November 2, 2023

1 Graphs

Formal definition of undirected graphs:

- Called G(V, E).
- has a set *V* of vertices, and *E* of edges.
- Edges are denoted as a pair of vertices $\{v_i, v_i\}$, and undirected mean that each pair of $\{v_i, v_i\}$ is unique.
 - e.g. Facebook friendship graphs, where the friendship must be mutual. (Has nearly 3 billion nodes, and each node has 330 vertices)
- Directed graph is denoted in the same way, except that $\{v_i, v_j\}$ means specifically that $v_i \to v_j$.

1.1 Size of Graphs

- Generally denoted as the number of vertices (denoted by n) and the number of edges (denoted by m).
- m can be at most n^2 , in the case that every vertex is connected to every other vertex (in which case we have n(n-1) edges)
- Degree: the number of edges that a vertex has. With directed graphs, we can specify in-degrees and out-degrees.

1.2 Representing Graphs

- Represented either as an adjacency matrix or an adjacency list.
 - Adjacency matrix: have the vertices listed out on both row and column, put a 1 if $\{v_i, v_j\}$ are connected on our graph.
 - Adjacency list: List of length n, and each cell is a linked list that points to all the neighbours of v_i .
- There are tradeoffs for both ways:

	Adjacency Matrix	Adjacency List
Storage Size	$O(n^2)$	O(n+m)
Checking whether $(u, v) \in E$	O(1)	$O(\deg(u))$
Enumerate all of <i>u</i> 's neighbours	O(n)	$O(\deg(u))$

· We will work with adjacency lists, since the storage size for adjacency matrices get too large too quickly.

1.3 Graph Connectedness

- We first have to solve the problem of graph traversal. Our approach will use "string and chalk," so we mark when we've reached a dead end and the "string" will allow us to backtrack.
- Algorithm description:

```
def explore():
    visited[u] = true

For all edges of $u$: if visited[v] = false then run explore(v)
```

- Explore guarantees that all the vertices that are visited by explore have a path from *u* to that vertex, and vice versa.
 - We prove the other direction: if there's a path, then that node is visited.

Assume that this is false: assume $\exists v$ that hasn't been explored but there is a path from u to v. Instead of looking at u to v, we look at the path from u to v_k , the first node along the path from u to v that is unexplored. This means that the algorithm reached v_{k-1} , then failed to explore v.

This is a contradiction, since explore must have been called on v_{k-1} , but failed to recurse down v_k .

1.4 Depth First Search (DFS)

Essentially calling explore, except it does it recursively on all the remaining nodes that haven't been visited yet.

- We can also use DFS to find the connected components of a graph! When we explore with DFS, we are implicitly exploring connected components, this uses the transitive fact of the explore() function.
- The edges that are visited by DFS are categorized into tree edges and back edges. Tree edges are edges that are used when the algorithm runs, and back edges are ones that exist within the original graph but aren't used.
- There's a third class called a *cross edges*, but they cannot exist for an undirected graph.
 - The proof is by contradiction: imagine that it did exist, then DFS would have called explore on that ancestor; but it can't possibly have since *u* and *v* do not exist in the same branch.
 - They *can* exist in a directed graph, since we only traverse through out edges (so there can be an in-edge that never gets traversed).

1.4.1 DFS Runtime

Explore() is called only once per node, and the runtime for each node for explore is proportional to deg(u), so the total is:

$$\sum_{u \in V} O(1 + \deg(u)) = O(n + m)$$

1.5 DFS for Directed Graphs

It's the same principle, our explore still only runs recursively on unvisited neighbours, but we need to also keep track of the amount of time at which we start and finish processing that node.

- Every time we enter a node, we stamp it with the time of the start clock. When we come out, we stamp again with the end clock. Every time we progress the clock, we increment it by 1.
- The point of the clock will be revealed later on in future lectures.
- The edges we keep track of here are forward, back and cross edges. Forward means we go down the tree from
 ancestor to descendant (not its immediate child), back means we go backwards (can be immediate) and cross
 edges are defined exactly as before.

The edges (pre, post, cross) are useful in determining properties of the graph G.

- Cross edges can only go from a later point in one branch to earlier than the other branch and not the other way around, due to the way that DFS executes depth-first.
- Suppose $(u, v) \in E$ is a tree edge. Then, we know that pre(u) < pre(v) (since u is hit first), and post(u) < post(v).
- Suppose that $(u, v) \in E$ is a back edge. Then, we have pre(v) < pre(u) < post(u) < post(v).

2 Strongly Connected Graphs

Recall that we saw earlier that DFS basically just calls explore repeatedly, to find all the vertices reachable from a vertex u. We also introduced two clocks, where when we first visit a node we stamp it with the start clock, and that vertex will have a pre(u) quantity that stores its value, then increments clock. When we leave the vertex, we stamp again with a post(u).

- For cross edges (the only thing we didn't finish last time), we have pre(v) < post(v) < pre(u) < post(u) (we get to v and finish exploring v before we ever touch v).
- As a recap:

Edge type	Relation
Tree edge	pre(u) < pre(v) < post(v) < post(u)
Back edge	pre(v) < pre(u) < post(u) < post(v)
Cross edge	pre(v) < post(v) < pre(u) < post(u)

2.1 Topological Sort

- The process of finding an ordering of vertices so that no edges go backwards. This is used in software package loading, to make sure that things aren't being and that are being downloaded in the chronological order.
- Mathematically, if u comes before v in the ordering, then there is no edge (v, u).
- Two types of special nodes: sources and sinks
 - **Source:** A node that has no incoming edges and has only outgoing edges.
 - Sink: A node that has no outgoing edges and only has incoming edges
- Note that a node with no edge at all is considered both a source and a sink.

2.2 DAGs

- Called a directed acyclic graph, or basically just a graph without any directed cycles. If we have a cycle, then we can't toplogically sort.
- We will find that if we run DFS on a graph, it is a DAG iff it has no back edges. We prove that if we have a back edge, then it cannot be a DAG: since they go from something that's already been visited to something that's been visited earlier, then we have already found a cycle.
 - Now we prove that if we don't have a DAG (i,e, has a cycle), then it has a back edge. Since DFS visits every vertex in a graph, it will eventually enter our cycle at some v_i . Then, it will traverse through the cycle until it visits all the nodes and eventually gets to v_k , the node right before it comes back to v_i . Then, $v_k \to v_i$ will become a back edge, since it was visited earlier in the graph.
- Back edges are really special! Recall that we only have a back edge when post(u) < post(v), since the other edges have post(v) < post(u). We then conclude that if we have a DAG, then it has the property that post(v) < post(u) (since it can't have back edges).

Does the logic work the other way around? Does this mean that if post(u) < post(v) then we have a DAG? Not necessarily. Just because post(v) > post(u) doesn't necessarily imply that a back edge exists.

• Our algorithm for topological sort: do a DFS on graph G, then enumerate all $v \in V$ in the decreasing order of post(v).

2.3 Strongly Connected Components

- To find DAGs on undirected graphs we can run DFS on them, but what about directed graphs?
- Definition: Vertices u and v are strongly connected if there is a path from u to v and there is also a path from v to u.
- A graph is *strongly connected* when all of its vertices are strongly connected.
- Generally, we partition a graph into *strongly connected components*, since it's very rare that the entire graph is strongly connected.
- Strong connectivity is an example of an equivalence relationship, since it satisfies all the properties: reflexive, symmetric and transitive.
 - Every vertex is strongly connected to itself
 - If *A* is strongly connected to *B*, then *B* is strongly connected to *A*.
 - If *A* is strongly connected to *B* and *B* is strongly connected to *C*, then *A* is strongly connected to *C*.
- If we flip all the edge (i.e. reverse the direction), the strongly connected components don't change at all!
- We can then divide this into a *meta graph*, where we group the graph by strongly connected components (try to be as general as possible when you do this).
 - The meta graph can't have cycles! Had a cycle existed, then it would imply that vertices from one strongly connected component can reach vertices of another, and vice versa!
 - Therefore, the meta graph *must* be a DAG.
- Why care about SCC? It is useful in many different fields, since SCCs naturally imply a strong equivalence of
 objects in a graph.

2.4 Finding SCCs

- Attempt 1: Consider all possible decompositions and check (BAD!)
- Attempt 2: Consider pairs of nodes, then run explore to see if they reach the other. (ALSO BAD!)
- There exists an algorithm that runs this in O(n + m) time, where we have to run DFS (smartly) only twice!

2.4.1 More Properties of SCCs

- It actually matters where we start our DFS, since sometimes we can't exit the particular connected component. Specifically, if the node is part of a source in the meta graph, then it will visit many nodes, whereas if the node is a sink then we never exit that particular connected component.
- The "right" place to start the DFS is in the **sink of the meta graph!** The key thing is that we shouldn't exit that connected component, so running explore() on any vertex within this CC will give us the whole SCC.
- Idea: Do the topological sort on the meta graph, then run DFS on the sinks of the meta graph.
- Suppose we run DFS on a graph G. Let C and C' be two connected components such that $C \to C'$ in the meta graph. Then, $\max(\text{post}(v \in C)) > \max(\text{post}(u \in C'))$.
 - *Proof:* Split into cases, based on where in the graph we started:

Case 1: Suppose DFS visited C first ($u \in C$ is the first node visited by DFS. Then, DFS will explore C' first before finishing its exploration of C. This means that DFS finishes C' before finishing C, meaning that max(post($v \in C$)) > max(post($v \in C$)).

Case 2: Suppose DFS visited C' first. Then, we will never visit C since there is no directed edge between $C' \to C$. Therefore, the post of C' stops before DFS even reaches C, Hence, we have $\max(\text{post}(u \in C')) < \max(\text{post}(v \in C))$.

- Immediate corollary of this: Suppose we ran DFS on a graph G. The highest post(v) belongs to a node v that is in the source SCC of the meta graph!
 - Imagine $\max(\text{post}(C))$ is not the largest post(v) value. Then, this means that there exists another C'' whose post(C'') is larger than that of C!
- So now we have a way of finding the source of the vertices in *C*, but we want to start from sinks, so we **flip the edges, then run DFS!** The only difference betweeh *G* and *G*^R (the reversed graph) is the direction of the edges; the connected components remain the same.

Instead of flipping edges, why not run the algorithm from the minimum post value instead?

Because the minimum isn't guaranteed to be a SCC in the same way the max is.

2.5 The Algorithm

- Compute G^R .
- Run DFS on G^R .
- Store the post numbers of this DFS in array called post-r
- Run DFS on *G*, but explore unvisited nodes in *decreasing* order of post-r. For every DFS we run, we find a new SCC.

3 Paths in Graphs

3.1 Single Source Shortest Path (SSSP)

- We want to compute the distances from a source $s \in V$ to other nodes in V.
- · We don't use DFS here because DFS might explore much longer paths first, so it might be very inefficient.
- Solution: use Breadth-First Search (BFS)
 - Analogous to a bird's eye perspective, where we explore successively outward in "neighbourhoods."
 - Start at exploring from distance 1, then when everything at distance 1 is explored, continue to explore at distance 2, etc.
- The type of BFS that we use depends a lot on what kind of graph we're dealing with:
 - Unweighted graphs: Ordinary BFS works
 - Positive Weights: Dijkstra's algorithm
 - Negative Weights allowed: Bellman-Ford Algorithm
- Going down this list makes the graph more general, but they are less efficient than the ones above.

Would a more correct statement be that BFS works if all the edges have the same weight?

3.2 Breadth First Search (BFS)

• Start at *s*, and add all the neighbours of *s* to a queue. For every vertex in the queue, we visit all the unvisited nodes from that vertex, and add it to the queue. Repeat until all nodes have been visited.

3.2.1 Runtime of BFS

- We enqueue and deque every node exactly once if the node is connected, otherwise we don't do it at all. This takes *O*(1) time.
- Once an item is dequeued, we need to check all the neighbours of a graph, costing $O(\deg(u))$ time.
- In total, our runtime is:

$$\sum_{u \in V} O(1 + \deg(u)) = O(n + m)$$

This is the same runtime as DFS, which is not a coincidence! DFS and BFS are actually related, except the queue is replaced by a stack.

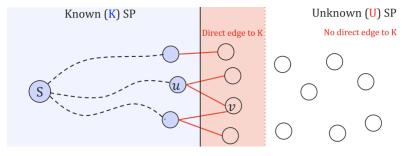
• We didn't implement it as a stack in lecture, but the idea is the same.

3.3 Weighted Graphs

- BFS doesn't work here because it ignores the weights of the graph. It is possible that a graph ends up being shorter but goes through more nodes, a possibility that BFS doesn't catch.
- Useful Fact: Any sub path of a shortest path is also a shortest path. This is rather obvious.
- So what we should think about is that to build the shortest path, we build the shortest path from other, shotest paths but add in the shortest edge. This guarantees that our shortest path remains the shortest.

3.4 Dijkstra's Algorithm

• Let K denote the set of "known" nodes where the length of shortest path is computed. To determine node we should add to K, we should select the vertex that gives the smallest dist $(s, u) + \ell(u, v)$. Visually:



- The red region is the set of nodes that we look at.
- We don't need to recompute all distances at every iteration instead we can just store the distances as we go
 along. Initial overestimates are fine, since eventually we will explore the shortest path, and its distance will
 eventually be updated.
- If we find a shorter path later on, we can update dist(s, u) to reflect that.
- If we want to find the shortest path from *S*, then we can add a new variable that stores the previous node in the sequence from *S* to *u*. Therefore, when we want to find the shortest path, then we are continually looking backward until we get back to *S*.

3.4.1 Runtime of Dijkstra's

• The runtime of Dijkstra's depends on the kind of data structure we used to keep track of the distances:

Implementation	Insert	Delete Min	Decrease Key	Runtime
Array	O(1)	O(n)	O(1)	$O(n^2 + m) = O(n^2)$
Binary Heap	$O(\log n)$	$O(\log n)$	$O(\log n)$	$O((n+m)\log n)$
Fibonacci Heap	O(1)	$O(\log n)$	O(1)	$O(n\log n + m)$

- The best known runtime of Dijkstra's algorithm is $O(n \log \log n + m)$.
- At the end of the day, this is slower than DFS, by the $\log n$ term.

3.5 Negative Weights: Bellman-Ford Algorithm

- Sometimes, having negative weights is possible, for instance when traversing an edge is more beneficial to you in some way.
- Shortest paths don't really make sense if a cycle has negative length (since then we'd be infinitely descending)
- All we need to do is modify Dijkstra's update function!
 - Call an update "safe" if dist(w) is an overestimate of the true shortest path between s and w. In other words, dist(w) ≥ d(s, w) for all $w \in V$.

see lectures for Bellman-Ford

4 Greedy Algorithms

- Are algorithms that build their solution piece by piece, and always takes the piece that offers the most obvious and immediate benefit.
- Some applications of where Greedy algorithms do work: Scheduling, satisfiability, Huffman Coding MSTs.

4.1 Scheduling

- Input: collection of jobs specified by their time intervals $[s_1, e_1], ..., [s_n, e_n]$. We want to find the largest subset of jobs that have no time conflicts.
- To do this, after choosing an interval, we'd want to choose the next interval that has the *earliest end time*. Jobs that finish earlier give us more opportunities to slot in more jobs later in the day.
 - This is not achieved by selecting the shortest job, because it does not give us freedom in where s_i and e_i
 - This is also not achieved by selecting the earliest job, since we don't know where e_i is.
- So our code is as follows:

While set of intervals is not empty:

Add interval j with the earliest finish time e_j .

Remove any conflicted interval *i* from the set, i.e. $[s_j, e_j] \cap [s_i, e_i] \neq \emptyset$.

- The runtime of this algorithm is O(n) if the intervals are already sorted by the end time, otherwise, we'd need $O(n \log n)$ time since we'd need to sort the intervals first.
- To show that the greedy algorithm works, we need to show that this algorithm doesn't rule out an optimal solution.

• Induction is very nice to prove these algorithms, since we'd just need to prove that the algorithm selects optimally at every time step! Let's try this for our scheduler:

Claim: For any $m \le k$ there is an optimal schedule OPT that agrees with the Greedy solution G on the first m intervals. More formally, if the OPT can be labelled as a list of i_1, \ldots, i_m and G has a list of j_1, \ldots, j_m , then we require that $i_1 = j_1, \ldots, i_m = j_m$.

Base case: m = 0, this is fairly trivial. Any two schedules agree on 0 things.

Inductive Hypothesis: This claim holds true for m. Now we show m + 1.

Inductive Step: There are two cases we need to consider:

- Case 1: when $i_{m+1} = j_{m+1}$, in which case we are done.
- Case 2: $i_{m+1} \neq j_{m+1}$. Then let's define another schedule OPT' which is the same as *OPT* except for the fact that i_{m+1} is replaced with j_{m+1} .

Note that j_{m+1} does not conflict with j_1, \ldots, j_m , since the greedy algorithm does not produce time conflicts. Also, j_{m+1} does not conflict with i_{m+2} since j_{m+1} ends earlier than i_{m+1} (by the greedy algorithm). Hence, placing j_{m+1} into this algorithm instead of i_{m+1} produces an *equally valid solution* for the schedule, since the size of OPT' is the same as that of OPT. Therefore, OPT' is also optimal, completing the proof.

Does this proof by induction assume that the Greedy solution gives a correct schedule?

• In essence, the proof is showing that there is no choice that the Greedy algorithm makes which rules out an optimal solution.

4.2 Horn Formulas

- Variables x_1, \dots, x_n are either true or false.
- Clauses:
 - "Implication clause" where $(x_i \wedge x_j \wedge ...) \implies x_k$. This is equivalent to $\overline{x}_i \vee \overline{x}_j \vee ... \vee x_k$.
 - "Pure Negative Clauses" where $(\overline{x}_i \vee \overline{x}_i \vee ...)$
- · A Horn formula is an AND of all Horn clauses, which are either implication or pure negative.
- There is a problem called Horn-SAT which asks us to find an assignment of variables that makes all Horn formulas to be true, if an assignment exists.
- Greedy Algorithm:
 - For all i, set x_i to be false.
 - While there exists an implication $(x_i \wedge \cdots \wedge x_j) \implies x_k$ being set to false, set x_k to be true.
 - If every pure negative clause $(\overline{x}_i \vee \cdots \vee \overline{x}_i)$ is set to true, we return (x_1, \dots, x_n) .
 - Otherwise, return "not satisfiable."

4.2.1 Proof of Correctness

- We want to show that whenever the greedy algorithm sets a variable x_i to true, it does not ruin a satisfying assignment. In other words, whenever a satisfying assignment exists, then Greedy will output one.
- We can show a stronger statement: the set of variables set to True by the greedy algorithm has to be set to true in any other assignment. We prove this by induction

Base case: In the 0th iteration of the while loop, nothing is set to true, so we're fine.

Inductive Hypothesis: The first *m* variables set to true by Greedy are also true in every satisfying solution.

Inductive Step: Let x_{m+1} be the next variable set to True by the greedy algorithm. This means that there was an unsatisfied implication $(x_i \wedge \cdots \wedge x_j) \implies x_k$ where the LHS was true, and x_{m+1} is false. This only happens when $x_i, \dots x_j$ are all set to True, by the greedy algorithm on m steps (which we know to match the optimal solution by inductive hypothesis).

Then the only way to satisfy this condition MUST have x_{m+1} set to true as well, and that completes the proof.

- Now we have to prove correctness. If Greedy outputs a solution, then it must be satisfiable this is fairly obvious, since the while loop and if condition makes sure that all clauses are satisfied.
- · We also want to

4.3 Codes

- Usually (things like ASCII) encode English characters using a fixed length of bits per character.
- If our goal is to save space, then we probably don't want that. Particularly, there are letters that appear more often than other characters, so if we were to use the same space for every character that'd be fairly wasteful.
- Assume that we have four letters with varying frequencies:

Frequency	Letter	Encoding 1	Encoding 2	Encoding 3
0.4	A	00	0	0
0.2	В	01	00	110
0.3	С	10	1	10
0.4	D	11	01	111
Total Cost:		2 <i>N</i>	N(0.4 + 0.3) + 2N(0.1 + 0.2)	0.4N + 2N(0.3) + 3N(0.2 + 0.4)
Total Cost.		21 V	= 1.3N	= 1.9 <i>N</i>

- There are issues with encoding 2: it's lossy in the sense that AB is encoded in the same way that BA is coded.
- These issues are solved in encoding 3, and we found that we can still do better than the 2N from our naive application where every letter gets the same number of bits.

5 Huffman Coding, MSTs

Recap of Greedy algorithms:

- Our goal is to prove that whenever a choice is made, that an optimal solution still exists, proven to exist via induction.
- Base case: at the beginning, achieving optimal choice is always possible
- Inductive Hypothesis is the same as normal induction, and inductive step is to show that if an alternate choice is made it doesn't violate the optimal solution.

5.1 Prefix codes and Trees

• Prefix codes can be represented as a binary tree with k leaves.

- the code is the "address" of a letter in the tree (i.e. the string of numbers leading from the root to that leaf).
 We want to order the tree from highest to lowest frequency, so that the letters with the highest frequency uses less characters.
- In general, the cost for such a tree is:

$$cost = \sum_{i=1}^{n} f_i \cdot depth(leaf i)$$

• Our goal is to find an optimal subtree. What does such a tree look like?

Answer: Even if we don't know what the frequencies are, the optimal code should be a full binary tree.

Now we need to prove that there exists an optimal tree when the two lowest frequency symbols are siblings
of each other.

Proof: By contradiction, let x, y be symbols with lowest frequencies and assume they aren't siblings. Let a, b be the deepest pair of siblings. Since x, y aren't siblings of each other, then only one of a, b are one of x or y. WLOG, let x = a. What happens if we swap x, y and a, b? Well, we know that f_a , $f_b \ge f_x$, f_y , and we've reduced the length of a, b while also reduced frequency of the deepest entries in the tree, meaning that we've ended up with a cheaper tree! Hence, the original tree could not have been an optimal tree.

5.2 Algorithm (Huffman Coding)

See below for pseudocode:

```
Node a object with a.freq = f_a and no children a.freq = f_a and no children Insert the node in a priority queue Q use key f_a While len(Q) > 1

x and y \leftarrow the nodes in Q with lowest keys create a node Z, with Z. freq = X. freq + Y. freq Let Z. left = X and X are X and X and X and X are X and X and X and X are X and X and X are X and X and X are X and X are X and X are X and X and X are X and X and X are X are X and X are X and X are X are X and X are X are X and X are X and X are X are X and X are X are X are X are X are X and X are X are X and X are X are X are X are X and X are X ar
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- The idea is to recursively generate a tree using the lowest frequencies, and combining them together by adding the frequencies of each tree's children.
- Runtime Analysis: Storing in our priority queue can be optimized by using a binary heap, which takes $O(n \log n)$ time. Combination also takes $O(n \log n)$ time, so total runtime is $O(n \log n)$.
 - Inserting into priority queue takes $O(n \log n)$ time.
 - At every step of the while loop, we perform 2 deleteMin instructions, which is constant time.
 - There is 1 insert each time (after the connection). This means that on every iteration, we are halving the number of nodes (hence $O(n \log n)$).
 - So total time complexity is $2O(n \log n) = O(n \log n)$.
- This generates a full binary tree with optimal coding.

Proof: We show that a greedy selection (which is what Huffman Coding is doing) does not rule out an optimality coding.

Base case: n = 2. We can generate optimal code using 0 for first letter and 1 for second letter. Huffman coding does the same.

Inductive Hypothesis: Assume that this works for n-1 letters.

Inductive Step: Let T be the optimal tree for the frequencies f_1, \ldots, f_n . WLOG, let $f_1 \le \cdots \le f_n$. Assume that the two lowest frequency codes are siblings (proven from earlier), and merge the two into a single node, where $f = f_1 + f_2$. Looking at the cost of the new tree T', we know that

$$cost(T) = cost(T') + f_1 + f_2$$

Huffman coding also does this merging process, and our inductive hypothesis guarantees that this tree is optimal on n-1 letters, so when we split back into the two characters it is still guaranteed to be optimal. Formally, if H' is the cost of the reduced tree, then

$$cost(H) = cost(H') + f_1 + f_2$$

which is the same as the cost relationship with T, so Huffman coding does indeed give an optimal coding.

What if $f_1 + f_2$ happens to be large enough such that $f_1 + f_2 > f_n$?

Apparently it doesn't matter?

5.3 Minimum Spanning Trees (MSTs)

- Tree also has edges, so we can assign a cost as well: $cost(T) = \sum_{e \in T} w_e$ (the sum of the weights).
- Suppose we're given a graph G(V, E) with non-negative weights. We want to find a set of edges that connects the graph, and has the smallest cost.
- Why do we care? This gives the notion of connectivity in a network, so you can think of cell towers or roads/railways as practical applications.
- We will use the same approach: first we ask about what an MST looks like.
 - It will be an acyclic graph, since removing an edge that's part of a cycle still preserves the connectivity in the graph.

5.3.1 Graph Structures and Cuts

- Cuts: a way to partition a graph that splits up the vertices into two groups.
- They're important because cuts go through edges to divide vertices into groups.
- Imagine we've already discovered some edges *X* of the MST. Consider the cut that doesn't cut any edges *X*. Now we look at the edges that are being cut. The edge from a larger MST is being cut, and it's the lowest weight edge that we should add to our MST!

This is a very important property, it shows that any cut that we make is an edge that can be added to our MST, so therefore regardless of which vertex we start searching at, an MST is still guaranteed.

- Therefore, we should add this edge and its corresponding vertex to our MST.
- We can formalize this argument via a proof, but I'm too lazy to write it here.
- Turns out that any algorithm that fits the following properties forms an MST:
 - Start with *X*, an empty list.
 - Pick S ⊆ V such that X has no edges from S to $V \setminus S$

- Choose the lighest weight edge from S to $V \setminus S$.
- Add edge to *X*.
- The proof for why this does give an MST can be done via induction.

5.3.2 Kruskal's Algorithm

- Instead of doing S and $V \setminus S$, it instead selects edges, and checks whether the edge forms a cycle. If it does, we don't add this edge. This process of checking a cycle actually does split our graph into S and $V \setminus S$, albeit implicitly.
- · We show correctness by showing that Kruskal's algorithm fits the meta algorithm given above.

6 A different greedy algorithm for MSTs

6.1 Prim's Algorithm

- The idea is to draw a tree by greedily adding the cheapest edge that can grow the tree.
- Start from some vertex, and repeatedly pick the lightest edge (u, v) such that $u \in S$ and $v \in V \setminus S$.

How exactly is this different from Kruskal's algorithm? Aren't both using cuts in the same way?

Kruskal's doesn't start from a given vertex, but instead just selects edges. Prim's starts with vertices and looks at edges that connect from S to $V \setminus S$

- Remember: the shape of the MST is dependent on the node that we start at, but an MST will always exist no matter which vertex we start at.
- Both Prim's and Kruskal's algorithm works on negative edge weights. This is because the cut property still holds, and the notion *minimum* spanning tree is not broken with negative edge weights.

6.2 Implementation

- The naive implementation of Prim's is actually quite slow, since on every added vertex we are looking for new
 cuts and checking edges every time.
- We can optimize by using priority queues (basically a max heap based on priority).

Is this true about priority queues?

- So here are the things we need to keep track of:
 - − For every edge $v \in V \setminus S$, check whether v has a direct edge of the set S of "visited" vertices, and also the cost of the lightest edge connecting v to the set S of visited vertices.
 - We had the same dilemma before, with Dijkstra's algorithm!
- So let's follow the same procedure as Dijkstra's!
 - First start with dist(v) set to infinity, and prev(v) to null for every vertex.
 - If a neighbor u is added to S (visited set) and $\operatorname{dist}(v) > w_{u,v}$, then we update $\operatorname{dist}(v) = w_{(u,v)}$, and set $\operatorname{prev}(v) = u$.
 - This is slightly different from Dijkstra's, where the dist array instead marks the minimum edge between two visited nodes, instead of the total distance from a certain vertex.

- Part of the reason for this is that MSTs don't care about where you start.
- The "cut" in this case is actually the process of adding adjacent edges from visited nodes to unvisited ones into the priority queue.

6.3 Runtime

- For a priority queue: we can either use a binary heap $(O(\log n))$ for each operation or fibonacci heap (a little bit better, since $\log(n)$ for inserts but O(1) for everything else.
- So because of the constant time for Fibonacci heap, it has $O(m + n \log n)$ time, whereas a binary heap has $O((m + n) \log n)$.
- · Comparing both algorithms:
 - Kruskal's: $O((m + n) \log n)$
 - Prim's (with Fibonacci heap) $O(m + n \log n)$.
 - For sparse graphs (so ones with not many edges), both are equally as good.
 - For dense graphs, Prim's is much better.

6.4 Set Cover Problem

- Input: the universe of *n* elements $U = \{1, ..., n\}$, and subsets $S_1, S_2, ..., S_m \subseteq U$, such that $\bigcup_{i=1}^m S_i = U$.
- Output: A collection of S_i of minimal size.
- This is an example of a problem where the greedy algorithm is not optimal! Instead, it is approximately optimal.
- Claim: if the optimal solution uses k sets, then the greedy algorithm uses at most $k \ln n$ sets.
- We will prove this recursively: let n_t be the number of elements not covered by the greedy algorithm after t choices. Then, we can reframe the problem to be that when $t = k \ln n$, we want $n_t < 1$.
 - Subclaim 1: $n_1 \leq n_0 \frac{n_0}{k}$.
 - *Proof:* the optimal solution requires k sets to cover U, so the average number of elements in any set is $\frac{n}{k}$. Hence, there is a set that counts more than $\frac{n}{k}$ elements (if not equal).
 - Subclaim 2: for any t, $n_{t+1} \le n_t(1-1/k)$.
 - *Proof:* This is a natural extension of claim 1.
- With this proven, we introduce an **approximation factor**, which is a way to say that Greedy is optimal, with an approximation factor of ln *n*.

7 Dynammic Programming I

7.1 Fibonacci Numbers, revisited

- Imagine computing Fibonacci numbers; there's a lot of repeated calculations! For instance, F(1) is computed 2^n times when we're looking for F(n)!
- To optimize this, store each successive computation of F(n) into an array that we access, so that we only need to compute each F(k) exactly once.
- This is called **memoization**, where we store things in a "memo," to be accessed by our algorithm later on.

7.2 Elements of Dynamic Programming

- There are a couple hallmarks of DP:
 - 1. Subproblems, or "optimal substructure". Refers to the fact that large problems can be broken up into smaller subproblems. For Fibonacci, this means that F(n) is recursively expressed in terms of smaller subproblems.
 - 2. Overlapping subproblems: A lot of subproblems overlap with one another. We recurse to smaller subproblems, and in doing so we see that a lot of computation is repeated. The solution to this is to use memoization, so that each computation is done only once.
 - 3. There are two ways to do DP:
 - Top-Down: start from the largest subproblem and recurse to smaller subproblems. This often involves recursion.
 - Bottom-up: start from the smallest subproblems then work to larger subproblems. Memoization still
 happens; we just fill the table from the small to largest problems. In this method, this doesn't need
 a recursive call.
- The mathematical runtime of top-down and bottom-up are the same.
- The computation structure for DP actually looks awfully similar to a DAG.
- If we view every subproblem as a node in the graph: construct it in such a way that an edge $i \to j$ exists if the solution to subproblem j directly depends on the solution to of subproblem i.
- Consider a topological sort on this DAG: then the bottom-up solution directly follows the conputation of this DAG!
 - In the top-down framework, we are filling up the memo table in topological sort order, since that table is still being filled from bottom up.

7.3 Shortest Paths on DAGs

- We're given a DAG with a source s. We want to find the cost of the shortest path from s to u for all $u \in V$. We also want to do this in linear time, O(n + m).
- We can always run a topological sort on this DAG in O(n + m) time. Our subproblems are the distances from s to u for every node u.
- After ordering in topological sort, we can just go down this graph *in topological order!* This means that the structure of the DP tree is the same as that of the topological sort.
- In terms of our recurrence relation, $\operatorname{dist}(u) = \operatorname{dist}(v) + \ell(u, v)$. Here, $\operatorname{dist}(v)$ is implied to be memoized, since it's already a solved problem.
- This is an O(n + m) solution to this problem!

7.4 DP Recipe

- a) Identify the subproblems (i.e. find the optimal substructure)
- b) Find a recursive formulation for the subproblems: just try to solve it via recursion and see where it gets you.
- c) Design the DP algorithm fill in a table, starting with the smallest sub-problems and building up.

7.5 Shortest Paths with k

- Here we consider the same problem of finding shortest path, but we're restricted to use at most k edges.

Fill this out from lecture recordings

7.6 All-Pair Shortest Paths

- Here, instead of finding the shortest path from a singular source node, we want to find it for all pairs of nodes.
- Input: again a graph with no negative cycles.
- Naively, we can run Bellman-Ford on all nodes, but this would take O(nm) a total of n^2 times, so our total runtime could be as large as $O(n^4)$ for dense graphs. Therefore, we're looking for a better algorithm.
- Identify the subproblem: subproblem k: for all pairs, find the shortest $u \to v$ path whose internal vertices (so the path they take) only use nodes $\{1, 2, ..., k\}$.
 - In other words, there's a collection of k nodes, and the path from $u \rightarrow v$ only uses these nodes.
- Recursion: When we have the set from $\{1, ..., k\}$, we want to find the relation between this set and how to expand this set. There are a couple ways that the new node can be added:
 - Case 1: The new node added does not lie on the path: then nothing really changes, so $\operatorname{dist}_{k+1}(u,v) = \operatorname{dist}_k(u,v)$.
 - Case 2: The shortest path uses the added node: this path can be broken into two parts: the shortest $u \to (k+1)$ path and then the shortest $(k+1) \to v$ path. Both of these paths are already computed (by definition of them only using the set $\{1, \dots, k\}$), so we just have to add these two up.
 - To combine these two, we find the minimum of these two to find whether the path from u to v has changed or not.
- Runtime: Each update is O(1) time, and we have to loop over u, v a total of k times, so overall $O(n^3)$ runtime.
- This is called the Floyd-Washall Algorithm.

8 Dynamic Programming II

We will look more at how to choose subproblems (step 1 of our "recipe" to solve DP problems) Some problems we'll look at today:

- · Longest increasing subsequence
- Edit distance
- Knapsack Problem

Pay attention to the information our subproblems need to be storing.

8.1 Longest Increasing Subsequence

- Given an array of integers $[a_1, ..., a_n]$, and we want to return the length of the longest increasing subsequence of the input. (the selection of the indices **doesn't have to be contiguous**)
- We're going to deal with 1-indexed arrays here.
- Why is this useful? This problem is one processing step used in a lot of other algorithms, and even the game of Solitaire (also called Patience Sorting)

8.1.1 Subproblems

- Which of the following is better?
 - L[j] is the length of the LIS in the array $[a_1, ..., a_j]$ for j = 1, ..., n.
 - L[j] is the length of the LIS in array $[a_1, ..., a_j]$ that ends in a_j for j = 1, ..., n.

The second one is far better, because we keep track of a_i information.

• The second subproblem is better, because keeping track of a_j is very valuable when we are trying to recurse back in our DP problem.

If we don't keep track of a_j , we don't have any information of the last element of our LIS subproblem, so we don't know how to attach it.

- Whatever the subproblem is not storing/not stating is going to be taken away from you. You can only observe things that the subproblem is storing/stating
- To add, there are two cases:
 - Suppose a[j] ≤ a[i]. Then we cannot add a_j because it's not part of the longest increasing subsequence. Therefore, L[j] (the LIS up to j) is the same as L[i].
 - Otherwise, a[j] > a[i], so we can add it to the LIS (so L[j] = L[i] + 1).

How is it possible that $a[j] \le a[i]$?

There could be an increasing sequence that grows slower that isn't counted in the *i*-th iteration

- Therefore recursively, we have to apply the following:

$$L[i] = \max_{i < j} \{L[i] : a_j > a_i\} + 1$$

 $L[1] = 1$

We want the maximum because we want the *lonegest subsequence* to tack a_i onto.

• In total, there are O(n) subproblems, and each subproblem has O(n) time, so in total we have an $O(n^2)$ runtime.

8.2 Edit Distance

- Given a string S[1,...,m] and T[1,...,n], we want to find the smallest number of edits to get us from S to T.
- Allowed operations: insert a character, delete character, change character.
- Why is this useful? Autocorrect, autocomplete in search engines and also DNA analysis of similarities.

8.2.1 Cost of Alignment

• Rather than thinking about distance in terms of *moves*, instead we can think about the cost of alignment. In other words, we look at the number of columns that don't agree.

S-NOWY
SUNN-Y
SUNN-Y
Alignment of cost 3
Alignment of cost 4

8.2.2 Subproblems

• We define a 2D array to keep track of subproblems, where

$$E(i, j) = \text{EditDist}(S[1, \dots, i], T[1, \dots, j])$$

So this defines the cost of optimal alignment for strings S[1,...,i] into T[1,...j].

- What are the different ways we can align the subproblems?
 - S[i] is dangling, T[j] is dangling, S[i] and T[j] are both fully aligned. Visually:

Case 1	Case 2	Case 3
$S[1 \cdots i-1]$ $S[i]$	$S[1\cdots i]$ –	$S[1 \cdots i-1]$ $S[i]$
$T[1\cdots j]$ -	$T[1\cdots j-1]$ $T[j]$	$T[1\cdots j-1] T[j]$

We add in one at a time, so there can only be one dangling letter (or none) at a time!

- We recurse based on the cases:
 - Case 1: S(i,j) = S(i-1,j) + 1.
 - Case 2: S(i,j) = S(i,j-1) + 1.
 - Case 3: $S(i,j) = S(i-1,j-1) + 1(S[i] \neq T[j])$.
 - Base cases: S(i, 0) = i, S(0, j) = j.

The 1 represents an indicator function that counts the number of misaligned characters.

- During the recursion step, we want to fill S(i, j) with the *minimum* value of these three, since we want to minimize the number of edits. We will store this information (memoizing) in a 2D array.
- We have to traverse our array either row by row, column by column, or diagonally. This is because we need to ensure that all three subproblems that we're considering have already been computed.
- Pseudocode:

```
Edit-Distance(S[1 ... m], T[1 ... n])
(m+1) \times (n+1) \text{ array } E
For i = 0, 1, ..., m, E[i, 0] = i
For j = 0, 1, ..., n, E[0, j] = j
For i = 1, ..., m
\text{For } j = 1, ..., n
E(i, j) \leftarrow \min \left\{ \begin{array}{c} E(i-1, j) + 1, \\ E(i, j-1) + 1, \\ E(i-1, j-1) + 1, \\
```

• Runtime: there are O(mn) subproblems, since we have a 2D array of dimension mn. At each subproblem, we are only computing a minimum, which is O(1) runtime, so we just have O(mn) runtime.

Isn't the indicator also computed at this step? Why is that not accounted in the runtime?

The indicator is run in constant time, since we're only looking at the *i*-th character in comparison to the *j*-th character.

8.3 Knapsack (with repetition)

- A weight capacity W and n items with weights and values (w_i, v_i) . We want to output the most valuable collection of items whose total weight is at most W.
- We will be selecting with repetition here, but there is an easier variation where we don't consider repetition.

8.3.1 Subproblems

- For all $c \le W$, we want to consider the best achievable arrangement for knapsack of capacity c.
- Recurrence: For a given item i, then once we put it in the knapsack there are only $c w_i$ capacity that remains to be optimized. This is our recurrence relation.

$$K(c) = \max_{i: w_i \le c} \{v_i + K(c - w_i)\}$$

This is a maximum over the value because we want to maximize the value being put in our knapsack.

• We will store this information in an array of size W + 1, since we need a K(0) element.

```
Knapsack-with-repetition(W, (w_1, v_1), ..., (w_n, v_n))
An array K of size W+1.
K[0] = 0
For c = 1, ..., W,
K[c] = \max_{i:w_i \le c} \{v_i + K(c - w_i)\}
return K[W]
```

- Runtime: There are O(W) subproblems, and at each subproblem, we have maximally n items we need to check. So in total, we have O(nW) runtime.
- Generally we want to think of the runtime in terms of the input. For graph problems, we looked at the input size |V| and |E|. But for this problem, W takes $\log(W)$ bits to represent W. So the input size is $\log(W)$. For the weights of the items themselves, they also only have at most $\log(W)$ bits, Therefore, the total runtime is $O(n \log W)$.

This kind of algorithm is polynomial in n, but not W. This is called a <u>pseudo-polynomial</u> algorithm, since it's an algorithm that's polynomial given the numerical value of the input but not in the input size.

9 Dynamic Programming III

We'll look at more examples today of DP.

9.1 Knapsack (without repetition)

- Start with a recap with knapsack: had a weight capacity W, and a set of items with individual weights (w_i , v_i), and we wanted to look at the most valuable combination of items.
- · Now, we're going to look at this problem with the with the constraint that we cannot choose with repetition
- To solve, look at how we solved the problem with repetition: introduced K(c) which gets us the best achieveable value for a capacity $c \le W$. The issue with trying the same thing is that our subproblems don't track which items have already been used. Why not keep track of both?

9.1.1 Subproblems

- Introduce a 2D array: essentially solve the problem for smaller knapsacks and also smaller capacities. Then expand in two directions: in terms of the number of items and also the capacity.
- So keep track of all weights $c \le W$ and all items $j \le n$. Define K(j, c) to be the optimal solution to the knapsack for capacity c and items $\{1, 2, ..., j\}$. (It doesn't need to use all the items from 1 to j.
- For each K(j, c), we recurse smaller subproblems:

Case 1: The optimal solution on items 1 through j doesn't use item j. Here, K(j, c) = K(j - 1, c).

Note that this is not equivalent to $K(j-1, c-w_i)$, since the w_i could be distributed among other items.

Case 2: the optimal solution on items 1 through j uses item j. Here, $K(j, c) = K(j - 1, c - w_j) + v_j$. We add v_j to K since we're now using item j.

The intuition here is that we use the optimal solution without item j, then add in item j at the end.

9.1.2 Implementation

• So let's formalize this:

$$K(j, c) = \max\{K(j-1, c), v_j + K(j-1, c-w_j)\}.$$

with base cases K(0, c) = 0 and K(j, 0) = 0. The base cases make sense since with no items our optimal value is 0, and with no allowed weights then the optimal value is also 0.

- Looking at K(j, c) it only relies on the subproblems K(j 1, c), or $K(j 1, c w_j)$, so we're only looking at row j 1, and different elements in that row. This tells us about the order in which we should be solving the subproblems: we could either do this row by row or column by column.
- For runtime, ther eare O(nW) subproblems, and in each subproblem we're doing constant work (memory access), so therefore the total runtime is O(nW), just like knapsack with repetition.
- For space complexity, notice that each K only depends on the previous row, so once we've moved onto the 3rd row, we no longer need the first. We can delete this from memory, so the optimized space complexity is O(W).

9.2 Traveling Salesperson Problem

- A notoriously difficult problem, and DP helps us get a *slightly* better runtime.
- Input: Cities 1, ..., n and pairwise distances d_{ij} between cities i and j. We want to find a "tour" of minimum total distance (so we need to visit every city exactly once and return to the city we started at).
- The naive brute-force algorithm basically is the one where we have to go through all possible tours: there are $n! \in O(n^n)$ possible tours, which makes this computation very expensive.
- Dynamic programming gives us $O(n^2 2^n)$. (this is nearly optimal, beating $O(n2^n)$ is theorized to be impossible)
 - − To give an illustration of the difference DP makes, if n = 25, then $O(n!) \approx 10^{25}$, whereas $O(n^2 2^n) \approx 10^{10}$, so we're already better by 15 orders of magnitude.

9.2.1 Subproblems

• One challenge of TSP is that subproblems aren't exactly solving the problem. If we just look at TSP for a subset of our graph, that doesn't necessarily give us a solution to the larger problem, since we're looking for cycles. Instead, we think of "partial solutions" to our graph.

• We think of the subprobles as starting from city 1, ends in city *j*, and passes thorugh all cities in a set *S* (which includes city 1 and *j*). Visually:

$$1 \rightarrow i_1 \rightarrow i_2 \rightarrow \cdots \rightarrow j$$

So we want to formally define T(S, j) to be the length of the shortest path visiting all cities in S exactly once, starting from 1 and ending at j.

9.2.2 Recurrence Relation

• How to compute T(S, j) using smaller subproblems? Well, look at the string again:

$$\underbrace{1 \longrightarrow i_1 \longrightarrow i_2 \longrightarrow \cdots \longrightarrow i}_{T(S \setminus j, i)} \longrightarrow j$$

To actually talk about T(S, j), then we need to add d_{ij} onto every $T(S \setminus j, i)$. However, what is annoying is that we actually don't know which city is second to last, so we'll need to consider every possible city $S \setminus j$.

• So, we'll have to pick the minimum over all $i \in S$ such that $i \neq j$. Formally:

$$T(S,j) = \min\{T(S \setminus j, i) + d_{ij} | i \in S \land i \neq j\}$$

- Our base cases are $T(\{1\}, 1) = 0$, this is fairly trivial. We also want that $T(S, 1) = \infty$. The reason we want this is because we're talking about incomplete paths, so T(S, 1) is not a valid non-cycle. Hence, we want to set it to ∞ .
- We're not done though, because we have to do something to get us back to a cycle! At the end of the recursion step, we'll want to add the final edge (j, 1) back, but adding only the minimum:

$$T(S,1) = \min_{j \neq 1} \{ T(\{1,\ldots,n\},j) + d_{j1} \}$$

9.2.3 Implementation

• Want an array of size $2^n \times n$, and start with base cases. Then work on the recursion:

```
\begin{aligned} \operatorname{TSP}(d_{ij} \colon i,j \in [n]) \\ & \text{An array } T \text{ of size } 2^n \times n. \\ & T[\{1\},1] = 0 \\ & \mathbf{For} \text{ set size } s = 2, \dots, n \\ & \mathbf{For} \text{ sets } S, \text{ s.t. } |S| = s, 1 \in S \\ & \mathbf{For} \ j \in S \\ & T[S,j] = \min_{i \in S: \ i \neq j} \{T[S \setminus \{j\},i] + d_{ij}\} \end{aligned} \mathbf{return} \min_{j \neq 1} T[\{1, \dots, n\}, j] + d_{j1}
```

• For runtime, there are $O(2^n \times n)$ subproblems, and on each layer we're doing O(n) work, since we're checking the minimum across n nodes every iteration. So, we have $O(n^2 2^n)$ as the final runtime.

How do we explain that $n^2 2^n$ is the number of subproblems?

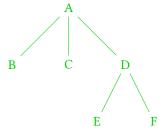
O(n) work at every step, then $n \times 2^n$ subproblems. There are 2^n subsets, and in each subset we can choose a j to exclude, which we can upper bound by saying that there are n of these. So $n \times 2^n$ is a tight upper bound on the number of subproblems.

9.3 Independent Sets in Trees

- We're given an undirected graph G = (V, E), and want to output the largest independent set of G.
- Recall that a set $S \subseteq V$ is considered independent if there are no edges between $u, v \in S$.
- This is also a notoriously hard problem, for general graphs. There isn't a polynomial time algorithm that does this. But for trees, we're in luck!

Why isn't the solution just selecting every other layer?

There are instances where we can pick from two consecutive layers and still not have an edge. Consider the tree:



Our greedy algorithm would select either $\{A, E, F\}$ or $\{B, C, D\}$, but the optimal set is actually $\{B, C, E, F\}$, so this proves that our algorithm isn't optimal.

• For trees, we know that they don't have cycles, so we can pick any node and say that that is the root. By doing this, we can get a "natural ordering" of the subproblems.

9.3.1 Subproblems

- Let I(v) be the size of the maximum independent set in the subtree that is rooted at v.
- Why is this a good subproblem? Becuase it's easy to write a recursion relation for it!
- For the subproblems, there are two cases:

Case 1: v (the root of the tree) is part of the optimal independent set. This means that the children aren't allowed to be part of the independent set. So if we take v, we can't take any of the subproblems. So we need to look instead at the *grandchildren* of v to join. Here, we'd write this as:

$$I(v) = 1 + \sum_{u \in \text{grandchildren}} I(u)$$

We add 1 here because we're including v now.

Case 2: v is not part of the optimal independent set. Here, we would just take the maximum of the children. Then:

$$I(v) = \max_{u \in \text{children}} \{I(u)\}\$$

So we'll take the max of these two cases:

$$I(v) = \max \big\{ 1 + \sum_{u \in \text{grandchildren}} I(u), \sum_{u \in \text{children}} I(u) \big\}$$

Also, base cases is that I(leaf) = 1.

9.3.2 Implementation

- We need a data structure to store the tree easily, and also make sure that every child is processed before the parents are. Well, we can iterate through the graph in post decreasing post order!
- The runtime of DFS on trees is O(|V|), and each edge is looked at ≤ 2 times once for the children and also once for its grandchildren, so each subproblem takes constant time.
- So that the total work is O(|E|) = O(|V|), since |E| = |V| 1.

10 Linear Programming

• Just like dynamic programming, there are different paradigms of programming, so for the following lectures we'll explore linear programming.

10.1 Example: Making Cake

- Suppose we're trying to make cake, and each item has a cost and associated profit. How much of each item should we produce in order to maximize profit?
- We're also given constraints, in terms of our supply of raw ingredients.
- This falls under the category of *constrained optimization*. They essentially ask us to maximize a quantity, while satisfying certain constraints. We've actually done this before!
 - When finding MSTs, the constraint was that our algorithm should have outputted some spanning tree.
 - For longest increasing subsequence, the constraint was that it was a subsequence and it is an increasing
 one.
- The difference between these and linear programming is that the objective function (our goal) is a minimalization (or maximization) of a linear function of the decision variables.
- Our goal is to find an algorithm that can solve all linear programs efficiently.

10.2 Decision Variables

- With decision variables we generally say that they are real values, and we cannot have integer constraints in linear programs. If they do exist, then it's now called an Integer Linear Programs (ILP), which has a completely different solution method.
- We now relax the constraints of x and y and allow them to take any value within \mathbb{R}

10.3 LP Standard Form

• A linear program in standard form can be written as follows:

We want to maximize the equation: $c_1x_1 + c_2x_2 + \cdots + c_nx_n$, subject to the constraints:

$$\begin{aligned} a_{11}x_1 + a_{12}x_2 + \cdots + a_{1n}x_n &\leq b_1 \\ a_{21}x_1 + a_{22}x_2 + \cdots + a_{2n}x_n &\leq b_2 \\ &\vdots \\ a_{m1}x_1 + a_{m2}x_2 + \cdots + a_{mn}x_n &\leq b_m \end{aligned}$$

and also $x_1, x_2, ... x_n \ge 0$. In general, we say that we have m constraints and n variables to satisfy. We can also write this in matrix form, where we want to maximize $\mathbf{c}^{\top}x$, subject to the equations $A\mathbf{x} \le \mathbf{b}$ and $\mathbf{x} \ge 0$. Here, \mathbf{c} and \mathbf{c}, \mathbf{x} are $1 \times n$ column vectors and b is a $1 \times m$ column vector.

Why aren't we taking $c^{\top}x^{\top}$?

10.3.1 Classroom Allocation

- We have a set of courses and possible classrooms, and every course needs a classroom, but not every course fits in every classroom! We want to find the maximum number of courses allocated to a classroom.
- We can construct a graph G = (V, E) where course c can be assigned to classroom r if and only if $(c, r) \in E$.
- We can make our decision variable $x_{c,r}$ (ideally, boolean of 0 or 1, but this condition must be relaxed) for each pair $(c,r) \in E$.
- Now for the constraints:
 - − Instead of constraining $x_{c,r}$ to 0 or 1, we can constrain $x_{c,r} \in [0,1]$. As we'll see, even though we've assumed $x_{c,r}$ can be real, they will turn out to be integers.
 - We want the room to not be simultaneously assigned to more than one course (fully) at a time. So, for all rooms r,

$$\sum_{c:(c,r)\in E}x_{c,r}\leq 1.$$

In other words, for every room, the number of courses assigned to it must be less than 1.

- We want every class to exist in one classroom only (so we don't have a situation where half the class is in Wheeler and the rest is in Pimentel) , so for all classes *c*,

$$\sum_{r:(c,r)\in E}x_{c,r}\leq 1.$$

Why isn't the constraint of $x_{c,r} \in [0,1]$ sufficient to guarantee this condition?

Because there's multiple $x_{c,r}$ being assigned to *every* pair! We want the sum of all of these to be less than 1, so that part of the class isn't somewhere else.

• Our objective is to find:

$$\max \sum_{(c,r)\in E} x_{c,r}.$$

10.4 Geometric Intuition

• Suppose we have the following LP when we're maximizing x + 2y with the following constraints:

$$x \le 3$$

$$x + y \le 5$$

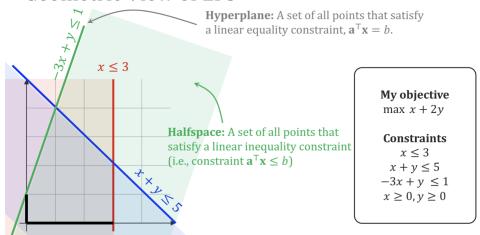
$$-3x + y \le 1$$

$$x \ge 0$$

$$\ge 0$$

We can graphically draw all of these constraints out, which will look as follows:

Geometric View of LPs



The overlap of these three regions defines the set where all our constraints are satisfied. We will call this region the **feasible region**. This region is always going to be a **convex region**.

• Definition: A set S is convex if for any two points $x, y \in S$, all points $z = \alpha x + (1 - \alpha)y$ for $\alpha \in [0, 1]$ satisfies $z \in S$.

In other words, all points on the line connecting *x* and *y* are also in *S*. We can prove (proof in slides) that every linear program will have a feasible set that is convex.

- Suppose now we're trying to maximize the equation x + 2y. To find the optimal solution, we have to find *level* sets of the feasible region, by considering x + 2y = c and incrementing c.
- *Definition:* A vertex (or extreme point) is a point found at the intersection of hyperplanes in R.
- *Claim:* Any linear program has an optimal solution that coincides with one of these vertices. Otherwise, the linear program could achieve unbounded values.

So what exactly happens when one of the hyperplanes is parallel to our objective equation?

We can reasonably take any point on that face, but note that a corner will appear there too, so there's no reason to not output the corner. In fact, almost all algorithms end up finding corners anyway, since that's where the extreme points are guaranteed to exist.

10.5 Algorithm for Finding Vertices

- Given m constraints, for each subset of n constraints, we find the intersection x^* of these constraints (do this by switching the inequalities with equalities and performing Gaussian elimination). If x^* satisfies all constraints, then add it to the list of extreme points.
- We could just go through all vertices and find the one with the highest payoff, but this would be incredibly slow:
 - There are $\binom{m}{n}$ different vertices, and this number grows very quickly.
 - In fact, with m = O(n) constraints, we can generate a situation where we have 2^n different vertices: hypercubes!

10.6 Simplex

- An algorithm that allows us to find the best neighboring vertex. Starting at x^* , look at all neighbours of x^* . If x^* is larger than all of these, return x^* . Otherwise, move to the highest neighbour and repeat.
- In higher dimensions, we define the neighbour is to say that two points are neighbours if they are formed by swapping out a *single* constraint.
- Simplex is quite interesting, in the sense that in the worst case it does exactly what the naive solution does, so its worst case runtime is still exponential. However, practically speaking, Simplex is one of the fastest algorithms we know of!

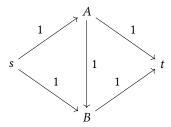
It's even faster than some algorithms that provably run in polynomial time!

11 Network Flow

- Recently declassified (1999) document about the USSR's shipment capacity from east to west. This was crucial information at the time since had a war broke out, the US could identify which supply routes they could bomb.
- They devised a greedy algorithm called "flooding," but this algorithm wasn't really optimal. It was finally solved by Ford and Fulkersson, and is now called the Ford-Fulkersson algorithm.
- Given a directed graph G = (V, E), one source vertex s and a sink t, and for each edge $e \in E$, we're given a capacity c_e which are integers.
- We want to find the maximum amount of water from $s \to t$.
- *Definition:* A flow assigns a number f_e to each directed edge $e \in E$ such that:
 - nonnegativity: $f_e \ge 0$
 - capacity: $f_e \le c_e$
 - flow in and flow out are equal: $\sum_{u \to v} f_{u,v} = \sum_{v \to w} f_{v,w}$
- Let's also define the size of the flow f to be the total quantity set from s to t. Using this definition, then the maximum flow is the one that maximizes size(f). This can be solved using linear programming!

11.1 Greedy (suboptimal) algorithm

- We'll find a path *P* from *s* to *t*, and send flow until it's saturated. We'll do this as much as we can. We repeat this until we run out of paths.
- This algorithm fails on some graphs, because it uses edge $A \to B$ when that edge is suboptimal! Consider the graph:



Our algorithm just looks at flow rate, so a possible path to take is $s \to A \to B \to t$, but this is clearly suboptimal! Instead, we should be going from $s \to A \to t$ and $s \to B \to t$.

11.2 Greedy Fix

- We instead consider a residual graph, where we subtract the flow given by greedy (s → A → B → t), and also
 generate a back edge that travels in the reverse order of the flow given by greedy, so that we can backtrack if
 needed.
- Formally, given a graph G and a flow f on G, the residual raph G_f is defined as: For all edges (u, v), if f goes from $u \to v$, then the residual graph will flow from $v \to u$ and the edge will have capacity $c_{u,v} f_{u,v}$.
 - By doing this, we allow our graph to backtrack along our suboptimal path if needed.
- This is the approach that Ford Fulkerson uses to find the optimal flow.

11.3 Ford-Fulkerson Algorithm

- Find a path *P* from *s* to *t* in the residual graph which is not yet saturated, and send more flow along *P*. We keep repeating this until everything's saturated, and this happens when all edges along one particular cut is zero.
- To show that this algorithm terminates, lets' first define an s-t cut is a partition of the graph into two sets of vertices L and R such that $s \in L$ and $t \in R$. We define the capacity of this cut to be the sum of all capacities from the edges that cross from L to R.
- Therefore, for any flow f and any cut (L, R), then $size(f) \le capacity(L, R)$. Then, the flow is actually upper bounded by the minimum cut along this graph (this is our "bottleneck" introduced at the outset)
- Then, this means that the max flow is also given by the minimum cut, and we can show that Ford-Fulkerson
 outputs a maximum flow by considering this relation between the flow and a cut. The proof of this is outlined
 in lecture.

Review the proof for this later

11.4 Runtime

- The number of augmenting paths must be less than U, where U denotes the maximum flow, so the update is less than $O(m+n) \cdot U$.
- But what this means
- There are other algorithms out there that optimizes this a little more: Edmonds-Karp gives us a runtime of $O(nm^2)$, which is much better than what we have.
- The best runtime was discovered last year, where we have $O(m^{1+o(1)} \cdot \log U)$

12 Network Flow II

We're going to explroe more LP problems today.

12.1 Bipartite Perfect Matching

• Input: a bipartite (undirected) graph G = (L, R, E) with |L| = |R| = n (so each node has a corresponding pair), and want to output a perfect matching from L to R.

A perfect matching is one where we can pair every vertex from *L* matches with exactly one vertex in *R*.

- Use cases include matching courses and classrooms: you can model a graph where *L* denotes the courses and *R* are classrooms, and *E* denotes whether a classroom can fit a course. So the example of perfect matching is basically asking whether we can assign every course to a classroom.
- To solve this, we convert this problem into one of max flow.

12.1.1 Algorithm

- First copy the graph G to make G', and make all the edges directed from L to R.
- Introduce a source vertex *s* that connects to every vertex in *L*, and introduce a sink that every vertex in *R* connects to. The capacity of each edge will be 1, so this is called "unit flow."
- *Claim: G* has a perfect matching if and only if the max flow on *G'* is *n*. (intuitively this also makes sense, since had it not been *n*, then it would mean that some vertex can't be matched)

Proof: First assume that G has a perfect matching, and let M denote that perfect matching. We can use this to construct a flow of size n, by putting 1 unit of flow along every edge in M. Then add 1 unit of flow going from S to every vertex, and add one unit of flow going from every vertex in S to S. There are S pairings, so the max flow here is S.

First recall from last lecture that if the capacities on our graph are integral, then the max flow is also integral. Now, let f be an integral flow of size n in G'. If it's an integral max flow, then we can only assign an integral amount to every edge, and since every edge has capacity 1, then our flow basically selects a subset of the edges (since flow is either 0 or 1 along the edges).

Each vertex $u \in L$ always has 1 unit of flow on 1 outgoing edge, since the flow from $s \to u$ is 1, and that flow needs to go somewhere. Similarly, each $v \in R$ has 1 unit of flow on 1 incoming edge. So one vertex in u has one outgoing edge in R and one vertex in R has a corresponding matching in L, so this specifies a matching of size n.

• This is a technique called a **reduction** from perfect matching to max flow.

12.2 LP Duality

- Recall that we said last lecture that the max-flow in a graph corresponds to the min-cut on that graph. So we
 could prove that a flow was optimal by showing a cut of the same value. This idea of correspondence is called
 duality.
- Suppose we wanted to maximize $5x_1 + 4x_2$, subject to the constraints:

$$2x_1 + x_2 \le 100$$

$$x_1 \le 30$$

$$x_2 \le 60$$

$$x_1, x_2 \ge 0$$

If we were to solve this, we'd get $x_1 = 20$, $x_2 = 60$ for a maximum value of 340. (You can check that this is a valid assignment). How do we check that this is the maximum value? Well, we can combine the inequalities in a clever way, which would end up getting us $5x_1 + 4x_2 \le 340$, which proves that our solution was optimal.

To show that this was optimal, we essentially multiplied the inequalities by values y_1 , y_2 , y_3 that gave us the optimal value. This meant that we were generating the equation:

$$(2y_1 + y_2)x_1 + (y_1 + y_3)x_2 \le 100y_1 + 30y_2 + 60y_3$$

We want to set y_1 , y_2 , y_3 such that the LHS is larger than our objective function, while making the RHS as small as possible. Basically, this means that we want to minimize $100y_1 + 30y_2 + 60y_3$ while also requiring that

$$0 \le y_1, y_2, y_3$$
$$5 \le 2y_1 + y_2$$
$$4 \le y_1 + y_3$$

So this is basically *another* LP problem! So the original problem is called the *primal LP*, and the second one is called the *dual LP*. Because of the way we set this up (with the maximization/minimization), it guarantees that any satisfying assignment of the primal LP will be less than that of the dual LP. This is the *magic trick*: the fact that we can always convert any maximization LP into a minimization in the dual LP.

Turns out that if we take the dual twice, we get back the original LP.

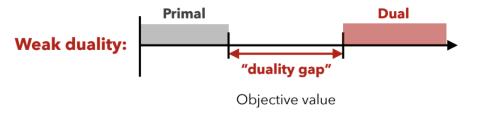
- This is a fairly standard exam problem: given a primal LP, compute the dual.
- There is also a matrix representation of the same problem. Recall the standard form for expressing LPs as maximizing $c^{\top}\vec{x}$ subject to the constraints given by $A \cdot \vec{x} \le \vec{b}$ and $\vec{x} \ge 0$.
- To construct the dual, we want to minimize the quantity $\vec{b}^{\top} \cdot \vec{y}$, where b used to be the constraints earlier. Our constraints are given by $A^{\top} \cdot \vec{y} \ge \vec{c}$ and $\vec{y} \ge 0$
- The theorem we saw earlier is called **weak duality**, which says that all feasible solutions *x* to the primal LP are less than the solutions *y* to the dual LP. The proof is as follows:

The primal LP maximization is written as $\vec{c}^{\top}x$, but we know that \vec{c} has another representation as $A^{\top}\vec{y}$, so we can write:

$$\vec{c}^\top \vec{x} \leq (\vec{y}^\top A) \vec{x} \leq y^\top b$$

Then, since $y^{\top}b = b^{\top}y$, then we have the inequality $c^{\top}x \leq b^{\top}y$.

• Graphically we can visualize this as:



• *Theorem*: If the primal LP is bounded (by any number), then the optimal solution to the Primal LP is equal to that of the dual LP. So in this case, the duality gap is 0

Doesn't this apply to most cases?

• The max-flow min-cut principle we had earlier is an example of an LP problem and a corresponding dual. So another way of proving what we did was to show that the min cut can also be written as a LP. Then, we can show that they are the duals of each other, and since the max-flow problem is always bounded, then we know that the optimal solution is when they are equal by strong duality.

12.3 Zero-Sum Games

• Input: a "payoff" matrix *M*, and two players: a row and column player.

• The matrix *M* specifies the amount that the row player wins, and the column specifies how much the column player wins. An example of this is rock paper scissors:

	Rock	Paper	Scissors
Rock	0	-1	1
Paper	1	0	-1
Scissors	-1	1	0

- In general, the row player selects a row r and the column player picks a column c. Then, we say that the row player wins $M_{r,c}$, while the column player wins $-M_{r,c}$.
- This is a zero-sum game because the sum of every row and column is 0.
- There are two strategies the players are allowed:
 - Pure strategy: pick one row/column and play their selection (e.g. row player always picks rock)
 - Mixed strategy: a probability distribution over pure strategies (e.g. we can have $P[\text{rock}] = \frac{1}{3}$, $P[\text{paper}] = \frac{1}{3}$, $P[\text{scissors}] = \frac{1}{3}$)

Also notice that the average score across all these strategies is 0, no matter what the column player does.

12.3.1 Game 1

· The game is described as:

- We want the row player to announce their strategy first, then the column player announces their strategy second. Because this is slightly unfair to the row player, we let the row player announce a mixed strategy $P = (p_1, p_2)$ where p_i is the probability that row i is chosen.
- The column player will respond by choosing a mixed strategy of their own, with distribution $Q = (q_1, q_2)$ where q_i is the probability of choosing row i.
- Then, the row player's average score is S(p, q) is the expected value that they get. So in this case:

$$S(p,q) = 3p_1q_1 - p_1q_2 - 2p_2q_1 + p_2q_2$$

So, the column player's best strategy is to minimize S(p, q), while the row player wants to maximize S(p, q).

• For the column player, they will choose the best strategy for themselves, after having seen *p*. So, since the column player knows what *p* the row player chose, then their minimization is the same as just minimizing over pure strategies:

$$\min_{\text{mixed strategies}} \{S(p,q)\} = \min_{\text{pure strategies}} \{3p_1 - 2p_2, p_2 - p_1\}$$

• The row player goes first, so their strategy is that they would want to maximize the column player's response. So for them they want to compute:

$$\max_{\text{mixed strategies}} \{\min\{3p_1 - 2p_2, p_2 - p_1\}\}\$$

· We'll see next time that this is exactly the same as computing two LPs, which are duals of each other!

13 Zero Sum Games II

13.1 Game 1

• We saw last time that given a ZSG structured like this:

then we can calculate that the payoff for column 1 is $3p_1 - 2p_2$, and the payoff for column 2 is $-p_1 + p_2$.

- This meant that the column player's best strategy was to look at these two values, and minimize this.
- The row player will then pick p_1 , p_2 in order to maximize what the column player tries to minimize. Therefore, the row player wants to calculate:

$$\max_{\text{mixed strategies }p} \{\min\{3p_1 - 2p_2, -p_1 + p_2\}\}$$

- As we said last lecture, this can be solved using LP! We'll also see that the row player going first doesn't hurt them.
- So we can formulate our LP as follows:

Maximize a quantity z, subject to the constraints:

$$z \le 3p_1 - 2p_2$$

 $z \le -p_1 + p_2$
 $1 = p_1 + p_2$
 $0 \le p_1, p_2$

Note that $z = \min\{3p_1 - 2p_2, -p_1 + p_2\}$, so by maximizing z subject to these two constraints means that z is constrained to the smaller of these two.

13.2 Game 2

The exact same game board, except we allow the column player to go first and the row player goes second.

- We do the exact same thing, by considering the possibilities from the row player's perspective. This is the same as looking from the column player's perspective in the previous problem.
 - From the row player's perspective, payoff of row 1 is $3q_1 q_2$, and row 2 is: $q_2 2q_1$. So we want to choose the max of these two, i.e.:

$$\max\{3q_1-q_2,-2q_1+q_2\}$$

- The column player will try to minimize this score, so they'll want to choose:

$$\min_{\text{mixed strategies } q} \{ \max\{3q_1-q_2, -2q_1+q_2\} \}$$

30

• This is another linear program! Here, we want to minimize z, subject to:

$$3a_1 - a_2 \le z$$

 $-2q_1 + q_2 \le z$
 $q_1 + q_2 = 1$
 $q_1, q_2 \ge 0$

Again, note that *z* is equal to the larger of the two inequalities, using the same logic as before.

- Now we compare the two problems, and compare the final quantities that we wanted to compute.
 - In game 1, we wanted to find $\max_{p} \{ \min_{q} \{ S(p, q) \} \}$
 - In game 2, we wanted to find $\min_{q} \{ \max_{p} \{ S(p, q) \} \}$
- · Given this construction, we have the inequality

$$\max_{p} \{ \min_{q} \{ S(p, q) \} \} \le \min_{q} \{ \max_{p} \{ S(p, q) \} \}$$

If the dual of the dual is the original, doesn't this mean that we get a situation like $x \le y \le x$?

The best way to see that this is true is that it's always better to be the player that goes second. Therefore, finding the minimum over q is better because they're able to "react" to the column player's moves.

• It turns out that these two LPs are actually duals of each other (the proof is to construct the second LP from the first one). Now, recall the property of strong duality, which holds for any LP that is bounded. This game is clearly bounded, so we know that there is an optimal value of the game (denoted by Value(game)) is the same for both games!

This is called the **min-max theorem**.

• This tells us that the order of play doesn't change the value of the game, and there is an optimal probability distribution that the row player can choose without caring about what the column player does.

Note that this is a zero sum game not by the game table, but because the gain of one player is an equal loss in the other player. So the numbers in the table can be anything!

• This also says that all zero sum games are strongly dual.

13.3 P vs. NP

- So far, we've seen a lot of algorithms: polynomial multiplication, MSTs, APSP, etc.
- In theoretical CS, we consider all these problems to be "efficiently solvable," We define this to be the case when a problem can be solved in **polynomial time**.

This is only in theoretical CS. In practice, we'd want to get everything down to O(n) time, if possible. Even $O(n^2)$ is quite bad, since given an input of size 10^9 (as is the case with the facebook graph), then the computation is on the order of 10^{18} (very bad)!

- We define P (stands for polynomial) to be a set of computational problems that are considered to be efficiently solvable.
- We define NP (stands for non-polynomial) to be another complexity class that aren't efficiently solvable themselves, but whose solutions can be efficiently checked.
 - Example: 3-coloring problem. We want to find a 3-coloring on this graph. Naively, we can brute-force and try all possible combinations, which would correspond to checking 3ⁿ possible graphs.

- The best known algorithm for this solves the problem in 1.3289^n time.
- So this algorithm is not in P, but it is in NP, since any solution can be verified in polynomial time (by checking all edges)
- Example 2: Factorization. Given an *n*-bit integer N, we want to factorize it into two numbers p, q > 1 such that pq = N.
- Naively, we can divide N by every number from 1 to \sqrt{N} . In terms of time complexity, this is $O(\sqrt{N})$, which we can simplify this to $O(2^{n/2})$ due to the way numbers are represented as bitstrings.

The best algorithm runs in time $C^{n^{1/3}} \log(n)^{2/3}$, so this is not a problem that's known to be in P.

However, the problem is in NP, because we can just verify any solution by multiplying them together, which takes at most $O(n^2)$ time.

14 P, NP, Reductions

- Recall that P is a "complexity class" of problems that are efficiently solvable (by efficient, we mean polynomial time), and NP being the complexity class of problems that can be verified efficiently. (again, polynomial time)
- Examples of problems in NP are: 3-color problem, TSP, factoring integers.

14.1 Rudrata Cycle (Hamiltonian cycle)

- Input: a graph G = (V, E), and we output a tour that visits each vertex exactly once.
- We can just run through all n! cycles in the graph, which is highly inefficient. The best known algorithm is $O(1.657^n)$ (this is faster than the DP solution), so this is not a problem in P!
- This problem is in NP, since given a string of vertices to visit, we can just go to our graph and check. This can be done in polynomial time.
- There are often modifications we can make to the problem that seemingly don't make that big of a difference, but drastically simplify the problem: finding an Eulerian tour (which adds the constraint that every edge must be visited), for instance, is a problem in *P*.

14.2 Traveling Salesperson Problem (TSP)

- Input: a graph G = (V, E) with edge weight, and there are three variations to this problem:
 - **Optimization TSP:** We're asked to find the tour with the minimum total weight. Our DP solution gets a runtime to $O(n^2 2^n)$, but this isn't polynomial time, so this isn't in P.
 - It's also not in NP, since there's no way to check whether a given tour is minimized without knowing the weights of all other tours.
 - Search TSP: Find a tour with total weight ≤ B, where B denotes our "Budget". This problem is not in P since coming up with a tour is still hard, but it is in NP, since verifying that the budget is less than B, can be done in polynomial time.

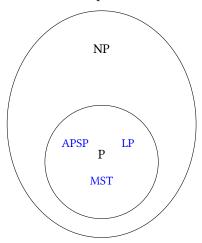
Notice that if we can solve Search TSP in polynomial time, this also gives a solution to Optimization TSP in polynomial time!

- **Decision TSP:** Asks whether there exists a tour with weight ≤ B. Again, just like Search TSP, this problem is not in P but is in NP.

- There are a list of problems that are known to not be in NP. Examples are: some optimization version of problems, counting problems (counting number of 3-colorings, for example), the halting problem.
- Formally, NP is only defined for **decision problems**. (CS172 material) For this class, we'll be looser and allow search problems as well.

14.3 P vs. NP

- **Theorem:** P ⊆ NP. In other words, any problem that can be *solved* in polynomial time can also be *verified* in polynomial time. The verification algorithm can literally just be to solve the problem again and check that the solutions match.
- A complexity diagram is usually used to show where problems lie:



• Largest open problem in theoretical computer science: is P = NP?

14.4 Reductions

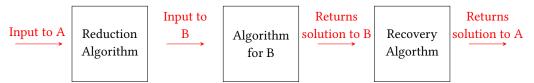
• We say that a problem A *reduces in polynomial time* to problem B if you can use any efficient algorithm for B to efficiently solve A. Mathematically, we'd write this as

$$A \leq_{p} B$$

Another way to say this is that "A's difficulty is less than B's difficulty", or that "A is at most as hard as B."

14.4.1 Zero Sum Games

- Recall the input: a payoff matrix M, and we want to output the Row player's optimal strategy.
- **Theorem:** zero sum games \leq_p Linear programming
 - In essence, we transfer the ZSG into a linear programming problem via a reduction algorithm. Then, take
 the LP input and run the LP algorithm on it (which is polynomial time), then run a recovery algorithm
 to go from the LP back to the ZSG.
 - As a general picture, we have



14.4.2 Rudrata Cycle vs. Min TSP

- It can also be shown that finding a Rudrata cycle reduces to the min-TSP problem, since both problems require us to find tours, then finding a Rudrata cycle is at most as hard as min-TSP.
- This is because finding a Hamiltonian tour is effectively finding a cycle on this graph that visits all vertices, which is exactly what min-TSP is doing except with more restrictions.

15 Reductions II

• Recap: Two computational problems A and B, and A reduces (in polynomial time) to B is written as $A \leq_p B$. This means that if an algorithm exists to solve B in polynomial time, then that same algorithm can be used to solve A in polynomial time.

Is this restricted to only polynomial time? Shouldn't any feasible algorithm that solves *B* also solve *A*?

Why are we so concerned about polynomial time? Do similar problems exist if we define "efficient" to be exponential time?

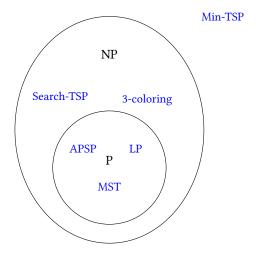
- Also recall the diagram we made to represent this process of reducing *A* to *B*, via a polynomial time reduction and recovery algorithm. Note that these two algorithms **must** execute in polynomial time.
 - We can prove that $A \leq_p B$ even if A, B are not known to be efficient.
- · We also saw two reductions: zero sum games reducing to LP, and Hamiltonian cycle reducing to min-TSP.
- Transitivity: If $A \leq_p B \leq_p C$, then $A \leq_p C$.

15.1 Common mistakes in Reductions

• If we're asked to prove that $A \leq_p B$, we need to come up with an algorithm that takes A to B, not B to A. Make sure you check that you're proving the correct direction!

15.2 Landscape of Problems

• We're going to use the below diagram to show the problems:



- We're not going to prove this, but it has been shown that factoring reduces to the 3-coloring problem. Similarly, factoring also reduces to the Rudrata Cycle problem.
- It turns out that every problem in NP reduces to Rudrata cycle!

- These are the most difficult problems in NP, and it can be shown that every problem in NP reduces to an NP-complete problem
- NP-Hardness: A problem A is NP-hard if every problem B in NP reduces to A.
- NP-Completeness: A problem A is NP-complete if $A \in NP$ and A is NP-hard.
- Problems in NP that aren't NP-complete are called an NP-intermediate problem
- Fact: Given two problems that are NP-complete, then $A \leq_p B$ and $B \leq_p A$. So this means that you can basically think of A and B are basically equivalent problems.

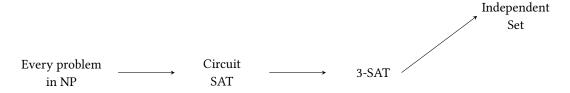
Is this a biconditional?

- There are thousands upon thousands of NP-complete problems, and by notion of reduction, they are (in some sense) the same problem.
- This also means that if there exists a polynomial time algorithm for any NP-problem, then this would imply that P = NP.

How is it that if P = NP then every problem becomes NP-complete?

15.3 Proving NP-Completeness

- Cook-Levin Theorem: showed that every problem in NP reduces in polynomial time to a circuit SAT problem.
- It can then be shown that circuit-SAT reduces to 3-SAT, making 3-SAT an NP-complete problem. In terms of a diagram:



Finish this Diagram Later

• To show that a problem is NP-complete, we first show that $A \in NP$, then pick some problem B that is known to be NP-complete and show that $B \leq_{p} A$.

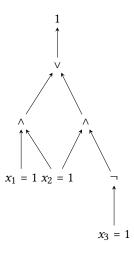
What if we show that $A \leq_p B$?

We don't need to, since $A \leq_p B$ is true already because B is an NP-complete problem!

15.4 Circuit SAT

- A Boolean circuit is a directed acyclic graph with:
 - Input nodes x_1, \ldots, x_n
 - one output node, with an output C(x)
 - gates marked OR, AND, NOT: ∨, ∧, ¬

A possible graph is:



- The input to circuit SAT is a circuit C with n inputs and m referring to the number of gates. We want to output an assignment of $(x_1, ... x_n)$ such that $x_i \in \{0, 1\}$ such that C(x) = 1.
- By the Cook-Levin theorem, circuit-SAT is NP-complete. As for a bit of intuition on why this is true, you can think of every problem as basically a collection of logical inputs, which basically means that every problem can be reduced to some complex circuit of logical gates.

15.4.1 3-SAT

- Here, we're given n Boolean variables x_1, \ldots, x_n such that $x_i \in \{0, 1\}$, and $m \le 3$ variable clauses that join the variables together.
- We want to output an assignment of x_1, \dots, x_n that satisfies all the clauses.
- **Theorem:** Circuit-SAT reduces to 3-SAT

Proof: Suppose we're given an input to a circuit-SAT problem.