

Algorithm Engineering

Federico Matteoni

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0.1 Introduction

Teacher: Paolo Ferragina

Exam: written + oral. Midterms in November and December, with exercises.

Classes will be recorded on Microsoft Teams. Also the book "The Magic of Algorithms" is very important: you must be used to talk about these things, not just be able to solve exercises.

Course Design and **analysis** of algorithms but also insights about **implementation**, with reference to libraries and considerations about what happens when using certain algorithms. The case of use is **big data**.

0.1.1 Algorithms

Algorithms Knuth's definition: "a finite, definite, effective procedure that takes some input and returns an output, with the output being the answer to the problem you want to solve."

Finite: a **finite sequence of steps**, not only a finite numbers of operations but also **the algorithm must terminate**. `while (true) do ...` will go on forever, so it's not an algorithm. The number of steps must be *very definite and reasonable*, which relates to the efficiency of the algorithm.

Definite: the steps are definite in an unambiguous way.

Effective: every step is basic, atomic, something that we can execute in small or constant time, constant is not a very precise word (seconds? Milliseconds?) so we will accept the "small time" rough definition. Also, the mapping input \rightarrow output must always be correct, which is the biggest difference with IA. An algorithm outputs the correct output for each input.

RAM Random Access Machine, CPU \leftrightarrow M, classical computing model (Von Neumann machine), the memory can read any place in constant time.

We will make a more sophisticated step. But without presenting very complicated models. We need a good balance, not a perfect but a *better* approximation than the RAM.

Analysis Let's take an algorithm A and let's find a function $T_A(n)$ that describes the time complexity of A. n is the input size, the number of items that the algorithm has to process. The time that A will take will be in hours, seconds or milliseconds based on the machine, but we approximate that time taken with the number of steps that are computed. Also, the number of steps depends not only on the number of items but on the items themselves, too. So we usually analyze the worst case scenario, or less often the average scenario. By analyzing the worst case scenario we can figure out the worst or "maximum" number of steps. **Asymptotic analysis**.

We want to exploit the characteristics of the various types of memory.

We will count not all the steps but the I/O ops, with a **2-level memory model**: the **first level** is the **internal fast memory** (cache + RAM) and the **second level** is the **mass unbound memory** (disk). In small memory situations, the first level can be interpreted as cache and the second level as internal memory, other times the first level is the internal memory and the second level is unbound slow memory.

Spatial Locality: access near items

Temporal Locality, or small working set: far apart items used often so we can exploit their presence in the cache

Poly vs Exp time complexity Let's say we have three algorithms with n , n^2 and 2^n in time complexity respectively. Let's express the time complexity fixing t time and counting how many items we can process in t time. In the first case is linear, so $n = t$, the second is $n = \sqrt{t}$ and the third is $n = \log_2 t$. If we have a k times faster machine, we can imagine that we are using the original machine for k more times, or k original machines in parallel. So in this case $n = kt$, $n = \sqrt{kt}$ and $n = \log_s kt$. The linear algorithm takes full advantage of the k times faster machine, the second algorithm has a small advantage of a multiplication factor \sqrt{k} and the last a negligible advantage of a sum factor of $\log_2 k$, which is basically none.

Example $A[1, n]$ integer array of which we want to compute the sum. The number of items, then, is n .

First situation: first we load B items and process them, then the next B items and so on. $\Rightarrow \# \text{ I/O} = n/B$, which we will see often and is called the **scan cost**, because we need to see each element, in batches of B elements. So B is the size of the memory page.

But we can follow a different approach: we take the first item of each batch of B elements, then the second items and so on, which will take n steps, but this will be possibly slower because this method takes more I/Os ops, n I/O ops. The larger the jumps the more I/O ops we do. The model doesn't distinguish between local and random I/Os.

Binary Search Array of n elements. We pick the middle element, we go to left/right, middle element of the section and so on. The time complexity is $T(n) = O(\log_2 n)$, but we have a lot of I/Os and big jumps, so elements in different and far pages. We have $\log_2 \frac{n}{B}$, because at a certain point the sub array where we search will be smaller than a page, so we have $\log_2 n$ steps $-\log_2 B$ the last steps inside the page, and for the properties of the logarithms we have $\log_2 \frac{n}{B}$. So the larger the page the smaller the number of I/Os. One consideration: n is the number of items, B is in kilobytes. So if we consider integers of 8 bytes, we have $\frac{B}{\text{size}}$ items, and with $B = 32 \text{ KB} = 2^{15} \text{ KB}$ we have circa 4000 times, or $2^{15}/2^3 = 2^{12}$.

How can we improve the search? We can consider the B^+ -trees. We split the array into the page size B and the array is sorted in ascendant order. The splits are called leaves, and each leaf has a key (one of the elements). We have a page with each key and the next element is the pointer to its page. Above one level, a page with a key of the first key list and a pointer to the key list, a key of the second and so on.

Fetch a page, binary search and follow the pointer. Number of I/Os is $\log_B \frac{n}{B}$ and the number of steps is a binary search for every page, so $(\log_B \frac{n}{B}) \text{ pages} \cdot \log_2 B$



Analysis $n = (1 + \epsilon)M$ with $\epsilon > 0$ data outside of the memory. So M is totally full and ϵM is stored in the unbound memory.

The first point is we want to find $P(\text{accessing the disk})$, with a totally random algorithm, is $= \frac{\epsilon M}{n} = \frac{\epsilon M}{(1 + \epsilon)M} = P(\epsilon)$

The second point is the average time of a step. $\sum_x P(X = x) \cdot x$ with X the variable of which we compute the average with that formula. X is time, in this case. The time is 1 in case of computing, and varies in case of accessing the memory. So we have the multiply the cost of access times the probability (of computing, going internal, going to the disk). Internal memory access costs 1, while on disk is larger and we say that the cost is c . So the average, with let's say a probability of memory access, $= (1 - a) \cdot 1 + a(P(\epsilon) \cdot c + (1 - P(\epsilon)) \cdot 1)$ but $0 < a < 1$ and $1 - P(\epsilon)$ are very small, so we can rewrite as $= a \cdot P(\epsilon) \cdot c = a \cdot \frac{\epsilon}{1 + \epsilon} \cdot c + O(1)$. If $a = 0$ then no memory access so the cost is constant. The larger is a the more memory access, the more the term is important, which is exactly what we want to capture. Usually, $a = 0.3 = 30\%$ and $c = 10^6$ the gap between accessing the disk and accessing the internal memory.

So $\frac{\epsilon}{1 + \epsilon} \cdot 0.3 \cdot 10^6 = \frac{\epsilon}{1 + \epsilon} \cdot 300000$. If $P(\epsilon) = 0.001$ the avg time of a step is $0.001 \cdot 300000 = 300$, so the disk has a lot of impact even with only a thousandth of memory access being on disk: the avg cost is 300 and not 1.

0.2 Sorting

Permuting problem Given an array $S[1, m]$ and a permutation π , the permutation problem asks to permute S according to π , creating $[S[\pi[1]], S[\pi[2]], \dots, S[\pi[m]]]$. For example $S = [A, B, C, D]$, $\pi = [3, 1, 2, 4]$ then π tells that the $\pi[0] = 3$ item goes to the first position. So $S_\pi = [C, A, B, D]$

```
1 for i = 1 to n:
2   S1[i] = S[pi[i]]
```

Which costs $\Theta(n)$

	PERM	SORT
RAM	n	$n \log n$
2-level memory	$\min\{n, C_{\text{sort}}\}$	C_{sort}

Solving the permuting in a scan + sort kind of way, otherwise we have to do a disk access per value.

scan $S : \langle S[i], i \rangle$ which is $\langle \text{item}, \text{position} \rangle$
 Which creates $\langle A, 1 \rangle \langle B, 2 \rangle \langle C, 3 \rangle \langle D, 4 \rangle$
 This costs $O(\frac{n}{B})$

scan $\pi : \langle \pi[i], i \rangle$ which is $\langle \text{src}, \text{dst} \rangle$
 Which creates $\langle 3, 1 \rangle \langle 1, 2 \rangle \langle 2, 3 \rangle \langle 4, 4 \rangle$
 This costs $O(\frac{n}{B})$

sort by first component of the sequence $\langle \pi[i], i \rangle$
 Which creates $\langle 1, 2 \rangle \langle 2, 3 \rangle \langle 3, 1 \rangle \langle 4, 4 \rangle$
 We take the item in source and move to destination

parallel scan $\langle A, 1 \rangle \langle B, 2 \rangle \langle C, 3 \rangle \langle D, 4 \rangle$
 $\langle 1, 2 \rangle \langle 2, 3 \rangle \langle 3, 1 \rangle \langle 4, 4 \rangle$
 Which creates $\langle A, 2 \rangle \langle B, 3 \rangle \langle C, 1 \rangle \langle D, 4 \rangle$

sort by second component
 Which creates $\langle C, 1 \rangle \langle A, 2 \rangle \langle B, 3 \rangle \langle D, 4 \rangle$

scan
 Which creates $[C, A, B, B]$
 This costs $O(\frac{n}{B})$

The I/O cost is 4 scan + 2 sort = $O(\frac{n}{B}) + 2 \cdot O(C_{\text{sort}})$ sorting cannot cost less than $\frac{n}{B}$ because we can't improve that, so $O(C_{\text{sort}})$

So we have proposed upper-bounds = algorithms for the sorting and the permuting problem. The permuting problem can be solved in $O(\frac{n}{B})$ + sorting I/Os. Moving n items takes $\Theta(n)$ I/Os.

Sorting n items in a two level memory of size M for internal memory and B for the disk page size, costs $O(\frac{n}{B} \cdot \log_{\frac{M}{B}} \frac{n}{M})$ with $L = \log_{\frac{M}{B}} \frac{n}{M}$, and often written as $\overline{O}(\frac{n}{B})$ with the overline or over tilde that means that is a scan.

L consists of the base $\frac{M}{B}$ and the argument $\frac{n}{B}$ which means that with a larger memory I'd like it to be faster, with bigger M the argument decreases and the base increases so the logarithm shrinks a lot. With a bigger B page size, $\frac{n}{B}$ decreases but the base increases.

$\frac{M}{B}$ = how many pages I can keep in the internal memory. Let's say $n = 2^{40}$, $M = 8\text{Gb} = 2^{33}$ and $B = 32\text{Kb} = 2^{15}$, then

$$\log_{\frac{n}{B}} \frac{n}{M} = \frac{\log_2 \frac{n}{M}}{\log_2 \frac{M}{B}} = \frac{\log_2 \frac{2^{40}}{2^{33}}}{\log_2 \frac{2^{33}}{2^{15}}} = \frac{\log_2 2^7}{\log_2 2^{18}} = \frac{7}{18} < 1$$

Let's see when is sorting preferred to moving items or viceversa

$$\frac{n}{B} \log_{\frac{M}{B}} \frac{n}{M} < n \Leftrightarrow \log_{\frac{M}{B}} \frac{n}{M} < B$$

We have $B = 1$ in the RAM model, $B = 32\text{Kb}$ in the 2-level model compared to $\log_{\frac{M}{B}} \frac{n}{M} = 2$ or 3. In practical situations with disk, sorting is better than moving numbers. If $B = 1$, in the RAM model, $M = O(1)$, then sorting is worse than moving.

Sorting Let's consider binary merge sort, which in the worst case costs $O(n \cdot \log_2 n)$.

```

1 def MergeSort(S, i, j):
2   if (i < j): # sort if at least two items, otherwise already sorted
3     m = (i+j)/2
4     MergeSort(S, i, m-1) # split into two halves
5     MergeSort(S, m, j)
6     Merge(S, i, m, j) # most of the cost is here

```

Let's evaluate the I/Os in the case that $n \gg M$ so the array can't be stored entirely in internal memory. Since it's based on the merge procedure, we can evaluate the cost of the merge and multiply by the number of levels. It loads a page every time it needs it and writes a page every time it fills one. So if the two arrays to merge are long l , it makes $\frac{l}{B}$ I/Os. So it takes $O(\frac{n}{B} \log_2 n)$ I/Os.

Sorting = computing the sorted permutation + implement the sorted permutation. So we can say that sorting \geq permuting, is at least difficult as permuting.

In the RAM model the \geq is $>$, it's strict, because sorting is $n \cdot \log n$ and computing is the real issue, because implement is linear. In the 2-level memory model they are almost equivalent. This considering **atomic items**, integers or non-splittable strings.

The mismatch is the base of the logarithm. The binary merge sort is $O(\frac{n}{B} \log_2 \frac{n}{M})$: partitioning the array in blocks of size M , the size of the memory, called runs. Can we generate runs longer than M in few I/Os? On average we will be able to create runs of size $2M$, which saves 1 full scan of data (which, for large data, is a lot of time saved).

Snow plow algorithm Sort item that you can, leave item that you cannot.

$S = 1, 7, 5, 3, 2$, with $M = 3$ items. The memory is divided in two parts: a min heap, items that are still unsorted, and an unsorted part, the snow that we cannot clean. Start from the memory with everything unsorted: min heap empty and only unsorted items. We put M unsorted items, so 1, 7, 5 in the unsorted part. We sort the unsorted items and move to the min heap, with the unsorted part left empty. We pick the minimum element and write it out of the memory. We have emptied a position, and we can fetch another item in the unsorted part, the 3. The new item is compared to the minimum. 3 is larger than the current minimum, so is written inside the min heap, which is now 3, 5, 7. We write again out the minimum, 3, and fetch another item. 2 is smaller than 3, so it goes in the unsorted part. At some point the min heap will be empty, the unsorted part will be full and we restart the phase. We pay 1 I/O as soon as we write out B items.



```

1 U = unsorted array of M items
2 H = min-heap over items of U
3 U = {}
4 while H != {}:
5     min = minimum from H
6     min -> output run
7     next = next item from input sequence
8     if next < min:
9         next -> U
10    else:
11        next -> H

```

Let's prove that the runs are $2M$ on average. We start with $|U| = M$ and $|H| = 0$, with U unsorted part and H heap part. Continuing we read items, with $\tau = \#$ items read. The phase has processed $\tau + M$ items, τ read and M already in memory. At the end of the phase, the min heap is empty $|H| = 0$ and $|U| = M$ and τ items are written out. So τ is the length of the run, we have to compute it, by making hypothesis about the distribution of the items, the probabilities of going to U and H . $P(\text{item read goes to } U) = \frac{1}{2}$, a totally random situation. By changing the probability we change "how much sorted" is the sequence. The more sorted is the sequence, the smaller the probability of going to U . $\text{avg}(|U|) = E[|U|] = \tau/2$ because we have τ items that go to U with probability $\frac{1}{2}$. Given that $|U| = M$, then on average $E[|U|] = \frac{\tau}{2} = M$ so average $\tau = 2M$

Exercise $M = 2$ and $S = 1, 8, 3, 2, 5, 0, 4, 6, \dots$

Issues Binary Mergesort doesn't always exploit all the memory. Because after creating the M -long first runs, by merging we need $3B$ for reading the memory (one for the first run, one on the second run and one page on the output), so $3B \ll M$, a lot of unused memory. We could fetch 2 pages per run, but the second page can be used only after the first page, so no advantage in allocating all data. Since there's a sequence of processing, even if a load immediately all the pages, we do not have much advantage in doing so. So we would like to merge k runs instead of two runs at a time. Since we want to fill the memory, we want $(k+1)B = M$, k pages for k runs plus an output page. So $k = \frac{M}{B} - 1$ which we can approximate with $\frac{M}{B}$. Each run generates $\frac{1}{B}$ I/Os, and the merged run is long $k \cdot \frac{M}{B}$ elements. So every run is of M elements which take $\frac{M}{B}$ I/Os, for $\frac{n}{M}$ runs. So the total cost is $O(\frac{n}{M} \cdot \frac{M}{B}) = O(\frac{n}{B})$ for creating runs. So the merge tree has $\frac{n}{M}$ leaves and every node has k runs to merge from the lower level, making it $\log_k \frac{n}{M}$ levels high.

The cost of mergesort is $O(\frac{n}{B} + \frac{n}{B} \cdot \log_k \frac{n}{M})$ which we have seen with $k = \frac{M}{B}$. It's the cost of creating the runs ($\frac{n}{B}$) plus the cost of merging (every merge is linear, total length divided by B , by the number of merge operations which is the levels of the tree)

With improve B, M by compression. So 3, 5, 10 is written as 3 (the first item) and storing the gaps (gap encoding) so 2, 5, ... with variable length encoders.

Let's rewrite the bound

$$\frac{n}{B} \cdot \log_{\frac{M}{B}} \frac{n}{M}$$

We don't know how items are consumed on the disks, given D disks. With $k = 2, D = 2$ we can take advantage of the parallelism on the first load. But once loaded in memory, we may consume the pages asymmetrically, so we may end up reading a lot from a disk and very few times from the other. Possibly, every page on the first disk is used before the second page on the second disk. So by adding more disks we take full advantage because they are the denominators. The know optimal bound is:

$$\Theta\left(\frac{n}{BD} \cdot \log_{\frac{M}{B}} \frac{n}{BD}\right)$$

For 1 disk is $O(\frac{n}{B} \cdot \log_{\frac{M}{B}} \frac{n}{B})$

For D disks we can consider one big disk with $B' = D \cdot B$ page size, so the heads of the disks are fixed together and move together. We do not change the algorithm, so the complexity is simply $O(\frac{n}{B'} \cdot \log_{\frac{M}{B'}} \frac{n}{B'}) = O(\frac{n}{DB} \cdot \log_{\frac{M}{DB}} \frac{n}{DB})$ but with D in the base of the logarithm makes the value larger, but the impact is small and the simplicity of the method makes it worth it. This is called **disk striping**.

Binary Decision Trees Every node (a, b) generates a comparison, the left branch is followed if $a < b$ and the right if $a > b$: it represents an algorithm. The leaves are sorted permutations according to the path of comparisons, total number of leaves is $n!$. The question is how big should be the tree so that it's a sorting algorithm, so to reach a number of leaves that's at least $n!$? It grows by 2^t nodes for every level t , so $2^t > n!$ and we have the lower bound. The left subtree, where the comparison is $a < b$, simply generates (by following a path) all the permutations in which a comes before b , whatever number of elements are between them. The right path, vice versa, generates permutations where b comes before a .

Lower bounds There are also lower bounds about permuting, an extra for the oral.

Binary decision trees proved lower bound of sorting in the RAM model. Now let's see the I/O lower bound, so not counting comparisons but read/written pages. The main idea is that whenever I fetch a page I fetch B items, not just two, so we can do many comparisons before loading another page. And the page goes in internal memory where there are more items which we compare. Whenever the items goes to internal memory we have to avoid repeating comparisons.

The memory M consists of several items, we bring the page of size B in internal memory and we have $M - B$ other items in the memory. One I/O executes many comparisons. If $B = 1$ and $M - B = 3$, so 3 items in M and 1 item brought from disk. How many comparisons? We can assume that the items in memory are sorted, so the item brought can go in 4 position (before the first, between the first two...). In general, the B items can enter the $M - B$ items in n ways. The memory consists of M cells, so $n = \binom{M}{B} \cdot B!$ with $B!$ because we have to account for the shuffling, so it needs to be counted only once. We have $\binom{M}{B}$ because every item from B can be inserted in one slot of M possible slots, so B slots out of M .

So two situations

1. A new read, a read of an input page for the first time
new reads = $\frac{n}{B}$
2. An old read, all other read-cases, $t - \frac{n}{B}$

t I/Os, can be a new or an old read. Either one of the $\frac{n}{B}$ pages for the first time or any other page (or one of those pages for the second/third/... time).

Two kinds of nodes, one node refers to a new I/O the other to an old I/O.

In a new I/O we have $\binom{M}{B} \cdot B!$ options, we have to account for the shuffling, in the second case we have $\binom{M}{B}$ options.

So in any path $\frac{n}{B}$ are new and $t - \frac{n}{B}$ are old. So how many steps or levels t are needed to guarantee that the number of leaves is $\geq n!$? Every node is no longer binary and can be one of two kinds. Still a decision tree. So in every step we have $\frac{n}{B}$ new I/Os and $t - \frac{n}{B}$ old I/Os.

permutations that the algorithm can distinguish is

$$= \left(\binom{M}{B} \cdot B! \right)^{\frac{n}{B}} \cdot \left(\binom{M}{B} \right)^{t - \frac{n}{B}} = \left(\binom{M}{B} \right)^t \cdot (B!)^{\frac{n}{B}} \geq n!$$

To compute, consider that $\log \left(\frac{a}{b} \right) = b \log_2 \frac{a}{b}$ and solving for t gives the multi way merge sort bound. $t = \Omega\left(\frac{n}{B} \log_{\frac{M}{B}} \frac{n}{M}\right)$

Two sorting paradigms The following:

merge-based paradigm: partition (computationally free, $O(1)$), recursion, recombination (merge, $O(\frac{n}{B})$)

distribution/partitioning-based paradigm: partition (costly part, $O(n)$), recursion, recombination (\mathcal{A})

How can we design a multi way quick sort? We can't take one pivot, because that'll be binary. We can take many pivot but we have to guarantee a good usage of internal memory, but more importantly we need that the partition have to be balanced: with k pivot I'd like that every partition has $\frac{n}{k}$ items. This requires a theorem.

Let's refresh the quick sort.

Time complexity of quick sort is $O(n \log n)$ Let's prove it by introducing a random variable

$X_{u,v} = \begin{cases} 1 & \text{if } S[u] \text{ is compared to } S[v] \\ 0 & \text{else} \end{cases}$ and we want the average number of comparisons

$$E \left[\sum_{u=1}^m \sum_{v>u}^m X_{u,v} \right] = \sum_{u=1}^m \sum_{v>u}^m E[X_{u,v}] =$$

The average of an indicator value (which $X_{u,v}$ is) is the sum of all possible values times their probability

$$\begin{aligned} &= \sum_{u=1}^m \sum_{v>u}^m (1 \cdot P(S[u] \text{ is compared to } S[v]) + 0 \cdot P(S[u] \text{ is not compared to } S[v])) = \\ &= \sum_u \sum_v P(S[u] \text{ is compared to } S[v]) \end{aligned}$$

Let's focus on the probability that taken two items they are compared. Comparison in quick sort is particular, may only occur in the partition part and when one of the elements is a pivot. With just one pivot and two parts, then $S[u]$ is compared to $S[v]$ only when one of them is the pivot. So $P(S[u] \text{ is compared to } S[v]) = P(S[u] \text{ or } S[v] \text{ is chosen as pivot in one call that involves both of them})$

Case a : the pivot is smaller or greater than both items, so both items go to the same partition: not significative because they can be compared in the next time

Case b : the pivot is between the two items, so no comparison because they'll go to different partitions.

Case c : is that one of the two items is taken as pivot, so **there's the comparison**.

So the probability is the number of times that c can happen (two items, so 2) divided by the total number of events possible, $b + c$, which is any item chosen between the two elements included, so $S[v] - S[u] + 1$. So $P(S[u] \text{ is compared to } S[v]) = P(S[u] \text{ or } S[v] \text{ is chosen as pivot in one call that involves both of them}) = \frac{2}{S[v] - S[u] + 1}$

k th ranked item In unsorted array S : is the item that, taken S , would go to the k th place is S would be sorted. The obvious way of finding out would be to sort and pick the item in the position k , but you'd be paying $n \cdot \log n$. This problem can be solved in linear time $O(n)$ on average: same base as quick sort with just one recursive call. Pick a random pivot and do a three-way partition: a partition of $n_{<}$ smaller elements than the pivot, a partition of $n_{=}$ elements of the same value as the pivot and a partition of $n_{>}$ larger elements than the pivot. Of course the left and right partition, of respectively smaller and bigger elements, are unsorted (the middle partition is of equal elements so is trivially sorted) but if the elements falls into the middle partition we can immediately know which elements it is. For example, $[3, 2, 0, 7, 7, 7, 10, 8, 9]$, if k is 4 we can answer that, being that there are 3 elements in the left partition, the 4th ranked element is a 7. Then if k falls in the $=$ partition, I'm done. If it falls in one of the other two, I cannot know but we can already discard some items: if the pivot is in the $<$ part we recurse on that part, otherwise if the pivot is on the $>$ part we recurse over there (in this case, dropping from k the items that are smaller, so $k = k - n_{=} - n_{<}$). Only one recursive call.

$T(n) = O(n) + [T(n_{<}) \text{ or } T(n_{>})]$ choice of going to the left or to the right, probability. Let's analyze that probabilities. Supposing that $n_{=} = \frac{n}{3}$, a balanced partitioning, with the pivot falling among the $n_{=}$ items (the middle partition). So this is the good case: the pivot is between the position $\frac{n}{3}$ and $\frac{2}{3}n$, and happens with probability $\frac{1}{3}$. The bad case is the pivot falling into the left or right partition.

In the good case $n_{<} \leq \frac{2}{3}n$ and $n_{>} \leq \frac{2}{3}n$. So

$$T(n) = O(n) + (P(\text{good case}) \cdot T() + P(\text{bad case}) \cdot T()) \Leftrightarrow$$

$$\Leftrightarrow T(n) = O(n) + \left(\frac{1}{3} \cdot T\left(\frac{2}{3}n\right) + \frac{2}{3} \cdot T(n-1) \right) \Leftrightarrow$$

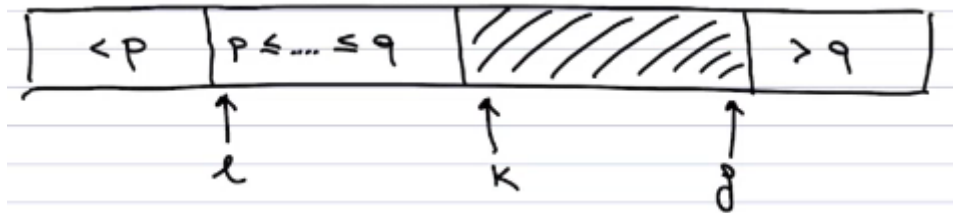
$$\Leftrightarrow T(n) \leq O(n) + \frac{1}{3} \cdot T\left(\frac{2}{3}n\right) + \frac{2}{3} \cdot T(n) \Leftrightarrow$$

$$\Leftrightarrow \frac{1}{3}T(n) \leq O(n) + \frac{1}{3} \cdot T\left(\frac{2}{3}n\right) \Leftrightarrow$$

$$\Leftrightarrow T(n) \leq O(n) + T\left(\frac{2}{3}n\right)$$

Dual Pivot (2012) 3-way partitioning made with 2 pivot taken at random: it reduces the number of comparisons and it increases the number of swaps. This seems totally ineffective (swaps impacts on memory and comparison on CPU), but the problem here is mispredictions on the CPU, modern CPU do look-ahead in the computing and if it badly predicts the future ifs it has to break the stack and go back. So with fewer comparisons it's better, it reduces the mispredictions. 10% circa speed-up factor.

The pivots maintain an invariant: the first partition, until index $l-1$, are items smaller than the first pivot (p), then there's a partition of items $p \leq \dots \leq q$ with q being the second pivot and this partition starts at l and ends at $k-1$. Then there's a partition of items that are still to be processed from index k to g , and at the end a partition of elements $> q$.



We take $S[k]$ and compare it to p . If $S[k] < p$ then swap (insert $S[k]$ in position l and increase l). Else, we compare $S[k]$ with q . If $S[k] \leq q$ we don't have to do anything, it's already in the correct position. Else, $S[k] > q$: we start from g and go back until we find an element $\leq q$ and we swap $S[k]$ with that element.

Bounded Quick Sort Bounding the recursive calls. Transform the deepness of the tree (the number of recursive calls) from n to $\log n$, but the time complexity remains n^2 in the worst case.

The trick is removing tail recursion. Techniques applied by compilers, we'll do by hand.



Generic call, between positions i and j . If $j - i > n_0$ (so if is sufficiently small, the array is in cache so we go fast) then we do insertion sort.

In the other case, with $\frac{i+j}{2}$ being the middle element, we do the partitioning: assuming the pivot on the left of the middle ($i \leq p < \frac{i+j}{2}$), the left partition of the pivot is smaller so contains less than half elements. So $i \leq p < \frac{i+j}{2}$, then recurse on the left because it has fewer elements, but we do not recurse on the right because it has a lot of elements so after the recursive call on the left we simply move i by doing $i = p + 1$. Else, with p on the right part, recurse on the right partition and then do $j = p - 1$

Multi Way Quicksort Key idea, given that we have a long array S to be sorted so doesn't fit in internal memory: we want to partition S among k buckets $B_1, B_2, B_3, \dots, B_k$. Every bucket is associated to a pivot: $k - 1$ pivots, a logical pivot $s_0 = -\infty$ at the beginning, $s_{k+1} = \infty$ at the end and $s_1, s_2, s_3, \dots, s_{k-1}$ between the buckets.

$B_i = \{S[j] \mid s_{i-1} < S[j] \leq s_i\}$, so the "right" pivot is part of the bucket ($s_i \in B_i$). The goal is having balanced $B_i \forall i$, so $|B_i| = \frac{n}{k}$.

The idea is similar to marge sort but in reverse: we assume that in memory M we have k buckets of size B , so we want to guarantee that $k \cdot B \leq M$. We sample at random $(k - 1)$ pivots from S (done with a scan $O(\frac{n}{B})$ I/Os). Not easy, sampling at random, we will see. We sort the the pivots $s_1 < s_2 < \dots < s_{k-1}$, with 0 I/Os because those are all in internal memory. But we cannot assume that the buckets in memory can contain the S 's buckets. So we fetch B items from S and distribute the items according to the pivots and do until a bucket is full. When a page is full, we write it as the first page of the corresponding buckets (according to the pivot). This takes $O(\frac{n}{B})$ I/Os.

1. Pick at random $(a + 1)k - 1$ samples from S , the pivots $\Rightarrow A$
2. Sort the samples in A
3. $s_i = A[(a + 1)i]$
Until the last block has a items



We sampled $(a + 1)(k - 1) + a = (a + 1)k - 1$ items

If $a = 0$ the number of samples is $k - 1$, no oversampling, fast sort ($O(k \log k)$) but no room to play around with the items. If $a > 0$ the sorting costs $O((ak) \log(ak))$. We chose $a = \Theta(\log k)$, only a logarithmic oversampling to get the balance

$$a + 1 = 12 \ln k$$

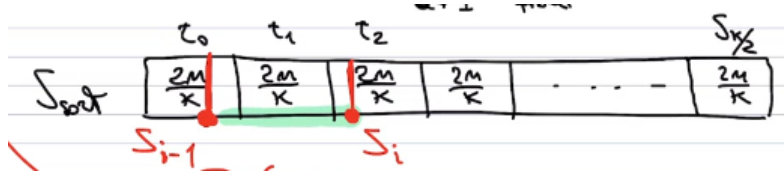
Balanced $\Rightarrow |B_i| < 4 \cdot \frac{n}{k}$

Proof By contradiction. We want to estimate $P(\exists B_i \mid |B_i| \geq \frac{4n}{k})$ and prove that it is $\leq \frac{1}{2}$ so that on average two extractions are enough to guarantee that one is correct.

$B_i = \{S[j] \mid s_{i-1} < S[j] \leq s_i\}$ by definition. A_{sort} contains samples from S .

S_{sort} is composed of $\frac{k}{2}$ cells of size $\frac{2n}{k}$: $t_0, t_1, \dots, t_{\frac{k}{2}}$.

Some pieces of t_i will be totally covered by a bucket B_j , because since B_i has size of at least $\frac{4n}{k}$ it's not possible for it to be totally included into a t_i . B_j has twice the size, so at least one t_i will be totally covered by it. So this means that $P(\exists B_i \mid |B_i| \geq \frac{4n}{k}) \leq P(\exists t_j \mid t_j \text{ is covered entirely by some } B_i)$. So the element before the beginning of B_i , which is s_{i-1} , it's contained in some point of some t_{j-1} and the last element, s_i , is in some part of t_{j+1} .



This means that the a samples are distributed in that part, between s_{i-1} and s_i : part in t_{j-1} , part in t_j and part in the beginning of t_{j+1} . How many samples inside t_j , the block covered by B_i ? They are surely $< a + 1$, so

$P(\exists t_j \mid t_j \text{ is covered entirely by some } B_i) \leq P(\exists t_j \mid t_j \text{ contains less than } a + 1 \text{ samples})$

We now apply the **union bound**: the probability of existence is upper bounded by the number of possibilities times the probability of one of the events. So $P(\exists t_j \mid t_j \text{ contains less than } a + 1 \text{ samples}) \leq \frac{k}{2} \cdot P(t_1 \text{ contains less than } a + 1 \text{ samples})$. We need to estimate the latter, how many samples will go to t_1 . $P(\text{sample occurs in } t_1) = \frac{2n}{k} = \frac{2}{k}$, and the average number of samples in t_1 , given $X_i = 1$ if the i th item is in t_1 , is $X = \#$ of samples in $t_1 = \sum_{i=1}^{(a+1)k-1} X_i$. So $E[X] = E[\sum_i X_i] = \sum_{i=1}^{(a+1)k-1} E[X_1] = ((a+1)k - 1) \frac{2}{k} \geq 2(a+1) - 1$ because $k \geq 2$ and $2(a+1) - 1 \geq \frac{3}{2}(a+1)$.

$$P(p \in D) = \text{positive events} / \text{total number} = \frac{|D|}{n} < \frac{m}{n} \leq \frac{n}{n} = 1$$

n is known, streaming model, $M = 2$, $n = 8$ items which are $[a, b, c, d, e, f, g, h]$, $P = [.5, .5, 0, .5, 1, 0, 1, 1] \leftarrow \text{Rand}(0, 1)$

$P(\text{sample } i_j) = \frac{m-s}{n-k+1}$ with s being

$s = 0, j = 1$ extracts a probability (.5) and compare P against the $P(\text{sample } i_j)$

$P = .5 \leq \frac{2-0}{8-1+1} = \frac{1}{4}$, first item not picked

$s = 0, j = 2, P = .5 \leq \frac{2-0}{8-2+1} = \frac{2}{7}$, not picked

$s = 0, j = 3, P = 0 \leq \frac{2-0}{8-3+1} = \frac{1}{3}$, c is picked

$s = 1, j = 4, P = .5 \leq \frac{2-1}{8-4+1} = \frac{1}{5}$, not picked

$s = 1, j = 5, P = 1 \leq \frac{2-1}{8-5+1} = \frac{1}{4}$, not picked

$s = 1, j = 6, P = 0 \leq \frac{2-1}{8-6+1} = \frac{1}{3}$, f is picked

$M = 2$ and we've picked two items, the end

In the reservoir sampling, still streaming model but n is unknown. $S = [a, b, c, d, e, f, g, h, i \dots]$, $M = 3$. Reservoir initialized with first M items, $R = [a, b, c]$. Each value in S has an integer $\rightarrow, -, \rightarrow, 2, 4, 1, 2, 3, 1$, with the last 1 is the h position, always drawn from the interval $[1, j]$. The number assigned are always smaller than the position j .

We consider the item d , extracts position $2 < 3 = M$ and substitutes b in the reservoir, $R = [a, d, c]$

e , position $4 > 3$ nothing changes

f , position $1 < 3$ and substitutes a , $R = [f, d, c]$

g , $R = [f, g, c]$

h , $R = [f, g, h]$

i , $R = [i, g, h]$

Disk Striping $D > 1$. Several disks, the pages of the disks are linked: first page of all D disks are a single page. So it's considered as a single disk with $B' = D \cdot B$

In sorting, the bound is $O(\frac{n}{DB} \cdot \log_{\frac{M}{DB}} \frac{n}{M})$ over D disks with disk striping. The lower bound is $\Omega(\frac{n}{DB} \cdot \log_{\frac{M}{B}} \frac{n}{M})$

$$\frac{\frac{n}{DB} \cdot \log_{\frac{M}{DB}} \frac{n}{M}}{\frac{n}{DB} \cdot \log_{\frac{M}{B}} \frac{n}{M}} \geq 1$$

$$\frac{\log_2 \frac{n}{M}}{\log_2 \frac{M}{DB}} \cdot \frac{\log_2 \frac{M}{B}}{\log_2 \frac{n}{M}}$$

$$\frac{\log_2 \frac{M}{B}}{\log_2 \frac{M}{B} - \log_2 D}$$

$$\frac{1}{1 - \frac{\log_2 D}{\log_2 \frac{M}{B}}}$$

$$\frac{1}{1 - \log_{\frac{M}{B}} D}$$

$\frac{M}{B}$ typically thousands, D at most 10-20, so $\log_{\frac{M}{B}} D < 1$. $M \rightarrow \infty \Rightarrow 1$ optimal. $M \rightarrow DB \Rightarrow \infty$, very very slow.

0.3 Randomized data structures

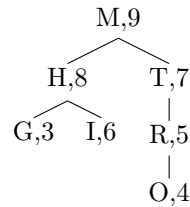
0.3.1 Treap

From binary search TREe and hEAP.

Every node is divided in two part: one part is the key and the other is the priority. A treap is a binary search tree according to the key and a heap according to the priority. To distinguish, letters for the key and numbers for the priority and we consider a maximum heap.

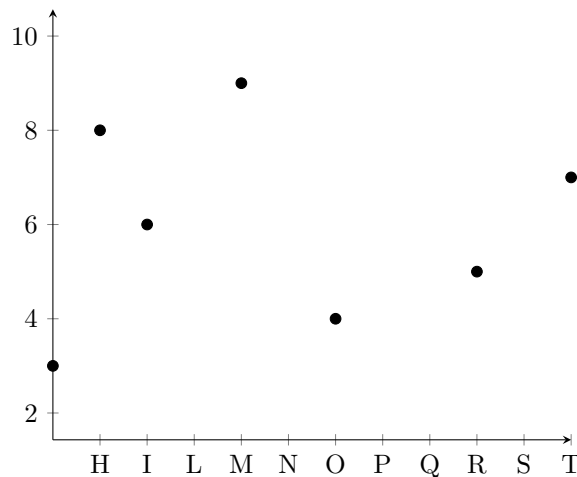
Taken a node, in the left subtree the keys will be smaller than the root, the right will have larger keys. The left and right tree, being a maximum heap, have smaller priorities: so the root has the maximum priority.

Let's consider the following Treap:

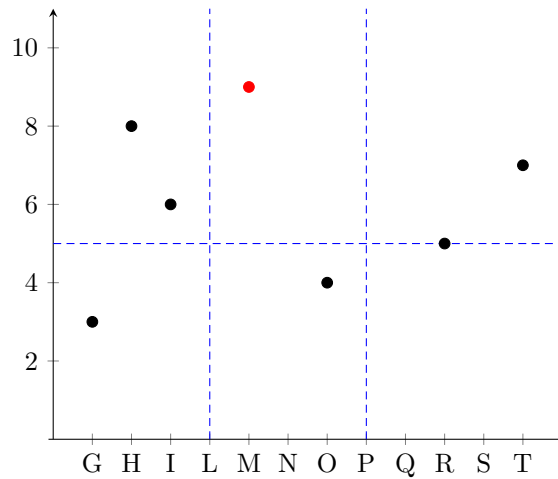


The keys are given, but the priorities are extracted at random. So the idea is that whenever you have a key to insert, you draw at random a number that you associate to the key as a priority. The randomized priorities makes the Treap balanced on average.

3-Sided Range Query We build a graph with the keys on the x axis and the priorities on the y axis.

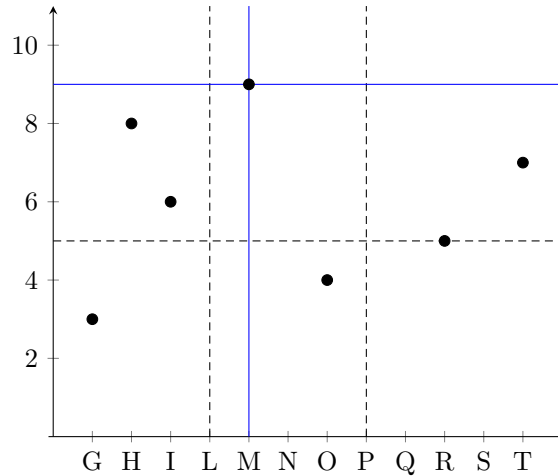


A typical query is indicated by $[a, b] \times [c, \infty]$, with $[a, b]$ on the x axis and $[c, \infty]$ on the y axis. The x axis so is bounded, for example by $[L, P]$, and for example $[5, \infty]$ makes the priority of the query we are looking for start from 5 upward. So the example becomes:

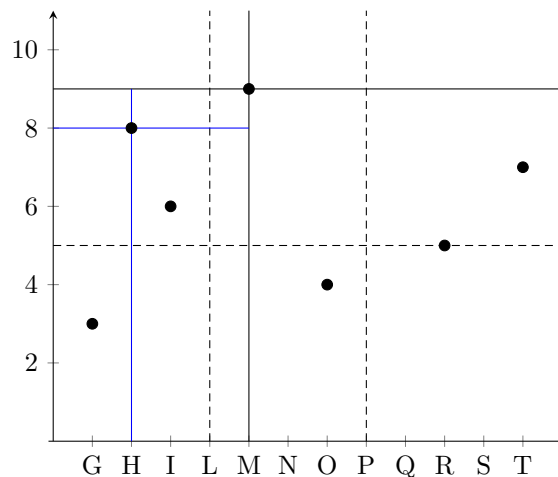


The data structure should return the $M, 9$ point, marked in the plot, because it's the only point in the set the resides in the query.

The root has highest priority, so in the plane is the top point. The left subtree is the subplot on the lower-left side of the root point, and the right subtree is the lower-right subplot



Then you recurse: find the root in the left subtree (topmost point) and create two other subplots **but limited to this subplot**. Example in the left subplot:



This was the approach in computational geometry, but the data structures are exactly the same.

So we start from the root, and we check if it's in the 3-sided range query. In this case is in the query, so we return it as a result. Since the root is inside the query, we go both left and right: this corresponds to going to the left and right subtree in the treap. In the left subtree, we recurs: is the root in the query? In this case, no but the height (priority) is still larger than the "bottom", the c value, we go down. But we don't go to the left subtree (it's totally discarded)

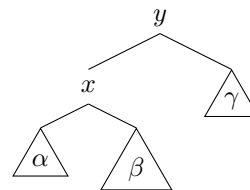
we go to the right subtree. If we went to the right subtree from the topmost root, that point it's still not in the query and still above c . Being in the right subtree, this time we go its left subtree and discard its right subtree. Basically we move toward the b value when we are on the right and toward a when we are on the left.

When we arrive in a node below the c value we know that, being a maximum heap, everything below that has a lower priority so we can drop everything.

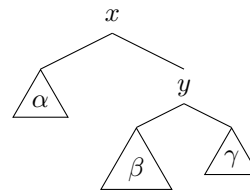
As for the **complexity**, we want to evaluate how much "wasted work" is done, as in how many nodes outside the query we evaluate. The number of points is equal to the depth of the tree, because we go only left or only right and always down one level, so given h height of the treap, the number of extra points we check is $O(h)$, one node extra per level at most. Inside the query we evaluate both left and right subtrees, but those are all "good" points that go into the result, so it's no wasted work. The particular case happens when we check the node just above c and in the query zone: for each of those nodes we check both subtrees, all of them will have priority $< c$ and will be cut out, but we check them. Those subtrees are at most twice the number of points of the result, the number of occurrences. So the total number of work wasted is h to the left $+h$ to the right $2 \cdot$ occurrences. The total cost then is $O(h + \text{the number of occurrences})$. Of course the occurrences cannot be avoided, as we have to list them.

The extra cost, h , is typically $\log n$ if the treap is balanced with n being the number of points.

Rotation Considering



In the right rotation, node x goes up and node y goes down. α was the left subtree of x and remains that, same for γ that remains the right subtree of node y . β cannot remain linked to x , so β will be relocated as the left subtree of y .



We claim that this relocation still guarantees the balancing of the treap. We can pick a node k inside β

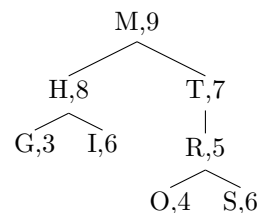
$$\text{key}(x) \leq \text{key}(k) \leq \text{key}(y)$$

because it's in the right subtree of x (therefore, larger key) but it's inside the left subtree of y (therefore, smaller key). When we go to β as left subtree of y after the rotation, so y inside the right subtree of x , we have that a node k inside β satisfies the same condition: it's still in the left subtree of y , therefore has smaller key, and still in the right subtree of x , therefore greater key. So the relocation preserves the conditions.

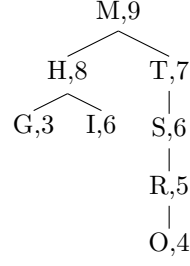
It's constant time, $O(1)$, because we change a constant number of pointers.

Search(k) It's just the search in the binary search tree. Based on the searched key, we go on the left if it's smaller and on the right if it's larger, so priority is not needed. Cost is $O(h)$ with h being the height of the tree.

Insertion(k, p) For example inserting $(S, 6)$. We start by searching for the position of S in the tree. Bigger than M, smaller than T, larger than R and empty so that's the spot.



It's still a binary search tree but it's not an heap because the priority 6 doesn't satisfy the condition. So like in the heap we have to check the balance: S,6 needs to go up but we have to preserve the heap property so we apply a rotation, in this case a left rotation because it's the right child.



So it's a search plus rotation to reestablish the heap property. The rotation is constant, but in the worst case there are h rotations. So the cost is again $O(h)$.

Delete(k) The deletion is trivial when we delete leaves. But for example let's remove T: we can make it a leaf by setting the priority to $-\infty$. So deletion is just: search for k , set the priority to $-\infty$ and then apply rotations to reestablish the heap property, then delete. Again, $O(h)$.

Of course in this particular case T has just one child, so we could remove it and link S as right subtree of M, but the procedure above handles every possible case.

Merge(T_1, T_2) $\forall x \in T_1$ and $\forall y \in T_2$ we have $\text{key}(x) < \text{key}(y)$.

We create a root with a logical key between the maximum key of T_1 and the minimum key of T_2 and $-\infty$ as the priority. The problem is the priority: it's a binary search tree (both T_i are, and the chosen key maintains that condition) but the $-\infty$ priority breaks the heap condition. But we know we can apply rotations to reestablish the condition: the root goes down until it reaches the leaf level, where it can be dropped. When this happens, the rest of the "stuff" will be a treap. So again it's $O(h)$: creating the node it's constant and the rotations are $O(h)$.

The reverse operation is the split.

Split(k, T) Splitting T according to the key k means creating a treap $T_{<k}$ with keys $< k$ and another treap $T_{>k}$ with keys $> k$. This assuming $k \notin T$. The split is trivial when the key is the root, so I need to put myself in a situation where k occupies the root. So we first insert k, ∞ , so that it goes to the root. At insertion, k, ∞ will be a leaf, so we rotate to reestablish the heap priority, arriving in a situation where we have k, ∞ as root with a left and a right subtree. The left subtree will be $T_{<k}$ and the right subtree will be $T_{>k}$ so the we delete the root. This costs $O(h)$ for insertion, $O(h)$ for rotations and constant for deleting the root, overall $O(h)$.

On average, a treap is balanced By drawing the priorities at random, the average treap is balanced. h height of the treap, it's n if it's just a path or $\log n$ if it's balanced, but $h = O(\log n)$ on average. Let's prove it.

Let's create an indicator variable $A_k^i = \begin{cases} 1 & \text{if } x_i \text{ is a proper ancestor of } x_k \\ 0 & \text{otherwise} \end{cases}$

$\text{depth}(x_k) = \sum_{i=1}^k A_k^i$ which counts the proper ancestors of x_k . $E[\text{depth}(x_k)] = E(\sum_{i=1}^k A_k^i) = \sum_{i=1}^k E[A_k^i]$ and the expected of a indicator variable is the probability of it being 1 so $= \sum_{i=1}^k P(A_k^i = 1)$ which is the probability of x_i being on the path to x_k . We consider $X(i, k) = \{x_i, x_{i+1}, \dots, x_{k-1}, x_k\}$ sorted. So $\sum_{i=1}^k P(A_k^i = 1)$ we consider $x_i < x_k$ or vice versa, is totally symmetric. So $\text{priority}(x_i) > \text{priority}(x_k)$

Theorem: $\forall i \neq k$ we have that x_i is a proper ancestor of $x_k \Leftrightarrow \text{priority}(x_i)$ is the largest in $X(i, k) = \{x_i, \dots, x_k\}$. In the minimum heap the priority would be the smallest.

Various cases:

1. x_r with largest priority, so $x_r, \dots, x_i, \dots, x_k$, so x_r becomes the root of the treap and the x_i, \dots, x_k go to the right subtree. So we consider the subtree.
Same for x_r with largest priority and $x_i, \dots, x_k, \dots, x_r$. x_r still root and x_i, \dots, x_k in the left subtree.
So we restrict the attention to the $X(i, k)$ set.
2. x_i has the largest priority $\Rightarrow x_i$ is the root of the treap and x_k goes to the right, so x_i is the ancestor.
3. x_j has the largest priority, where $x_j \neq x_i$ and $x_j \in X(i, k)$, it's one of the items $\Rightarrow x_j$ to the root, x_i to the left and x_k to the right. So x_i is surely **not an ancestor** of x_k .

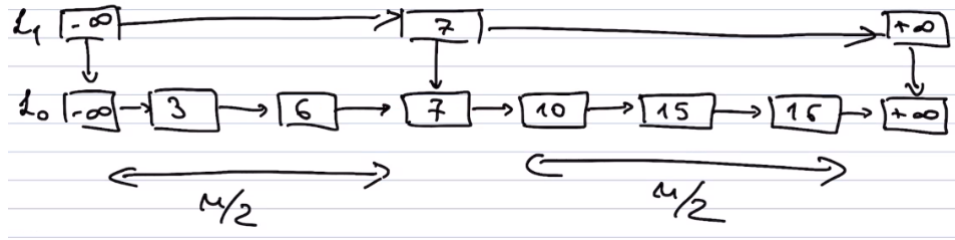
$P(x_i \text{ ancestor of } x_k) = P(x_i \text{ has the largest priority of } X(i, k))$ and because the priorities are picked are random $= \frac{1}{|X(i, k)|} = \frac{1}{k-i+1}$ if $i < k$ or $\frac{1}{i-k+1}$ if $i > k$. Not $\frac{1}{n}$ because we consider only the subset of $X(i, k)$

$\sum_{i=1, i \neq k}^n \frac{1}{|X(i, k)|} = \sum_{i=1}^{k-1} \frac{1}{k-i+1} + \sum_{i=k+1}^n \frac{1}{i-k+1}$ the first goes like $\log_2 k$ (because it goes like $\frac{1}{k} + \frac{1}{k-1} + \dots + \frac{1}{2}$) and the second goes like $\log_2(n - k + 1)$ (because it goes like $\frac{1}{2} + \frac{1}{3} + \dots + \frac{1}{n-k+1}$)

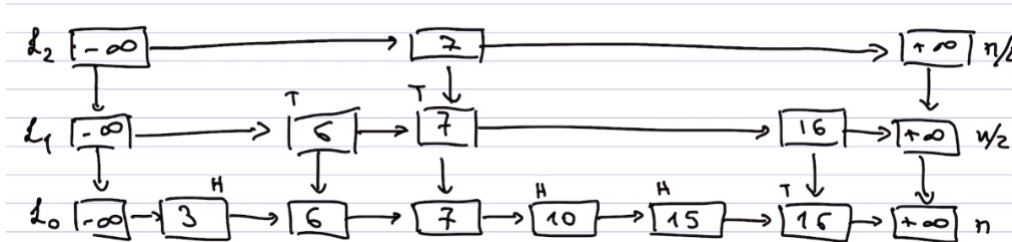
Keys are inserted in random order so that priority = order, so average depth is $O(\log n)$.

0.3.2 Skip lists

We have a list of items $L_0 = -\infty \rightarrow 3 \rightarrow 6 \rightarrow 7 \rightarrow 10 \rightarrow 15 \rightarrow 16 \rightarrow +\infty$ with $\pm\infty$ logical marks. The search in lists is very bad, the idea is to create levels of lists. List of level $L_1 = -\infty \dots +\infty$ with vertical pointers.



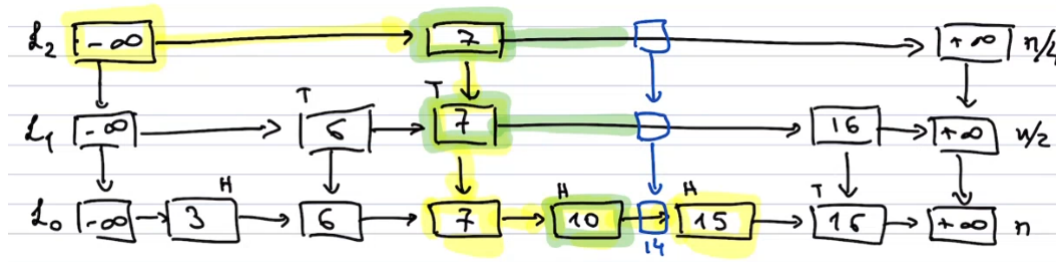
We can promote k items to L_1 , $k + \frac{n}{k} \Rightarrow k = \sqrt{n}$. The items are promoted randomly. For every item in L_0 , it is promoted with probability $= \frac{1}{2}$ (coin toss, tails promotes heads doesn't), promoting $\frac{n}{2}$ items on average. This can be done for more levels.



Search We search for the topmost list, compare and go to the correct portion in the list below. At every step I scan the list if the element I'm looking for is larger and go to the list below if it's smaller.

Deletion The deletion removes the column of items, in every list that item appears. In the worst case we reconstruct $\log n$ lists.

Insertion First you find the position in L_0 (search, beginning from the topmost list), you insert and then toss the coin to insert in the upper list. To identify the left/right part in the upper list, we look for the latest item for every level that I have seen before going down during the search phase, in a sense that's the frontier to the item to be inserted.



So keep track of the items that go down and adjust the pointers of those items.

Height of the skip list $L = \#$ of levels of the skip list on n items $= \max_k L(k)$ with $L(k) =$ level of the k th item. So the height of the skip list, the levels of the skip list, is the maximum level among the items.

$P(L(k) \geq l) =$ the probability of extracting l tails (independent coin tosses) $= (\frac{1}{2})^l$.

$P(L \geq l) = P(\max_k L(k) \geq l) = P(\exists k \mid L(k) \geq l) \leq n \cdot (\frac{1}{2})^l = \frac{n}{2^l}$ for the union bound.

1. $l \leq \log_2 n \rightarrow \frac{n}{2^l} \geq \frac{n}{2^{\log_2 n}} = \frac{n}{n} = 1$ so the inequality is not significant, because a probability is always ≤ 1

2. $l > \log_2 n \rightarrow \frac{n}{2^l} < \frac{n}{2^{\log_2 n}} = 1$

$$P(L \geq c \cdot \log_2 n) \leq \frac{n}{2^{c \cdot \log_2 n}} = \frac{n}{(2^{\log_2 n})^c} = \frac{n}{n^c} = \frac{1}{n^{c-1}}$$

In the search process we have, with high probability, $O(\log n)$ "down steps". For the number of "right steps", let's observe the process in reverse: from the bottom we go to the left for as many right pointers we find, but we go up as soon as we find a downward pointer. How many right pointers before a vertical pointer on average? We go left when we don't have a vertical pointer, which means with probability $\frac{1}{2}$ because of the coin toss, so it's a sequence of heads terminate by a tail (which means going up) so on average we have 2 right steps. Every level is independent, so it applies to every level.

So $O(\log n)$ down steps and $O(\log n)$ right steps for the search process, so the total cost of search is $O(\log n)$. Insertion is a search plus a constant number of pointers change, deletion is a search plus a constant number of pointers change so everything is $\log n$.

0.4 Set Intersection

From a mathematical point of view we have $A, B \subseteq N$ and we want to compute $A \cap B$ fast. With $|A| = n, |B| = m$, in the case of unsorted sets, the comparison has complexity of $O(n \cdot m)$. We assume sorted sets, so we first sort A and B and then apply one of the algorithms that we'll see.

0.4.1 Merge Based Intersection

Based on the merge procedure seen in merge sort. For example $A = \underline{1}, 5, 8, 9, \dots$ and $B = \underline{2}, 5, 9, \dots$ with pointers on $1 \in A$ and $2 \in B$, the first elements. In the merge sort we do the comparison and write the minimum, advance the pointer in the corresponding list and go on. In the example we would end up with $1, 2, 5, 5, 8, \dots$. But we want to simplify this. In this algorithm, whenever two items are compared, the minimum one is discarded and if the items are equal then it goes into the intersection.

It's linear in both lists, so the complexity is $O(n + m)$. This approach is optimal whenever n is in the same order of m , or equal.

0.4.2 Binary Search

Assuming $n > m$. We take an element from B and binary search in A , for each element of B . Every search takes $\log n$, so this costs $O(m \cdot \log n)$. This approach is better than the previous, of $O(n + m)$ complexity, when $m \log n < n \Leftrightarrow m < \frac{n}{\log n}$. So it's better up to a very long B , which means that doing a continuous binary search is better than the merge based intersection.

This is true in theory, in practice scanning (the merge based intersection) is typically very fast.

0.4.3 Mutual Partitioning

Still $n > m$. Let's say we take the first item of B , binary search over A and find that it's between two certain values in A . When we pick the second item of B , in the binary search we start from scratch, but it's nonsensical because the second item is larger than the previous item of B , so we can search only in the last right partition of A that's left. This way we always go to the right, but if the first item is at the beginning of A we don't advance very much so we try to balance the partitioning: instead of picking the first element of B , we pick the middle element in position $\frac{m}{2}$. That middle element is found to occur between two elements in A , and this splits A into A_1 , until the left element included, and A_2 from the right element included. This also splits B into two parts: the left half until the middle element excluded, B_1 , and the right half from the middle element excluded, B_2 .

After $\log n$ steps, the binary search, we have turned $A \cap B$ into two problems: $A_1 \cap B_1$ and $A_2 \cap B_2$, over which we can recurse.

We have that $|B_1| = |B_2| = \frac{m}{2}$, with a ± 1 in case of odd m . Those two partitions will be again divided, and so on. So we make $\log m$ recursive steps. So the complexity $T(n, m) = O(\log n)$ (for the binary search) $+ T(n_1, \frac{m}{2})$ (for recursion over $A_1 \cap B_1$) $+ T(n_2, \frac{m}{2})$ (for recursion over $A_2 \cap B_2$). To solve this we have to make some assumptions:

If the partitioning is fully unbalanced (the middle element of B goes either all the way to the left or all the way to the right of A)

It's a good case because, although we keep all A we still drop half of B . We either have $A_1 = \emptyset$ or $A_2 = \emptyset$ but still dropping $\frac{m}{2}$ elements. So in this case $n, m \rightarrow n, \frac{m}{2} \rightarrow n, \frac{m}{4} \dots$ supposing that every step is fully unbalanced. We have all n , so the cost is guided by the binary search over A with $|A| = n$ which we do $\log m$ times, so the total cost is $O(\log n \cdot \log m)$

The worst case is the opposite of this one, which is the best case.

If the item of B that we search ends up in the middle of A , so $|A_1| = |A_2| = \frac{n}{2}$ and $T(n, m) = O(\log n) + T(\frac{n}{2}, \frac{m}{2}) + T(\frac{n}{2}, \frac{m}{2})$ which if solved gives $T(n, m) = O(m(1 + \log \frac{n}{m}))$

We will comment on how this is an optimal bound and how we cannot do less comparisons than this one.

A theorem states that if we have $s(n)$ solutions, a comparison based algorithm must execute $\Omega(\log_2 S(n))$ comparisons. In this scenario, $|B| = m$ can intersect $|A| = n$ in $\binom{n}{m}$ ways: all the possible ways in which we can take m items out of n items. Inputting this as $s(n)$ gives the $T(n, m) = O(m(1 + \log \frac{n}{m}))$ seen before.

0.4.4 Doubling/Exponential/Galloping Search

Let's take A, B and assumes that they intersect up until the 12, in position $a_{i_{j-1}} b_{j-1}$. We have

$A = - - 12, 16, 19, 20, 25, 27, 30, 31, 34, 38, 40, 41, 44, 45, 47, 50, 60, \dots$

$B = - - 12, 41$

We compare the $41 \in B$ with 16, 19, 25 and then 60 $\in A$, respectively with step 1, 2, 4, 8, 16. So we don't go one-by-one but we jump exponentially, we go on until the element (41) is larger. When it's smaller, with 60, we stop. We discard the elements we skip until we encounter the larger element: now we know that 41, our element, is between this element and the previous element we compared with (34), so the longer the jump the more we discard.

41 is b_j , compared with $a_{i_{j-1}} + 2^k$ and we go ahead until $a_{i_{j-1}} + 2^{k-1} < b_j \leq a_{i_{j-1}} + 2^k$

When the element is found (let's say in position $a_f \leq a_{i_{j-1}} + 2^k$), the search for the next element, b_{j+1} will start from a_{f+1} . Also we always have $a_{i_{j-1}} \leq b_{j-1} < a_{i_{j-1}+1}$.

Complexity Let's see how much work is done in a repetitive way. $\Delta_j = \min\{n, 2^k\}$, because if $2^k > n$ we would jump outside the array, so $\Delta_j \leq 2^k$. Also, whenever I search for b_j we find it in the element a_{i_j} : we can say that $a_{i_{j-1}} + 2^{k-1} < a_{i_j} \leq a_{i_{j-1}} + 2^k$, so $2^{k-1} < i_j - i_{j-1} \leq 2^k$ which means $2^k < 2(i_j - i_{j-1})$. So we have that $\Delta_j < 2(i_j - i_{j-1})$
 $\sum_{j=1}^m \Delta_j < 2 \sum_{j=1}^m (i_j - i_{j-1})$ which is a telescopic sum: $i_1 - i_0 + i_2 - i_1 + i_3 - i_2 + \dots + i_m - i_{m-1}$, where we can simplify all the elements except for $i_0 = 0$ and i_m . So we have $\sum_{j=1}^m \Delta_j < 2i_m$, and i_m is the position of the last element of B inside A , which cannot be greater than m , so $i_m \leq n$, which means $\sum_{j=1}^m \Delta_j \leq 2n$

The algorithm does $\log_2 \Delta_j$ jumps plus the binary search inside the last portion (when we encounter the greater element): that portion is surely smaller than Δ_j . So the algorithm costs $\sum_{j=1}^m O(\log_2 \Delta_j + \log_2 \Delta_j)$, so $\sum_{j=1}^m O(\log_2 \Delta_j) = O\left(\sum_{j=1}^m \log_2 \Delta_j\right)$. The Jensen inequality says that this can be upper bounded by $O\left(m \cdot \log_2 \frac{\sum \Delta_j}{m}\right) = O(m \cdot \log \frac{2n}{m})$

0.4.5 Two-level memory approach

As usual A, B with $|A| = n > |B| = m$ and we partition A into logical blocks of size L (memory word or page size, for example). For every partition we take the first key, and we build A' which consists of only those first keys elements, so $|A'| = \frac{n}{L}$. We want to find out in what blocks of A we can find the elements of B , by exploiting A' : the arrays are sorted, so this is enough.

We merge A' and B , and so we find in which blocks of A can fall the elements of B , by using the keys as dividers in the merged $A' + B$ array. This merge costs $O\left(m + \frac{n}{L}\right)$.

This also produces buckets in B , and B_i will be intersected with A_i . How many blocks of A_i will be considered at most? m , because B has m elements, so in the worst case is one element of B per block of A . So we have $|B| + mL = m + mL = O(mL)$.

So the total cost is $O(mL + \frac{n}{L})$. Not optimal but very good, one of the best algorithms.

0.4.6 Interpolation search

Search in a set $X \subseteq N$, which is sorted $X = \{x_1, x_2, \dots, x_n\}$. Let's define $b = \frac{x_n - x_1 + 1}{n}$ and partition X into buckets of size b of the universe. So with $b = 3$ the bucket $B_1 = \{1, 2, 3\}$, $B_2 = \{4, 5, 6\} \dots$

For every bucket we have pointers in the index array $I = 1, 3, 0, 0, 4, 5 \dots$ with the indexes of the beginning and the end of the bucket in X . If I want to search for y , given that I partitioned the universe I know already the bucket of y : it's B_J with $J = \lfloor \frac{y-x_1}{b} \rfloor + 1$, so we do a binary search over B_J looking for y . It costs constant, for computing J , plus the binary search over B_J which costs $O(\log_2 b)$

Theorem Time complexity is $O(\log_2 \Delta)$ where $\Delta = \frac{\max_{2, \dots, n} x_i - x_{i-1}}{\min_{2, \dots, n} x_i - x_{i-1}}$, the first time that a complexity depends on the distribution of the data.

Proof The maximum is surely larger than the average so $\max_{2, \dots, n} x_i - x_{i-1} \geq \frac{\sum_{i=2}^n x_i - x_{i-1}}{n-1}$, again a telescopic sum, so $= \frac{x_n - x_1}{n-1} \geq \frac{x_n - x_1 + 1}{n} = b$. So

$$\begin{aligned} \max_{2, \dots, n} x_i - x_{i-1} &\geq b \\ |B_J| &\leq \frac{b}{\min_{2, \dots, n} x_i - x_{i-1}} \leq \frac{\max_{2, \dots, n} x_i - x_{i-1}}{\min_{2, \dots, n} x_i - x_{i-1}} = \Delta \end{aligned}$$

0.5 Data Compression

Entropy Of a source, assuming an alphabet Σ of symbols (characters, or digits in the case of integers), with a probability distribution $\{p_\sigma\}$ over Σ .

The entropy of a source is $H = \sum_\sigma p_\sigma \cdot \log_2 \frac{1}{p_\sigma} = H_0$, sometimes specified as H_0 which means no context, every symbol is independent from the previous one.

We know that $H_0 \geq 0$ and $\sum_\sigma p_\sigma = 1$ because is a probability distribution. In order for $H = 0 \Leftrightarrow$ either $p_\sigma = 0$ or $\log_2 \frac{1}{p_\sigma} = 0 \Leftrightarrow p_\sigma = 0$ or $p_\sigma = 1$. Because the sum is 1, this means that if $\exists \sigma' \mid p_{\sigma'} = 1 \Rightarrow \forall \sigma \neq \sigma' \mid p_\sigma = 0$. So entropy is 0 when one symbol has probability one and the other have probability 0.

An upper bound is $H \leq \log_2 |\Sigma|$. The formula of the entropy has its maximum value when all the p_σ are equal, so $H \leq \log_2 |\Sigma| \Leftrightarrow p_\sigma = \frac{1}{|\Sigma|}$, a uniform probability: we would have $\sum_\sigma \frac{1}{|\Sigma|} \cdot \log_2 |\Sigma| = |\Sigma| \cdot \left(\frac{1}{|\Sigma|} \cdot \log_2 |\Sigma| \right) = \log_2 |\Sigma|$. So we have found that

$$\text{Highly skewed probability } 0 \leq H \leq \log_2 |\Sigma| \text{ Uniform probability}$$

We also have a **theorem**, of Shannon: any prefix-code for source Σ takes an average number of bits $\geq H$ per symbol of Σ .

Considering a random variable X_σ that gets value $\log_2 \frac{1}{p_\sigma}$ with probability p_σ , $E[X] = \sum_\sigma p_\sigma \cdot \log_2 \frac{1}{p_\sigma} = H$, so $\log_2 \frac{1}{p_\sigma}$ defines the information content of σ : the **entropy is the average information content of each symbol**.

This also states that the optimal encoding for a symbol σ is a codeword of length $\log_2 \left(\frac{1}{p(\sigma)} \right)$

Code C Algorithm that assigns sequence of bits (codewords) to symbols $\sigma \in \Sigma$.

For example, if $\Sigma = \{a, b, c\}$, $C = \begin{cases} a \rightarrow 0 \\ b \rightarrow 10 \\ c \rightarrow 110 \end{cases}$ It's a variable length code, because every symbol is encoded with a

variable number of bits, and a prefix-free code, because if we compare the codewords I note that no codeword is prefix

of another one. This is important because if we designed the code like $C = \begin{cases} a \rightarrow 0 \\ b \rightarrow 01 \\ c \rightarrow 1 \end{cases}$ and we receive 01 we cannot

unequivocally say if the codeword was produced by the sequence ac or by b. We want uniquely decodable codes.

We will focus on variable-length prefix-free codes.

Looking at C , can we optimize it? We could reduce c and assign it the codeword 11, but we cannot encode b with just 1 because we couldn't then get the prefix-free for c .

We have a golden rule: more frequent symbols get shorter codewords. We want to relate the length of the codeword to the frequency of its symbol. The **average length of a code** C in bits is $E[C] = \sum_{\sigma} p_{\sigma} \cdot \text{length}(\text{codeword}(\sigma))$. In the example, with $p(a) = \frac{1}{4}, p(b) = \frac{1}{2}, p(c) = \frac{1}{4}$, then $E[C] = p(a) \cdot |cw(a)| + p(b) \cdot |cw(b)| + p(c) \cdot |cw(c)| = \frac{1}{4} \cdot 2 + \frac{1}{2} \cdot 2 + \frac{1}{4} \cdot 3 = 2$ bits

Is the golden rule satisfied for the C in the example? No because b is the most probable symbol but is not the shortest

codeword. A better code for those probabilities would be $C' = \begin{cases} a \rightarrow 00 \\ b \rightarrow 1 \\ c \rightarrow 01 \end{cases}$, with b being the most probable symbol

having the shortest length and a and c having same length, where we would get $E[C'] = \frac{1}{4} \cdot 2 + \frac{1}{2} \cdot 1 + \frac{1}{4} \cdot 2 = 1.5$ bits which is better (and optimal, we will prove it).

$\forall C$ prefix-free $E[C] \geq H$ (Shannon), so if we can make the length of the codeword of σ exactly $\log_2 \frac{1}{p_{\sigma}}$ bits we have optimality. Formally, if $|cw(\sigma)| = \log_2 \frac{1}{p_{\sigma}} \Rightarrow C$ is optimal. But $\log_2 \frac{1}{p_{\sigma}}$ is a real number, which cannot be a length. For it to belong to N , we should have $p_{\sigma} = 2^{-x}$.

0.5.1 Integer coding

$\Sigma = N$, we will consider sequences of integers $S = s_1 s_2 \dots s_n$ with $s_i \in N$ and possibly repetitive (one or more of s_i can repeat). Sometimes we have increasing sequences, such that $s_i < s_{i+1}$: if we have a sequence like this we can transform it into a sequence comprised of the first element and the next element encoded as the distance from the first, the third element encoded as the distance from the original second, and so on. So $S' = 1 \ 3 \ 5 \ 6 \ 9 \dots$ will be $S = 1 \ 2 \ 2 \ 1 \ 3 \dots$, these are called d -gaps, or gaps, and are used in place of the increasing sequences for the encoding. Of course we can revert S to its original sequence S' by making the prefix sum: a number is the sum of the previous numbers, so by summing the numbers from left to right.

This increasing sequences occur in search engines (page ids), databases (keys), hash tables (positions of unempty slots)...

Also if $x \in Z$ we can encode them as integers with $2x$ if $x \geq 0$ or $-2x + 1$ if $x < 0$.

Fixed length code Up to now we took the maximum $s_i^* = \max_i s_i$ and encoded each s_i with $1 + \lfloor \log_2 s_i^* \rfloor$ bits, a fixed-size encoding.

With $m = s^*$, the fixed length code encodes s_i with $1 + \lfloor \log_2 m \rfloor$ bits.

This is optimal when

$$1 + \log_2 m = \log_2 \frac{1}{p(x)} \Leftrightarrow \log_2 2m = \log_2 \frac{1}{p(x)} \Leftrightarrow p(x) = \frac{1}{2m}$$

Which is a uniform distribution from 1 to m , apart from the 2, which is the case where the fixed-length encoding is optimal.

Unary code With $x > 0$, $U(x) = 00 \dots 01$ with $x - 1$ zeroes before 1, therefore the overall length is x bits, assuming $x > 0$.

Very efficient for small x , but with large x it's inefficient. For which distribution it's optimal? We apply what we've seen before:

$$|U(x)| = x = \log \frac{1}{p(x)} \Leftrightarrow 2^x = \frac{1}{p(x)} \Leftrightarrow p(x) = 2^{-x}$$

So it's optimal for highly skewed distributions, very concentrated to small numbers.

γ -code Improvement over the unary code, still for $x > 0$.

$$\gamma(9) = \begin{matrix} 000 & 1001 \\ |\text{bin}(9)| - 1 & \text{bin}(9) \end{matrix} = 0001001 \text{ with } \text{bin}(x) \text{ being the binary representation of } x.$$

In general, $\gamma(x) = |\text{bin}(x)| - 1$ zeroes followed by $\text{bin}(x)$.

$\text{bin}(x)$ surely starts with 1, so to decode you read the number of zeroes until you find the first 1 and then you know how many bits to read after that 1

$$|\gamma(x)| = 2\text{bin}(x) - 1 = 2(|\log_2 x| + 1) - 1 = 2\lfloor \log_2 x \rfloor + 1 \text{ bits}$$

Ignoring the floor, in order to get the optimal distribution for the γ -code we need

$$|\gamma(x)| = \log_2 \frac{1}{p(x)} \Leftrightarrow 2 \log_2 x + 1 = \log_2 \frac{1}{p(x)} \Leftrightarrow \log_2 2x^2 = \log_2 \frac{1}{p(x)} \Leftrightarrow p(x) = \frac{1}{2x^2}$$

Which is a distribution smoother than the one optimal for the Unary code.

The sequence 01000111 1011 is decoded as 10 111 1 11 = 2 7 1 3

From the CPU point of view is a bad code because of bit shifts: scan the number of bits and then read the number, and bits shifts break the pipeline of the processor.

Also for very large numbers we have lots of zeroes so a long decoding time.

δ -code Again for $x > 0$, $\delta(x) = \gamma(|\text{bin}(x)|)$ followed by $\text{bin}(x)$

For example, $\delta(14) = \begin{matrix} \gamma(4) & 1110 \\ \gamma(|\text{bin}(14)|) & \text{bin}(14) \end{matrix} = 00\ 100\ 1110$

δ -code is advantageous with large numbers.

$$\begin{aligned} |\delta(x)| &= |\gamma(|\text{bin}(x)|)| + |\text{bin}(x)| = (2 \log_2 |\text{bin}(x)| + 1) (\log_2 x + 1) = 2 \log_2 (\log_2 x + 1) + 1 + (\log_2 x + 1) = \\ &= 2 \log_2 \log_2 x + \log_2 x + 2 \Rightarrow p(x) = \frac{1}{2x(\log_2 x)^2} \end{aligned}$$

Rice encoding Depends on a parameter k , given $x \geq 0$ computes $q = \lfloor \frac{x}{2^k} \rfloor$ and $r = x - q2^k$ with $0 \leq r \leq 2^k$. For example, given $x = 13, k = 2$, we have $q = 3, r = 1$.

The Rice encoding is the unary encoding of $q+1$ ($U(q+1)$) followed by the binary encoding of r on k bits ($B(r)$) which can always be encoded in $\leq k$ bits because $r \leq 2^k$, so we can use always k bits. In our example $R_2(13) = 0001\ 01$, while with $k = 3$ we have $R_3(13) = 01\ 101$ (because $q = 1$ and $r = 5$).

Geometric distribution p successes and $1-p$ failures, the geometric distribution is $P(\text{the first success is after } x \text{ experiments})$. $P(1) = p, P(2) = (1-p)p, \dots, P(x) = (1-p)^{x-1}p$. Also 2^k is almost equal to $\frac{\ln 2}{p}$

R_k is optimal for the geometric distribution.

The problem with Rice is the unary encoding, which is hard to process (bit by bit)

P for delta encoding For a bounded set of integers, in the range of $0 \leq x < 2^w$, so a w bits sized word. Assuming that most of the integers are small, in the range $0 \leq x < 2^b - 1$, so can be represented with $b \ll w$ bits. For example, $w = 32, b = 4$.

I can represent my sequence with two arrays of bits, the first array has every integer of b bits and the second has w bits integers. For every symbol, if it's small enough I represent it in the first array with b bits, otherwise I'll put a special escape value in the first array and the symbol is stored in the second array with w bits.

Variable Byte encoding Idea is that given a variable i there's a binary representation $B(i)$, for example $i = 2$, then $B(i) = 10$. We use 1 bit out of 8 to make the code prefix free. So $B(i)$ is splitted in blocks of 7 bits, $B(i) = 00011011001111$. The most important 7 bits become a byte of 10001101 and the other 0101111. Continuation bit is one if the encoding continues and 0 if the encoding ends.

Wasteful because we never have the initial byte of the representation (most significant bits) can never be all zero.

Not used in practice, lets see how to avoid the waste of space. 1 byte: 128 values (2^7), with 2 bytes we have 14 bits and 2^{14} values and so on. Using one or two bytes we can encode $2^7 + 2^{14}$ different values. This is done as follows: suppose to have $0 \leq i < 2^7 + 2^{14}$. If $i < 2^7 \Rightarrow$ one byte with continuation bit = 0. If $2^7 \leq i < 2^7 + 2^{14}$ I encode $i - 2^7$ in two bytes. Clearly $0 \leq i - 2^7 < 2^{14}$.

Reasoning in the same way with 3 bytes we can encode up to $2^7 + 2^{14} + 2^{21}$ different integers.

(s, c) dense codes Again byte oriented. s stoppers and c continuers.

We had every value with last bit 0 was a stopper, while every value with continuation bit 1 was a continuer. This means that the values in range $[0, 127]$ were all stoppers, and in range $[128, 255]$ are continuers.

We have chosen to have s stoppers and c continuers. We must have $s + c = 256$, so for example even $[0, 50]$ stoppers and $[51, 255]$ continuers is acceptable. With one byte we necessarily have a stopper, so s possible values because there are s stoppers.

With 2 bytes, first one is a continuer and the second a stopper so total number is cs possible values.

With three bytes, 2 continuers and the last a stopper so c^2s possible values.

Using up to three bytes the total number of possible values is $s + cs + c^2s = s(1 + c + c^2)$ which is a geometric progression of ratio c so $s = \frac{c^3-1}{c-2}$.

In general using up to k bytes we can encode $s \frac{(c^k-1)}{c-1}$ different values.

Suppose we are encoding all positive integers (≥ 0). Given i , how many bytes necessary to encode it? I have to find the value $k \mid s \frac{c^{k-1}-1}{c-1} \leq i < s \frac{c^k-1}{c-1}$, cannot be encoded with $k-1$ bytes but can be encoded with k bytes. Found k we can conclude that i will be encoded with k bytes.

Example $s + c = 8$ so the combinations are $s, c = 4, s = 6, c = 2, s = 5, c = 3$. Let's use $(s, c) = (5, 3)$ and define 000, 001, 010, 011, 100 as stoppers and 101, 110, 111 as continuers.

If I use one block, it can only consist of a stopper, so the values 0, 1, 2, 3, 4, so 5 values in total.

With 2 blocks, one continuer and one stopper, so each continuer can be paired with each stopper: 3 continuers · 5 stoppers is 15 values in total. This way we encode the next integers, so the values 5, 6, ..., 19. This makes $5 + 15 = 20$ values encodable with 2 3-bit blocks.

Supposing we have to encode 100 with $(5, 3)$ dense code. We have to find $k \mid s \frac{c^{k-1}-1}{c-1} \leq 100 < s \frac{c^k-1}{c-1}$ is satisfied.

$$k = 2 \Leftrightarrow 5 \frac{3^{2-1}-1}{3-1} \leq 100 < 5 \frac{3^2-1}{3-1} \Leftrightarrow 5 \leq 100 < 20$$

$$k = 3 \Leftrightarrow 5 \frac{3^{3-1}-1}{3-1} \leq 100 < 5 \frac{3^3-1}{3-1} \Leftrightarrow 20 \leq 100 < 65$$

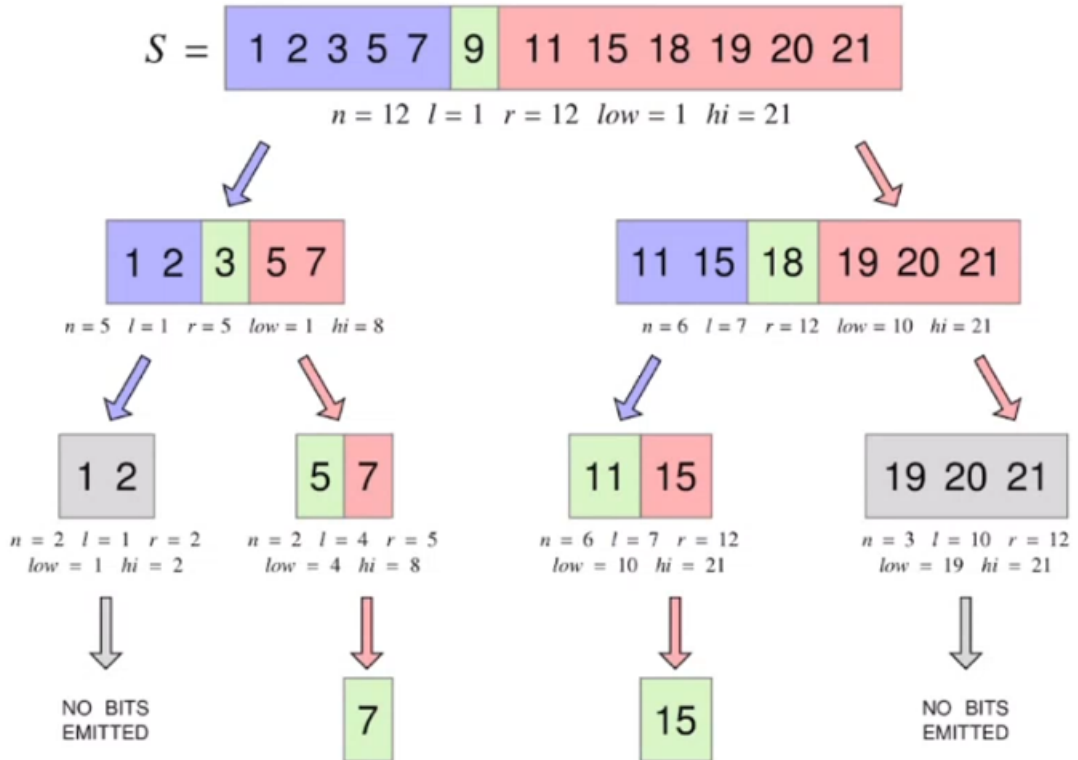
$$k = 4 \Leftrightarrow 5 \frac{3^{4-1}-1}{3-1} \leq 100 < 5 \frac{3^4-1}{3-1} \Leftrightarrow 65 \leq 100 < 200$$

I need $k = 4$ blocks, with 4 blocks we encode values in $[65, 200]$. 100 is the $100 - 65 = 35$ th value encoded with $k = 4$ blocks.

With two blocks we encode $5 + 15 = 20$ values, with three blocks we have another 45 values so 65 total values, with the fourth blocks we add 135 more values, for a total of 200 values.

Each codeword with exactly 4 blocks ($cccs$) encodes at most $3^3 \cdot 5 = 135$ values with $(5, 3)$ dense encoding.

Interpolative Code It requires that both the encoder and the decoder stays synchronized on what's going on.



The first encoded item is the one in the middle of the sequence S (9 in the example), then the interpolative encoding is recursively called on the first and second half of the sequence. Again we encode the middle item and so on.

The first important element is that the input sequence must be **strictly increasing**. If it's not, we can transform it by summing the item with the previous one. For example $S = 7, 1, 9, 2, 13, 7, 8$ become $S = 7$ (the first item as-is), 8 ($1 + 7$), 17 ($9 + 8$), 19 ($2 + 17$), 32 , 39 , 47 .

At the beginning, the encoder writes down the indexes of the first and last item l, r , and the values. In the example of the image, those values are $[1, 12, 1, 21]$. The decoder will read these values and know that there are 12 items, the first is 1 and the last is 21. Then the encoder writes down the middle element (index 6, value 9).

The decoder **knows that** between position 1 and 6 there are 6 increasing numbers, so the item in position 6 will have a value of at least 6. Also know that the last value is 21 and $21 - 6 = 15$. So the middle element has a value x such that $6 \leq x \leq 15$.

The encoder writes down the difference between the actual value of the middle element, 9, and the minimum possible value, 6, so writes down 3. It also knows that the difference is in the range $[0, 9]$ because the maximum value is 15, so the difference 3 is written out with 4 bits $\rightarrow 0011$.

The decoder reads 0011, so the difference is 3, so the actual value is 9.

Now the encoder goes to the left part, whose first index is $l = 1$ and last is $r = 5$. We just encoded 9, so the value in position 5 cannot be greater than 8. Now it writes the value in the middle position, 3, and the value it has to encode is 3.

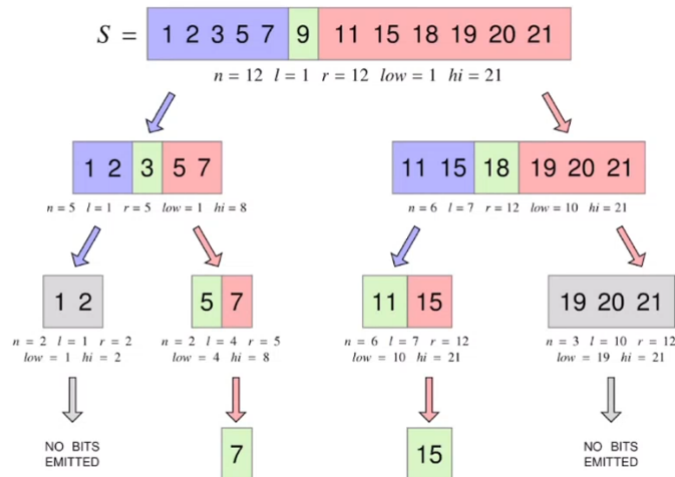
The decoder knows that the value in position 3 is ≥ 3 and reasoning from the right, the value is also ≤ 6 . Four possible values, so 2 bits to encode in that range. The encoder writes 00. Now it would proceed to the left part, but there are two elements from index 1 to 2 and the values are again from 1 and not greater than 2 because we just encoded 3, so it's information that the decoder already knows and it starts working on the right part.

It needs to encode 5, just encoded 3 and we're on the right part, so the value must be ≥ 4 . It also knows that the last value in this part is at most 8 because we encoded 9, so the middle element is between 4 and 8, so five elements so 3 bits where it's encoded 1, the difference between the actual value 5 and the minimum value 4. We proceed, with the 7. It's at least 6 (we just encoded 5) and at most 8, so two bits.

It proceeds like this.

Effectivness The bits represents values in ranges, so if the ranges are small we use fewer bits. So it's efficient when values are clustered, "dense".

Elias Fano codes Where we can access any value in the compressed sequence very efficiently. We have n increasing values in the range $[0, u)$, $b = \lceil \log_2 u \rceil$ bits for the representation and $l = \lceil \log_2(\frac{u}{n}) \rceil$, $h = b - l$. An example with $n = 8, u = 32, b = 5, l = 2, h = 3$



We work with two columns: the l least significant bits and the h most significant bits. The values in the l column are not compressed at all, so L is just the concatenation of those values. Because it has no compression, l is chosen also to make L small compared to the rest of the encoding.

The values in the h column are increasing, because the values of the sequence are increasing and the most significant bits remain increasing even when removing the least significant bits. To compress H the idea is the following: we know that $h = 3$, so $2^h = 2^3 = 8$ possible values and maybe some of them don't appear (for example: 2, 3, 5 do not appear in the example). We consider all 8 possible values and we record a unary representation of the number of items

in the h column with that value (in the example, only one 0 so 10 with as many ones as the occurrences ended by a 0 "separator"). The values are increasing, so the one 0 is the first, the two 1 are the second and third, zero 2, zero 3 and then one 4 so is the fifth.

Important: in H , $\#0 = 2^h$ (there are 2^h zeroes, separators) and $\#1 = n$ because we encode n values. In L , $\#bits = l \cdot n$. There's a one-to-one correspondence between the 1 in H and the groups of l bits in L : the first 1 in H corresponds, in the example, to the first two bits, and that 1 indicates a group of h bits with value 0, so 000 01.

Decompress We need a data structure supporting the operation $SELECT(H, i)$, which returns the position of the i th 1 in H . For example, $SELECT(H, 5)$ should return 11. Of course, $0 < i \leq n$. Assuming we want to find $S[5]$. Until now, with the other codes, we needed to decompress the whole sequence in order to get a single element. With this code we can access a single element from the compressed data. For the least significant bits we simply select the 5th group of $l = 2$ bits in L , which is 00 and we get this in $O(1)$. The 5th elements corresponds to the 5th 1 in H , and we get that position with the $SELECT$ operation stated before ($SELECT(H, 5) = 11$) but we need to know the bucket as well. To find the bucket it belongs, we simply count the number of 0 that are before that 1, and to count them knowing that it's the 11th position and the 5th 1, we simply do $11 - 5 = 6$. So we have that the 5th element is in the 6th buckets and $h = 3$, so the most significant bits are 110. Putting it all together, we get 110 00 = 24 which is correct.

Exercises

Do one level of recursion of interpolative code on $S = 11, 14, 16, 19, 20, 21, 22$

We begin with $l = 1, r = 7, min = 11, max = 22$, the middle element is the one in index $\lfloor \frac{7+1}{2} \rfloor = 4$ which is 19, so it must be $(min + m - l) = 11 + 4 - l = 14 \leq x \leq 19 = 22 - 7 + 4 = (max + m - r)$, and $19 - 14 = 5$ so it can be encoded with 3 bits: 19 will be encoded with 101.

The left part is 11, 12, 16 with $l = 1, r = 3, min = 11, max = S[4] - 1 = 18$, the middle element is $\lfloor \frac{3+1}{2} \rfloor = 2$ which is 14 and must be $12 \leq x \leq 17$, so we need 3 bits: 14 is encoded as 010.

The right part is 20, 21, 22 with $l = 5, r = 7, min = S[4] + 1 = 20, max = 22$, the middle is $\lfloor \frac{7+5}{2} \rfloor = 6$ which is 21, must be $21 \leq x \leq 21$, so 0 bits.

δ -code and Rice code with $k = 3$ on $S(1, 6, 15, 18, 21, 24, 30)$

$S' = (1, 5, 9, 3, 3, 3, 6)$, with $\delta(x) = \gamma(|bin(x)|) bin(x)$ and $\gamma(y) = 0 \dots 0bin(y)$ with $|bin(y) - 1|$ zeroes, so:

$$\begin{aligned}\delta(1) &= 11 \\ \delta(5) &= 011101 \\ \delta(9) &= 001001001 \\ \delta(3) &= 01011 \\ \delta(6) &= 011110\end{aligned}$$

As for Rice, with $k = 3$ and remembering $q = \lfloor \frac{x-1}{2^k} \rfloor$ and $r = (x - 1) - q2^k$ with $R_k(x) = unary(q + 1) bin(r)$ and $U(x) = 00 \dots 01$ with $x - 1$ zeroes, so:

$$\begin{aligned}R_3(1) &= 1000 \text{ with } q = \lfloor \frac{1}{2^3} \rfloor = 0 \text{ and } r = 1 - 1 - 0 \cdot 2^3 = 0 \\ R_3(5) &= 1100 \text{ with } q = 0 \text{ and } r = 4 \\ R_3(9) &= 01000 \text{ with } q = 1 \text{ and } r = 0 \\ R_3(3) &= 1010 \text{ with } q = 0 \text{ and } r = 2 \\ R_3(6) &= 1101 \text{ with } q = 0 \text{ and } r = 5\end{aligned}$$

Show this (1, 3) codeword of 8 on 2-bit words

$s = 1$ stoppers $\Rightarrow \{00\}$, $c = 3$ continuers $\Rightarrow \{01, 10, 11\}$. So:

Values	Codewords
0	00
1	01 00
2	10 00
3	11 00
4	01 01 00
5	01 10 00
6	01 11 00
7	10 01 00
8	10 10 00 \leftarrow

Encode $S = (1, 3, 4, 5, 9, 16, 23, 27, 28, 31, 40)$ **with Elias-Fano and show how to answer ACCESS(4), NEXTGEQ(8) and NEXTGEQ(32).**

We have $n = 11, u = 41, b = \lceil \log_2 u \rceil = 6$ and $l = \lceil \log_2(\frac{u}{n}) \rceil = 2$ so $h = b - l = 4$. The table is

Value	H	L
1	0000	01
3	0000	11
4	0001	00
5	0001	01
9	0010	01
16	0100	00
23	0101	11
27	0110	11
28	0111	00
31	0111	11
40	1010	00

$L = 01\ 11\ 00\ 01\ 01\ 00\ 11\ 11\ 00\ 11\ 00$

$H = 110\ 110\ 10\ 0\ 10\ 10\ 10\ 110\ 0\ 0\ 10\ 0\ 0\ 0\ 0\ 0$

As for the operations:

Access(4): 4th group in L for the least significant bits, 4th 1 in H (SELECT₁(4)) for the most significant bits. We get 01 from L and the 4th 1 corresponds to the 2nd bucket, which is 1 on $h = 4$ bits \Rightarrow 0001, so we get 000101 = 5

NextGeq(8) = NextGeq(0010 01) so we look for the starting position in H for the bucket 0010 = 2 ($\eta = \text{SELECT}_0(2) + 1 = 6 + 1 = 7$) $H[\eta] = 1$ so is a non empty bucket, we scan its elements from $\eta - H(8) = \eta - 2 = 5$ and return 9

NextGeq(32) = NextGeq(1000 00)

$\eta = \text{SELECT}_0(8) + 1 = 18 + 1 = 19$ and $H[19] = 0$ so bucket empty, we return the next element which is given by Access($\eta - H(32)$) = Access(19 - 8) = Access(11)

Encode $S = (11, 14, 16, 19, 20, 21, 22)$ **with Elias-Fano and answer Access(5) and NextGeq(17).** Reduce the Universe size by subtracting 11 (so 5 + 11 in the first and 17 - 11 in the second question)

0.6 String Sorting

Until now: atomic items. Now: sequence of symbols/characters/letters (**strings**), drawn from a alphabet Σ with $|\Sigma| = \sigma$ symbols.

We have a set of strings $S[1, n]$ with N the total length and n the number of strings. Also $L = \frac{N}{n}$ is the average length. Often $\sigma = 256$

Usually **qsort**, compared-based algorithm: need to specify the comparison function. So **qsort** is agnostic to the kind of objects it's sorting, we can use it to sort strings: it costs $O(L \cdot n \cdot \log n) = O(N \cdot \log n)$ time on average (worst case if using **mergesort**, **heapsort** or other very efficient sorting algorithms). Whenever the portion between l and r in **qsort** is small, with atomic items we would do insertion sort because the elements would fit in few cache lines, but with strings each item is a pointer so the items are probably spread apart. So we will try to improve this.

The lower bound is composed by two parts:

I have to sort the first characters of these n objects and I cannot escape the comparison of the first objects, so the classic lower bound $\Omega(n \cdot \log n)$

The first character only is too limiting, how many characters I have to surely look at? d_i is the **distinguishing prefix** and is the minimum number of characters that we have to consider in order to distinguish the string s_i from the others. We do not need the rest of the characters of s_i in order to distinguish it from the other strings. We can say that $d = \sum_s d_s$ is the total distinguishing prefix.

So the second part of the lower bound is $\Omega(d)$, I cannot look at less than d symbols.

So the **lower bound for comparison-based sorting algorithms for strings** is $\Omega(n \log n + d)$ time or character comparisons, which in asymptotic notation means the maximum between the two parts.

State of qsort In the case of n strings s_i , composed by l zeroes followed by $\log n$ other characters

$$\begin{array}{ccc}
 & l & \log n \\
 s_1 & 00 \dots 0 & 00 \dots 0 \\
 s_2 & 00 \dots 0 & 00 \dots 1 \\
 \vdots & \vdots & \vdots \\
 s_n & 00 \dots 0 & 11 \dots 1
 \end{array}$$

The **qsort** is $O(L \cdot n \cdot \log n)$, $L = l + \log n$ because they have all the same length. So **qsort** is $O(l \cdot n \cdot \log n + n \cdot \log^2 n)$. It is at least a $\log n$ factor from the lower bound, which can be very impactful due to the cache misses and the memory accesses.

0.6.1 Radix sort

Two families

Most significant digit sort (MSD), where we scan from the left to right

Least significant digit sort (LSD), where we scan from the right to left
Used in the punch cards

An example: $S = \{007, 017, 042, 911, 999\}$, $n = 5$, $\sigma = 10$, $N = 3 \cdot 5 = 15$. The idea of MSD is trivial: we scan the strings from left to right and position them according to the first character. So we have an array of size σ and we put each item in the corresponding bucket according to the first symbol. Then we consider the second digits for every bucket, and do the same thing.

$$\begin{array}{ccccccccc}
 0 & 1 & 2 & \dots & 9 & & & & \\
 \downarrow & & & & \downarrow & & & & \\
 007 & & & & 911 & & & & \\
 017 & & & & 999 & & & & \\
 042 & & & & & & & & \\
 \\
 0 & 1 & 2 & 3 & 4 & \dots & 0 & 1 & \dots & 9 \\
 \downarrow & \downarrow & & & \downarrow & & \downarrow & & & \downarrow \\
 007 & 017 & & & 042 & & 911 & & & 999
 \end{array}$$

We go down until we have one item per bucket in each level. If we had "019", it would've gone in the second bucket of the second level, so we would have done a third level.

For each element s_i we create d_i nodes (arrays), its distinguishing prefix, but both nodes are in general shared with other strings. So \forall string s we create d_s nodes. The total number of nodes (arrays) $\leq d = \sum_s d_s$ because some nodes are shared between strings. The arrays are of size σ , so in space terms we have $O(d \cdot \sigma)$. The sorting time is composed of: the time required to create the tree of arrays of $d \cdot \sigma$ nodes, so $O(d \cdot \sigma)$, then we have to scan the final set of strings, the last level, which is at most every possible node of the last level, another $O(d \cdot \sigma)$. So the cost is $O(d \cdot \sigma)$. This is not a comparison-based algorithm, so the lower bound stated before doesn't apply.

Trie A data structure. It's a tree, every node is an array of σ positions where each position have a pointer to other nodes (σ -ary tree). The **MSD algorithm builds a trie**.

Good and bad properties. The good is that the array is very sparsely populated, with big σ , so it's very efficient in time for searches, but it's inefficient in space ($d \cdot \sigma$).

Given a pattern $P[1, p]$ of p characters, we can check if P is in the trie or not in $O(p)$, optimal in time. With arrays it's very inefficient in space, though, because most of the arrays would be empty or almost empty especially in the leaves. We need another way of implementing an array, there are two possibilities: binary search tree or hashtable, let's see the second one.

Trie with hashtables We insert $\langle \text{digit}, \text{pointer} \rangle$ pairs. The idea is to create an hashtable for every node whose size is proportional to the number of distinct pairs ($\equiv \#$ edges, the pointers in the arrays). For example, an hashtable T of 3 positions where we insert the pairs $\langle 1, p_1 \rangle, \langle 2, p_2 \rangle, \langle \sigma, p_\sigma \rangle$ can be populated by applying the hash function to the

$$\begin{array}{lcl}
 T & & \\
 \text{keys and end up like} & 0 & \rightarrow \langle 2, p_2 \rangle \\
 & 1 & \\
 & 2 & \rightarrow \langle 1, p_1 \rangle, \langle \sigma, p_\sigma \rangle
 \end{array}$$

So the size of the table is proportional to the number of keys, $|T| = O(\# \text{ keys})$, and the cost of searching in an hashtable is $O(1)$ on average.

Trie with binary search tree In this case, the worst case time cost is $O(\log \sigma)$

In both cases we have a space complexity of $O(\# \text{ keys}) = O(\# \text{ edges})$

MSD radix sort with hashtable In every node we have and hashtable of size the number of edges but not sorted. The algorithm goes like this:

1. Builds the trie
Search and insertion of hashtables are both $O(1)$, so we pay 1 for every string for every level we need to go down on. So constant time \cdot the number of traversed nodes, which is d_s for the string $s \Rightarrow$ total time is $\sum_s O(1) \cdot d_s = O(d)$ on average time.
In a binary search tree, the $O(1)$ becomes $O(\log \sigma)$ so the total cost is $O(d \cdot \log \sigma)$
2. Sorting the hashtables to sort the edges. We can use **qsort** on the pairs, which costs $O(\# \text{ edges} \cdot \log_2 \# \text{ edges})$. Let's define l_u the number of edges in the node u . So the total cost of sorting is $\sum_u O(l_u \cdot \log l_u) = O(\sum_u l_u \log l_u) = O(\sum_u l_u \log \sigma)$ because the number of edges in node u is $\leq \sigma$ for every u . So we have $= O((\log \sigma) \cdot \sum_u l_u)$, the sum over all the nodes of the number of edges, which is $\leq d$. So the total cost of sorting is $O(d \cdot \log \sigma)$ in time and $O(d)$ in space.

LSD radix sort LSD scans the opposite way, from right to left, we sort every column using what's called a **stable-sort**: a sort that respects the original order, if two columns are equal. We used the counting sort, with a complexity of $O(\# \text{ keys} + \sigma)$ with $|\Sigma| =$ size of the alphabet.

An example assuming to have the following strings

017	111	007	007
042	042	111	017
665	first digit \rightarrow 665	second digit \rightarrow 017	third digit \rightarrow 042
111	017	042	111
007	007	665	665

By respecting the order thanks to the stable sort, we get a ordered sequence. The total complexity is the $\#$ digits per string times the cost of counting sort. So $O(L \cdot (n + \sigma)) = O(Ln + L\sigma) = O(N)$ time and space.

Proof of why it works Let's focus on two strings: x, y , with $x < y$. Let's say they have the same prefix α , a character a for x and b for y and then the rest of the string:

$$\begin{aligned} x &= \alpha & a & \dots \\ y &= \alpha & b & \dots \end{aligned}$$

$x < y \Rightarrow a < b$, for example x is "abaco" and y is "abate": $\alpha = \text{"aba"}$, $a = \text{c}$, $b = \text{t}$.

$$\begin{aligned} x &= \text{aba} & \text{c} & \text{o} \\ y &= \text{aba} & \text{t} & \text{e} \end{aligned}$$

In the first phases, we compare the right part of both strings and x could end up before or after y , it's not important. At some point, it will compare a, b and since $a < b$ we will have a situation where x comes before y . Then we compare α , but they are equal and thanks to the stable sort the relative order between x and y will not change (possibly changing distance, but they will not "invert" order). This applies to every pair of strings, so the final order of strings will be a sorted order.

More precise complexity Let's assume binary strings, with 0, 1 digits and of b length. We can apply the LSD radix sort to groups of r bits each: $\frac{b}{r}$ groups. Each group of r bits can be interpreted as a symbol, so $\sigma = 2^r$ maximum configurations of r bits (from all zeroes to all ones)
 $\frac{b}{r}$ phases each costing $n + 2^r$, so the total cost is $O(\frac{b}{r}(n + 2^r))$: the minimum is in $r = \log n$ and so the complexity becomes $(\frac{b}{\log n}) \cdot (n + n) = O\left(\frac{n \cdot b}{\log n}\right) = O\left(\frac{N}{\log n}\right)$ time.

0.6.2 Multi-key Quicksort

The invariant is that there is a prefix $[0, i - 1]$ shared by the strings in R , i is the character we compare and at the beginning $i = 1$

```

1 def MKQ(R, i):
2     if (|R| <= 1):
3         return R
4     else:
5         choose a pivot string p in R
6         Rless = {s in R | s[i] < p[i]}
7         Req = {s in R | s[i] = p[i]}
8         Rgreat = {s in R | s[i] > p[i]}
9         A = MKQ(Rless, i)
10        B = MKQ(Req, i+1)
11        C = MKQ(Rgreat, i)
12        return A + B + C

```

We have the strings, and we know that all of them share $i-1$ characters. The i th character of the pivot strings realizes the three-way partitioning of R : if the i th character of the string is $<$ than the i th character of the pivot, then that string goes into $R_{<}$, if it's $=$ goes into $R_{=}$ and if it's greater goes into $R_{>}$.

So returning the three partitions in order gives us the sorted order of the original R . Some considerations:

$|R_{<}|, |R_{>}| < |R|$ because p is not in them, so their size is surely shrinking

$|R_{=}| \leq |R|$, could be equal whenever the i th character of all the strings in R is equal to the i th character of the pivot string p . In any case, i is increased by one (also $i \leq$ maximum string length)

So **the recursion is well-posed**.

Let's reason about complexity, taking $s \in S$ the original set, with $R = S$ at the beginning. For a generic step we have R, i and there are two cases that can occur to s :

$s[i] < p[i] \rightarrow s \in R_{<}$ or $s[i] > p[i] \rightarrow s \in R_{>}$ and in both cases $|R_{<}| < |R| \wedge |R_{>}| < |R|$

$s[i] = p[i] \rightarrow s \in R_{=}$ and $i++$

How many times we do $i++$ on the string s ? $i \leq d_s$ because d_s is the maximum number of symbols shared with another string plus 1 (the distinguishing prefix), so the prefix of s $s[1, d_s]$ is distinct from every other string. So on average, picking a random pivot, we have that $|R_{<}| \simeq \frac{|R|}{2}$ and $|R_{>}| \simeq \frac{|R|}{2}$. This means that we have a logarithm of recursive steps, and since we're starting from S we have $O(\log |S|)$ recursive steps on average.

For the string s , the numbers of recursive calls is $O(\log n + d_s)$, and $\sum_s O(\log n + d_s) = O(n \log n + d)$ average time, so it's optimal on average.

0.6.3 Ternary Search Tree

It's a **ternary trie**.

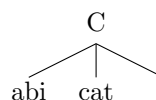
Each node has a character and may have three outgoing edges: one is $<$, one is $=$ and one is $>$. If the node corresponds to the i th position and has character "c", then all the nodes in the $=$ subtree have "c" in the i th position, all the nodes in the $<$ subtree have characters $<$ "c" in the i th position and in the $>$ subtree the i th characters are $>$ "c". In the $<$ and $>$ subtree you still remain in the i th character, so the root of those subtrees still correspond to the i th character, while in the $=$ subtree you advance so that root corresponds to the $i+1$ th character (much like the MKQ code before).

But we don't want to explode the entire string. Let's do an example with $S = \{\text{cat}, \text{abi}, \text{cast}, \text{cam}, \text{at}\}$:

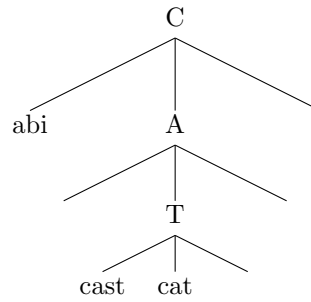
We insert "cat", in an empty tree, so simply



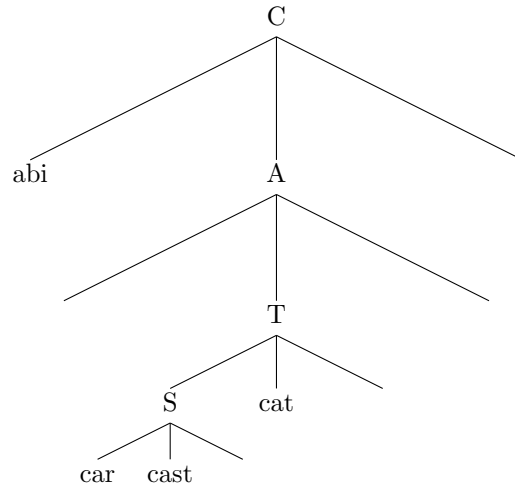
"abi" in the $<$ subtree



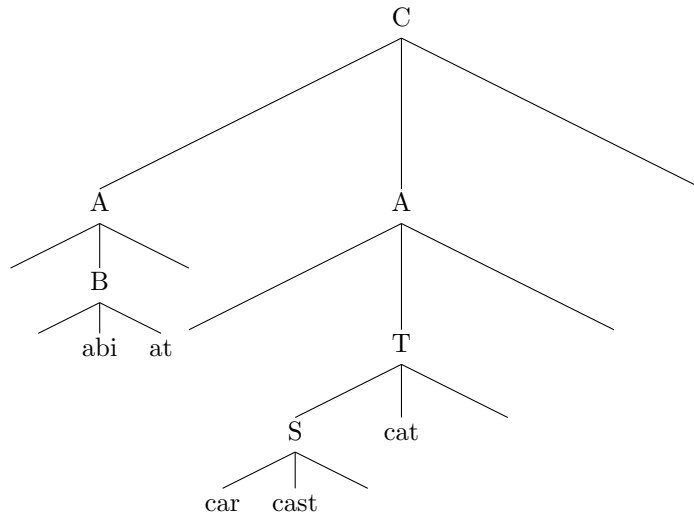
"cast", but we have "cat" so we have to distinguish the two string. The standard way is to take the character of the string that was already in the tree, so we distinguish them based on the "T" in "cat"



"car", again a *collision* with "cast" (remember that the $<$ and $>$ subtree match still the i th character, and only the $=$ subtree match the $i + 1$ th character)



"at", *collision* on the $<$ subtree of "C"



The TST could be unbalanced, but if strings are inserted randomly then we have that the string s gets a path of length $O(|s| + \log n)$ on average. Also every node needs only an array of three positions and one character, very compact in space.

0.7 Compressed Data Structures

Able to store the data in compact form, often significantly reducing the size of the data. The questions are about what we can do for the query operations: asymptotically identical to the uncompressed data structures or we lose a little bit. Very useful structures in edge computing and smartphone, where there's not much space available and the energy consumption must be under control.

Assume a binary array $B = 0100110\dots$ of size $|B| = m$ and with $n = \#$ of ones. This could be just a binary array or, for example, the characteristic vector of a set: we have a set of integers from 1 to m and we have just the integer 2, the integer 5, the 6 and so on. If we want to store the position of the ones, the typical solution is to encode the position. With **standard encoding** we have a space complexity of $O(n \cdot \log_2 m)$ bits (pointer-based programming). We will propose a pointer-less programming.

Two operations, fundamental basic blocks for modern applications:

$\text{Rank}_b(i)$ = number of b in the prefix $B[1, i]$, with $b \in \{0, 1\}$

E.g.: $\text{Rank}_1(4) = 1$

Property: $\text{Rank}_0(i) = i - \text{Rank}_1(i)$

$\text{Select}_b(i)$ = position of the i -th b in B , with $b \in \{0, 1\}$

E.g.: $\text{Select}_1(3) = 6$

These two primitives can be used to implement all the common queries on data structures. If a data structure is implemented with Rank and Select as primitives, then it can use all the optimization that exists of those two data structures.

Theorem \exists data structures that takes $o(m)$ bits **in addition** to B (read-only) and supports Rank_1 in $O(1)$ time. $o(m)$ means sublinear but in the universe, so its performance scales with the size of the universe.

With B array, we partition B into blocks of size Z , so we end up with $O(\frac{m}{Z})$ blocks. For each block, I store Rank_1 of all previous blocks (absolute, from the beginning). The space needed for the blocks to store the counting is the number of the blocks times the counter, an integer which counts up to n , so $\log_2 n$, but we want in terms of m and $n \leq m$ so $\log_2 m$ is surely enough bits. So the space occupied by the big blocks is $O(\frac{m}{Z} \cdot \log_2 m)$ bits.

The size of the big blocks, Z , impacts on the space: we would like a larger Z to reduce space but if it becomes larger the time we need to compute Rank_1 also increases. We make another partition of smaller blocks to increase the granularity. We consider small blocks inside the big blocks, of size z . So every big block is partitioned in $\frac{Z}{z}$ small blocks, and for each of them we store the relative rank from the beginning of the enclosing big block. The space needed by small blocks is $O(\frac{m}{z} \cdot \log_2 Z)$ bits.

So the total space is $O(\frac{m}{Z} \cdot \log_2 m + \frac{m}{z} \cdot \log_2 Z)$ bits. The time complexity is $O(z)$

Four-russians tricks, because we work in a RAM model with word size of $O(\log m)$ bits. The idea is as follows: in the small blocks of z size, we start from the beginning of that block and scan counting the ones. For every small block configuration X and for every position $Y \in [1, z]$, we store the number of ones in $X[1, Y]$: basically, for the sequence X currently in z we store the number of ones up to position Y . In some sense, we create a table of all possible situations (configurations, X s), and all possible answers. Time is constant, how big is this table? We have 2^z rows and z columns, with every entry requiring $\log_2 z$ bits.

$O(\frac{m}{Z} \log_2 m + \frac{m}{z} \log_2 Z + 2^z \cdot z \cdot \log_2 z)$. How to solve the query $\text{Rank}_1(i)$: we take the counter stored at the big block including i , the counter stored at the small block including i and let X be the small blocks including i and $Y = i \bmod z$, we get $T[X, Y]$ (the entry in the table), all constant time. The space is the problem, if $Z = (\log_2 m)^2$ and $z = \frac{1}{2} \log_2 m$ (less than the word) let's use in the formula

$$\begin{aligned}
& O\left(\frac{m}{Z} \log_2 m + \frac{m}{z} \log_2 Z + 2^z \cdot z \cdot \log_2 z\right) \\
& O\left(\frac{m}{(\log_2 m)^2} \log_2 m + \frac{m}{\frac{1}{2} \log_2 m} \log_2 ((\log_2 m)^2) + 2^{\frac{1}{2} \log_2 m} \cdot \left(\frac{1}{2} \log_2 m\right) \cdot \log_2 \left(\frac{1}{2} \log_2 m\right)\right) \\
& \quad \vdots \\
& \frac{m}{\log m} + \frac{m}{\log m} \cdot \log \log m + \sqrt{m} \cdot \log m \cdot \log \log m
\end{aligned}$$

Which is a sum of $o(m)$, so it's $o(m)$.

Theorem \exists data structures using $o(m)$ bits in addition to B which actually support Select_1 and Select_0 in constant time.

This two theorems do not depend on n , only on the universe size m . A solution is that if we take Elias-Fano combined with Select_1 and Select_0 , we get a data structure which occupies uses $2n + n \lceil \log \frac{m}{n} \rceil + o(n)$ bits, so we can support Rank_1 in $O(\log \frac{m}{n})$ time and Select_1 in $O(1)$ time.

So how I can construct a data structure which space depends on the number of keys rather than the universe of the

keys? The idea is: let's take the array $B = 01001011\dots$, and Elias-Fano can represent integers so we build S as the set of positions of the ones (starting from position 0 for Elias-Fano), so $S = \{1, 4, 6, 7, \dots\}$ and $|S| = n$. We know that the universe $u = s_m + 1 \leq m$ because the maximum values is $m - 1$, and $m - 1 + 1 = m$.

If I apply Elias-Fano, $l = \lceil \log_2 \frac{u}{n} \rceil \leq \lceil \log_2 \frac{m}{n} \rceil$ and $h = b - l$, L array of blocks of l bits and H array of $2n$ bits. The $\text{Access}(i)$ uses Select_1 in H and $\text{Geq}(x)$ uses Select_0 and something else.

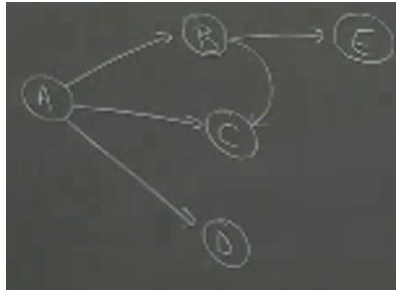
In B , $\text{Select}_1(i)$ is the i -th 1, which is $\text{Access}(i)$ in S so a $\text{Select}_i(i)$ in H , so it requires the space needed for the H so an increase of $+o(2n)$ bits. So $|H| + |L| + o(2n)$ (the constant 2 can be dropped so just $o(n)$ is fine).

$\text{Rank}_1(x)$ which can be implemented as a binary search for x in S , and with Access we can do it. So it costs $O(\log n)$ time using binary search + $\text{Access}()$. There's another way, NextGeq . So $\text{Rank}_1(x)$ can be implemented with $\text{NextGeq}(x)$ which finds $y \in S \mid y \geq x$ and returns the position, which is the Rank. To implement NextGeq , as seen before, we need Select_0 , so space of $|H| + |L| + o(2n)$ for Select on H . We need to go on L too, and scan the bucket but scanning is too slow. The values are increasing so we can do binary search, if the position is inside the bucket. If it's larger, we take the next bucket and same as before. So is $\log \text{bucket size} = \log 2^l = l = \log \frac{m}{n}$, more tricky but better.

Example With pointer-less programming, what we've seen up until now can be applied in several situations and not only integers. The idea is to have an array of strings S and to manage them we take an array P of pointer: the first pointer points to the first string in S , the second to the second string and so on. But P can store just the offset of the starting index of each string, so if the first string starts in index 0 and the second in index 4 and so on, P can be $\{0, 4, \dots\}$. So starting from the problem of storing strings we returned to the problem of storing increasing integers, with the solutions already seen.

With P of pointers we pay $n \cdot w$ with w being the word size (32 or 64 bits). With P of offsets, instead, the cost is $2n + n \lceil \log \frac{|S|}{n} \rceil + o(n)$ but without the disk access due to the pointers.

Free Encoding and Traversal (Binary Trees)



A graph have nodes, labels and edges. How to store it? We know to store it as an adjacency list or matrix, typically list for space reasons. An adjacency list is an array of nodes and a list of adjacent nodes for each node. The nodes in the lists are sorted according to some rule. In the example:

```

A  →  B, C, D
B  →  C, E
C  →
D  →
E  →

```

We can't use strings, due to variable length and a lot of pointers. We can assign each node an id, for example A is 1, B is 2 and so on based on the order of appearance in the graph. The list will become

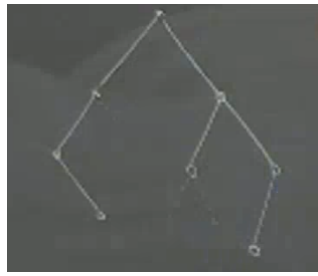
```

1  A  →  2, 3, 4
2  B  →  3, 5
3  C  →
4  D  →
5  E  →

```

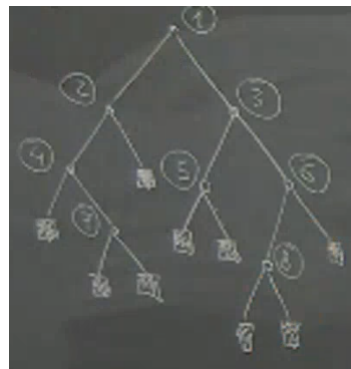
Same trick: taking the problem and turning it into the problem of storing sequences of integers, so typically apply Elias-Fano to the adjacency lists. But I could use something more compressed but less powerful, since every time I access the adjacency list I do it from left to right, without the need of NextGeq and other operations. So I could use P for delta or other codes.

Let's assume we have this tree (with left-right childs that might be null)



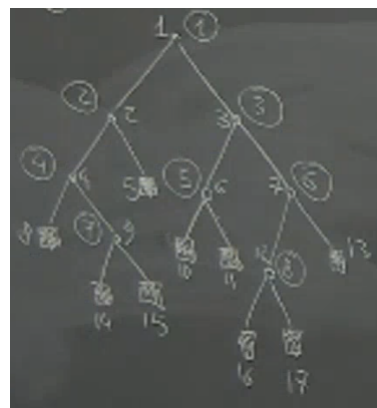
With $n = 8$ nodes. We could implement it with pointers, every node with a left and right pointer and possibly another pointer for the parent, and with 8 bytes per pointer that's 24 bytes per node just for the structure. We can do better in an elegant and brilliant way, consisting of several steps.

1. **Tree completion:** transforming the tree into a full binary tree



The circled numbers are relative to the original tree

2. **Node labeling** for the full binary tree



3. **Serialization of the node labels** of the full binary tree

B of 17 positions, I scan the tree and every time I find an original node I record 1 and every time I find a dummy node I record 0.

Positions	1	2	3	4	5	6	7	8	9	10	11	12	13	14	15	16	17
$B =$	1	1	1	1	0	1	1	0	1	0	0	1	0	0	0	0	0

Compress and index B using Rank/Select data structures, using a space of $|B| + o(|B|)$ for $\text{Rank}_1 + o(|B|)$ for Select_1

B is 17 bits, but where does this number comes from? It's $2n + 1$, because we have n ones (original nodes) and $n + 1$ dummy nodes. So the space is $2n + 1 + o(n)$, the lower bound is the **catalan number**.

Catalan Number The number of binary trees of n nodes $= 2^{2n}$, then the number of bits surely needed to represent a tree is $\log_2 2^{2n} = 2n$, so you can't do better than $2n$ bits in space. So we are far from the lower bound by $1 + o(n)$ bits.

$o(n)$ can be interpreted as $o(1)$ bits per node, so I'm far from the lower bound of less than 1 bit per node. This solution is called **succint**, because it's not compressed but we get the minimum possible plus something that is vanishing.

Let's see how, in the original tree,

to take the right child from node x

to take the left child from node x

to take the parent from node x

If we consider $x \rightarrow 2x + 1$ it works on the circled nodes to the non-circled nodes (children of 1 are 2 and 3, children of 3 are 6 and 7...). So the right child is $2x + 1$ and the left child is $2x$, both uncircled but I want the circled nodes so from each dummy node I have to find the corresponding original one or a "not existing in the original tree" value. If I take the number 9, for example, I can find it in B , which numerations corresponds to the dummy nodes. The corresponding original node is the Rank_1 , if the value is 1

Positions	1	2	3	4	5	6	7	8	9	10	11	12	13	14	15	16	17
$B =$	1	1	1	1	0	1	1	0	1	0	0	1	0	0	0	0	0
Rank_1	1	2	3	4		5	6		7				8				
									↑								

So I just do $\text{Rank}_1(9)$. So the left child is $\text{Rank}_1(2x)$ if $B[2x] = 1$ and the right child is $\text{Rank}_1(2x + 1)$ if $B[2x + 1] = 1$. The parent comes from inverting the child operations, and it's $\lfloor \frac{\text{Select}_1(x)}{2} \rfloor$. On the uncircled nodes, the parent is the half, but we are reasoning for uncircled values. So first we turn x into an uncircled value, the opposite of before, then we divide by two and find the parent (circled). Of course not in the root, so $x \neq 1$. Let's see an algorithm that, given a path Π and a tree described by its parameters "...", answers if the requested path exists in the given tree starting from the root or not.

```
def match(path, ...):
    x = 1
    if (L[1] != path[1]):
        return false
    for i in range(1, len(path)-1):
        l = Rank1(2x)
        r = Rank1(2x+1)
        if (B[2x]=1 and L[l]=path[i+1]):
            x = l
        else if (B[2x+1]=1 and L[r]=path[i+1]):
            x = r
        else:
            return false
    return true
```

With L an array of labels such that the original node with id i has label $L[i]$.

0.8 Data Compression

Families of compressors

Statistical Compressors

Lempel-Ziv parsings

Burrows-Wheeler Transform, which we will not see

0.8.1 Statistical Compressors

Most important representatives of statistical compressors, based on frequencies, are Huffman and arithmetic.

Notation

Σ , the alphabet of symbols that we will encode.

Can be binary, formed by bytes, words. What changes is the size and the number of possible values of the symbols: 2 for bits, 256 for bytes, thousands/millions for words. It could be an alphabet of integers, so the size is ∞

By changing the alphabet we change how we manage the data. The text **abaabbaab** can be interpreted with an alphabet of bits, of bytes, of letters, of binomials... in the case of letters we have 5 "a"s and 4 "b"s, but in the case of binomials we have 1 "ab", 2 "aa", 1 "bb" and 1 "b#". Changing the size of the alphabet can help introducing the "context" around the symbol. With single letters you can permute them in the text and still get the same frequency, but with couple of letters getting the same frequency by randomly shuffling the letters it's harder. The downside is the increasing in size of the alphabet, which we have to take into account when we decompress.

In fact, by compressing T we obtain $C(T)$ composed of:

Preamble, which contains infos about Σ , frequency...

Depends on the family of compressors and also the programmer.

Body, a sequence of bits

The frequencies can be in a:

Static model: the frequencies are given, you know the frequencies of each symbol

Semi-static model: the frequencies are computed from the data to compress (like the example before)

You assume that the entire text behaves the same way, but can't take into account variations (for example, first part of the text in english and second part in italian)

Dynamic model: consider a uniform starting model and encode while updating the model

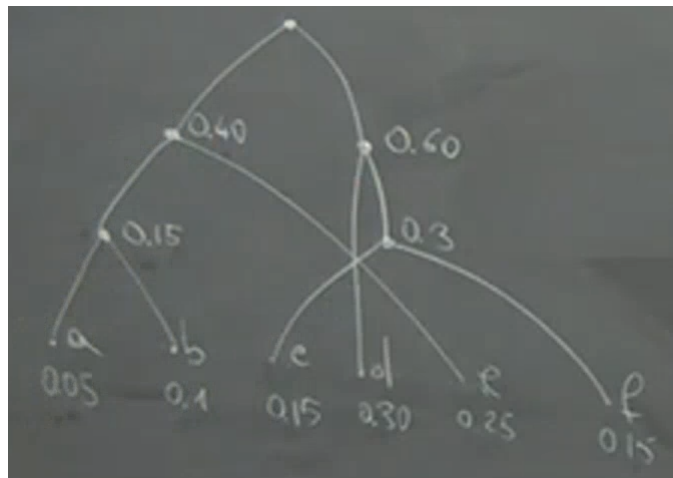
Huffman algorithm Greedy but also optimal. It repeats the following step: pick the two lightest symbols (less probability) and merge them in one new symbol whose probability is the sum of the probabilities of the merged symbols.

Example $\Sigma = \{a, b, c, d, e, f\}$ with corresponding frequencies of $\{0.05, 0.1, 0.15, 0.3, 0.25, 0.15\}$, so a **static-model**.

So the algorithm starts with

a	b	c	d	e	f
0.05	0.1	0.15	0.3	0.25	0.15

 and chooses to merge the two with the lowest probability, a and b obtaining a new symbol with probability 0.15. Now the choice is between c, f and the new symbol and can pick only two: whatever the choice, the algorithm is optimal, so we can pick the oldest symbols so we merge c and f into a symbol of probability 0.30. Next choice is the ab symbol and e , with a new symbol of 0.4 probability. Next is d and cf , with a probability of 0.6. We are left with just two symbols, 1.0 of probability.



The result is a **binary tree**, which is a way to generate codewords, so codes for symbols. We assign 0 to the left subtree and 1 to the right (or viceversa, the order is not important, and can also change the assignment for each subtree). With this we can compute the codewords, from the root to the symbol:

a is 000

b is 001

c is 110

d is 10

e is 01

f is 111

Important things:

It's a **prefix-free code**: considering two codewords, no one is a prefix of the other

Golden Rule of Data Compression: the most frequent symbols get the shortest codewords

This because it starts merging from the lightest symbols, which will end up on the bottom of the tree.

To measure the "goodness" of a code we need the **average codeword length** $L = \sum_{\sigma \in \Sigma} f(\sigma) \cdot l(c_{\sigma})$ with $\sigma \in \Sigma$ symbols, $f(\sigma)$ the frequency of the symbol and $l(c)$ the length of its codeword. In our example, $L = 3 \cdot 0.05 + 3 \cdot 0.1 + 3 \cdot 0.15 + 2 \cdot 0.3 + 2 \cdot 0.25 + 3 \cdot 0.15 = 2.45$ bits.

Whats the relation between L average codeword length and h height of the tree? L means that on average every symbol will emit L bits. The height of the tree is equal to the codeword length, a 1 : 1 relation between length of the codeword and depth of the symbol so that is also called trie of codeword. This means that L can be interpreted, in terms of the height, as the average path length \equiv average depth of the tree.

For this algorithm, the preamble would contain the tree and Σ . So for the example **aaabb**, the body would be 000000000001001. With **aaabb** I use 5 bytes, so 40 bits. Now I use 15 bits for the body plus the size of the preamble. So a good comparison is fundamental.

Theorem Huffman code is optimal among the prefix-free codes for Σ . In other words $L_H \leq L_C \forall C$ prefix-free code.

Lemma 1 Let F be the set of binary trees corresponding to optimal prefix-free codes, thus binary trees with minimum average depth. Assuming they have $|\Sigma|$ leaves.

\exists tree $\in F$ | the minimum weighted leaves are at the largest depth.

Lemma 2 Considering a tree corresponding to a code C , with x, y leaves among the $|\Sigma|$ leaves, and considering its **reduced tree** R_C with the two leaves x, y merged into a leaf with probability $P(x) + P(y)$ (so that R_C has $|\Sigma| - 1$ leaves). We have $L_C = \sum_{\sigma \neq x, y} l(c_{\sigma}) \cdot f(\sigma) + l(x)P(x) + l(y)P(y)$ and $L_{R_C} = \sum_{\sigma \neq x, y} l(c_{\sigma}) \cdot f(\sigma) + l(\langle x, y \rangle)(P(x) + P(y))$. Given that $l(x) = l(y) = l + 1$ then $l(\langle x, y \rangle) = l$ so that $L_C = L_{R_C} + P(x) + P(y)$

Proof The base case in on $|\Sigma| = 2$, obviously optimal. By induction, from $|\Sigma| - 1 \rightarrow |\Sigma|$ symbols, assume that Huffman constructs the optimal code for $|\Sigma| - 1$ symbols.

Let's consider L_H , for lemma 2 we have $L_H = L_{R_H} + P(x) + P(y)$, with x, y being the symbols of the smallest frequency merged by Huffman. Let's pick an optimal code, O . From lemma 1, there's a tree whose smallest probability leaves are at the deepest level. So in O we know that x, y are at the bottom, and we can apply lemma 2 obtaining R_O with the leaves x, y merged.

H and O would have $|\Sigma|$ leaves, and R_H, R_O would have $|\Sigma| - 1$ leaves. For inductive hypothesis, we know that Huffman of $|\Sigma| - 1$ leaves, so R_H , is optimal. So the average depth of R_H , which is L_{R_H} , is \leq than the average depth or R_O . So we have $L_{R_H} \leq L_{R_O}$

So we have $L_H = L_{R_H} + P(x) + P(y) \leq L_{R_O} + P(x) + P(y)$ and for lemma 2 we have $= L_O$, concluding that $L_H = L_O$, proving that Huffman is optimal.

Entropy The entropy $H = \sum_{\sigma} P(\sigma) \cdot \log_2 \frac{1}{P(\sigma)}$. In Huffman, $L_H = \sum_{\sigma} P(\sigma) \cdot l(c_{\sigma})$.

Very similar formulas, differing just by one factor: $\log_2 \frac{1}{P(\sigma)} \in \mathbb{R}$ and $l(c_{\sigma}) \in \mathbb{N}$, so in general not equal.

If the probabilities are diatic, in the form of 2^{-k} , the two formulas could correspond, but it's not the general case, but surely we have that

$$H \leq L_H < H + 1$$

So Huffman loses up to 1 bit per symbol: is this bad or good? This formula is per symbol, let's see it for the entire sequence (multiply by n)

$$nH \leq \text{Huffman length} < nH + n$$

So we lose n bits over the overall file. Good or bad depends on nH : if very small entropy, so H close to zero meaning the text is highly repetitive, then $+n$ is very large, but if the text is not repetitive, then n is possibly smaller than nH and so it's ok. So the loss depends on the entropy. Arithmetic is able to drop that $+1$.

This comes from the **Shannon bound** ($H \leq L_C$) and it's for all prefix-free code C . The property of Huffman is the right part.

Canonical Huffman We want to create an Huffman code with some properties: more succinct in space (not regarding the codewords, which are already optimal, but regarding what we are storing in the preamble) and with an increased speed in the compression.

The main idea is to turn the Huffman tree into an unbalanced tree with the deepest codewords in the left part, and the more to the right the less deep it is. A heap-like structure: more to the left, less to the right.

1. Build an Huffman tree
2. Compute codeword lengths
3. Compute the number of symbols per level/length (**num** array)
So for each level, the number of codewords per level
4. Compute the symbol table (**symb**)
For every level, keep a list of symbols per each level
5. Computing the new codewords.

It's sufficient to find the first codeword for each level, because all the codewords in that level are increasing $l = \text{max level}$, the first codeword of level max $fc[l_{\text{max}}] = 0$ and $l < l_{\text{max}}$ such that $fc[l] = \frac{fc[l+1] + \text{num}[l+1]}{2}$. Mathematically, by summing the first plus the num array element, I'm going to the next available leaf not occupied by any symbol on that level, and dividing by two in binary is just making a shift so we're dropping the leaf and going up one level. So it's the first codeword of the level $+1$ plus the number of leaves divided by two to go to the parent.

With fc and the length which is l , we represent the value on l bits and get the codeword.

An example For the codeword lengths

a	4
b	4
c	3
d	2
e	2
f	2

					0	1	2	...
num =	pos	1	2	3	4			
	val	0	3	1	2			
						symb =		
						2	d, e, f	
						3	c	
						4	a, b	

fc	1	2	3	4
	$\frac{fc[2] + \text{num}[2]}{2} = \frac{1+3}{2} = 2$	$\frac{fc[3] + \text{num}[3]}{2} = \frac{1+1}{2} = 1$	$\frac{fc[4] + \text{num}[4]}{2} = \frac{0+2}{2} = 1$	0
codeword	dummy value for the algorithm	01	001	0000

The first codeword at the deepest level is 0, and the levels without codewords are still computed due to the functioning of the algorithm. The values are used as integers, so the $fc[1] = 2$ cannot be represented on $l = 1$ bit, but it's not a problem because it's a dummy value and the fact that it's outside the tree is fundamental for the algorithm.



So we don't store the tree, we store the first codeword and the number of codewords in each level. To reconstruct the tree, we assign the first symbol to the first codeword and incrementally the rest of the symbols that are left for each codeword on that level, then move up one level to the first codeword of that level and assign the rest of the symbols and so on. So we store *fc* and *symb*, *num* is a temporary array just for computing.

For **encoding**, given $T = bcf$ we see that b is position/rank 1 of level 4, we take $fc[4] + rank[b] = 0 + 1 = 1$ on 4 bits so 0001. Same for $c \Rightarrow fc[3] + rank[c] = 1 + 0 = 1 \Rightarrow 001$

The space needed is the space for *fc* plus the space for *symb* plus the space for the array $symb \rightarrow level$. *fc* occupies l_{max} the number of bits to represent the codewords which is at most l_{max} , the table *symb* occupies $|\Sigma| \cdot \log |\Sigma|$ because it is a table of lists containing each symbol that cost $\log |\Sigma|$ to store, the array has to store a level (on $\log_2 l_{max}$ bits) for each symbol, which are $|\Sigma|$, so occupies $|\Sigma| \cdot \log_2 l_{max}$. So the total space is

$$l_{max} \cdot l_{max} + |\Sigma| \cdot \log |\Sigma| + |\Sigma| \cdot \log_2 l_{max}$$

The tree has $|\Sigma|$ leaves and $|\Sigma| - 1$ internal nodes, so it occupies $2|\Sigma| - 1$ pointers, which is a lot more bits.

For **decoding**, we can only use *fc* and *symb* (the $symb \rightarrow level$ array is not needed for decoding). To decode an array v of l bits, we must check if the node coded by the l bits it's an internal node, so we need to read one more bit and repeat, or if the node is a leaf. We check if the value $v < fc[l]$: if it's less than the first key, it's "on the left" therefore it's an internal node, otherwise it's a leaf therefore we decode it. The dummy node of a level m without leaves it's always greater than any possible value v of the m bits, so we will always "go down". The code:

```

1 v = nextBit();
2 l = 1;
3 while (v < fc[l]):
4     v = 2*v + nextBit();
5     l++;
6 return symb[l][v-fc[l]]

```

0.8.2 Arithmetic Coding

Tools for the arithmetic Coding $x \in [0, 1)$ positional representation for x : $.b_1b_2 \dots b_k \dots$ with b_i representing 2^{-i} , with $b_i \in [0, 1]$. These are called **dyadic fractions**, because are of the form $\frac{v}{2^n}$

For example $.101 = \frac{1}{2} + \frac{1}{8} = \frac{5}{8}$

To represent, you add zeroes to the binary representation of v until you get k bits: $.0 \dots 0 bin(v)$ such that the length is k bits. For example, $\frac{5}{16} = \frac{5}{2^4}$ so 5 in binary over 4 bits and the dot $\Rightarrow .0101 = \frac{1}{4} + \frac{1}{16} = \frac{4+1}{16} = \frac{5}{16}$

We need another tool, which we can use to represent a fraction as a sequence of that type: an algorithm **converter(x, k)** with k being the precision/accuracy, how many bits we want to use and $x \in [0, 1)$. The algorithm works as follows: if $2x < 1$ we output 0 and repeat on $2x$, else if $2x \geq 1$ we output 1 and repeat on $2x - 1$.

An example $x = \frac{1}{3}$

$$2x = \frac{2}{3} < 1 \rightarrow 0$$

$$2x = \frac{4}{3} > 1 \rightarrow 1 \text{ and } \frac{4}{3} - 1 = \frac{1}{3}$$

$$2x = \frac{2}{3} \rightarrow 0$$

$$\rightarrow .0\bar{1}$$

$$x = \frac{5}{8}$$

$$2x = \frac{10}{8} > 1 \rightarrow 1 \text{ and } \frac{10}{8} - 1 = \frac{1}{4}$$

$$2x = \frac{1}{2} < 1 \rightarrow 0$$

$$2x = \frac{2}{2} = 1 \rightarrow 1 \text{ and } \frac{2}{2} - 1 = 0$$

When we have 0 we stay with 0 forever so $\rightarrow .101\bar{0} = .101$

When we stop after k bits we have to compute the error. **Lemma:** given $x = .b_1b_2 \dots b_db_{d+1} \dots$, we compute $\bar{x} = .b_1b_2 \dots b_d0 \dots 0$ which is x truncated at the first d digits. So the difference is the value of the bits from b_{d+1} onward, so

$$x - \bar{x} = \sum_{i=1}^{\infty} b_{d+i} \cdot 2^{-(d+i)} = 2^{-d} \cdot \sum_{i=1}^{\infty} b_{d+i} \cdot 2^{-i} \leq 2^{-d} \cdot \sum_{i=1}^{\infty} 1 \cdot 2^{-i} = 2^{-d}$$

Because the worst case is when every b_i with $i \geq d + 1$ is 1 and $t \sum_{n=1}^{\infty} \frac{1}{2^n} = 1$. This means that `converter(x, k)` makes an error of at most 2^{-k}

Arithmetic coding Let's assume we have an alphabet of 3 symbols with $P(a) = \frac{1}{2}, P(b) = P(c) = \frac{1}{4}$. We assume to map the symbols in the $[0, 1)$ interval alphabetically and according to the probability. So $f_a = 0, f_b = \frac{1}{2}, f_c = \frac{3}{4}$ are the cumulative frequencies of the previous symbols. a is the first so 0, b is preceded by a so is $P(a)$, c so is $P(a) + P(b)$. We do some trivial (for computers) calculation on the interval, depending on the text. Let's assume $T = abca$

The size at the beginning is the size of the whole interval $[0, 1)$, so $s_0 = 1$

The first symbol is a , we go in its interval which is $[0, \frac{1}{2})$ and the size of the new interval is $s_1 = s_0 \cdot P(a) = \frac{1}{2}$. This new interval must be partitioned according to the probabilities, so the first partition is of size $P(a) = \frac{1}{2}$ of the current interval so it's $\frac{1}{4} = s_1 \cdot P(a)$, the second and third are of size $P(b) = P(c) = \frac{1}{4}$ so they are $\frac{1}{8} = s_1 \cdot P(b) = s_1 \cdot P(c)$

The first value, the border between the first two partitions, is 0 (the beginning of the new interval) + the size of the first partition $\frac{1}{4} = \frac{1}{4}$, the second border value is $0 + \frac{1}{4}$ for the first partition + $\frac{1}{8}$ for the second = $\frac{3}{8}$, then the last border is the last value which is $\frac{1}{2}$

Programmatically, the interval starts at $0 + f_a \cdot s_1 = 0$. The second partition starts at $0 + f_b \cdot s_1 = 0 + \frac{1}{2} \cdot \frac{1}{2} = \frac{1}{4}$ and so on. So the starting position of a letter is the current size times the cumulative frequency of that letter.

The second symbol is a b , so we zoom in it's interval inside s_1 , which goes from $[\frac{1}{4}, \frac{3}{8})$ of size $s_2 = s_1 \cdot P(b) = \frac{1}{2} \cdot \frac{1}{4} = \frac{1}{8}$ and we partition in three parts again like before. The first partition is again for a and starts at $\frac{1}{4}$, the beginning of the interval. The second partition, of b , starts at $\frac{1}{4} + f_b \cdot s_2 = \frac{1}{4} + \frac{1}{2} \cdot \frac{1}{8} = \frac{5}{16}$ and the third starts at $\frac{1}{4} + f_c \cdot s_2 = \frac{11}{32}$

So the arithmetic coding keeps the two integers describing the interval $[l_n, l_n + s_n)$: the left and the right endpoints of the interval. We start with $l_0 = 0, s_0 = 1$, and in every step $s_{n+1} = s_n \cdot P(T[n+1])$ and $l_{n+1} = l_n + f_{T[n+1]} \cdot s_n$, with $T[n+1]$ being the character in position $n+1$ which we are encoding.

What Arithmetic Coding Sends on the Network Let's start with an example: $T = abac \rightarrow$ will output $[\frac{19}{64}, \frac{5}{16})$. With a semi-static model, where we count occurrences, $f(a) = \frac{2}{4} = \frac{1}{2}, f(b) = f(c) = \frac{1}{4}$. Arithmetic coding sends the number in the middle of the interval, so in this case the size of the interval is $s = \frac{1}{64}$, the left extreme is $l = \frac{19}{64}$ therefore the picked number is $l + \frac{s}{2} = \frac{19}{64} + \frac{1}{128} = \frac{39}{128}$. To represent this number we can do the classical representation of $\frac{v}{2^k}$ which is $\text{bin}(v)$ over k bits, but we could do with less bits. So we represent a truncation of the number but guaranteeing that the truncation is larger than l .

Remember we send the number and the length of T , because the number could correspond to an infinite number of texts.

Theorem We truncate to $d = \lceil \log_2 \frac{2}{s} \rceil$ bits. The truncated number $\text{converter}(l + \frac{s}{2}, d) \in [l, l + s)$. Remember that, according to a lemma, the number of information lost due to truncation at d bits is 2^{-d} .

$$2^{-d} = 2^{-\lceil \log_2 \frac{2}{s} \rceil} = \frac{1}{2^{\lceil \log_2 \frac{2}{s} \rceil}} \leq \frac{1}{2^{\log_2 \frac{2}{s}}} = \frac{1}{\frac{2}{s}} = \frac{s}{2}$$

So this is a safe truncation, the number is still $\geq l$.

In the example $d = \lceil \log_2 \frac{2}{s} \rceil = \lceil \log_2 \frac{2}{\frac{1}{64}} \rceil = \lceil \log_2 128 \rceil = 7$ bits, so we do $\text{converter}(\frac{39}{128}, 7) = 0100111$ which is just 39 over 7 bits but it's a particular case, 39 can be represented over 7 bits without truncation.

How good is the arithmetic compressor? For Huffman, we had $H \leq L_{Huff} < H + 1$ bits, giving Huffman compression $< nH + n$ with $n = |T|$, because Huffman can only use an integer number of bits per codeword while the entropy H uses real numbers.

Arithmetic compression $< nH + 2$, with a theoretical constant 2 in the case of infinite arithmetic precision. In reality, it's something like $< nH + c \cdot n$ with $c \simeq \frac{1}{100}$.

Theorem $d < nH + 2$

$$d = \lceil \log_2 \frac{2}{s} \rceil = \lceil 1 - \log_2 s \rceil < 2 - \log_2 s = 2 - \log_2 \prod_{i=1}^{|T|} p(T[i]) = 2 - \sum_{i=1}^{|T|} \log_2 p(T[i]) =$$

In the previous example of $T = abac$ this would be $2 - [\log_2 p(a) + \log_2 p(b) + \log_2 p(a) + \log_2 p(c)] = 2 - [2 \cdot \log_2 p(a) + 1 \cdot \log_2 p(b) + 1 \cdot \log_2 p(c)]$ so the number of occurrences times the probability

$$= 2 - \sum_{\sigma} \text{occ}(\sigma) \cdot \log_2 p(\sigma) = 2 - n \cdot \sum_{\sigma} \frac{\text{occ}(\sigma)}{n} \cdot \log_2 p(\sigma) =$$

For sufficiently large texts, the number of occurrences $\text{occ}(\sigma)$ over n converges to the frequency of σ $f(\sigma)$, and in the semi-static model $p(\sigma)$ is $f(\sigma)$

$$2 - n \sum_{\sigma} p(\sigma) \cdot \log_2 p(\sigma) = 2 + n \sum_{\sigma} p(\sigma) \log_2 \frac{1}{p(\sigma)}$$

and $H = \sum_{\sigma} p(\sigma) \log_2 \frac{1}{p(\sigma)}$.

Recently, ANS (Asymmetric Numeral Systems) was proposed as a variation of the Arithmetic Encoding we've seen and it's actually better than this bound.

So in every case you have to look at the entropy: if it's very small, so the source is highly compressible, then the nH part is close to 0 and additive part $+n$ or $+c \cdot n$ makes the difference. When the entropy is very large then nH dominate and we can use Huffman.

Decoding Whenever we decompress, we have the preamble and the body of the compressed file. The preamble of the arithmetic encoders usually consists of the probabilities of the symbols and the length of the text.

For example $P(a) = \frac{1}{2}, P(b) = P(c) = \frac{1}{4}$ and the text length = 2 with the body being $(111)_2$. The first thing to do is to transform the body into a real number (keeping the fraction, because then it's easier to do the calculations by hand): $(111)_2 = .111 = \frac{1}{2} + \frac{1}{4} + \frac{1}{8} = \frac{7}{8}$.

Now we start from $[0, 1)$ and partition according to the probabilities: from 0 to $\frac{1}{2}$ is a , from $\frac{1}{2}$ to $\frac{1}{2} + \frac{1}{4} = \frac{3}{4}$ is b and from $\frac{3}{4}$ to $\frac{3}{4} + \frac{1}{4} = 1$ is c . With $\frac{7}{8}$ we go in the c partition (zoom into $[\frac{3}{4}, 1]$) and partition. Now the size is $s_1 = \frac{1}{4}$ so a goes from the beginning $\frac{3}{4}$ to $\frac{3}{4} + s_1 \cdot P(a) = \frac{3}{4} + \frac{1}{4} \cdot \frac{1}{2} = \frac{7}{8}$, same reasoning for b which goes from $\frac{7}{8}$ to $\frac{15}{16}$.

Every interval is closed below and open up, so $\frac{7}{8}$ belongs to the interval b . We stop because we found text len = 2 symbols, and return $T = cb$.

0.8.3 Lempel-Ziv Family

LZ Compressors

1. Parsing, transform the text into a sequence of triples.
2. Encoding

Parsing $T = aacaacab|caaaaaaac$, where we've compressed up until the line and need to compress the next part. So the first part is the **dictionary**, all the substrings that starts in the first part are the dictionary. It parses the next part and finds the longest substring that starts in the first part:

c is in the first part

ca too

caa too

$caaa$ doesn't

So it takes caa plus one character: $caaa$ is called a **phrase**, it's represented as a copy of the corresponding substring in the dictionary, and is represented with $\langle d, l, \text{char} \rangle$ corresponding to the distance of the copy (in this case, from caa , so $d = 6$), the length (caa has $l = 3$) and the next character (a). We then move the bar after the a and repeat for $T = aacaacabcaaa|aaaaac$.

Now the important part is that the copy start before: in this case, we have 4 a s, copied beginning from just the last a before the bar: we copy $T = aacaacabcaaa|aaaaac$ from $T = aacaacabcaaa|aaaaac$ and it's not important that overlaps, becoming $\langle 1, 4, c \rangle$

What happens at the beginning? We start from $T = |aacaacabcaaaaaaac$, encode $\langle 0, 0, a \rangle$ and the bar goes after the last character encoded so $T = a|acaacabcaaaaaaac$. We emit $\langle 1, 1, c \rangle$ and $T = aac|aacabcaaaaaaac$, then $\langle 3, 4, a \rangle$ giving $T = aacaaca|bcaaaaaaac$ and so on.

Encoding The crucial part is how to encode the numbers and the characters, we cannot represent them in full because we can end up with a larger space complexity than the original data.

For example, we could define three "texts":

text of distances $T_{dist} = 01361$

text of lengths $T_{len} = 01434$

text of characters $T_{char} = acaac$

and we could encode these three texts in any way we want as seen before.

Window to copy Whenever we look for a substring in the dictionary to copy, we cannot look arbitrary before: typically the compressors have a window of words of length w and we consider only copies that starts at most w character before so $d \leq w$. In **gzip** we have options $-1, \dots, -9$ which specifies the windows from 100 Kb to 900 Kb. The larger the windows the more you compress, so less phrases and **faster in decompression** because every phrase tells you to go back and copy, it's a random jump, so with less random jumps the caching is more effective. But with larger windows we have slower compression.

The approach discussed is called **LZ 77**.

How to find the copies There are several algorithms in literature, let's see the one used by **gzip** due to it being interesting from a programming point of view. Let's say we are in the situation $T = \dots | caaa \dots$: a brute-force way would be to take c and search for it going back and find it, then take ca and go back and search and find it so on until we don't find the occurrence of the substring. This is really slow, we need something faster: filtering. We want to find the candidate positions and then check them. So I want some tool that tells me the (hopefully few) interesting positions to check. This filter can be built with an hashtable over triples of characters. Let's see an example with $w = 7$: $T = acaacab|caaa \dots$ from position 2, 3, 4, 5, \dots . For every previous position, it keeps a key in the hashtable of three characters:

$\langle bca, 8 \rangle$

$\langle abc, 7 \rangle$

$\langle cab, 6 \rangle$

$\langle aca, 5 \rangle$

$\langle aac, 4 \rangle$

$\langle caa, 3 \rangle$

$\langle aca, 2 \rangle$

So we can already see that two $acas$ will collide in the same slot, and we have to keep bot satellite informations 5 and 2. **gzip** finds copies only if they are longer or equal than three characters: it's useless to substitute one or two characters with a triple.

So it takes the next three characters caa and look for in the hashtable: it can find a collision, for example $\langle caa, 3 \rangle$, $\langle aac, 4 \rangle$ and $\langle cab, 6 \rangle$ can collide, these are the candidate positions so it compares over them.

After encoding the triple, we shift the window so the table is cut: each triple points to the next one so we can start from the head and simply cancel triples. We of course also add the new triples.

LZSS Doesn't use triples but pairs by making the following observation: either we copy, because $d > 0$ or we don't copy and meet one single character. So whenever $d = 0$ only the character is significant, and whenever we copy the character is useless. So LZSS considers just $\langle d, l \rangle$ (no extra characters) or $\langle 0, \text{char} \rangle$.

So for example $T = aacaacabcaaaaaaac$ we emit $\langle 0, a \rangle$ and $T = a|acaacabcaaaaaaac$, we can only copy a so we output $\langle 1, 1 \rangle$ and $T = aa|caacabcaaaaaaac$, then $\langle 0, c \rangle$ and $T = aac|aacabcaaaaaaac$, then we can copy $aaca$ starting from the beginning (3 characters before) and copying 4 characters, so we output $\langle 3, 4 \rangle$ and $T = aacaaca|bcaaaaaaac$.

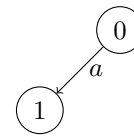
The decoder expects pairs, reads the first number and if it's zero then knows that the next element is a character, otherwise knows that the next one is a number.

LZ78 Different dictionary, this time is explicit. $T = aabaacabcabcb$, we have and output and a explicitly constructed dictionary that starts with ϵ , the empty string.

Output

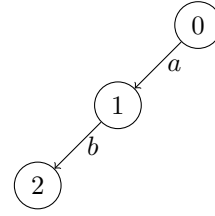
$T = aabaacabcabcb$

Since we don't find anything in the dictionary, which at this time is just the node 0 corresponding to ϵ , we emit $\langle 0, a \rangle$ and since we are processing one character we append it to the dictionary.



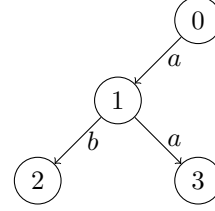
$T = a|abaacabcabcb$

We start from the root, match a and can't find b so we stop after b , emit $\langle 1, b \rangle$ and add b to the trie.



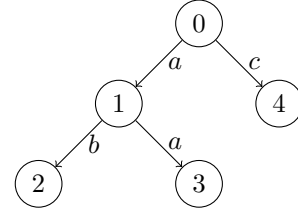
$T = aab|aacabcabcb$

We have aa , find just one a so we encode $\langle 1, a \rangle$ and add the second a to the first a subtree.



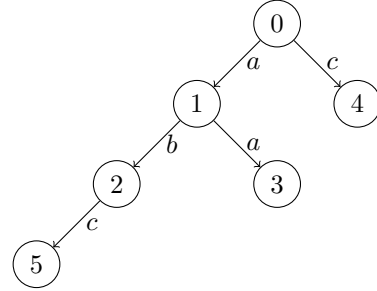
$T = aabaa|cabcabcb$

We have no c from the root, encode $\langle 0, c \rangle$ and add it to the root.



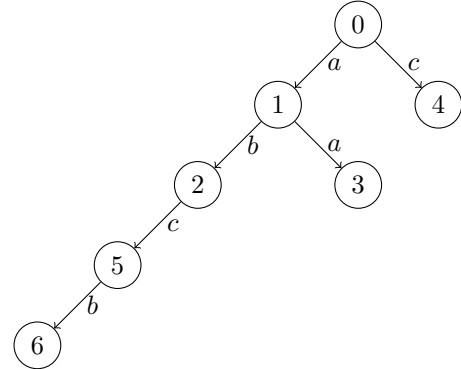
$T = aabaac|abcabcb$

We find ab in the trie, so we emit $\langle 2, c \rangle$ and add c to the b subtree.



$T = aabaacabc|abcb$

We emit $\langle 5, d \rangle$ and add d trie (although it's useless since we've finished)

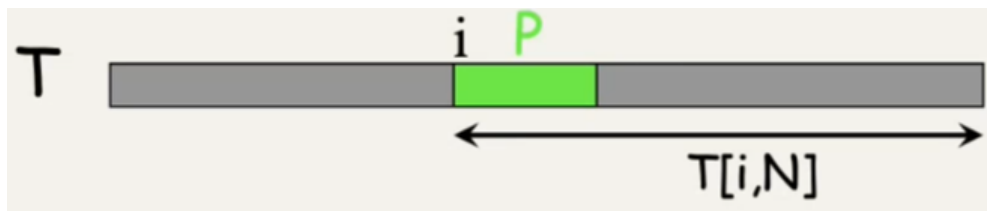


The tries of course is then dropped and we just encode the sequence of pairs: we have two texts (node numbers and characters) that can be encoded as we like.

ZLIB Implementation of LZ77 which requires the array to compress A , its length n , and an array of output Out of length m passed as a pointer (so $\&m$) initialized as $m = 1.2 \cdot n$ so a 20% increase in size of A . This because the system can't know if A is compressible or not, so the extra size is to account for the uncompressible cases (more because we use pairs or triples). When the procedure returns, m has been updated with the occupied size which, possibly, is much smaller than $1.2 \cdot n$

0.8.4 The Suffix Array

Notations Pattern P occurs at position i of $T \Leftrightarrow P$ is a prefix of the i -th suffix of T (i.e. $T[i, N]$)



So we can transform the problem of searching for occurrences into a problem of searching for prefixes. Occurrences of P in T are all the suffixes of T having P as a prefix.

$\text{SUF}(T)$ = sorted set of all suffixes of T . Algorithmic reduction from substring search to prefix search: for example, using the trie data structure. We have a problem in space, because the total length of the suffixes is n^2 .

We consider an ending character, which sometimes is noted as $\#$ and sometimes as $\$$. So for example $T = \text{mississippi}\# \Rightarrow \text{SUF}(T)$:

$\#$

$i\#$

$ippi\#$

$issippi\#$

$ississippi\#$

$mississippi\#$

$pi\#$

$ppi\#$

$sippi\#$

$ssissippi\#$

$ssippi\#$

$ssissippi\#$

So all suffixes in $\text{SUF}(T)$ having prefix P are in a contiguous portion of the sorted set. Also the starting position of the contiguous portion is the lexicographic one of P , meaning that if I can find the position of my prefix in $\text{SUF}(P)$ I can simply scan ahead to find all the suffixes with P prefix. So we can consider the suffix array, where every element is a pointer to the position in T :

12

11

8

5

2

1

10

9

7

4

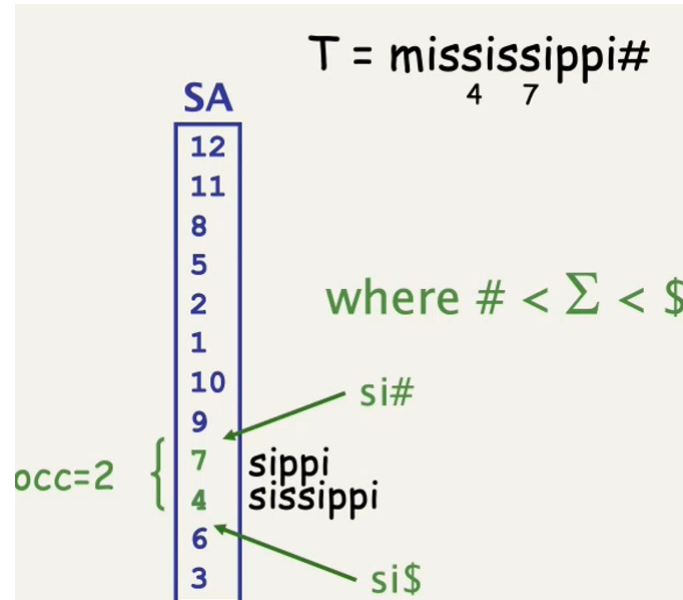
6

3

The suffix array SA occupies a lot less space than the n^2 of $SUF(T)$, it's $O(n)$ in $n = |T|$ and every element is a number of at most $\log_2 n$ bits, so is $\Theta(n \log_2 n)$ bits. In practice, is $5n$ bytes.

To look for a pattern, indirected binary search on SA: $O(p) = O(|P|)$ time per suffix comparison. So go in the middle element, compare P to the substring pointed by the element and so on, binary search. The total search cost is $O(p \log_2 n)$ time.

Listing the occurrences is easy: take the element, access the position and compare, and go to the next element if the comparison is successful. This costs the number of occurrences times p , which can be costly with many occurrences. We can speed this up by finding the entire block: we look for $P\#$ with $\# < \sigma \forall \sigma \in \Sigma$ and for $P\$$ for $\sigma < \$ \forall \sigma \in \Sigma$. Therefore, the block of occurrences of P is delimited by those two elements.



With a total time of $O(p \log_2 n + \text{occ})$.

Longest Common Prefix Array The LCP array has elements that indicates the length of the longest prefix shared by adjacent suffixes. For example, the first and second suffixes in SA share no prefix (the first is " $\#$ " and the second is " $i\#$ "), so the first element of LCP is 0. LCP:

0
0
1
4
0
0
1
0
2
1
3

How long is the common prefix between $T[i, \dots]$ and $T[j, \dots]$? The minimum element between them in LCP, so for example between 5 and 10 the common prefix is long 0 characters. So the min of the subarray $LCP[h, k]$ such that $SA[h] = i$ and $SA[k] = j$.

0.8.5 Burrows-Wheeler Transform

Two very simple compressors

MTF Move-To-Front

Start with a list of symbols, a text $L = [a, b, c, d, \dots]$ and for each input symbol s :

1. Output the position of s in L
2. Move s to the front of L

For example, $L = [a, b, c, l], S = \text{cabala} \Rightarrow \text{MTF}(S) = 3\ 2\ 3\ 2\ 4\ 2$.

It's a **dynamic code** (the same letter can be encoded with different values) and has memory (if the symbol has been read recently, its number will be small).

RLE Run-Length-Encoding

Specifically designed for spacial locality.

Example: $\text{abbbaacccca} \Rightarrow (a, 1), (b, 3), (a, 2), (c, 4), (a, 1)$

In binary string just numbers and the first bit is sufficient.

Burrows-Wheeler Transform Let's take a given text $T = \text{mississippi}\#$ and we start by rotating T :

```
mississippi#
ississippi#m
ssissippi#mi
sissippi#mis
issippi#miss
ssippi#missi
sippi#missis
ippi#mississ
ppi#mississi
pi#mississip
i#mississipp
#mississippi
```

This produces $|T|$ rows, giving a square matrix (**BWT matrix**). Then we sort the rows ($\#$ is the smallest character):

```
# mississipp i
i #mississip p
i ppi#missis s
i ssippi#mis s
i ssissippi# m
m ississippi #
p i#mississi p
p pi#mississ i
s ippi#missi s
s issippi#mi s
s sippi#miss i
s sissippi#m i
```

Again a square matrix. We notice that the first column, F , is ordered and contains all the characters of T : this because we are considering circling rotations and fixing a column every character will pass by that column. We cannot compress just F because every text comprised of those characters (in the example, 4 "i"s, 1 "m", 2 "p"s and 4 "s"s) would give the same F . The trick is done with L , the column of the last characters: the transformation is invertible if we consider the sequence of characters in the last column.

So given a string T , we return L , which in this case $L = \text{ipssm}\#\text{pississ}$.

Note that the character in L is the one that precedes the character in F . With "mississippi" this transformation can't show its true power. With natural language texts, there are a certain number of characters that precedes other characters: for example, in english the letter "n" succeeded by a space can be preceded by an "a", an "o" and other few vowels ("an", "on", "in"...), so in these portions of texts the array L will contain just few repeated characters. This means that L is **locally homogeneous**, so is **highly compressible**.

Bzip Algorithm

1. Move-to-Front of L
2. Run-Length coding
3. Statistical coder

Computing the BWT matrix If we look at the matrix and notice the position of $\#$, we notice that we need to sort just the suffixes of T :

```
#mississipp i
i#mississip p
ippi#missis s
issippi#mis s
ississippi# m
mississippi #
pi#mississi p
ppi#mississ i
sippi#missi s
sissippi#mi s
ssippi#miss i
ssissippi#m i
```

So $\#$ is a dividing character. So these are the sorted suffixes of T , and we can compute it with a suffix array (SA) construction algorithm, with a linear space instead of quadratic. Given the items of the SA , the corresponding item of L is $L[i] = T[SA[i] - 1]$. This requires many jumps in memory, so it's the main reason why bzip is significantly slower than gzip. So this spurred a number of publications in '94-'10 on SA construction.

SA with qsort Initialize the array with the text T and then just sort with qsort. The difference is that every position is treated as a suffix, not as a character. So with qsort we will sort every suffix (from position i of the original T up to the $\#$), giving SA at the end of the procedure.

The procedure is inefficient: $\Theta(n^2 \log n)$ in the worst case with $\Theta(n \log n)$ cache misses.

L→F Mapping Can we map L 's characters onto F 's characters? Need to distinguish between equal characters. Every column i is a permutation of the original T given by the first $i - 1$ characters. Given a character on L , I have to find the equal character on F , so I will know the previous character. Iterating this procedure gives the original T from right to left.

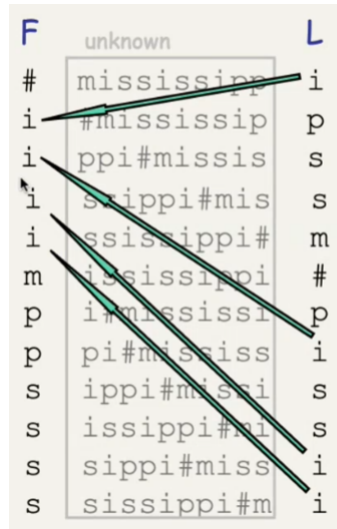
Taken two characters in L , e.g. two "i"s, the first to appear in L is alphabetically smaller than the second. I bring each character to the front of its corresponding rectangle: the alphabetical order doesn't change.

F	unknown	L
#	mississipp	i
i	#mississip	p
i	ppi#missis	s
i	ssippi#mis	s
i	ssissippi#	m
m	ississippi	#
p	i#mississi	p
p	pi#mississ	i
s	ippi#missi	s
s	issippi#mi	s
s	sippi#miss	i
s	sissippi#m	i

Each row will correspond to another row with the rotated character in F and another character in L which is the previous character. Note that the rows may come closer, but the green will always come before the red.

F	unknown	L
#	mississipp	i
i	#mississip	p
i	ppi#missis	s
i	ssippi#mis	s
i	ssissippi#	m
m	ississippi	#
p	i#mississi	p
p	pi#mississ	i
s	ippi#missi	s
s	issippi#mi	s
s	sippi#miss	i
s	sissippi#m	i

These stability in the ordering guarantees that the mapping is trivial.



These operations are just $\text{Rank}_{\text{char}}(\text{pos})$ and $\text{Select}_{\text{char}}(\text{pos})$ and the reason why these operations are so popular nowadays. $\text{Rank}_i(\text{pos})$ tells how many "i"s are above pos.

Decompression After decompressing in the inverse order of the compression, we end up with L . First thing to do is to find F : sort L and find F . We now need to construct the string, and the first character from which we start, constructing from right to left, is # which is the first character of F . So the last character of T is the corresponding character in L , which in the example is an "i". Whenever I have an item in L , I compute the corresponding character in F and iterate. The corresponding "i" in F of the first "i" in L is the first i we find in F , in the second row. And so on. Several issues regarding efficiency in time and space.

0.9 Hashing and Dictionary Problem

Hashing Typically related to the **Dictionary problem**: we have a set of objects $D = \{o_1, \dots, o_n\}$ with every o_i associated to $\langle \text{key}, \text{satellite information} \rangle$. We want to solve queries like:

Membership(k): is $k \in D$?

Insertion(k): $D = D \cup \{k\}$

Deletion(k): $D = D \setminus \{k\}$

The hashing function $h : U \rightarrow [0, m - 1]$ from the universe U , with a table $T[0, m]$. The hash function is applied to the key which gives a position inside the table. Two objects can collide in the same position, and stay "outside" the table in a overflow list. This is called **hashing with chaining**. If U is small, then a direct table is better.

This approach has a space complexity of the size of the table (m) plus the number of objects outside the table (n), so $O(m + n)$. Note that m are pointers, the cells of the table, while n are pointers and keys: if the key is an integer then is 4-8 bytes like the pointers, but if it's a string then it's another pointer.

Loading factor $\alpha = \frac{n}{m}$, ratio between the number of keys and the size of the table.

Analysis Up until now in our studies we've seen h that were **simple uniform hashing**, meaning that $\forall k P(h(k) = i) = \frac{1}{m}$. This means that every key is independent and is **computationally unfeasible**, although possible it takes too much space and time. Let's consider this kind of function.

The insertion is $O(1)$ because we simply append in T .

The Membership and Deletion both require a search, and the search costs the average overflow list length: a constant time to access the position required and a scan of the list in that position to search for the key.

The average list length is

$$E[\text{length of a list}] = \sum_{k' \in D, k' \neq k} P(h(k') = h(k)) = \sum_{k' \in D, k' \neq k} P(h(k') = i) = \sum_{k' \in D, k' \neq k} \frac{1}{m} = \frac{n}{m} = \alpha$$

So the average time for a search is the loading factor α . Typically we assume that the computation of $h(k)$ is $O(1)$ so the cost of both Membership and Deletion is $O(1 + \alpha)$ to account for the computation.

If the size of the table is chosen proportional to n then α is constant. We have a problem: how much is n ? It's certainly dynamic.

Constant Time in Dynamic Context Let's consider the evolution over time. At the beginning, $t = 0$ we have n_0 keys and a table of size $m_0 = 2n_0$ so we allocate a table of twice the size, meaning $\alpha_0 = \frac{1}{2}$. Then we add and delete keys.

Let's consider a time $t = 1$ where either $\# \text{ keys} \geq 2n_0$ or $\# \text{ keys} \leq \frac{n_0}{2}$. Before this time, we certainly have $\frac{m_0}{4} = \frac{n_0}{2} < \# \text{ keys} < 2n_0 = m_0$. At $t = 1$ the number of keys is getting either too large or too small, so we **resize the table**: if $\# \text{ keys} \geq 2n_0$ then $m_1 = 2m_0$ so the table doubles, else if $\# \text{ keys} \leq \frac{n_0}{2}$ then $m_1 = \frac{m_0}{2}$ the table halves. The 2 is just a constant for example. What's important is that between $t = 0$ and $t = 1$ the number of keys is proportional for a given constant to the size of the table. But resizing the table requires destroying the current table and creating a new one, so in $t = 1$ we have a peak of complexity because we execute a lot of operations. The cost is the cost of moving keys into the new table and destroying the old table.

Moving keys: it's dependent on how many keys we have to move and how much moving a key costs.

We have to move $2n_0$ keys, and the cost of insertion is still constant, so the complexity is $O(2n_0 \cdot 1)$

Destroying a table: it's its size so $O(m_0)$

Constructing the new table requires $O(m_1)$, so the total cost is $O(2n_0 + m_0 + m_1)$. m_0 and m_1 are always $O(n_0)$, so the cost is $O(n_0)$ and in fact is $\Theta(n_0)$.

Amortized Analysis So we can't say that an hashtable is constant, because pointwise it's $O(1)$ then $O(n_0)$ and so on. Instead, we do an **amortized analysis**: amortized meaning that we don't count the pointwise complexity but the complexity over a sequence of operations.

Let's consider the window of time between $t = 0$ and $t = 1$ in the case where $n_1 = 2n_0$. We can have insertions and deletion, but at least we did n_0 insertions (in general, operations). So n_0 operations of $O(1)$ plus $\Theta(n_0)$ at $t = 1$ for the resize, meaning that in this window of time we did $O(2n_0) = O(n_0)$.

So in the amortized analysis we take the worst case cost of a sequence of operations and divide it by the number of operations.

Hashing functions

Simple uniform: every key can go everywhere in the table

This means that k_1 can go everywhere in principle, same for k_2 and so on up to $k_{|U|}$, same thing (can go to 0, ..., $m - 1$). I can have a function that maps every key in 0, a function that maps k_i in $i \bmod m - 1$...any combination is possible, so the number of simple uniform hashes is at least the number of mappings, which are $m^{|U|}$. So to be represented every function needs $\Omega(|U| \log m)$ bits. So considering $|U| = 2^{64}$, as in the case of keys of 8 bytes, we would need 2^{64} bits which is *a lot*.

So these function exists but they require too much space (\Rightarrow time) to be represented.

Universal hashing: a class H of hash functions $h : U \rightarrow [0, m]$ is called **universal**

$\Leftrightarrow \forall x \neq y \in U \quad \#\{h \mid h(x) = h(y)\} \leq \frac{|H|}{m}$, so for every pair of items there are at most $\frac{1}{m}$ hash functions that collide them.

Universal Hashing with Chaining Let's compute $E[\text{list length in cell } i]$, remembering that $i = h(k)$:

$$E[\text{list length in cell } i] = \sum_{k' \in D, k' \neq k} 1 \cdot P(h(k') = h(k))$$

The probability is linked to the function I pick in H , so it's the random factor, so it's $P_{h \in H}$ and h is taken from an universal class so the probability of $k' \neq k$ colliding in h is $\frac{1}{m}$, so

$$E[\text{list length in cell } i] = \sum_{k' \in D, k' \neq k} \frac{1}{m} = \frac{n}{m} = \alpha$$

So the key concept is that the h is from an universal class and we have two different keys.

Does this class of functions exists? How practical is it?

Example of universal hashing $h : U \rightarrow [0, m]$ with $u = |U|$. So a key k consists of $\log_2 u$ bits.

Let's divide the key k into blocks of $\log_2 m$ bits: so blocks from k_0 to k_{r-1} where $r = \frac{\log_2 u}{\log_2 m}$ = the number of blocks.

So $\forall k_i \ 0 \leq k_i < m$

Let's consider a parameter of h : a of size $\log_2 u$ bits with $0 < a < u$ and the same partitioning (blocks of $\log_2 m$ bits giving a_0, \dots, a_{r-1}). Note that a is **picked at random**.

We get

$$h_a(k) = \sum_{i=0}^{r-1} a_i \cdot k_i \bmod n$$

with $a \neq 0$ and n prime number. So h_a is a class of functions, $\#h_a = u - 1$

Let's prove that it's universal, so the property $P(h_a(x) = h_a(y)) = \frac{u-1}{m}$