieve the reduction factor of $\Theta(n/\log n)$ without broadcast channel.

this system, we compare the performance of different shuffling algorithms under various setups, and show that the coded shuffling algorithms can achieve significant speed-up gains in practice.

1.1 Related Works

Distributed Machine Learning and Data Shuffling Distributed gradient decent methods on networked systems have been extensively studied in the literature [4,5,7,9–11,15,18,21,23,31]. When running distributed gradient descent algorithms, periodic shuffling of the training data is observed to achieve large statistical gains [6,12,14,26,27,32,33] Further, it has been shown that the stochastic gradient method with without-replacement sampling converges faster than the other with withreplacement sampling [25,28]. Our shuffling algorithm can be viewed as a cost-efficient enabler of the without-replacement sampling across the distributed nodes.

Coding Shuffling for Distributed Machine Learning Indeed, our work falls under the umbrella of a recent paradigm of 'codes for distributed machine learning', proposed in [16, 17]. In the aforementioned work, the authors propose two different algorithms based on coding-theoretic ideas, called 'coded shuffling' and 'coded computation'. The coded shuffling algorithm alleviates communication bottlenecks of distributed computing networks by leveraging excess storage. In [16], the authors analyze the transmission rate of the coded shuffling algorithm assuming that 1) the number of data points grows to infinity and 2) a perfect broadcast channel is given between master and workers. They also provide simulation results based on the expected performance of the coded shuffling algorithm and the statistical gains of without-replacement sampling, showing that coded shuffling algorithm can help parallel/distributed gradient methods converge faster. However, the authors did not provide the actual experimental results, and hence it is not clear how well the coded shuffling algorithm would perform in practice, especially when the number of data points is finite and a perfect broadcast channel is not given. Our UberShuffle algorithm is a strict improvement over the original coded shuffling algorithm, which can improve the performance with a finite number of data points. Further, we also show that one can achieve the promised performance gain up to a factor of 2 even without a perfect broadcast channel by leveraging an efficient multicast algorithm. The details are provided in Sec. 3.

A few recent works also study the fundamental limits of the coded shuffling problem [1,2]. In [1], the authors characterize the information-theoretic limits of the coded shuffling problem for n=2 and n=3, and in [30], the authors consider the worst-case shuffling performance and propose a new coded shuffling algorithm, designed to perform well under the worst-case scenario.

We note that a similar idea that uses coding ideas to reduce the shuffling traffic is also proposed in [19]. More precisely, the authors consider the MapReduce framework, in which distributed workers first perform 'map' phase, then shuffle the results of the map phase between the workers, and then perform 'reduce' phase to finish the computation. The key idea is to assign duplicate map tasks across the workers to generate redundant map results. Viewing these redundant computation results as side-information, they apply the coding-theoretic ideas to reduce the bandwidth required to shuffle the map results between the workers. Even though this algorithm seems very similar to ours, there are a few key differences. First, our algorithm is specifically designed to speed up the convergence of distributed machine learning algorithms while theirs is for MapReduce tasks. Further, our algorithm makes use of the excess memory/storage to generate common information while their algorithm generates common information by increasing the computational load of the map phase by a constant factor. Thus, our algorithm can always speed up distributed machine learning algorithms as long as excess memory/storage is available while theirs is applicable only when the computational load of the map phase is lower than a certain threshold. For instance, the load of the map phase is observed to be less than 0.2% of that of the shuffling phase in their experiments [20].

Efficient Multicast Algorithms In [3], the authors propose a multicast algorithm that transmits data from a node to multiple other nodes in about twice of the unicast time. Using this algorithm, we analyze the shuffling time for coded shuffling algorithm when used without a perfect broadcast channel. In Sec. 3, we provide the details of the multicast algorithm and new guarantees of the coded shuffling algorithm when used without a broadcast channel.

2 System Model and Data Shuffling Algorithms

In this section, we first describe the system model and then provide informal illustrations of the shuffling algorithms, proposed in [17], using a running example. Consider a master-worker distributed computing environment with n distributed workers and a master. The master has access to the entire data set, consisting of q (unit-sized) data points $(d_1, d_2, ..., d_q)$. At the beginning of each epoch, the system requires all of these data points to be randomly redistributed across n workers. The master node first draws a random assignment of each data point such that all workers are assigned the same number of data points. Let us denote the destination of data point d_i by $w_i \in [n]$. We denote by D_i the set of data indices assigned to worker i, i.e., $D_i = \{j \mid w_j = i\}$. We also assume that each worker can cache up to $\lfloor \alpha q \rfloor$ data points. We denote the set of data indices that is cached in worker i's cache by C_i .

2.1 The uncoded shuffling algorithm

At the beginning of epoch t+1, the master node first draws a new data assignment w^t , and then computes $U_i^{t+1} := D_i^{t+1} \setminus C_i^t$ for each i. Note that U_i^{t+1} is a set of data points that are required by worker i for epoch t+1 but are not stored in worker i after epoch t. Under the uncoded shuffling algorithm, the master node shuffles data points simply by transmitting U_i^{t+1} to worker i for each i. It is clear that every worker is able to obtain its own data set because $D_i^{t+1} = U_i^{t+1} \cup \left(C_i^t \cap D_i^{t+1}\right)$ for all i.

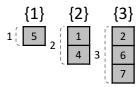


Figure 1: Illustration of the encoding table for uncoded shuffling algorithm.

Example 1 (The uncoded shuffling algorithm). Consider a case where n=3, q=9, and $\alpha=0.44$. Further, assume that $C_1=\{2,3,4,8\}, C_2=\{6,7,8,9\}, C_3=\{1,3,4,5\}$ and $D_1=\{3,5,8\}, D_2=\{1,4,9\}, D_3=\{2,6,7\}$. The uncoded shuffling algorithm will incur 6 unicasts: d_5 to worker 1, d_1 , d_4 to worker 2, and d_2 , d_6 , d_7 to worker 3. See Fig. 1 for visualization.

In general, the communication cost of the uncoded shuffling algorithm can be shown as following theorem [16].

Theorem 1 (Uncoded Shuffling Rate). Total communication rate (in data points transmitted per epoch) of the uncoded shuffling scheme is

$$R_u = q(1 - s/q). (1)$$

Proof. Since the new permutation S_i^{t+1} is picked uniformly at random, s/q fraction of the data rows in S_i^{t+1} are present in current cache C_i^t . Hence, the number of data that each worker need becomes $|C_i^t \cap S_i^{t+1}| = \frac{q}{n}(1-\frac{s}{q})$. Thus, the master node needs to transmit $n \times \frac{q}{n}(1-\frac{s}{q}) = q(1-\frac{s}{q})$ data points in total.

2.2 The coded shuffling algorithm

The key idea of the coded shuffling algorithm is simple: instead of transmitting data points one by one, the master node linearly combines multiple data points and broadcasts a fewer number of *coded* data points. The coded shuffling algorithm judiciously encodes the data points such that each worker can obtain new data points by decoding the received message with the aid of cached data points.

 $^{^2}$ When $\alpha=1$, every node can store the entire dataset, and hence shuffling is not needed anymore. However, this is not feasible if the dataset is too large to fit in a single memory. (For instance, the size of Google Books Ngram is 2.2TB [22].) One may achieve $\alpha=1$ by storing the entire dataset in large storage units (such as SSD/HDD) but this approach is either cost-inefficient or slow due to excessive storage I/O overhead.

Algorithm 1 Coded Shuffling: Indexing algorithm at master

The coded shuffling algorithm consists of 3 parts: indexing algorithm where encoding table is generated, encoding algorithm where data are encoded based on encoding table, and decoding algorithm where encoded data is decoded to obtain needed data.

Indexing Algorithm. At the beginning of epoch t+1, the master node first constructs a table that contains the information about which data points are cached by which workers after epoch t. This information can be represented by a binary-valued matrix $\mathbf{C}^t \in \{0,1\}^{n \times m}$, whose (i,j)-th element represents whether data point j is cached at worker i (1) or not (0). Consider data point j, which is assigned to worker $\pi(j)$ for epoch t+1. The master node finds the set of workers in which this data point is cached, i.e., $\{i: \mathbf{C}_{ij}^t = 1\}$, and assigns it to a tuple $(\mathcal{I}, \pi(j))$, where $\mathcal{I} := \{i: \mathbf{C}_{ij}^t = 1\} \cup \pi(j)$. We define $\mathcal{I}(\mathcal{I}, i)$ as the set of data points that are assigned to tuple (\mathcal{I}, i) . Note that all the data points in $\mathcal{I}(\mathcal{I}, i)$ are required by worker i, and cached by the workers in $\mathcal{I} \setminus \{i\}$.

Encoding Algorithm. After every data point is indexed in a similar way, the master node starts encoding the data points as follows. For each subset $\mathcal{I} = \{i_1, i_2, \dots, i_{|\mathcal{I}|}\}$ of [n], it first looks up the following assignment sets: $Z(\mathcal{I}, i_1), Z(\mathcal{I}, i_2), \dots, Z(\mathcal{I}, i_{|\mathcal{I}|})$. All the data points in these assignment sets will be multicasted to workers in \mathcal{I} . Given these sets, the master node removes one data point from each of the $|\mathcal{I}|$ assignment sets, and then generates an encoded data point by adding them up. This procedure is repeated until all the assignment sets become empty.

Decoding Algorithm. After multicasting is done, the worker nodes obtains the required data points by decoding the received packets based on the encoding table. For decoding, each worker i identifies the encoded data points including D_i^{t+1} , and decodes the required data by subtracting the data rows cached in the worker. Otherwise, the worker simply discards the data received.

Note that cache information binary matrix \mathbf{C}_{ij}^t and data assignment for next epoch π , generated by the master node, are broadcast to every workers. After receiving this information, each worker identifies assignment sets $Z(\mathcal{I},i)$ of the encoding table by running the indexing algorithm, given in Algorithm 1.

The pseudocode of the indexing, encoding and decoding algorithms are provided in Alg. 1, Alg. 2 and Alg. 3. We also note that the computational overhead of the indexing and encoding procedures is $\mathcal{O}(nq)$, which is observed to be negligible compared to the time to communicate the training data in Sec. 4.

We now illustrate the indexing, encoding, and decoding procedures of the algorithm in following toy example.

Example 2 (The coded shuffling algorithm). Consider the same scenario considered in the previous example shown in Example 1. Assume that the master node *broadcasts* $d_2 + d_5$. Since d_2 is stored in worker 1 and d_5 is stored in worker 3, worker 1 can obtain d_5 by subtracting d_2 from $d_2 + d_5$, and worker 3 can obtain d_5 similarly. Thus, the master node can reallocate two data points using a single broadcast transmission, reducing the communication overheads. The coded shuffling algorithm is a straightforward generalization of this idea, which we briefly explain below. Consider data point d_1 , which is required by worker 2 for the subsequent epoch of the learning algorithm. Observe that this data point is *exclusively* cached in worker 3. On the other hand, the data points

Algorithm 2 Coded Shuffling: Encoding algorithm at master

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\begin{array}{l} \textbf{procedure} \ \mathsf{ENCODING}(Z,A) \\ \textbf{for} \ \mathsf{each} \ \mathcal{I} \in [n]^n, \textbf{do} \\ Y \leftarrow (Z(\mathcal{I},\mathcal{I}_1), Z(\mathcal{I},\mathcal{I}_2), \dots Z(\mathcal{I},\mathcal{I}_{|\mathcal{I}|})) \qquad \rhd \mathcal{I}_i \ \mathsf{is} \ \mathsf{the} \ i^\mathsf{th} \ \mathsf{element} \ \mathsf{of} \ \mathcal{I} \ \mathsf{in} \ (\mathsf{some} \ \mathsf{order}) \\ \ell \leftarrow \max_{i=1}^{|\mathcal{I}|} |Y_i| \\ \textbf{for} \ t \in [1,2,\dots,\ell] \ \textbf{do} \\ \qquad \qquad \mathsf{broadcast} \ b_{\mathcal{I},t} = \sum_{i=1}^{\mathcal{I}} a_{(Y_i)_t} \ (\mathsf{with} \ \mathsf{metadata} \ Y_t) \qquad \qquad \rhd \mathsf{if} \ |Y_i| < t, \ a_{(Y_i)_t} = 0 \\ \textbf{end for} \\ \textbf{end procedure} \end{array}
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Algorithm 3 Coded Shuffling: Decoding algorithm at worker *i*

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\begin{array}{ll} \textbf{procedure} \ \mathsf{DECODING}(b,Y,\pi) & \rhd i \ \text{is the worker index} \\ J \leftarrow \{j:\pi(j)=i\} \cap Y & \rhd |J| \ \text{is always} \ 0 \ \text{or} \ 1 \\ \textbf{if} \ |J| = 0 \ \textbf{then} \\ \text{return null} & \textbf{else} \\ \text{return } b - \sum_{i \in Y \setminus J} a_i \\ \textbf{end if} \\ \textbf{end procedure} \end{array}
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 d_6 and d_7 are required by worker 3, and they are cached in worker 2. Hence, the master node can combine the data points from each of these groups and broadcast the encoded data points. One can visualize this indexing result in the table labeled as $\{2,3\}$ in Fig. 2. The first column of the table corresponds to the data points that are required by worker 2 and exclusively cached in $\{2,3\}\setminus\{2\}$, and the second column is for the data points required by worker 3 and exclusively cached in $\{2,3\}\setminus\{3\}$. By repeating this procedure for every data point, one can obtain all the encoding tables shown in Fig. 2. Once the encoding table is obtained, the master node simply generates one encoded packet per row of the encoding tables, and broadcasts them to the workers. By the property of the indexing procedure, every worker is guaranteed to obtain the desired data points by decoding the received packets. Here, note that the communication cost of the coded shuffling algorithm for this example is 4 transmissions.

The communication cost of the coded shuffling algorithm is characterized in [17].

Theorem 2 ([17] The coded shuffling algorithm). As q approaches infinity, the number of broadcasts of the coded shuffling algorithm is $R_c = \frac{q}{(np)^2} \left((1-p)^{n+1} + (n-1)p(1-p) - (1-p)^2 \right)$ per epoch, where $p = \frac{s-q/n}{q-q/n}$.

Proof. The authors of [17] shows that cardinality of set $S_i^{t+1} \cap \tilde{C}_{\mathcal{I} \setminus \{i\}}^t$ for $\mathcal{I} \in [n]$ with $|\mathcal{I}| \geq 2$ is

$$|S_k^{t+1}\cap \tilde{C}_{\mathcal{I}\backslash\{i\}}^t|\simeq \frac{q}{n}\times \frac{|\mathcal{I}|-1}{n}p^{|\mathcal{I}|-2}(1-p)^{n-(|\mathcal{I}|-1)},$$

when q gets large (and n remains sub-linear in q).

Since the master node will send $\sum_{i \in \mathcal{I}} A(S_i^{t+1} \cap \tilde{C}_{\mathcal{I} \setminus \{i\}}^t)$ for each subset \mathcal{I} with $|\mathcal{I}| \geq 2$, the total rate of coded transmission is

$$R_c = \sum_{i=2}^{n} {n \choose i} \frac{q}{n} \frac{i-1}{n} p^{i-2} (1-p)^{n-(i-1)}.$$

To complete the proof, above expression can be simplified using the following identity:

$$\sum_{i=2}^{n} \binom{n}{i} (i-1)x^{i-2} = \frac{1 + (1+x)^{n-1}(nx - x - 1)}{x^2}.$$

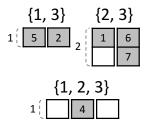


Figure 2: Illustration of the encoding table for coded shuffling algorithm.

This theorem implies that as q gets large, and n grows sublinearly in q, $R_c \to \frac{q(1-\alpha)}{\alpha n}$. Thus, assuming a broadcast channel of bandwidth B between the master node and the workers and denoting the shuffling time by T, we can characterize the shuffling times as follows.

Corollary 3 (Shuffling times with broadcasts). Given the broadcast bandwidth B, the shuffling time of the coded shuffling algorithm is $T_{broadcast, coded} \simeq \frac{q}{B} \frac{(1-\alpha)}{\alpha n}$. Similarly, the shuffling time of the uncoded shuffling algorithm is $T_{broadcast, uncoded} = \frac{q}{B} (1-\alpha)$, and hence $T_{broadcast, coded} \simeq \frac{1}{\alpha n} T_{uncoded}$.

3 Coded Shuffling in Practice

The coded shuffling algorithm cannot be immediately deployed in practice due to a few limitations. First, it assumes a perfect broadcast channel between the master and the workers, which is not available in many practical applications. Further, the theoretical guarantee holds only when the number of data points approaches infinity, and its performance with a finite number of data points is unknown. In this section, we propose a few modification to the coded shuffling algorithm to address these limitations, making the algorithm more applicable in practice.

3.1 Van de Gejin's pseudo-broadcast algorithm

In [16], it is assumed that the master can transmit a packet to all n workers in one time slot. However, in practice, nodes in clusters are usually connected to network via (bidirectional) single-ported network interfaces, i.e., at most one message can be sent to another node at full bandwidth and at most one message can be received from another node at full bandwidth. Hence, a natural question is whether one can still achieve the promised gain of the coded shuffling algorithm under such a practical setup. It turns out that under a mild assumption, one can make use of Van de Gejin's pseudobroadcast algorithm [3], which fully exploits interconnections between the workers, to achieve the gain of the coded shuffling algorithm.

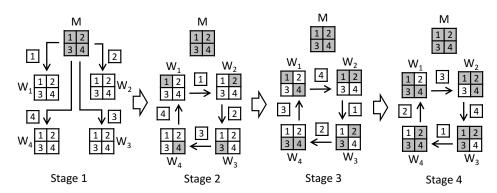


Figure 3: Illustration of Van de Gejin's pseudo-broadcast algorithm for n=4. The packet is divided into n subpackets, indicated by square boxes. Dark boxes indicate that corresponding subpackets are already received.

In [3], Van de Gejin et al. propose the following pseudo-broadcast algorithm.

Theorem 4. Assume a cluster with n workers and a master node. There exists a multicast algorithm such that the time to multicast a single packet of unit size to n workers is

$$T_{mutlicast} = \frac{2n-1}{n} T_{unicast}.$$
 (2)

Proof. Consider the following multicast algorithm. (See Fig. 3 for visual illustration.)

In the first stage, the master node divides the packet into n subpackets and scatters them: the i^{th} packet is sent to worker i for $1 \leq i \leq n$. In the subsequent stages, the n workers circulate the subpackets, forming a ring structure. More precisely, for $1 \leq j \leq n-1$, in stage j+1, worker i transmits $\{((i+j-2) \bmod n)+1\}^{\text{th}}$ subpacket to worker $\{(i \bmod n)+1\}$. The time taken for the first stage of the algorithm is $n \times \frac{1}{n}T_{\text{unicast}} = T_{\text{unicast}}$. The subsequent stages can be completed in time $(n-1) \times \frac{1}{n}T_{\text{unicast}}$ since transmissions of each subpacket can be run in parallel. Hence, the total transmission time is $\frac{2n-1}{n}T_{\text{unicast}}$.

Applying the above theorem to our coded shuffling scheme, it is clear that for each packet, $1.5T_{\rm unicast} < T_{\rm mutlicast} < 2T_{\rm unicast}$ since the number of destinations of each packet in our scheme is always larger than 1. Also, for large n, using Van de Gejin's broadcast algorithm, one can achieve $T_{\rm mutlicast} \simeq 2T_{\rm unicast}$. Thus, assuming a fully connected mesh network of link bandwidth B between the nodes, we can characterize the shuffling times for the multicast environment as follows.

Corollary 5 (Shuffling times with multicasts). Given the unicast bandwidth B, the shuffling time of the coded shuffling algorithm with multicast channel is $T_{multicast,coded} = \frac{q}{B} \frac{(1-\alpha)}{\alpha n} \frac{2n-1}{n} \simeq \frac{q}{B} \frac{2(1-\alpha)}{\alpha n}$. Hence, $T_{multicast,coded} \simeq 2T_{broadcast,coded} = \frac{2}{\alpha n} T_{uncoded}$.

3.2 UberShuffle: The coded shuffling algorithm with carpool

Thm. 2 makes an implicit assumption that the number of data points q grows faster than a certain growth rate. Recall that the master node generates one encoded packet per row of the encoding tables. If q grows fast enough, the number of allocated packets in every column of the table is the same. However, if q is finite, the number of allocated packets in the columns of the table are different from each other, and this results in resource wastage. More rigorously, one can show that the maximum number of packets allocated to any column of the table is almost equal to the average number of packets allocated to any column of the table if $q = \omega(e^n)$.

To improve the performance of the coded shuffling algorithm in practice, we propose a variation of it, which we call UberShuffle. We illustrate the new encoding algorithm via the running example.

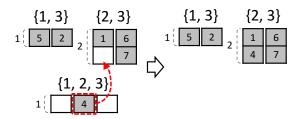


Figure 4: Illustration of the encoding table for UberShuffle algorithm.

Example 3 (UberShuffle). See Fig. 4 for visualization. Recall that in Fig. 2, there are a few missing entries in the encoding tables. The key idea of UberShuffle is to fill these gaps by reallocating data points between the encoding tables in order to reduce the number of packets. For instance, consider d_4 . This data point is exclusively stored in $\{1,3\}$ and required by worker 2. The original coded shuffling algorithm assigns this data point to the encoding table $\{1,2,3\}$ since this can maximize the coding gain in the asymptotic regime. However, in this example with a finite number of data points, this may not be the optimal choice. For instance, one can reallocate d_4 to the first column of the encoding table $\{2,3\}$. Note that such reallocation violates the 'exclusive' requirement of the original shuffling algorithm but does not compromise the decoding conditions. As a result of the reallocation, the total number of broadcast packets can be reduced from 4 to 3.

³A similar phenomenon (though not identical) is also observed in the coded caching literature. (See [29].)

In general, any data point can be reallocated from column i of table \mathcal{I}_1 to column i of table \mathcal{I}_2 if $\mathcal{I}_1 \supset \mathcal{I}_2$. The UberShuffle algorithm finds such reallocations between the encoding tables to fill the missing slots in them. (And this is why the algorithm is named 'UberShuffle': it resembles the famous carpool matching system.) Roughly, the UberShuffle algorithm first constructs a directed acyclic graph (DAG) between the columns of the encoding tables. In this DAG, a directed edge from column i of table A to column j of table B exists if and only if i=j and $A \subset B$. (For instance, in the previous example, a directed edge exists from column 1 of table $\{1,3\}$.) After constructing the DAG, our algorithm finds a flow, which is corresponding to the packet reallocations, on the DAG to reduce the total number of encoded packets. More specifically, the UberShuffle algorithm greedily optimizes the number of encoded packets by considering each layer of the DAG.

To explain the algorithm in detail, the algorithm first constructs n-1 arbitrarily ordered lists of index sets: $\mathcal{J}_2, \mathcal{J}_3, \ldots, \mathcal{J}_n$, where $\mathcal{J}_i = \{\mathcal{I} \in [n]^n | |\mathcal{I}| = i\}$ for all $2 \leq i \leq n$. It then check each of the elements in \mathcal{J}_2 one by one. Consider an element of \mathcal{J}_2 , say \mathcal{I} , and denote by \mathcal{I}_1 and \mathcal{I}_2 the elements of \mathcal{I} . If $|Z(\mathcal{I},\mathcal{I}_1)| = |Z(\mathcal{I},\mathcal{I}_2)|$, the packets destined for the subset \mathcal{I} is maximally efficient, and hence the algorithm proceeds to the next element in \mathcal{J}_2 . If $|Z(\mathcal{I},\mathcal{I}_1)| \neq |Z(\mathcal{I},\mathcal{I}_2)|$, this implies that the packets destined to the subset \mathcal{I} has some rooms to include more packets. Without loss of generality, assume that $|Z(\mathcal{I},\mathcal{I}_1)| + m = |Z(\mathcal{I},\mathcal{I}_2)|$. That is, there are m more packets destined to node \mathcal{I}_2 than to node \mathcal{I}_1 . For this case, the algorithm greedily finds packets that can be 'carpooled'. As mentioned above, one can safely reallocate packets from $Z(\mathcal{I}',\mathcal{I}_1)$ to $Z(\mathcal{I},\mathcal{I}_1)$ if $\mathcal{I}' \supset \mathcal{I}$. Denote the set of (strict) supersets of set \mathcal{I} by \mathcal{I}^S . Our algorithm searches for valid reallocations by iterating the elements of $\mathcal{I}^S \cap \mathcal{J}_2, \mathcal{I}^S \cap \mathcal{J}_3, \ldots, \mathcal{I}^S \cap \mathcal{J}_{2+\nu}$. The algorithm stops the search process when it makes m reallocations, when it finishes iterating all the elements, or when it reaches the maximum depth ν . After the search process, the algorithm proceeds to the next elements in \mathcal{J}_2 , and the same procedure is repeated until all elements of the set are examined. The algorithm then moves on to lists of larger index sets \mathcal{J}_3 , \mathcal{J}_4 , and so on. See Algorithm 4 for the pseudocode of our algorithm.

Algorithm 4 Coded Shuffling: UberShuffle algorithm at master

```
procedure UBERSHUFFLE(Z, \nu)
        for each i \in [2, \ldots, n], do
                                                                                                                \triangleright \mathcal{J}_i := \{ \mathcal{I} \in [n]^n | |\mathcal{I}| = i \} \text{ for all } 2 < i < n \}
                for each \mathcal{I} \in \mathcal{J}_i, do
                       \begin{array}{l} \ell \leftarrow \max_{j=1}^{|\mathcal{I}|} |Z(\mathcal{I}, \mathcal{I}_j)| \\ \text{for each } \mathcal{I}_j \in \mathcal{I}, \text{do} \end{array}
                                if |Z(\check{\mathcal{I}},\mathcal{I}_j)| < \ell, then
                                         m \leftarrow \ell - |Z(\mathcal{I}, \mathcal{I}_i)|
                                        for each k \in [i+1, i+2, \dots, i+\nu], do for \mathcal{I}' \in \mathcal{I}^S \cap \mathcal{J}_k, do |\mathcal{I}'| = |\mathcal{I}'| = |\mathcal{I}'| if |\mathcal{I}(\mathcal{I}', \mathcal{I}_j)| > 0, then
                                                                                                                            \triangleright \mathcal{I}^S is set of (strict) supersets of set \mathcal{I}
                                                                 m' \leftarrow \min \{m, |Z(\mathcal{I}', \mathcal{I}_j)|\}
                                                                 reallocate m' elements from |Z(\mathcal{I}', \mathcal{I}_i)| to |Z(\mathcal{I}, \mathcal{I}_i)|
                                                                 m \leftarrow m - m'
                                                         end if
                                                         if m=0, then
                                                                 break
                                                         end if
                                                 end for
                                         end for
                                end if
                        end for
                end for
        end for
end procedure
```

To observe the gain of UberShuffle, we simulate the number of broadcast transmissions of various shuffling algorithms. Shown in Fig. 5 are simulation results with $n=20, \nu=2$, and $q\in\{10^4,10^5,10^6\}$.

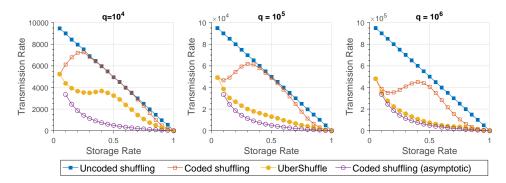


Figure 5: Number of packets used in Coded Shuffling algorithm for n=20. Storage rate q/s is denoted by α .

One can observe that even with 10^6 data points $(q=10^6)$, the performance of the original coded shuffling algorithm is far from its theoretical guarantees (Thm. 2), and this gap is even larger as q decreases. Further, observe that the UberShuffle algorithm significantly outperforms the original coded shuffling algorithm. For instance, the UberShuffle algorithm is observed to reduce the number of packets by a factor of 5.4 when $q=10^6$ and $\alpha=0.55$ and by a factor of 2.58 when $q=10^5$ and $\alpha=0.325$.

4 System Implementation

In this section, we implement a generic distributed machine learning system using Open MPI C. Further, we implement various shuffling algorithms in our system so that we can compare the performances of different shuffling algorithms. The overall architecture of Ubershuffle system is given in Fig. 6.

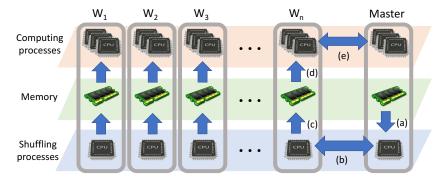


Figure 6: The system architecture of Ubershuffle system. (The figure indicates a only some of nodes for simplicity.) Each of the stage indicates follows.(a) Master node shuffling process reads data, (b) and distributes shuffled data to worker shuffling process using shuffling algorithms. (c) Each worker's shuffling processes saves shuffled data received from master. (d) Each worker's computing processes runs SGD algorithm using data saved by its shuffling process. (e) Models processed by computing process in each worker are merged and distributed by master's SGD process.

4.1 System Architecture

We design the system to use two cores in each node (including the master node), each of them is used for two different processes: computing process and shuffling process.

Initially, the entire training data is stored in master node's memory, and each of worker node stores a subset of master's training data, with jointly covering the entire training data across distributed nodes.

The computing processes in master and worker nodes perform distributed SGD operations as a whole. On each epoch of the SGD procedure, each computing process runs SGD over existing local

training data. After each epoch, the master node simply collects all the trained model from the n workers, find a synchronized (averaged) model, and share the new model with the workers.

The shuffling processes in master and worker nodes shuffles the data across the nodes based shuffling algorithms explained in Sec. 2. On each epoch, The master node's shuffling core encodes the new data rows to be shuffled, and these encoded data rows are decoded by the workers' shuffling cores. The workers' shuffling cores are able to access the locally-cached data rows, which are necessary for decoding the encoded packets for coded shuffling.

We pipeline the epochs of SGD and shuffling procedures. That is, computing processes of all nodes runs SGD operation until newly shuffled data is ready in its local worker's shuffling process. When shuffled data is ready, shuffling process copies its new data to computing process via memory in order for computing process to run SGD with newly received data. In this way, the system minimizes performance loss on SGD procedure due to waiting for shuffled data to be ready in each node.

4.2 Implementation of Efficient Multicasting Algorithms

While our theoretical analysis assumes the Van-De-Gejin algorithm for multicasts, the Van-De-Gejin algorithm is not available as a multicast protocol in <code>OpenMPI v1.6</code>. Instead, the split-binary-tree algorithm [24] is implemented as a multicast protocol in <code>OpenMPI v1.6</code>. We experimentally demonstrate that the split-binary-tree algorithm indeed behaves similar to the Van-De-Gejin algorithm. Plotted in Fig. 7 is the average packet transmission time as a function of the number of workers, measured on an Amazon EC2 cluster. One can note that the packet transmission time is bounded by $3.5T_{\text{unicast}}$ for all tested values of n <= 20. That is, the multicast protocol available in <code>OpenMPI v1.6</code> is at most 1.5 times worse compared to the Van-De-Gejin algorithm. 4

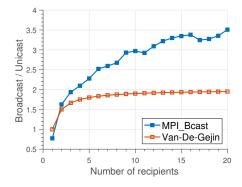


Figure 7: Gains of broadcasting over uni-casting in distributed systems

4.3 Algorithm Modifications: Communicator Merging

Under OpenMPI framework, a *communicator* has to be defined for each set of multicasting group. Our coded shuffling algorithm considers all possible multicasting groups, which are 2^n combinations of n workers. In OpenMPI v1.6, the large computational overhead is associated with communicator declaration. Hence, in the beginning of the algorithm, it is inevitable to declare all the communicators and keep them in memory. However, such an approach makes the coded shuffling algorithm infeasible due to the exponentially large memory footage. Moreover, in OpenMPI v1.6, only up to 2^{15} communicators can be declared at the same time.

In order to resolve this issue, we implement the communicator merging procedure, which merges 2^n communicators into $2^{n'}$ communicators, where n' is a chosen parameter which divides n. For instance, when n=20 and n'=10, the merging procedure reduces the number of communicators

 $^{^4}$ In [17], the authors experimentally show that the broadcasting time increases logarithmically in n while our results show that it is bounded by a constant. The difference is due to the different choices of multicasting algorithms: In [17], the multicasting algorithm is not specified, and hence rather inefficient algorithm is used by default; In this work, we specifically choose the split-binary-tree multicasting algorithm to achieve the constant multicasting time.

from $2^{20} \simeq 10^6$ to $2^{10} \simeq 10^3$, reducing the memory footage associated with communicators by a factor of 1000. We partition 20 workers into 10 nodal subsets, with k^{th} subset containing worker node 2k-1 and 2k. When master process of shuffling layer broadcasts encoded data to its target worker processes, it concatenates data that has target node of same nodal subsets, and sends concatenated data based on 10 nodal subsets.

Define a function that maps an index set to an augmented index set as follows:

$$\psi_2(\mathcal{I}) = \{x | x \in [n], \left| \frac{x+1}{2} \right| = \left| \frac{i+1}{2} \right| \text{ for some } i \in \mathcal{I} \}.$$
 (3)

With the above definition, one can see that the communication merging concatenates two encoded packets $b_{\mathcal{I}_1,t}$ and $b_{\mathcal{I}_2,t}$ if and only if $\psi_2(\mathcal{I}_1)=\psi_2(\mathcal{I}_2)$. For instance, $b_{\{2,3\},t}$ and $b_{\{1,3\},t}$ are concatenated since $\psi_2(\{2,3\})=\psi_2(\{1,3\})=\{1,2,3,4\}$.

We note that the communicator merging process slightly increases the shuffling time of coded shuffling since it increases the number of destinations per packet.

5 Experimental Results

In this section, we briefly introduce machine learning algorithms used in UberShuffle system. We then evaluate the performances of various shuffling algorithms on several applications.

5.1 Machine learning algorithms and Hardware Setup

Distributed SGD for Matrix Completion We evaluate the performance of our shuffling algorithms for the distributed SGD algorithm for a low-rank matrix completion problem, proposed in [26].

The goal of the low-rank matrix completion problem is to fill the missing entries of a given matrix X of size $n_r \times n_c$ assuming that the rank of X is low compared to the dimension of it. This problem can be reduced to finding matrix L of size $n_r \times r$ and matrix R of size $n_c \times r$ by iteratively computing sequence of $L^{(k)}$ and $R^{(k)}$ as follows:

$$L_{i_k}^{(k+1)} = \left(\left(1 - \frac{\mu_1 \gamma_k}{|\Omega_{i_k}|} \right) L_{i_k}^{(k)} - \gamma_k f_{i_k j_k}^{'} \left(L_{i_k}^{(k)} R_{j_k}^{(k)} \right) R_{j_k}^{(k)} \right), \tag{P5}$$

$$R_{j_k}^{(k+1)} = \left(\left(1 - \frac{\mu_1 \gamma_k}{|\Omega_{\star j_k}|} \right) R_{j_k}^{(k)} - \gamma_k f_{i_k j_k}' \left(L_{i_k}^{(k)} R_{j_k}^{(k)} \right) L_{i_k}^{(k)} \right), \tag{P6}$$

where Ω is a subset of $\{1,\ldots,n_r\} \times \{1,\ldots,n_c\}$, denoting the observed entries, and $\Omega_{i\star} = \{j|(i,j)\in\Omega\}$ and $\Omega_{\star j} = \{i|(i,j)\in\Omega\}$. The index pair (i_k,j_k) is chosen uniformly at random from Ω for k^{th} iteration, and γ_k is the step size for k^{th} iteration. Detailed proof of this property is given in [26].

In [26], the authors propose a way of distributing the observed entries of X across workers and updating the parameters in parallel without incurring any conflicts. First, the master process generates a random permutation of rows and columns, say π_{row} and π_{col} . It then partitions the entire data points into n^2 sets, where n is the number of worker nodes. Once the permutation is given, each worker node is assigned an exclusive set of n sets. The authors of [26] show that it is possible to compute the exact SGD updates in parallel even though n workers work with their own individual data points in parallel. Further, the authors show that shuffling improves the convergence speed, and hence our shuffling algorithm can be used to reduce the communication cost of this procedure. For more detailed description, see [26].

For the experiments, we run our system on both synthetic data and real data. For the synthetic data, we generate low-rank matrices with random Gaussian factor matrices. For a fixed choice of number of rows n_r , number of column n_c , rank r, and noise level σ^2 , we generate a low-rank matrix $M = Y_L Y_R^T$, where the entries of Y_L and Y_R are i.i.d. Gaussian random variables. We then normalize the generated matrices so that ||M|| = 1. Then, for each row of matrix M, a fixed

number of entries are selected as training set (we denote the number of entries in each row by m). For the real data, we use the Movielens 20m dataset [13]. We preprocess the dataset by randomly sampling 68 data points from each row of the uses, resulting in a sampled dataset with $n_r = 68682$, $n_c = 16622$, and m = 68.

We split the data set into training and test sets, and measure $(X_{ij} - M_{ij})^2 / (M_{ij})^2$ on the test set. We also measure the runtime of each algorithm.

On each instance, we run our system for 7 epochs, and the step size is reduced by a factor of 0.9 after every epoch, i.e., $\gamma_k = \gamma_0 (0.9)^k$. We evaluate the performances of the algorithms with varying r and μ , and here, for each configuration, we report the result with r and μ that perform the best only.

Parallel SGD for Linear Regression We also run the parallel SGD (PSGD) algorithm [33] for linear regression, described as follows. The master node randomly initialize x, and then broadcasts it to all workers. At the same time, the master node reads the data matrix A and shuffles partial row vector a_i 's of A and the corresponding y_i 's to workers. Once the data rows are shuffled, the workers independently run the SGD algorithm using the local data rows. That is, for each data point a_i , the master node updates x as

$$x = x - \gamma a_i (a_i \cdot x - y_i), \tag{4}$$

where γ is the step size. After all the data points are used to update the parameter x, each worker sends the final parameter x to the master. The master receives the updated parameters from the workers, and then computes the global parameter x by averaging the received updated parameters. At the end of each epoch, we measure $\|x-x_{\rm ans}\|/\|x_{\rm ans}\|$ and its runtime. Synthetic data for our experiment is made by randomly generating a matrix A of size $q\times m$ and a vector x of size m, and y is generated by y=Ax.

System Design and Cluster Setup We implement a generic distributed machine learning system using Open MPI C. Further, we implement various shuffling algorithms in our system so that we can compare the performances of different shuffling algorithms. We remark that our system is inherently fault-tolerant: when nodes fail or straggle, the master node ignore them and proceed to the next iteration with the other nodes. All experiments are run on an Amazon EC2 cluster with the following configuration. We use a m3.2xlarge (8-core 2.5GHz Intel Xeon E5-2670 v2 with 30GB RAM) instance for the master, and 20 m3.xlarge (4-core 2.5Ghz Intel Xeon E5-2670 v2 with 15GB RAM) instances for the workers. Note that the master is configured to have a sufficiently large RAM since it has to hold the entire training dataset. Further, though it has 8 cores, we use only 2 cores for our experiments. Also, the workers are configured to have maximum network bandwidth, and to use only 2 cores as well.

5.2 Experimental Results

Table 1: Experimental setups and shuffling time comparison. 'CS = coded shuffling', 'US = UberShuffle', and 'UN = uncoded shuffling'; 'MC = matrix completion' and 'LR = linear regression'.

	a m		0	Shuffling Time (sec)				Setup
	q	m	α	UN	CS	US	(CS-US)/CS	Setup
(a)	10^{5}	10^{4}	0.2	105.5	123.6	65.2	47.2%	MC, Synthetic
(b)	3×10^{5}	1000	0.2	35.0	28.8	25.3	12.2%	MC, Synthetic
(c)	6.8×10^{4}	68	0.2	0.42	1.66	1.52	8.4%	MC, Real
(d)	7×10^{5}	2×10^5	0.14	112.90	94.60	60.82	35.7%	LR, Synthetic

We summarize various experimental scenarios in Table 1 along with the measured shuffling times. In most cases, we observe that the coded shuffling with carpool achieves the fastest shuf-

⁵Note that our coding schemes are not applied to other communication patterns such as model synchronization. However, if the size of model parameters is large, those communication overheads are not negligible anymore, and whether or not it can be reduced via similar coding techniques is an interesting open problem.

⁶Indeed, this approach is observed to converge well or even faster [8].

fling times. Note that these shuffling times include all the extra overheads such as the encoding/decoding/indexing times. For the experimental setup (c), the uncoded shuffling time achieves the best performance: This is because the packet size for each data point (m=68) is too small, and hence the communication costs of the shuffling algorithms are almost negligible. On the other hand, the coded shuffling algorithms cost non-negligible amounts of times to encode/decode/index data points, making themselves slower than the uncoded shuffling algorithm.

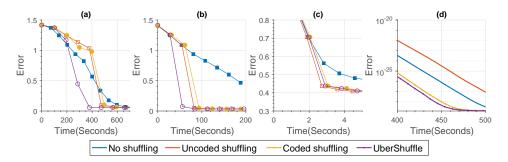


Figure 8: Convergence performances. (a) and (b) are for the matrix completion with synthetic data; (c) is for the matrix completion with real data; and (d) is for the linear regression with synthetic data.

We also report the convergence performance of the learning algorithms in Fig. 8. For scenarios (a) and (b), the best performance is observed with UberShuffle, and the convergence time is reduced by at least 19.8% and 32.1%, respectively. For scenario (c), while all of the shuffling algorithms outperform the one without shuffling, the coded shuffling algorithms indeed perform slightly worse than the uncoded shuffling algorithm. As discussed earlier, this is due to the relatively large computational overheads of the coded shuffling algorithms compared to the very low communication costs. For scenario (d), both the coded shuffling algorithms significantly outperform the other algorithms in terms of convergence time. Interestingly, the uncoded shuffling algorithms fails to perform better than the no shuffling algorithm due to its excessive communication costs.

Under our experimental setup where nodes are connected via 1Gbps bandwidth ethernet, transmission time is observed to be the dominant factor of the shuffling time, and hence we do not attempt at minimizing UberShuffle's computational overhead (indexing/encoding/decoding time), which is negligible anyway. However, if much faster broadcast environment (such as RoCE with 50Gbps bandwidth) is available, this may not be the case anymore, and the shuffling gain might not be observed unless one minimizes the computational overhead of UberShuffle algorithm.

5.3 Data shuffling via shared storage systems

To deal with a large-scale data set in distributed machine learning systems, one may store the entire data set in a shared storage system, which can be accessed by the distributed workers. If such a shared storage system is available, one can shuffle the data points simply reading data points from the shared storage system, without relying on the master node's shuffling mechanism. Compared with our data shuffling scheme, which fully exploits the network bandwidth among the distributed workers, such an approach based on a shared storage system may incur various bottlenecks at the storage side, eventually slowing down the shuffling process. In this section, we compare the performance of our shuffling systems with other distributed learning systems with a shared storage system.

We implement a new shuffling process that reads data points directly from a shared storage instead of receiving from the master node. At the beginning of each epoch, each shuffling process reads the data points that are required for the subsequent epoch but are not stored in the current cache, i.e., $S_i^{t+1} \setminus C_i^t$ for worker i. Once the read process is completed, the new data points are copied to the SGD process of the same worker. All the other experimental setups are kept the same as in Sec. ??.

For the shared storage system, we consider two options available on Amazon Web Service (AWS): EBS (Elastic Block Storage) and EFS (Elastic File System). For the EBS case, we attach a storage unit to the master node, and then let the master node share the disk with the distributed workers using the NFS (Network File Storage) protocol. The EFS is a new virtual storage solution provided

by AWS with which one can attach via NFS an networked storage system to compute units such as EC2 (Elastic Compute Cloud). There are two available operations modes: "General Purpose" mode and "Max I/O" mode. The "Max I/O" mode offers higher storage thoughput at the cost of higher file operation latency. On the other hand, "General Purpose" mode offers relatively low storage throughput but guarantees reasonable file operation latency.

We compare the performance of our systems with shuffling algorithms and those with shared storage systems. Especially, we run the SGD algorithm for low-rank matrix completion problem, described in Sec. 5.1, with $n_r=300k, n_c=300k, r=10, m=1000, \alpha=0.05$. Table 2 summarizes the average per-epoch shuffling time.

Table 2: The average per-epoch shuffling time for UberShuffle and data shuffling by direct reading from shared storages.

Setup	Approx. Epoch Shuffling Time (sec)
UberShuffle	34
EBS Storage	110
EFS Storage (General purpose)	490
EFS Storage (Max I/O)	1100

One can clearly observe that the average per-epoch shuffling time is much larger when shared storage systems are used instead of our shuffling algorithm: the UberShuffle system is 3.2 times faster than the fastest storage-based alternative. As briefly mentioned above, this is due to the network and disk bottlenecks at the storage side and the large number of file operations incurred during the shuffling procedure. Especially, for the case of the EFS storage, each data file is split into multiple pieces and stored as multiple file units in order to maximize the throughput. This significantly degrades the perforamnce of the shuffling procedure for the following reason. In every epoch, each worker sends its own $15000\,$ file queries to the shared storage system, and this simultaneously happens at all the workers. Hence, in total, about $3\times 10^5\,$ files quires are received at the storage side, significantly slowing down the shuffling procedure.

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