

4. THE GREEDY METHOD

Serge Gaspers

Katie Clinch

serge.gaspers@unsw.edu.au

k.clinch@unsw.edu.au

Course admin: Song Fang cs3121@cse.unsw.edu.au

School of Computer Science and Engineering, UNSW Sydney

Term 3, 2023

- 2. Assorted problems
- 2.1 Activity selection
- 2.2 Cell tower placement
- 2.3 Minimising job lateness
- 2.4 Tape storage
- 2.5 Interval stabbing
- 2.6 Knapsack problem
- 2.7 The Huffman Code
- 3. Applications to graphs
- 3.1 Directed graph structure
- 3.2 Single source shortest path
- 3.3 Minimum spanning trees
- 4. Puzzle

What is a greedy algorithm?

Answer

A greedy algorithm is one that solves a problem by dividing it into stages, and rather than exhaustively searching all the ways to get from one stage to the next, instead only considers the choice that appears best.

This obviously reduces the search space, but it is not always clear whether the locally optimal choice leads to the globally optimal outcome.

Does the greedy method always work?

Answer

No!

Suppose you are searching for the highest point in a mountain range. If you always climb upwards from the current point in the steepest possible direction, you will find a peak, but not necessarily the highest point overall.

Is there a framework to decide whether a problem can be solved using a greedy algorithm?

Answer

Yes, but we won't use it.

The study of <u>matroids</u> is covered in CLRS. We will instead prove the correctness of greedy algorithms on a problem-by-problem basis. With experience, you will develop an intuition for whether the greedy method is useful for a particular problem.

How do we prove that a greedy algorithm is correct?

Answer

There are two main methods of proof:

- Greedy stays ahead: prove that at every stage, no other sequence of choices could do better than our proposed algorithm.
- **Exchange argument**: consider an alternative solution, and gradually transform it to the solution found by our proposed algorithm without making it any worse.

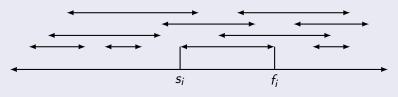
These are reminiscent of induction and contradiction respectively.

- 2. Assorted problems
- 2.1 Activity selection
- 2.2 Cell tower placement
- 2.3 Minimising job lateness
- 2.4 Tape storage
- 2.5 Interval stabbing
- 2.6 Knapsack problem
- 2.7 The Huffman Code
- 3. Applications to graphs
- 3.1 Directed graph structure
- 3.2 Single source shortest path
- 3.3 Minimum spanning trees
- 4. Puzzle

- 2. Assorted problems
- 2.1 Activity selection
- 2.2 Cell tower placement
- 2.3 Minimising job lateness
- 2.4 Tape storage
- 2.5 Interval stabbing
- 2.6 Knapsack problem
- 2.7 The Huffman Code
- 3. Applications to graphs
- 3.1 Directed graph structure
- 3.2 Single source shortest paths
- 3.3 Minimum spanning trees
- 4. Puzzle

Problem

Instance: A list of n activities, with starting times s_i and finishing times f_i . No two activities can take place simultaneously.



Task: Find a <u>maximum size</u> subset of compatible activities.

Possible ideas:

- Choose shorter activities first?
- Choose activities which have fewest conflicts?
- Choose activities with earlier end times first?

Attempt 1

Always choose the shortest activity which does not conflict with the previously chosen activities, then remove the conflicting activities and repeat.



In the above example, our proposed algorithm chooses the activities in blue, then has to discard all the red activities, so clearly this does not work.

Possible ideas

- C/hdøse/shokter/activities/fyrst?
- Choose activities which have fewest conflicts?
- Choose activities with earlier end times first?

Attempt 2

Maybe we should always choose an activity which conflicts with the fewest possible number of the remaining activities? It may appear that in this way we minimally restrict our next choice . . .



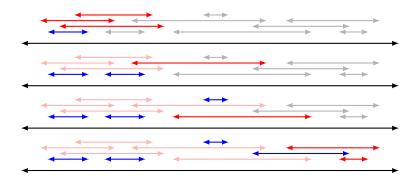
As appealing this idea is, the above figure shows this again does not work!

Possible ideas

- C/hdøse/shokter/activities/first?
- C/h/døse/activitives/which/have/fewest/døhflicts?/
- Choose activities with earlier end times first?

Solution

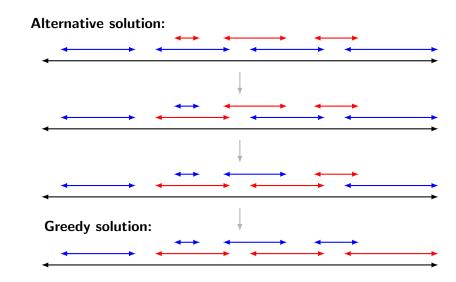
Among those activities which do not conflict with the previously chosen activities, always choose the activity with the earliest end time (breaking ties arbitrarily).



Sketch proof. To prove the correctness of our algorithm, we will use an <u>exchange argument</u>. We will show that any alternative solution can be transformed into a solution obtained by our greedy algorithm with at least as many activities.

- Find the first place where the chosen activity violates the greedy choice.
- What if we replace that activity with the greedy choice?

Alternative solution: Greedy solution:



- Does the new selection have any conflicts? No!
- Does the new selection have the same number of activities? Yes!
- So the greedy choice is actually just as good as the choice used in the alternative solution!
- We replace it and repeat.
- Continuing in this manner, we can eventually "morph" any alternative solution into the greedy solution, thus proving the greedy solution is optimal.

- What is the time complexity of the algorithm?
- We represent activities by ordered pairs of their starting and their finishing times and sort them in increasing order of their finishing time (the second coordinate), in $O(n \log n)$ time.
- We go through this sorted list in order. How do we tell whether an activity conflicts with the already chosen activities?

- Suppose we are up to activity i, starting at s_i and finishing at f_i , with all earlier finishing activities already processed.
- If all previously chosen activities finished before s_i , activity i can be chosen without a conflict. Otherwise, there will be a clash, so we discard activity i.
- We would prefer not to go through all previously chosen activities each time.

- We need only keep track of the latest finishing time among chosen activities.
- Since we process activities in increasing order of finishing time, this is just the finishing time of the last activity to be chosen.
- Every activity is therefore either chosen (and the last finishing time updated) or discarded in constant time, so this part of the algorithm takes O(n) time.
- Thus, the algorithm runs in total time $O(n \log n)$, dominated by sorting.

A related problem

Instance: A list of n activities with starting times s_i and finishing times $f_i = s_i + d$; thus, all activities are of the same duration. No two activities can take place simultaneously.

Task: Find a subset of compatible activities of <u>maximal total</u> duration.

Solution

Since all activities are of the same duration, this is equivalent to finding a selection with a largest number of non conflicting activities, i.e., the previous problem.

What happens if the activities are not all of the same duration?

Solution

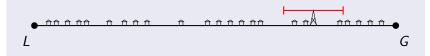
The greedy strategy no longer works - we will need a more sophisticated technique.

2. Assorted problems

- 2.1 Activity selection
- 2.2 Cell tower placement
- 2.3 Minimising job lateness
- 2.4 Tape storage
- 2.5 Interval stabbing
- 2.6 Knapsack problem
- 2.7 The Huffman Code
- 3. Applications to graphs
- 3.1 Directed graph structure
- 3.2 Single source shortest paths
- 3.3 Minimum spanning trees
- 4. Puzzle

Problem

Instance: Along the long, straight road from Loololong (in the West) to Goolagong (in the East), houses are scattered quite sparsely, sometimes with long gaps between two consecutive houses. Telstra must provide mobile phone service to people who live alongside the road, and the range of Telstra's cell tower is 5km.



Task: Design an algorithm for placing the minimal number of cell towers alongside the road, that is sufficient to cover all houses.

- Let's attempt a greedy algorithm, processing the houses west to east.
- The first house must be covered by some tower, which we place 5km to the east of this house.
- This tower may cover some other houses, but eventually we should reach a house that is out of range of this tower. We then place a second tower 5km to the east of that house.
- Continue in this way until all houses are covered.

Cell Towers 2

- At each house, we need to decide whether to place a new tower. This can be done in constant time by referring to the most recently created tower, which can itself be updated in constant time if necessary.
- Therefore this algorithm runs in O(n) time if the houses are provided in order, and $O(n \log n)$ time otherwise.

Exercise

Prove the correctness of this algorithm using an exchange argument.

Cell Towers 2

One of Telstra's engineers started with the house closest to Loololong and put a tower 5km away to the east. He then found the westmost house not already in the range of the tower and placed another tower 5 km to the east of it and continued in this way till he reached Goolagong.

- His junior associate did exactly the same but starting from Goolagong and moving westwards and claimed that his method required fewer towers.
- Is there a placement of houses for which the associate is right?

2. Assorted problems

- 2.1 Activity selection
- 2.2 Cell tower placement
- 2.3 Minimising job lateness
- 2.4 Tape storage
- 2.5 Interval stabbing
- 2.6 Knapsack problem
- 2.7 The Huffman Code
- 3. Applications to graphs
- 3.1 Directed graph structure
- 3.2 Single source shortest paths
- 3.3 Minimum spanning trees
- 4. Puzzle

Problem

Instance: A start time T_0 and a list of n jobs, with duration times t_i and deadlines d_i . Only one job can be performed at any time; all jobs have to be completed. If a job i is completed at a finishing time $f_i > d_i$ then we say that it has incurred lateness $l_i = f_i - d_i$.

Task: Schedule all the jobs so that the lateness of the job with the largest lateness is minimised.

We have two criteria we can order by: duration time t_i , and deadline d_i . Which is more important to get the optimal solution?

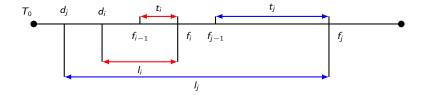
- Perform quicker tasks first?
- Prioritise tasks with earlier deadlines?
- Maybe both t_i and d_i need to be considered?

Solution

Ignore job durations and schedule jobs in the increasing order of deadlines.

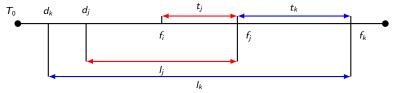
Proof of optimality

Consider any alternative schedule. We say that jobs i and j form an inversion if job i is scheduled before job j but $d_j < d_i$.

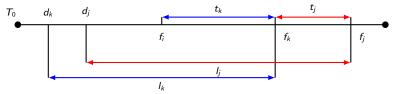


- We will show that there exists a scheduling without inversions which is at least as good.
- Recall that bubble sort only swaps adjacent array entries, and eventually sorts the array.
- Similarly, if we manage to resolve all inversions between adjacent jobs, eventually all the inversions will be resolved.

Solution with jobs j and k inverted:



Solution with one fewer inversion:



Note that swapping adjacent inverted jobs reduces the larger lateness!

2. Assorted problems

- 2.1 Activity selection
- 2.2 Cell tower placement
- 2.3 Minimising job lateness

2.4 Tape storage

- 2.5 Interval stabbing
- 2.6 Knapsack problem
- 2.7 The Huffman Code
- 3. Applications to graphs
- 3.1 Directed graph structure
- 3.2 Single source shortest paths
- 3.3 Minimum spanning trees
- 4. Puzzle

Problem

Instance: A list of n files of lengths l_i which have to be stored on a tape. Each file is <u>equally likely</u> to be needed. To retrieve a file, one must start from the beginning of the tape and scan it until the file is found and read. **Task:** Order the files on the tape so that the average (expected) retrieval time is minimised.

Solution

Store files in order of increasing length.

- Suppose we store the files on the tape in the order $l_1, l_2, ..., l_n$
- The probability we want to read any one of the files is $\frac{1}{n}$.
- So the expected retrieval time is:

$$E = \sum_{i=1}^{n} \frac{1}{n} (I_1 + \dots + I_i) = \frac{1}{n} \sum_{i=1}^{n} (I_1 + \dots + I_i)$$

$$= \frac{1}{n} (I_1 + (I_1 + I_2) + (I_1 + I_2 + I_3) + \dots + (I_1 + I_2 + \dots + I_n))$$

$$= \frac{1}{n} (nI_1 + (n-1)I_2 + (n-2)I_3 + \dots + 2I_{n-1} + I_n).$$

■ This is minimised if $l_1 \le l_2 \le l_3 \le ... \le l_n$, so we simply sort the files by increasing length for an $O(n \log n)$ solution.

But what if the files are not equally likely to be picked?

Problem

Instance: A list of n files of lengths l_i and probabilities to be needed p_i , $\sum_{i=1}^n p_i = 1$, which have to be stored on a tape. To retrieve a file, one must start from the beginning of the tape and scan it until the file is found and read. **Task:** Order the files on the tape so that the **expected** retrieval time is minimised.

Now we have two criteria we could order by: length l_i and probability p_i .

- \blacksquare Files with small length l_i should appear near the start
- Files with large p_i should appear near the start.

Solution

Store files in decreasing value of the ratio $\frac{p_i}{l_i}$.

■ If the files are stored in order $l_1, l_2, ... l_n$, then the expected time is proportional to

$$E = l_1 p_1 + (l_1 + l_2) p_2 + (l_1 + l_2 + l_3) p_3 + \dots + (l_1 + l_2 + l_3 + \dots + l_n) p_n$$

- We now show that this is minimised if the files are ordered in a decreasing order of values of the ratio p_i/l_i .
- Suppose that the files do not appear in this order. Then there are two adjacent files k and k+1 such that $\frac{p_k}{l_k} < \frac{p_{k+1}}{l_{k+1}}$
- What happens if we swap the order of these two files?

■ Before the swap, the expected retrieval time is:

$$E = l_1 p_1 + (l_1 + l_2) p_2 + (l_1 + l_2 + l_3) p_3 + \dots$$

$$+ (l_1 + l_2 + l_3 + \dots + l_{k-1} + l_k) p_k$$

$$+ (l_1 + l_2 + l_3 + \dots + l_{k-1} + l_k + l_{k+1}) p_{k+1}$$

$$+ \dots + (l_1 + l_2 + l_3 + \dots + l_n) p_n$$

After the swap, the new expected retrieval time is:

$$E' = l_1 p_1 + (l_1 + l_2) p_2 + (l_1 + l_2 + l_3) p_3 + \dots + (l_1 + l_2 + l_3 + \dots + l_{k-1} + l_{k+1}) p_{k+1} + (l_1 + l_2 + l_3 + \dots + l_{k-1} + l_{k+1} + l_k) p_k + \dots + (l_1 + l_2 + l_3 + \dots + l_n) p_n.$$

■ Thus, $E - E' = I_k p_{k+1} - I_{k+1} p_k$ is positive whenever $I_k p_{k+1} > I_{k+1} p_k$, i.e. when $p_k / I_k < p_{k+1} / I_{k+1}$.

- Consequently, E > E' if and only if $p_k/l_k < p_{k+1}/l_{k+1}$.
- This means that the swap decreases the expected time whenever $p_k/I_k < p_{k+1}/I_{k+1}$, i.e., if there is an inversion: file k+1 with a larger ratio p_{k+1}/I_{k+1} has been put after file k with a smaller ratio p_k/I_k .
- As long as the sequence is not sorted, there will be inversions
 of consecutive files, and swapping will reduce the expected
 time. Consequently, the optimal solution is the one with no
 inversions.

1. Introduction

2. Assorted problems

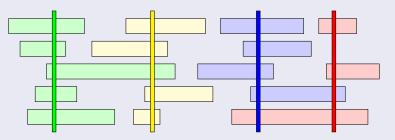
- 2.1 Activity selection
- 2.2 Cell tower placement
- 2.3 Minimising job lateness
- 2.4 Tape storage

2.5 Interval stabbing

- 2.6 Knapsack problem
- 2.7 The Huffman Code
- 3. Applications to graphs
- 3.1 Directed graph structure
- 3.2 Single source shortest paths
- 3.3 Minimum spanning trees
- 4. Puzzle

Problem

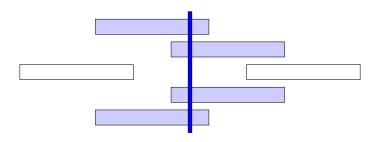
Instance: Let X be a set of n intervals on the real line, described by two arrays $X_L[1..n]$ and $X_R[1..n]$, representing their left and right endpoints. We say that a set P of points stabs X if every interval in X contains at least one point in P.



Task: Describe and analyse an efficient algorithm to compute the smallest set of points that stabs X.

Attempt 1

Is it a good idea to stab the largest possible number of intervals?

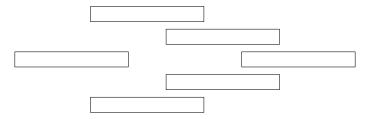


No! In the above example, this strategy needs three stabbing points rather than two.

Hint

The interval which ends the earliest has to be stabbed somewhere.

What is the best place to stab it?



1. Introduction

2. Assorted problems

- 2.1 Activity selection
- 2.2 Cell tower placement
- 2.3 Minimising job lateness
- 2.4 Tape storage
- 2.5 Interval stabbing
- 2.6 Knapsack problem
- 2.7 The Huffman Code
- 3. Applications to graphs
- 3.1 Directed graph structure
- 3.2 Single source shortest paths
- 3.3 Minimum spanning trees
- 4. Puzzle

Problem

Instance: A list of n items described by their weights w_i and values v_i , and a maximal weight limit W of your knapsack. You can take each item any number of times (not necessarily integer).

Task: Select a non-negative quantity of each item, with total weight not exceeding W and maximal total value.

- Items with <u>low</u> weight are better.
- Items high value are better.
- This is the first problem in this section where we can take fractions of items.

Solution

Fill the entire knapsack with the item of highest value per unit weight!

In other words, select the item with largest ratio $\frac{V_i}{W_i}$, and pack $\frac{W}{W_i}$ many copies of that item into your knapsack.

Problem

Instance: A list of n discrete items described by their weights w_i and values v_i , and a maximal weight limit W of your knapsack.

Task: Find a subset S of the items with total weight not exceeding W and maximal total value.

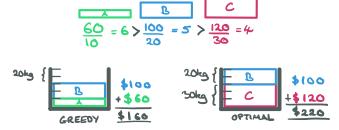
- As before, we want large value and low weight items.
- But this time we can't take fractions of an item, and we can't take multiple copies of an item. How does that change things?

Example

Suppose we have a knapsack of capacity $W = 50 \,\mathrm{kg}$, and three items with weights and values: $A(10 \,\mathrm{kg},\$60)$, $B(20 \,\mathrm{kg},\$100)$, $C(30 \,\mathrm{kg},\$120)$.

Will our previous greedy strategy give an optimal solution?

Greedy method: Pack the item with highest value / weight ratio that will fit.



- So a greedy algorithm based on the value / weight ratio fails for 0-1 knapsack.
- Is there an alternate greedy approach that works instead? No.
- This is a problem that can't be solved by a greedy algorithm.
- So when does the Greedy Strategy work??
- Unfortunately there is no easy rule . . .

For those interested, it relates to the following problem...

Exercise (\$1,000,000 prize if you solve it!)

Is P = NP?

- Fractional knapsack is a problem in P.
- The 0-1 knapsack problem is an example of an NP-hard problem.
- Greedy algorithms work on <u>some</u> problems in P, but they do not work on NP-hard problems.
- Later in the course we'll learn methods that can be applied to NP-hard problems such as dynamic programming.

1. Introduction

2. Assorted problems

- 2.1 Activity selection
- 2.2 Cell tower placement
- 2.3 Minimising job lateness
- 2.4 Tape storage
- 2.5 Interval stabbing
- 2.6 Knapsack problem

2.7 The Huffman Code

- 3. Applications to graphs
- 3.1 Directed graph structure
- 3.2 Single source shortest paths
- 3.3 Minimum spanning trees
- 4. Puzzle

- \blacksquare Assume you are given n sorted arrays of different sizes.
- You are allowed to merge any two arrays into a single new sorted array and proceed in this manner until only one array is left.

Exercise

Design an algorithm which achieves this task and moves array elements as few times as possible.

Give an informal justification why your algorithm is optimal.

This problem is somewhat related to the next problem, which is arguably among the most important applications of the greedy method!

Problem: unambiguous code

You are given a set of symbols, for example the English alphabet plus punctuation marks and a blank space (to be used between words).

abcdefghijklmnopqrstuvwxyz ,.?!

You want to encode these symbols using binary strings, so that sequences of such symbols can be decoded in an unambiguous way.

"Unambiguous" means we there is only one way to decode the binary string.

original message o binary code o decoded message

■ How long do these binary strings need to be? We have 31 symbols and $2^5 = 32$. So 5 bits per symbol will work.

A possible solution

Assign a different 5 bit sequence to each symbol in abcdefghijklmnopgrstuvwxyz ,.?!

- This works! Every symbol can be encoded, and any message we encode can be decoded unambiguously.
- But this code is inefficient. Common symbols and rare symbols are all given binary encodings of the same length. It would be more efficient to encode <u>common</u> symbols with <u>short</u> strings.
- Is there a way to construct an and efficient and unambiguous code?

Example

■ Let's look at a smaller alphabet with just 6 symbols:

acekts

- Since $2^3 = 8 > 6$, we could assign each symbol a 3-bit binary string.
- But the letters *e* and *t* are very common in English, so let's give them shorter strings:

$$e = 00, t = 11$$

And all the other letters get strings of length 3:

$$a = 010, c = 101, k = 111, s = 100$$

■ Is this binary code unambiguous? No. Let's see why

Code: a = 010, c = 101, e = 00, k = 111, s = 100, t = 11

Original message: cake

original message o binary code o decoded message

Encoding step: $cake \rightarrow (101)ake \rightarrow (101)(010)ke \rightarrow (101)(010)(111)e \rightarrow (101)(010)(111)(00)$ Binary code. 101010111100

Decoding step. 10101011100

Two possible decodings:

- 101 010 11 100 \rightarrow cats
- 101 010 111 00 \rightarrow cake

Our code was ambiguous because the encoding of t=11 is a prefix of k=111.

- A code in which no coded symbol is a prefix of another is called a prefix code.
- Prefix codes are unambiguous.

Problem: efficient, unambiguous code

You are given a set of symbols, for example the English alphabet plus punctuation marks and a blank space (to be used between words).

abcdefghijklmnopqrstuvwxyz ,.?!

Each symbol has a frequency (probability of occurring). You want to encode these symbols using binary strings, so that sequences of such symbols can be decoded in an unambiguous way. And so that the expected length of the encoded text is minimised.

Optimal solution

Huffman code!

- Huffman's method uses a greedy algorithm to construct an optimal prefix code.
- This makes sure that the average number of bits in an encoding of an "average" text is as small as possible.
- We'll next go through an example of how to apply Huffman's code. For the details of the algorithm, and the proof of optimality, see the textbook.
- MATH3411 Information, Codes and Ciphers covers this and much more!

Example. Constructing a Huffman code

Symbols	a	С	е	k	s	t
Probabilities	0.20	0.08	0.32	0.05	0.10	0.25

a	0.20	a	0.20
C	0.08	ck	0.13
e	0.32	→ e	0.32
k	0.02	,	
۴	0.22	٤	0.22
S	0.10	S	0.10



Example. Constructing a Huffman code

Symbols	a	С	е	k	s	t
Probabilities	0.20	0.08	0.32	0.05	0.10	0.25

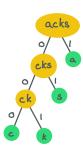
a	0.20	a	0.20
ck	0.13	cks	0.23
e	0.32	و	0.32
٤	0.22	→ ا	0.22
S	0.10		



Example. Constructing a Huffman code

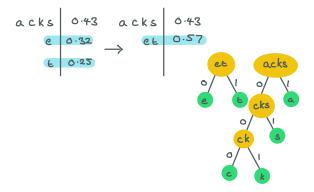
Symbols	a	С	е	k	s	t
Probabilities	0.20	0.08	0.32	0.05	0.10	0.25

a	0.20	ac	ks	0.43
cks	0.23			
e	0.32	\rightarrow	e	0.32
٤	0.25		٤	0.52



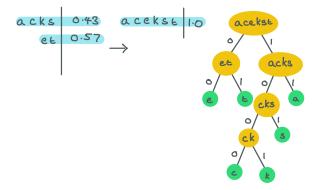
Example. Constructing a Huffman code

Symbols	а	С	e	k	S	t
Probabilities	0.20	0.08	0.32	0.05	0.10	0.25

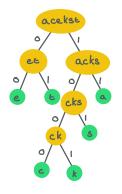


Example. Constructing a Huffman code

Symbols	a	С	e	k	S	t
Probabilities	0.20	0.08	0.32	0.05	0.10	0.25



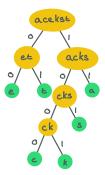
Example. Constructing a Huffman code



Symbols	a	С	e	k	S	t
Binary string	11	1000	00	1001	101	01

Example. Decoding using the Huffman code

Symbols	a	С	e	k	S	t
Binary string	11	1000	00	1001	101	01



Code received: $100011100100 \rightarrow cake$

- Is our Huffman code efficient?
- We saw at the beginning that giving each symbol a 3 bit code results in a prefix code.
- So what's the average number of bits in our Huffman code? Is it better than 3?

Symbols	a	С	e	k	s	t
Probabilities	0.20	0.08	0.32	0.05	0.10	0.25
Binary string	11	1000	00	1001	101	01
Number of bits	2	4	2	4	3	2

avg. bits per symbol =
$$\sum$$
 (number of bits)(probability)
= $2(0.2) + 4(0.08) + 2(0.32) + 4(0.05) + 3(0.1) + 2(0.25)$
= 2.36

- 1. Introduction
- 2. Assorted problems
- 2.1 Activity selection
- 2.2 Cell tower placement
- 2.3 Minimising job lateness
- 2.4 Tape storage
- 2.5 Interval stabbing
- 2.6 Knapsack problem
- 2.7 The Huffman Code
- 3. Applications to graphs
- 3.1 Directed graph structure
- 3.2 Single source shortest paths
- 3.3 Minimum spanning trees
- 4. Puzzle

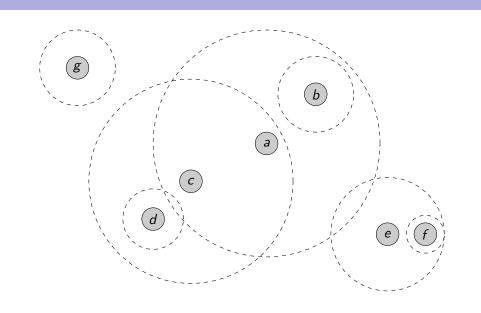
- 1. Introduction
- 2. Assorted problems
- 2.1 Activity selection
- 2.2 Cell tower placement
- 2.3 Minimising job lateness
- 2.4 Tape storage
- 2.5 Interval stabbing
- 2.6 Knapsack problem
- 2.7 The Huffman Code
- 3. Applications to graphs
- 3.1 Directed graph structure
- 3.2 Single source shortest paths
- 3.3 Minimum spanning trees
- 4. Puzzle

Problem

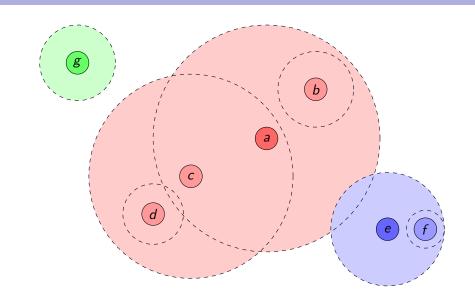
Instance: There are n radio towers for broadcasting tsunami warnings. You are given the (x_i, y_i) coordinates of each tower T_i and its radius r_i of range. When a tower is activated, all towers within the radius of range of the tower will also activate, and those can cause other towers to activate and so on.

You need to equip <u>some</u> of these towers with seismic sensors. When these sensors are activated, they activate the tsunami warning at their tower. These activated towers then activate all towers in their range, and so on.

Task: Design an algorithm which finds the fewest number of towers you must equip with seismic sensors so that all towers will become activated.



- Activating a causes the activation of b and c, and therefore d is activated also.
- Activating e causes the activation of f.
- g must be activated separately.
- Therefore a minimum of three sensors are required.
- Note that we could have placed the first sensor at c instead of a.



How to construct a greedy algorithm for placing sensors?

What inputs were we given?

- Locations (x_i, y_i)
- radii *r_i*.

What did we learn from the example?

- Towers with very small radius could not activate others. So big radius is helpful.
- Towers can only <u>directly</u> activate towers whose locations are inside their radius. So maybe location is also important?
- A tower T_i can indirectly activate another tower T_j if there is a sequence of activations from T_i to T_j . Is there a way to identify when one tower can indirectly activate another?

Attempt 1 (only consider radius)

Find the unactivated tower with the largest radius (breaking ties arbitrarily), and place a sensor at this tower. Find and remove all towers activated as a result. Repeat.

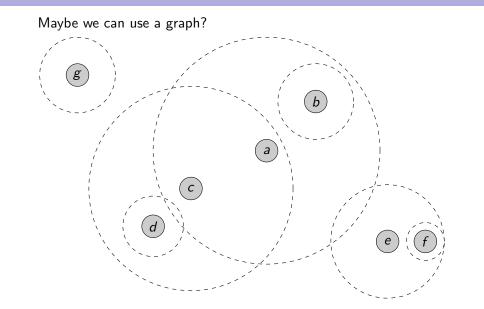
Attempt 2 (only consider direct activations)

Find the unactivated tower with the largest number of towers within its range (breaking ties arbitrarily), and place a sensor at this tower. Find and remove all towers activated as a result. Repeat.

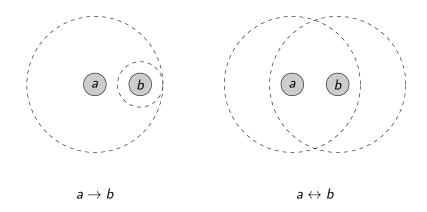
Exercise

Give examples which show that neither of these algorithms solve the problem correctly.

Our greedy algorithm needs to consider indirect activations. How?



It is useful to consider the towers as vertices of a directed graph, where an edge from tower a to tower b indicates that the activation of a directly causes the activation of b, that is, b is within the radius of a.



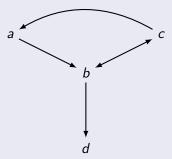
Observation

Suppose that activating tower a causes tower b to also be activated, and vice versa. Then we never want to place sensors at both towers; indeed, placing a sensor at a is equivalent to placing a sensor at b.

How can we can extend this notion to a larger number of towers?

- Cycles also have this property.
- Can we do better?

Example



All four towers can be activated by placing just one sensor at a, b or c.

Observation

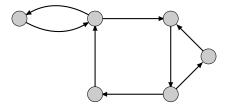
Let S be a subset of the towers such that that activating \underline{any} tower in S causes the activation of all towers in S.

We never want to place more than one sensor in S, and if we place one, then it doesn't matter where we put it.

In this way, we can treat all of S as a unit; a super-tower.

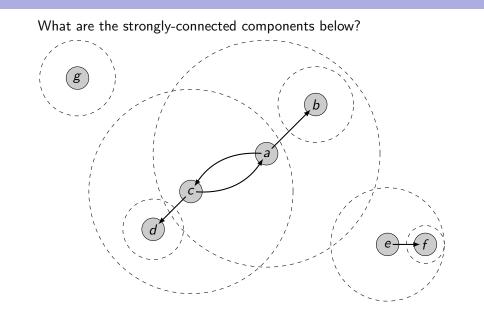
Definition

Given a directed graph G = (V, E) and a vertex v, the strongly connected component of G containing v consists of all vertices $u \in V$ such that there is a path in G from v to u and a path from u to v. We will denote it by C_v .



In the terms of our problem, strongly connected components are maximal super-towers.

Tsunami warning



How do we find the strongly connected component $C_v \subseteq V$ containing v?

Claim

u is in C_v if and only if u is reachable from v and v is reachable from u.

We can split this into two subproblems:

- Find all the vertices that can be reached from v.
- Find all the vertices which can reach v.

Then a vertex u is in C_v if and only if it appears in both of these sets.

To find all the vertices that can be reached from v:

■ Do a BFS or DFS of G to find the set R_v of vertices of the directed tree rooted at v.

Finding the vertices that can reach v is harder:

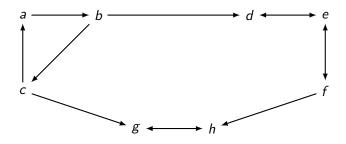
- Construct another graph $G_{rev} = (V, E_{rev})$ consisting of the same set of vertices V but with the set of edges E_{rev} obtained by reversing the direction of all edges E of G.
- Find the set R'_{v} of vertices of the directed tree of G_{rev} rooted at v.

Claim

u is in C_v if and only if u is reachable from v and v is reachable from u.

In other words, u is reachable from v in both G and G_{rev} . Equivalently, u is in both R_v and R_v' .

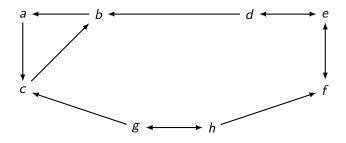
Example. Suppose the original graph G = (V, E) is



Then the set of vertices reachable from e in G is

$$R_e = \{d, e, f, g, h\}.$$

The reverse graph $G_{rev} = (V, E_{rev})$ is



Then the set of vertices reachable from e in G_{rev} is

$$R'_e = \{a, b, c, d, e, f\}.$$

Combining

$$R_e = \{d, e, f, g, h\}$$

with

$$R'_e = \{a, b, c, d, e, f\},\$$

we have

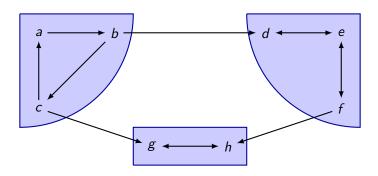
$$C_e = R_e \cap R'_e = \{d, e, f\}.$$

- Note that C_d and C_f are also the same set, namely $\{d, e, f\}$.
- Similarly, we find that $C_a = \{a, b, c\}$ and $C_g = \{g, h\}$.

To recap:

- Use BFS to find the set $R_v \subseteq V$ of all vertices in V which are reachable in G from v.
- Similarly find the set $R'_v \subseteq V$ of all vertices which are reachable in G_{rev} from v.
- The strongly connected component of G containing v is given by $C_v = R_v \cap R'_v$.

Therefore the decomposition of G into strongly connected components is

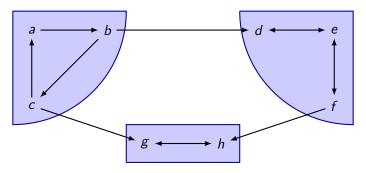


Time complexity.

- Finding all strongly connected components in this way could require O(V) traversals of the graph.
- Each of these traversals is a BFS, requiring O(V + E) time.
- Therefore the total time complexity is O(V(V + E)).

- Faster algorithms exist! Famously, Kosaraju's algorithm and Tarjan's algorithm find all strongly connected components of a directed graph in linear time, i.e. O(V + E).
- We won't cover them in this course, but we will in COMP4128. You can however quote them in assessments (e.g. Kosaraju's algorithm is presented in CLRS §22.5).
- A linear time algorithm is asymptotically "no slower than" reading the graph, so we can run these algorithms "for free", i.e. without worsening the time complexity of our solution to a problem.

But what should we do next?



- We've found a decomposition into strongly connected components.
- How do we use this to choose where to place the sensors?

- It should be clear that distinct strongly connected components are disjoint sets, so the strongly connected components form a partition of V.
- Let C_G be the set of all strongly connected components of a graph G.

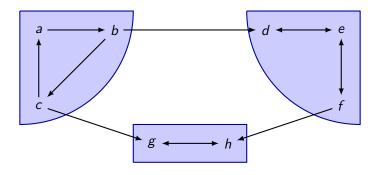
Definition

Define the condensation graph $\Sigma_G = (C_G, E^*)$, where

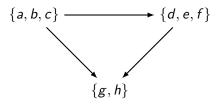
$$E^* = \{(C_{u_1}, C_{u_2}) \mid (u_1, u_2) \in E, C_{u_1} \neq C_{u_2}\}.$$

The vertices of Σ_G are the strongly connected components of G, and the edges of Σ_G correspond to those edges of G that are not within a strongly connected component, with duplicates ignored.

Recall our earlier example



The condensation graph is simply



If this were our tsunami problem,

- Where would we put sensors on the condensation graph?
- Why?

Let's recap

- We begin our solution to the tsunami warning problem by finding the condensation graph.
- Now we have the set of super-towers, and we know for each super-tower which others it can activate.
- Our task is to decide which super-towers need a sensor installed in order to activate all the super-towers.
- We need to know one more property about the condensation graph.

Claim

The condensation graph Σ_G is a directed acyclic graph.

Proof Outline

Suppose there is a cycle in Σ_G . Then the vertices on this cycle are not $\underline{\text{maximal}}$ strongly connected sets, as they can be merged into an even larger strongly connected set.

Solution

The correct greedy strategy is to only place a sensor in each super-tower without incoming edges in the condensation graph.

Proof

These super-towers cannot be activated by another super-tower, so they each require a sensor. This shows that there is no solution using fewer sensors.

Proof (continued)

We still have to prove that this solution activates all super-towers.

Consider a super-tower with one or more incoming edges. Follow any of these edges backwards, and continue backtracking in this way.

Since the condensation graph is acyclic, this path must end at some super-tower without incoming edges. The sensor placed here will then activate all super-towers along our path.

Therefore, all super-towers are activated as required.

So the steps we used to solve the tsunami problem were:

Algorithm

- **Input:** towers with locations (x_i, y_i) and radii r_i .
- Construct a directed graph G, where (u, v) is an edge if tower u can directly activate tower v.
- Construct the reversed graph G_{rev} of G.
- Use DFS and BFS on G and G_{rev} to find the strongly connected components of G.
- Form the condensation graph Σ_G , whose vertices are the strongly connected components of G.
- **Solution:** Place a sensor on every vertex of Σ_G that has no incoming edges.

Why is this a greedy algorithm? Here's one answer...

- We choose to place a sensor on a vertex v whose strongly-connected component C_v has the smallest number of incoming edges in Σ_v .
- We then delete v and all vertices that are reachable from v.
- And repeat.
- It just so happens that this smallest number of incoming edges is always 0.

There is another answer, but it requires a complicated definition...

In the special case where G has no cycles, then we can order the vertices of G by the following rule.

Definition

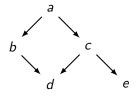
Let G = (V, E) be a directed graph, and let n = |V|. A topological sort of G is a linear ordering (enumeration) of its vertices $\sigma: V \to \{1, \ldots, n\}$ such that if there exists an edge $(v, w) \in E$ then v precedes w in the ordering, i.e., $\sigma(v) < \sigma(w)$.

Property

A directed acyclic graph permits a topological sort of its vertices.

Note that the topological sort is not necessarily unique, i.e., there may be more than one valid topological ordering of the vertices.

Example. Find a topological sort of the vertices in this graph:



There are many solutions, such as:

- abcde
- acebd

Why is abdce not a solution? This is because \overrightarrow{cd} is an edge, but d appears before c in the list.

How do we topologically sort the vertices?

Algorithm

Maintain:

- a list L of vertices, initially empty,
- an array *D* consisting of the in-degrees of the vertices, and
- lacksquare a set S of vertices with no incoming edges.

Algorithm (continued)

While set S is non-empty, select a vertex u in the set.

- \blacksquare Remove it from S and append it to L.
- Then, for every outgoing edge e = (u, v) from this vertex, remove the edge from the graph, and decrement D[v] accordingly.
 - If D[v] is now zero, insert v into S.

If there are no edges remaining, then L is a topological ordering. Otherwise, the graph has a cycle.

- This algorithm runs in O(V + E), that is, linear time.
- Once again, we can run this algorithm "for free" as it is asymptotically no slower than reading the graph.
- In problems involving directed acyclic graphs, it is often useful to start with a topological sort and then think about the actual problem!
- A topological ordering is often a natural way to process the vertices. We'll see more of this in Dynamic Programming.

- 1. Introduction
- 2. Assorted problems
- 2.1 Activity selection
- 2.2 Cell tower placement
- 2.3 Minimising job lateness
- 2.4 Tape storage
- 2.5 Interval stabbing
- 2.6 Knapsack problem
- 2.7 The Huffman Code
- 3. Applications to graphs
- 3.1 Directed graph structure
- 3.2 Single source shortest paths
- 3.3 Minimum spanning trees
- 4. Puzzle

Problem

Instance: a directed graph G = (V, E) with <u>non-negative</u> weight w(e), and a designated source vertex $s \in V$.

We will assume that for every $v \in V$ there is a path from s to v.

Task: find the weight of the shortest path from s to v for every $v \in V$.

Example. Suppose the graph G = (V, E) models Sydney's road network with roads E and intersections V. The weight of an edge \overrightarrow{uv} records how much time it takes to drive down the road from u to v. Your home is located at intersection s, and you want to identify the quickest journey from your home to any other location in Sydney.

Problem

Instance: a directed graph G = (V, E) with <u>non-negative</u> weight w(e), and a designated source vertex $s \in V$.

We will assume that for every $v \in V$ there is a path from s to v.

Task: find the weight of the shortest path from s to v for every $v \in V$.

Observe that this problem makes the following assumptions:

- There is exactly one source vertex s.
- The graph is directed.
- The edge weights are non-negative.

There are other versions of this problem that make different assumptions, we'll see some of them later in the course.

Note

To find shortest paths from *s* in an <u>undirected</u> graph, simply replace each undirected edge with two directed edges in opposite directions.

Note

There isn't necessarily a unique shortest path from *s* to each vertex.

This task is accomplished by a very elegant greedy algorithm developed by Edsger Dijkstra in 1959.

Algorithm Outline

Maintain a set S of vertices for which the shortest path weight has been found, initially empty. S is represented by a boolean array.

For every vertex v, maintain a value d_v which is the weight of the shortest 'known' path from s to v, i.e. the shortest path using only intermediate vertices in S. Initially $d_s=0$ and $d_v=\infty$ for all other vertices.

At each stage, we greedily add to S the vertex $v \in V \setminus S$ which has the smallest d_v value. Record this value as the length of the shortest path from s to v, and update other d_z values as necessary.

This outline still leaves much work to do. So we'll examine the following four features of Dijkstra's algorithm.

- **Correctness.** Why is it correct to always add the vertex outside S with the smallest d_V value?
- **Updates.** When v is added to S, for which vertices z must we update d_z , and how do we do these updates?
- **Data structure.** What data structure should we use to represent the d_v values?
- **Time complexity.** What is the time complexity of this algorithm, and how is it impacted by our choice of data structure?

First, we will prove the correctness of Dijkstra's algorithm.

Claim

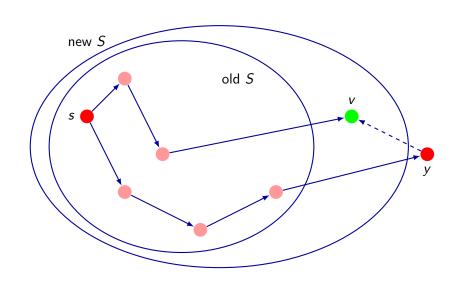
Suppose v is the next vertex to be added to S. Then d_v is the length of the shortest path from s to v.

Proof

- d_v is the length of the shortest path from s to v using only intermediate vertices in S. Let's call this path p.
- If this were <u>not</u> to be the shortest path from s to v, there must be some shorter path p' which first leaves S at some vertex y before later reaching v.

- Now, the portion of p' up to y is a path from s to y using only intermediate vertices in S.
- Therefore, this portion of p' has weight at least d_y .
- Since all edge weights are non-negative, p' itself has weight at least d_y .

- But *v* was chosen to have smallest *d*-value among all vertices outside *S*!
- So we know that $d_v \le d_y$, and hence the weight of path p is at most that of p'.
- Therefore, d_v is indeed the weight of the shortest path from s to v.



Question

Earlier, we said that when we add a vertex v to S, we may have to update some d_z values. What updates could be required?

Answer

If there is an edge from v to z with weight w(v,z), the shortest known path to z may be improved by taking the shortest path to v followed by this edge. Therefore we check whether

$$d_z > d_v + w(v, z),$$

and if so we update d_z to the value $d_v + w(v, z)$.

As it turns out, these are the only updates we should consider!

Claim

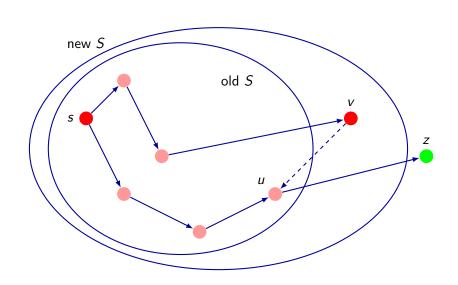
If d_z changes as a result of adding v to S, the new shortest known path to z must have penultimate vertex v, i.e. the last edge must go from v to z.

Proof

- Suppose that adding v to S allows for a new shortest path through S from s to z with penultimate vertex $u \neq v$.
- Such a path must include v, or else it would not be new. Thus the path is of the form

$$P = s \rightarrow \cdots \rightarrow v \rightarrow \cdots \rightarrow u \rightarrow z$$
.

- Since u was added to S before v was, we know that there is a shortest path P' from s to u which does not pass through v.
- Appending the edge from u to z to P' produces a path through S from s to z which is no longer than P.
- This path was already a candidate for d_z , so the weight of P is greater than or equal to the existing d_z value.
- This is a contradiction, so the proof is complete.



- Now, we are ready to consider data structures to maintain the d_v values.
- We need to support two operations:
 - find the vertex $v \in V \setminus S$ with smallest d_v value, and
 - for each of its outgoing edges (v, z), update d_z if necessary.
- We'll start with the simplest data structure: the array.

Let n = |V| and m = |E|, and suppose the vertices of the graph are labelled $1, \ldots, n$, where vertex s is the source.

Attempt 1

Store the d_i values in an array d[1..n].

At each stage:

- Perform a linear search of array d, ignoring those vertices already in S, and select the vertex v with smallest d[v] to be added to S.
- For each outgoing edge from vertex v to some $z \in V \setminus S$, update d[z] if necessary. (As before, we check whether d[v] + w(v, z) is smaller than d[z]).

Question

What is the time complexity of this algorithm?

Answer

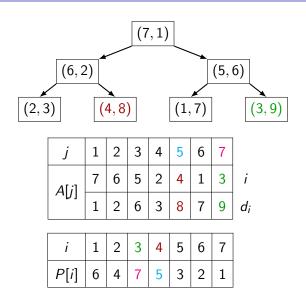
- At each of *n* steps, we perform a linear scan on an array of length *n*.
- We also run the update procedure (in constant time) at most once for each edge.
- The algorithm therefore runs in $O(n^2 + m)$.

- In a simple graph (no self loops or parallel edges) we have $m \le n(n-1)$, so we can simplify the time complexity expression to just $O(n^2)$.
- If the graph is dense, in other words when the number of edges $m = O(n^2)$, then this running time is actually pretty good. So using an array fine for dense graphs. But not all graphs are dense, for example road networks are not dense as they have m = O(n).
- Can we do better when $m \ll n^2$?

- Recall the two operations we need to support:
 - find the vertex $v \in V \setminus S$ with smallest d_v value, and
 - for each of its outgoing edges (v, z), update d_z if necessary.
- So far, we have done the first operation in O(n) using a linear search.
- How can we improve on this?

- The first operation isn't a pure 'find minimum' because we have to skip over vertices already in *S*.
- Instead, when we add a vertex v to S, we could try deleting d_v from the data structure altogether.
- We now have three operations to support: find minimum, delete minimum, and update any.
- The first two of these suggest the use of a min-heap, but the standard heap doesn't allow us to update arbitrary elements.

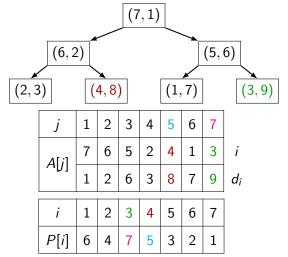
- We will use a heap represented by an array A[1..n]; the left child of A[j] is stored in A[2j] and the right child in A[2j+1].
- Every element of A is of the form $A[j] = (i, d_i)$ for some vertex i. The min-heap property is maintained with respect to the d-values only.
- We will also maintain another array P[1..n] which stores the <u>position</u> of elements in the heap.
- Whenever A[j] refers to vertex i, we record P[i] = j, so that we can look up vertex i using the property $A[P[i]] = (i, d_i)$.



With an augmented heap we can do all the usual heap operations (insertion, deletion and access top) with the usual time complexity. But now we can also change d-values.

- Changing the *d*-value of vertex i is now an $O(\log n)$ operation.
 - First, look up vertex i in the position array P. This gives us P[i], the index in A where the pair (i, d_i) is stored.
 - Next, we update d_i by changing the second entry of A[P[i]].
 - Finally, it may be necessary to bubble up or down to restore the min-heap property. In this algorithm, *d*-values are only ever reduced, so only bubbling up is applicable.
- Accessing the top of the heap still takes O(1), and popping the heap still takes $O(\log n)$.

Example. Suppose d[3] is updated from 9 to 5. How do we update our data structures?



Algorithm

Store the d_i values in an augmented heap of size n.

At each stage:

- Access the top of the heap to obtain the vertex v with smallest key and add it to set S.
- Pop the corresponding element d_v from the heap.
- For each outgoing edge from v to some $z \in V \setminus S$, update d_z if necessary. (This is the step where we need to use the position array).

Question

What is the time complexity of our algorithm?

Answer

- Initially building the heap can be done in linear time O(n). Or less clever algorithms can build the heap in $O(n \log n)$.
- Each of n stages requires a deletion from the heap (when a vertex is added to S), which takes $O(\log n)$ many steps.
- Each of the m edge causes at most one update of a key in the heap, also taking $O(\log n)$ many steps.
- Thus, in total, the algorithm runs in time $O((n+m)\log n)$. But since there is a path from v to every other vertex, we know $m \ge n-1$, so we can simplify to $O(m\log n)$.

Note

In COMP2521/9024, you may have seen that the time complexity of Dijkstra's algorithm can be improved to $O(m + n \log n)$. This is true, but it relies on an advanced data structure called the <u>Fibonacci heap</u>, which has not been taught in this course or any prior course.

Therefore, this improvement will be considered <u>external</u> to our course; you cannot use it in assessments unless you also detail the Fibonacci heap construction and operations, and prove the improved time complexity. This will not be the intended solution for any assessable problem in our course.

- 1. Introduction
- 2. Assorted problems
- 2.1 Activity selection
- 2.2 Cell tower placement
- 2.3 Minimising job lateness
- 2.4 Tape storage
- 2.5 Interval stabbing
- 2.6 Knapsack problem
- 2.7 The Huffman Code
- 3. Applications to graphs
- 3.1 Directed graph structure
- 3.2 Single source shortest paths
- 3.3 Minimum spanning trees
- 4. Puzzle

Definition

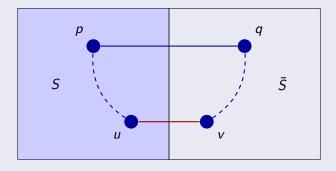
- G = (V, E) is a connected graph and each edge in E has a length.
 - A <u>spanning tree</u> T of G is any tree which is a subgraph of G on the same vertex set V. The <u>weight</u> of T is the sum of all edge lengths in T.
 - A minimum spanning tree *T* of a connected graph *G* is a tree which, out of all choices of spanning tree of *G*, has minimal weight (i.e. the total length of all edges is minimised).

Lemma

Let G = (V, E) be a connected graph with all lengths of edges in E distinct, and let S be a non-empty proper subset of the vertices V. Assume e = (u, v) is an edge with $u \in S$ and $v \notin S$ and that e has minimal length among all the edges having this property. Then e must belong to every minimum spanning tree T of G.

Proof

Assume that there exists a minimum spanning tree T which does not contain such an edge e = (u, v).



- Since T is a spanning tree, there exists a path from u to v within T, and this path must leave S by some edge, say (p,q) where $p \in S$ and $q \notin S$.
- However, (u, v) is shorter than any other edge with one end in S and one end outside S, including (p, q).
- Replacing the edge (p, q) with the edge (u, v) produces a new tree T' with smaller total edge weight.
- This contradicts our assumption that *T* is a minimum spanning tree, completing the proof.

- There are two famous greedy algorithms for the minimum spanning tree problem.
- Both algorithms build up a forest, beginning with all n isolated vertices and adding edges one by one.
- Prim's algorithm uses one large component, adding one of the isolated vertices to it at each stage. This algorithm is very similar to Dijkstra's algorithm, but adds the vertex closest to S rather than the one closest to the starting vertex.
- We will instead focus on Kruskal's algorithm.

- We sort the edges *E* in increasing order by weight.
- An edge *e* is added if its inclusion does not introduce a cycle in the graph constructed thus far, or discarded otherwise.
- The process terminates when the forest is connected, i.e. when n-1 edges have been added.

Claim

Kruskal's algorithm produces a minimal spanning tree, and if all weights are distinct, then such a Minimum Spanning Tree is unique.

Proof

We consider the case when all weights are distinct.

• Consider an edge e = (u, v) added in the course of Kruskal's algorithm, and let F be the forest in its state <u>before</u> adding e.

- Let S be the set of vertices reachable from u in F (i.e. in the same connected component). Then clearly $u \in S$ but $v \notin S$.
- The original graph does not contain any edges shorter than *e* with one end in *S* and the other outside *S*. If such an edge existed, it would have been considered before *e* and included in *F*, but then both its endpoints would be in *S*, contradicting the definition.
- Consequently, edge e is the shortest edge between a vertex of S and a vertex of \bar{S} and by the previous lemma it must belong to every minimum spanning tree.

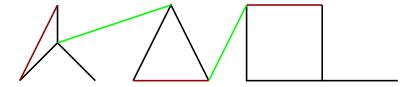
- Thus, the set of edges produced by Kruskal's algorithm is a subset of the set of edges of every minimum spanning tree.
- But the graph produced by Kruskal's algorithm by definition has no cycles and is connected, so it is a tree.
- Therefore in the case where all edge weights are distinct, Kruskal's algorithm produces the unique minimum spanning tree.

So Kruskal's algorithm is correct. But how can we implement it efficiently?

Kruskal's algorithm

- We sort the edges *E* in increasing order by weight.
- An edge *e* is added if its inclusion does not introduce a cycle in the graph constructed thus far, or discarded otherwise.
- The process terminates when the forest is connected, i.e. when n-1 edges have been added.

- To efficiently implement Kruskal's algorithm, we need to quickly determine whether a certain new edge will introduce a cycle.
- An edge e = (u, v) will introduce a cycle in the forest F if and only if there is already a path between u and v, i.e., u and v are in the same connected component.



In our implementation of Kruskal's algorithm, we store the vertices using the <u>Union-Find</u> data structure. It handles disjoint sets, supporting three operations:

- **1** MakeUnionFind(S), which returns a structure in which all elements (vertices) are placed into distinct singleton sets. This operation runs in time O(n) where n = |S|.
- 2 FIND(v), which returns the (label of the) set to which v belongs. This operation runs in time O(1).
- 3 UNION(u, v), which changes the data structure by merging the sets A and B (whose labels are u and v respectively) into a single set $A \cup B$. The first k UNION operations run in total time $O(k \log k)$.

- Note that we do not give the run time of a single UNION operation but of a sequence of *k* UNION operations.
- This is called <u>amortized analysis</u>; it effectively estimates the average cost of each operation in a sequence.
- Any one UNION operation might be $\Theta(n)$, but the total time taken by the first k is $O(k \log k)$, i.e. each takes 'on average' $O(\log k)$.
- This is different to average case analysis, because it's a statement about an aggregate, rather than a probability.

Union-Find 14

- We will label each set by one of its elements, called the representative of the set.
- The simplest implementation of the Union-Find data structure consists of:
 - an array A, where A[i] = j means that i belongs to the set with representative j;
 - **a** an array B, where B[i] contains the number of elements in the set with representative i;
 - an array L, where L[i] contains pointers to the head and tail of a linked list containing the elements of the set with representative i.

Union-Find 14

Note

If i is not the representative of any set, then B[i] is zero and the list L[i] is empty.

Note

The list array L allows us to iterate through the members of one of the disjoint sets, which is used in the UNION operation.

Given two sets I and J with representatives i and j, UNION(i,j) is defined as follows:

- assume $B[i] \ge B[j]$ (i.e. $|I| \ge |J|$); otherwise perform UNION(j,i) instead;
- for each $m \in J$, update A[m] from j to i;
- update B[i] to B[i] + B[j] and B[j] to zero;
- **a** append the list L[j] to the list L[i] and replace L[j] with an empty list.

Union-Find 151

Observation

The new value of B[i] is at least twice the old value of B[j].

Observation

Suppose m is an element of the smaller set J, so its label A[m] changed from j to i.

Then the observation above tells us that B[A[m]] (formerly the old B[j], now the new B[i]) at least doubled.

What's the significance of B[A[m]]? It's the number of elements in the set containing m.

- The first *k* UNION operations can touch at most 2*k* elements of *S* (with equality if they each merge a different pair of singleton sets).
- Thus, the set containing an element m after the first k UNION operations must have at most 2k elements.
- Since every UNION operation which changes A[m] at least doubles B[A[m]], we deduce that A[m] has changed at most log 2k times.
- Thus, since at most 2k elements have their label changed at all, we can conclude that the first k UNION operations will cause at most 2k log 2k label changes in A.

- Each Union operation requires only constant time to update the size array *B* and the list array *L*.
- Thus, the first k UNION operations take $O(k \log k)$ time in total.
- This Union-Find data structure is good enough to get the sharpest possible bound on the run time of Kruskal's algorithm.
- See the textbook for a Union-Find data structure based on pointers and path compression, which further reduces the amortised complexity of the UNION operation at the cost of increasing the complexity of the FIND operation from O(1) to $O(\log n)$.

- We now use the previously described Union-Find data structure to efficiently implement Kruskal's algorithm on a graph G = (V, E) with n vertices and m edges.
- We first have to sort m edges of graph G which takes time $O(m \log m)$. Since $m < n^2$, we can rewrite this as $O(m \log n^2) = O(m \log n)$.
- As we progress through the execution of Kruskal's algorithm, we will start with n isolated vertices, which will be merged into connected components until all vertices belong to a single connected component. We use the Union-Find data structure to keep track of the connected components constructed at any stage.

- For each edge e = (u, v) on the sorted list of edges, we use two FIND operations to determine whether vertices u and v belong to the same component.
- If they do not belong to the same component, i.e., if $\operatorname{FIND}(u) = i$ and $\operatorname{FIND}(v) = j$ where $i \neq j$, we add edge e to the spanning tree being constructed and perform $\operatorname{UNION}(i,j)$ to merge the connected components containing u and v.
- If instead FIND(u) = FIND(v), there is already a path between u and v, so adding this edge would create a cycle. Therefore, we simply discard the edge.

- We perform 2m FIND operations, each costing O(1).
- We also perform n-1 UNION operations, which in total cost $O(n \log n)$.
- The overall time complexity is therefore $O(m \log n + m + n \log n)$.
- The first term (from sorting) dominates, so we can simplify the time complexity to $O(m \log n)$.

Let's end by looking at an application of Kruskal's algorithm.

Problem

Instance: A complete graph G with weighted edges representing distances between the two vertices.

Task: Partition the vertices of G into k disjoint subsets so that the minimal distance between two points belonging to different sets of the partition is as large as possible. Thus, we want a partition into k disjoint sets which are as far apart as possible.

Solution

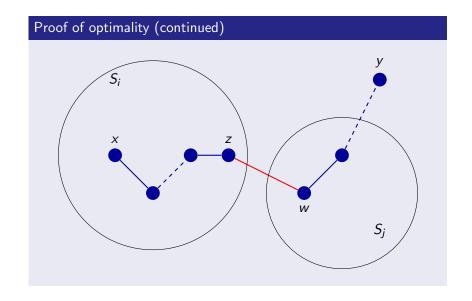
Sort the edges in increasing order and start performing the usual Kruskal's algorithm for building a minimal spanning tree, but stop when you obtain k connected components, rather than a single spanning tree.

Proof of optimality

- Let δ be the distance associated with the first edge of the minimal spanning tree which was not added to our k connected components.
- lacktriangleright It is clear that δ is the minimal distance between two vertices belonging to different connected components.
- All the edges included in the connected components produced by our algorithm are of length at most δ .

Proof of optimality (continued)

- Consider any partition S into k subsets different from the one produced by our algorithm.
- This means that there is a connected component produced by our algorithm which contains vertices x and y such that $x \in S_i$ and $y \notin S_i$ for some $S_i \in S$.



Proof of optimality (continued)

- Since x and y belong to the same connected component in the Kruskal solution, there is a path P in the Kruskal solution that connects x and y.
- Let z and w be two consecutive vertices on that path such that z belongs to S_i and $w \notin S_i$.
- Thus, $w \in S_j$ for some $j \neq i$.

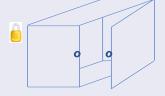
Proof of optimality (continued)

- Since (z, w) was an edge chosen by Kruskal's (partial) algorithm, we know that $d(z, w) \le \delta$.
- It follows that the distance between these two clusters $S_i, S_i \in \mathcal{S}$ is at most δ .
- Thus, such a partition cannot be a better clustering than the one produced by our algorithm.

- What is the time complexity of this algorithm?
- We have $\Theta(n^2)$ edges; thus sorting them by weight will take $O(n^2 \log n^2)$, which we can simplify to $O(n^2 \log n)$.
- Running the (partial) Kruskal algorithm requires $O(m \log n)$ steps, making use of the Union-Find data structure. Since $m = \Theta(n^2)$, this step also takes $O(n^2 \log n)$.
- So the algorithm has time complexity $O(n^2 \log n)$ in total.

- 1. Introduction
- 2. Assorted problems
- 2.1 Activity selection
- 2.2 Cell tower placement
- 2.3 Minimising job lateness
- 2.4 Tape storage
- 2.5 Interval stabbing
- 2.6 Knapsack problem
- 2.7 The Huffman Code
- 3. Applications to graphs
- 3.1 Directed graph structure
- 3.2 Single source shortest paths
- 3.3 Minimum spanning trees
- 4. Puzzle

Problem



Bob is visiting Elbonia and wishes to send his teddy bear to Alice, who is staying at a different hotel. Both Bob and Alice have boxes like the one illustrated above, as well as padlocks which can be used to lock the boxes.

Puzzle 167

Problem (continued)

However, there is a problem. The Elbonian postal service mandates that when a nonempty box is sent, it must be locked. Also, they do not allow keys to be sent, so the key must remain with the sender. Finally, you can send padlocks only if they are locked. How can Bob safely send his teddy bear to Alice?

Puzzle 168

Hint

The way in which the boxes are locked (via a padlock) is important. It is also crucial that <u>both</u> Bob and Alice have padlocks and boxes. They can also communicate over the phone to agree on the strategy.

There are two possible solutions; one can be called the "AND" solution, the other can be called the "OR" solution. The "AND" solution requires 4 mail one way services while the "OR" solution requires only 2.