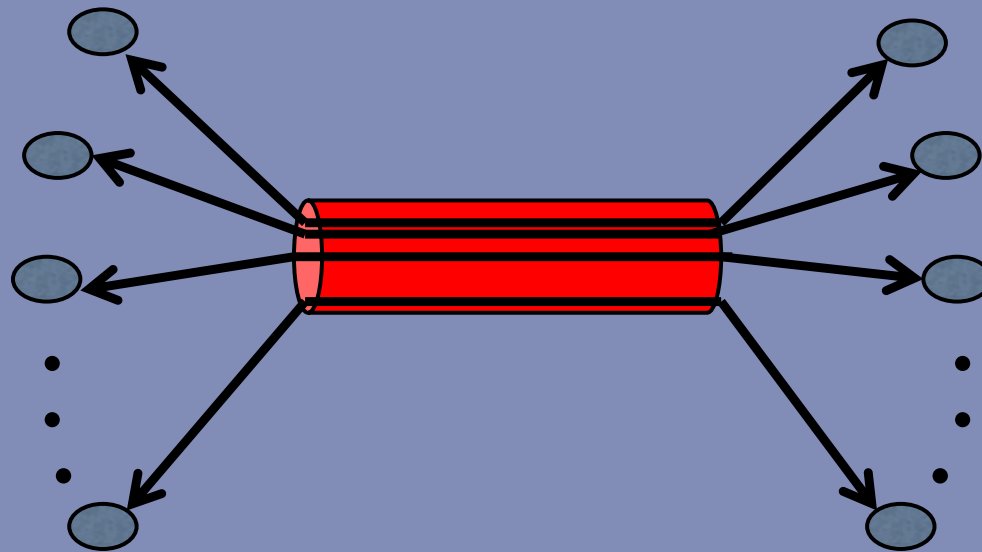


# Multiplexing

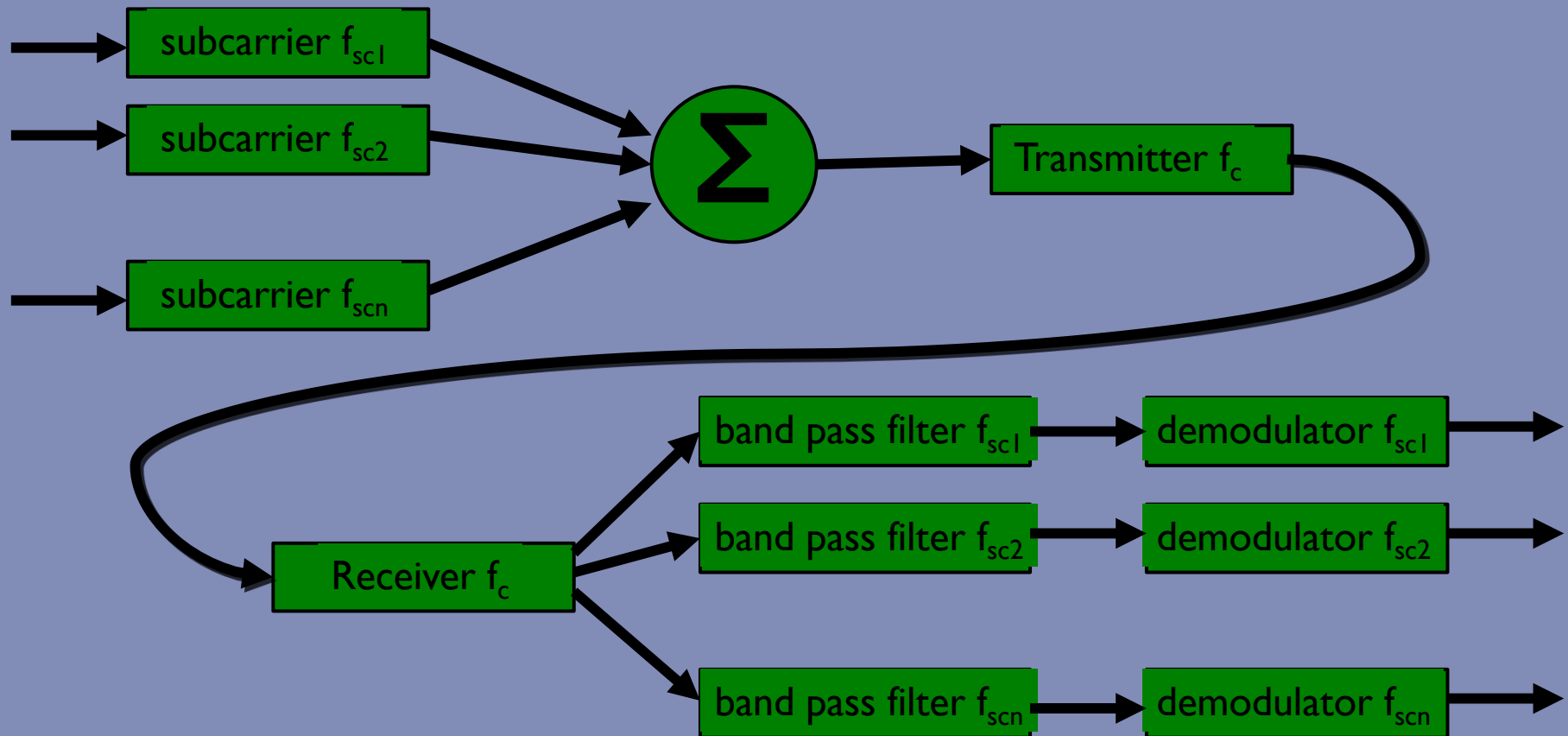
”From one channel to multiple channels”

How to share one medium while facilitating multiple channels of communication:



**Frequency Division and Time Division Multiplexing**

# Frequency Division Multiplexing (FDM alla FSK)



# Standards

# voice channel	bandwidth	spectrum	US/AT&T	ITU_T
12	48 kHz	60-108 kHz	Group	Group
60	240 kHz	312-552 kHz	Super Group	Super Group
300	1,23 MHz	812-2044 kHz	Super Group	Master Group
600	2,52 MHz	564-3084 kHz	Master Group	Master Group
900	3,87 MHz	8,52-12.39 MHz	Master Group	Super Master Group
3600	16,98 MHz	0.56-17,55 MHz	Jumbo Group	Jumbo Group
10800	57,44 MHz	3,12-60,57 MHz	Jumbo Group Multiplexed	Jumbo Group Multiplexed

# Example: ADSL

**ADSL** Asymmetric Digital Subscriber Line

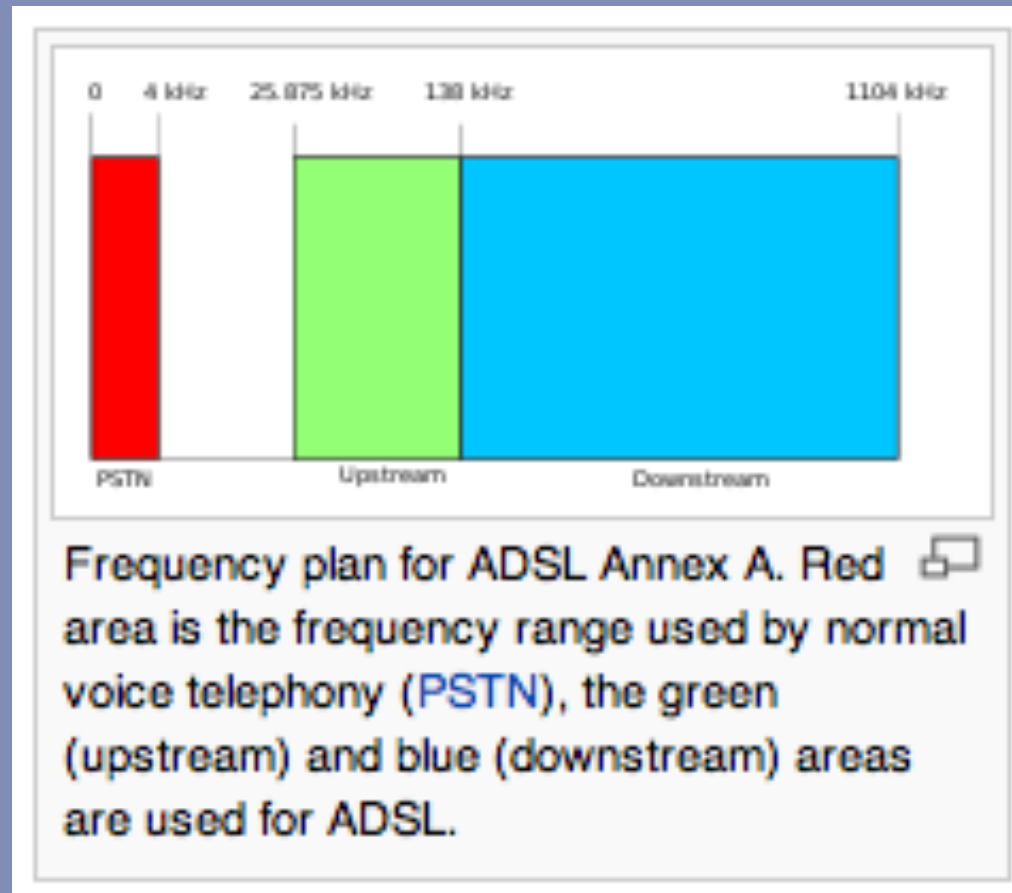
Originally for Video-on-Demand: less control  
going up – lots of image going down

➔ Very similar to internet usage !!!!!!!

**Multiple “regular” phone connections at  
the same time on which QAM (Quadratic  
Amplitude Modulation) is implemented**

- Reserve lowest 25 kHz for Voice (POTS, Plain Old Telephone Service) **25 instead of 5 to prevent cross talk between voice&data**
- Facilitate two bands: small upstream / big downstream
- Use FDM within upstream and downstream band

# ADSL 4 kHz channels

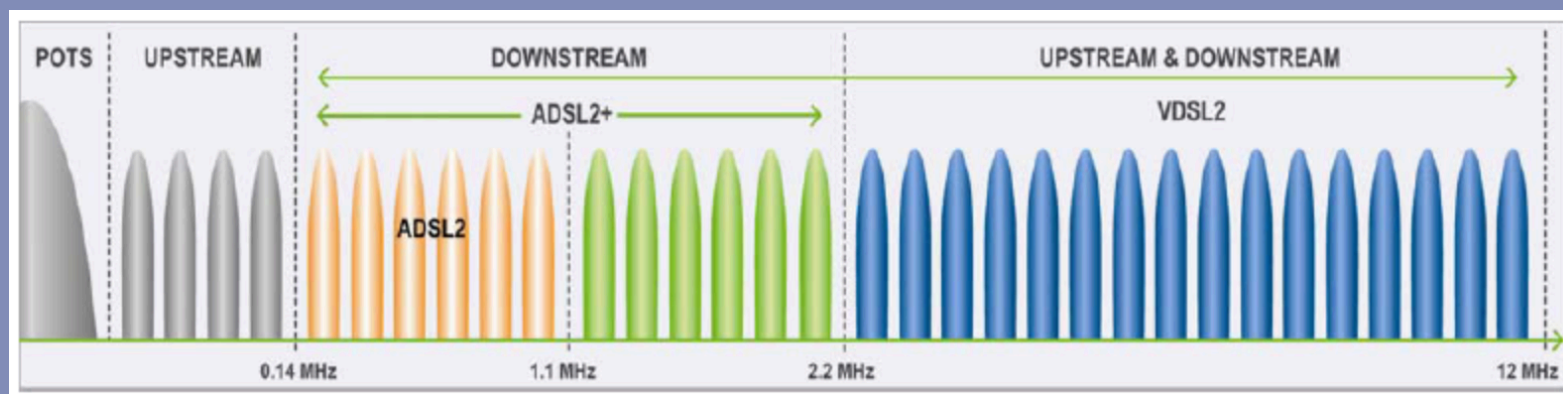


# ADSL standards

Version	Standard name	Common name	Downstream rate	Upstream rate	Approved in
ADSL	ANSI T1.413-1998 Issue 2	ADSL	8.0 Mbit/s	1.0 Mbit/s	1998
ADSL	ITU G.992.1	ADSL (G.DMT)	12.0 Mbit/s	1.3 Mbit/s	1999-07
ADSL	ITU G.992.1 Annex A	ADSL over POTS	12.0 Mbit/s	1.3 Mbit/s	2001
ADSL	ITU G.992.1 Annex B	ADSL over ISDN	12.0 Mbit/s	1.8 Mbit/s	2005
ADSL	ITU G.992.2	ADSL Lite (G.Lite)	1.5 Mbit/s	0.5 Mbit/s	1999-07
ADSL2	ITU G.992.3	ADSL2	12.0 Mbit/s	1.3 Mbit/s	2002-07
ADSL2	ITU G.992.3 Annex J	ADSL2	12.0 Mbit/s	3.5 Mbit/s	
ADSL2	ITU G.992.3 Annex L	RE-ADSL2	5.0 Mbit/s	0.8 Mbit/s	
ADSL2	ITU G.992.4	splitterless ADSL2	1.5 Mbit/s	0.5 Mbit/s	2002-07
ADSL2+	ITU G.992.5	ADSL2+	24.0 Mbit/s	1.1 Mbit/s	2003-05
ADSL2+	ITU G.992.5 Annex M	ADSL2+M	24.0 Mbit/s	3.3 Mbit/s	2008

# VDSL

- ADSL has been succeeded over the last years by **Very High Bit Rate Digital Subscriber Line (VDSL)**, and **Very High Bit Rate Digital Subscriber Line 2 (VDSL2)**
- Main difference: up to 12 MHz FDM instead of 1 MHz and use “noise cancellation techniques”



Version	Standard name	Common name	Downstream rate ⇄	Upstream rate ⇄	Approved on ⇄
VDSL	ITU G.993.1	VDSL	55 Mbit/s	3 Mbit/s	2001-11-29
VDSL2	ITU G.993.2	VDSL2	100 Mbit/s	100 Mbit/s	2006-02-17
VDSL2-Vplus	ITU G.993.2 Amendment 1 (11/15)	VDSL2 Annex Q VPlus/35b	300 Mbit/s	100 Mbit/s	2015-11-06



# DOCSIS

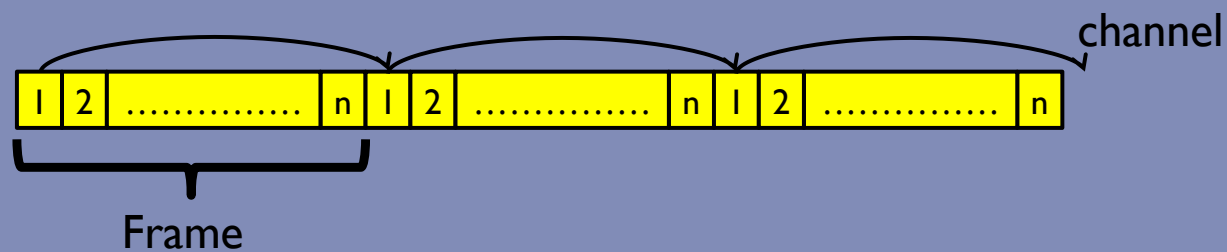
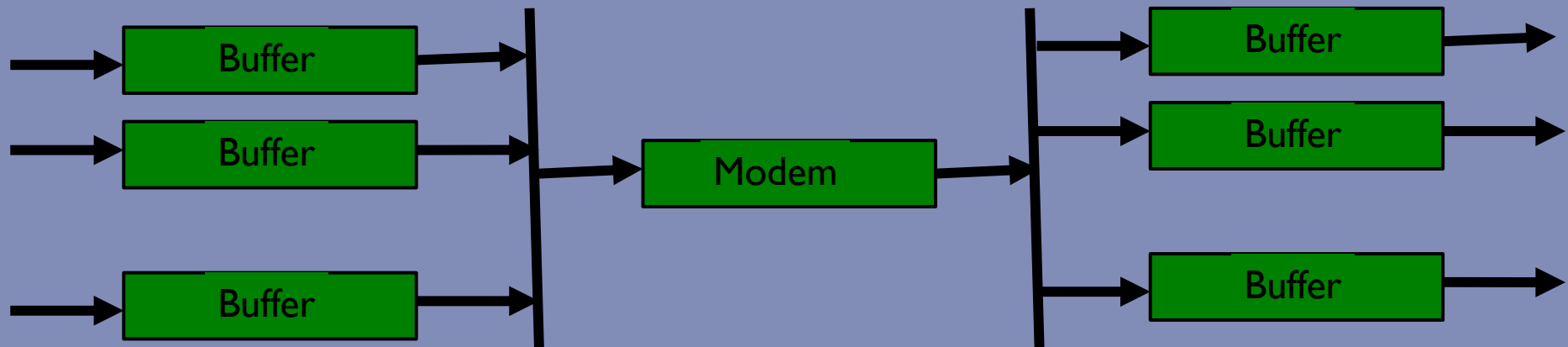
In the meantime we also have a Cable alternative for home usage:

## Data Over Cable Service Interface Specification

which relies on 256-QAM modulation for downstream and 64-QAM for upstream (DOCSIS 3.0), and 4096-QAM modulation for OFDM/OFDMA (DOCSIS 3.1). OFDM is Orthogonal FDM also used in VDSL. OFDMA is OFD Multiple Access.

DOCSIS version <sup>[13]</sup>	Production date	Maximum downstream capacity	Maximum upstream capacity	Features
1.0	1997	40 Mbit/s	10 Mbit/s	Initial release
1.1	2001	40 Mbit/s	10 Mbit/s	Added VOIP capabilities, standardized the DOCSIS 1.0 QoS mechanisms
2.0	2002	40 Mbit/s	30 Mbit/s	Enhanced upstream data rates
3.0	2006	1.2 Gbit/s	200 Mbit/s	Significantly increased downstream/upstream data rates, introduced support for IPv6, introduced channel bonding
3.1	2013	10 Gbit/s	1–2 Gbit/s	Significantly increased downstream/upstream data rates, restructured channel specifications
3.1 Full Duplex	2017	10 Gbit/s	10 Gbit/s	Introduces support for fully symmetrical speeds

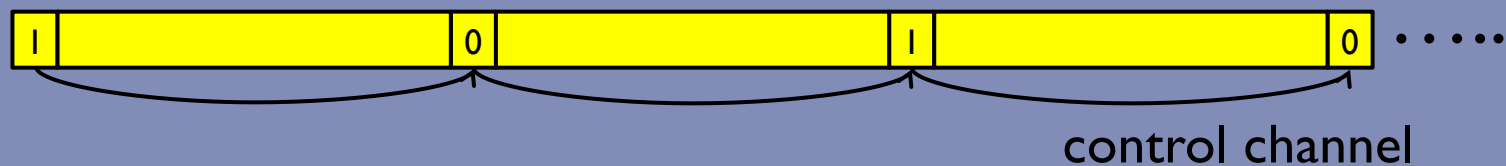
# Time Division Multiplexing(TDM)



**Synchronous TDM:** not synchronous but frames are fixed and slots are always filled

# How is framing implemented

## Added digit framing



## Pulse Stuffing

Frequency  $>$   $\Sigma$  (freq. of the sources)

additional bits are added at fixed position in the frame

# Relationship with data link framing



$F_1 F_2 A_1 F_2 C_1 A_2 d_1 C_2 d_1 d_2 \dots$

for



$F_1 f_1 f_1 d_1 d_1 d_1 C_1 A_1 F_1$  in characters  
↓ flag    ↓ FCS    ↓ information    ↓ control    ↓ address

# Example: ISDN (Integrated Services Digital Network)

## DS-1 transmission format

**Voice** → PCM (8000 samples per second, 8-bit)

TDM-frame = 24 (channels) × 8 bits + 1 (frame bit)  
= 193 bits

Data rate: 8000 × 193 = **1.544 Mbps**

**DATA** only 23 out of 24 channels used,  
24<sup>th</sup> channel has special SYNC BYTE  
per channel 1 bit for user/system data

→ 7 × 23 × 8000 = 56 × 23 kbps = **56 kbps** p. channel

# Standards Telephony

US/JAPAN			ITU-T		
	# channels	Mbps	Level	#channels	Mbps
DS-1	24	1.544	1	30	2.048
DS-1C	48	3.152	2	120	8.448
DS-2	96	6.312	3	480	34.368
DS-3	672	44.736	4	1920	139.264
DS-4	4032	274.176	5	7680	565.148

# Currently being phased out!!

- Initially used heavily by the broadcasting industry. Also introduced for home usage. For instance, 25 million installments in Germany in 2003
- Became obsolete due to ADSL (in 2018 it will be phased out in Germany)
- ***It Still Does Nothing, Innovations Subscribers Don't Need***

# Network **SWITCHING**

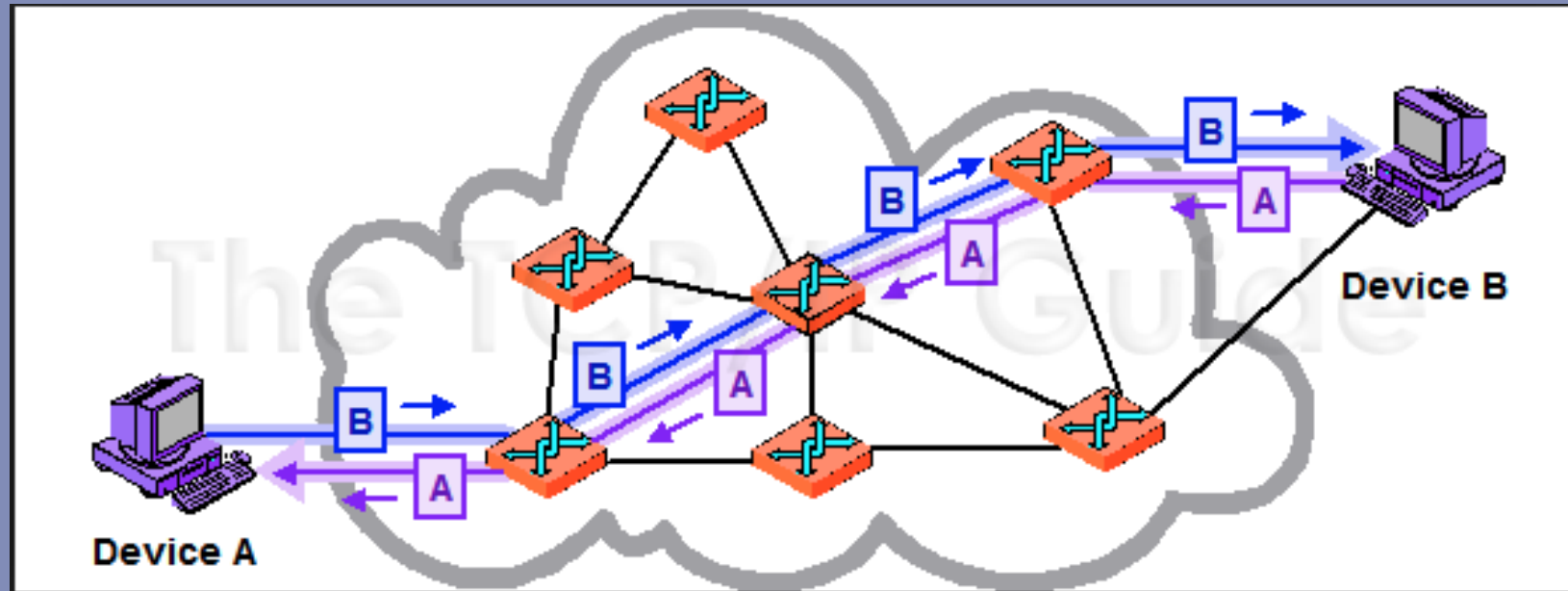
Next to multiplexing: **switching** is required to realize multi to multi connections

Especially needed in Wide Area Networks (WAN)

Also present in Local Area Networks (LAN) or in multi processors architectures.



# Circuit Switching



1. A dedicated path between two end stations is realized or channel (TDM/FDM)
2. Data is being transmitted (Switches don't inspect data)
3. Path is broken up

# Circuit switching

## IMPORTANT CHARACTERISTIC:

### **BLOCKING VS NON-BLOCKING**

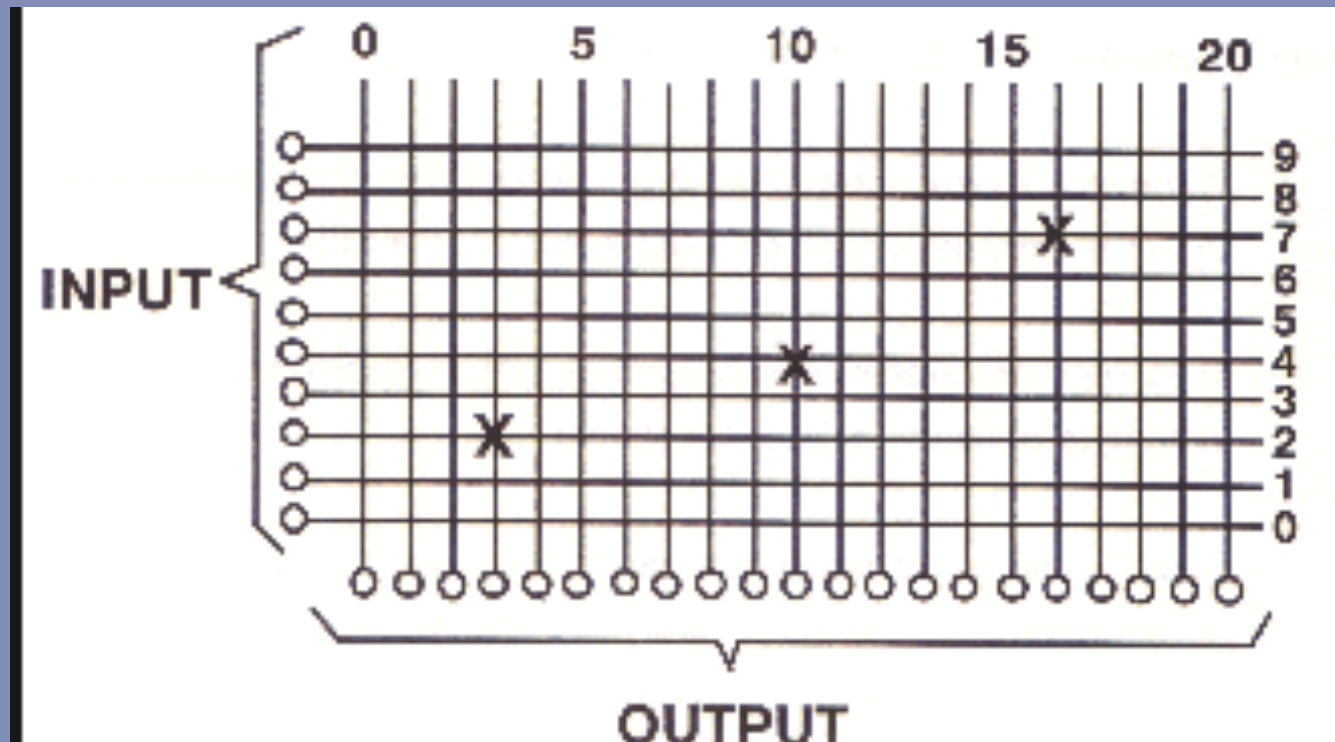


Connection cannot be realized  
because all paths are occupied



Connection can always be realized and at any time

# Space Division Switching: non-blocking **Crossbar Switch**

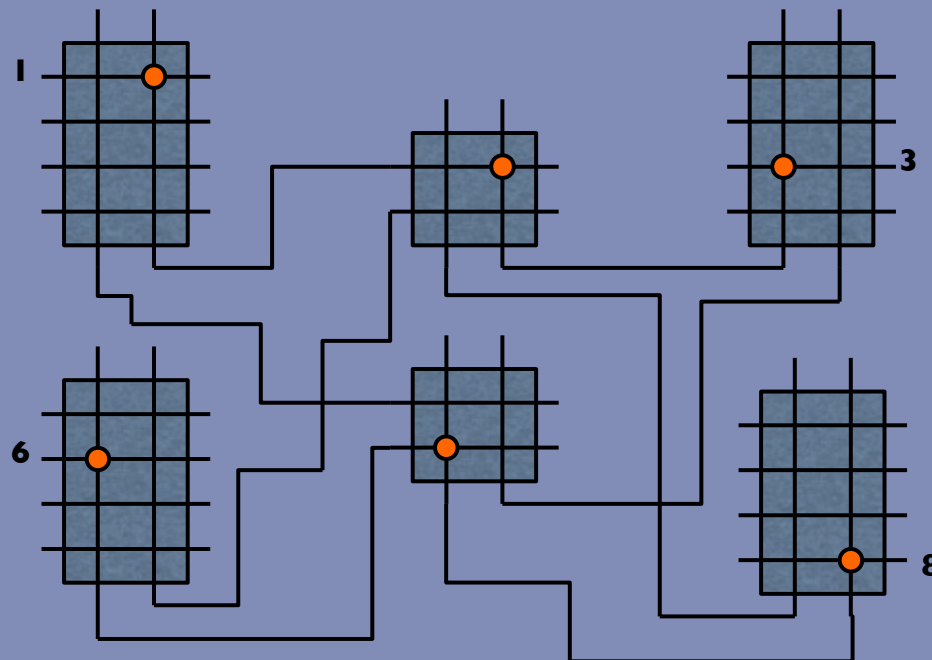


**Very costly:  $N^2$  switches**

# Multistage Networks

Use many small crossbar switches and connect them wisely.

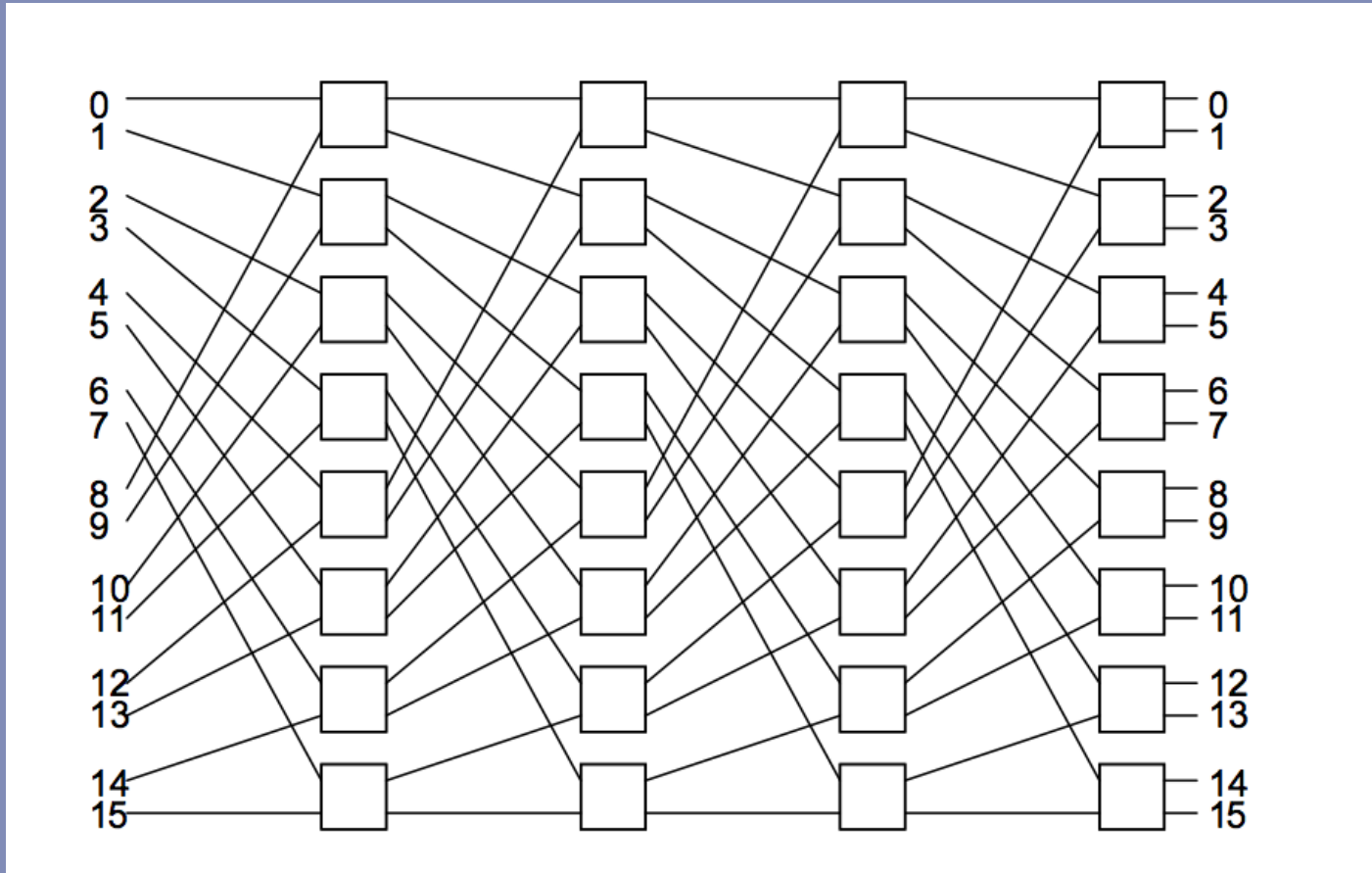
Blocking can occur!!!!



5 cannot be  
connected to 2

# Omega Networks

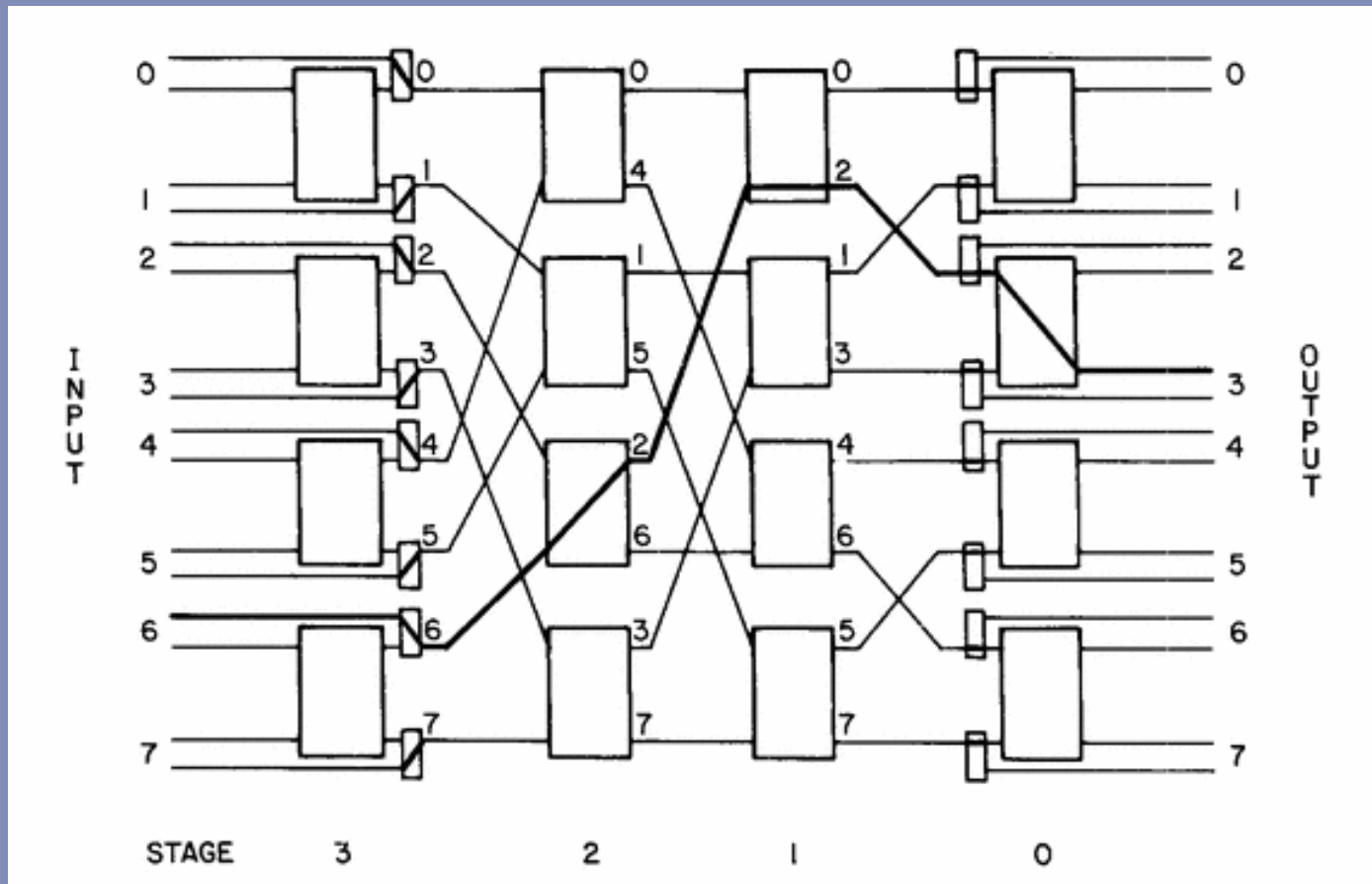
(based on Perfect Shuffles)



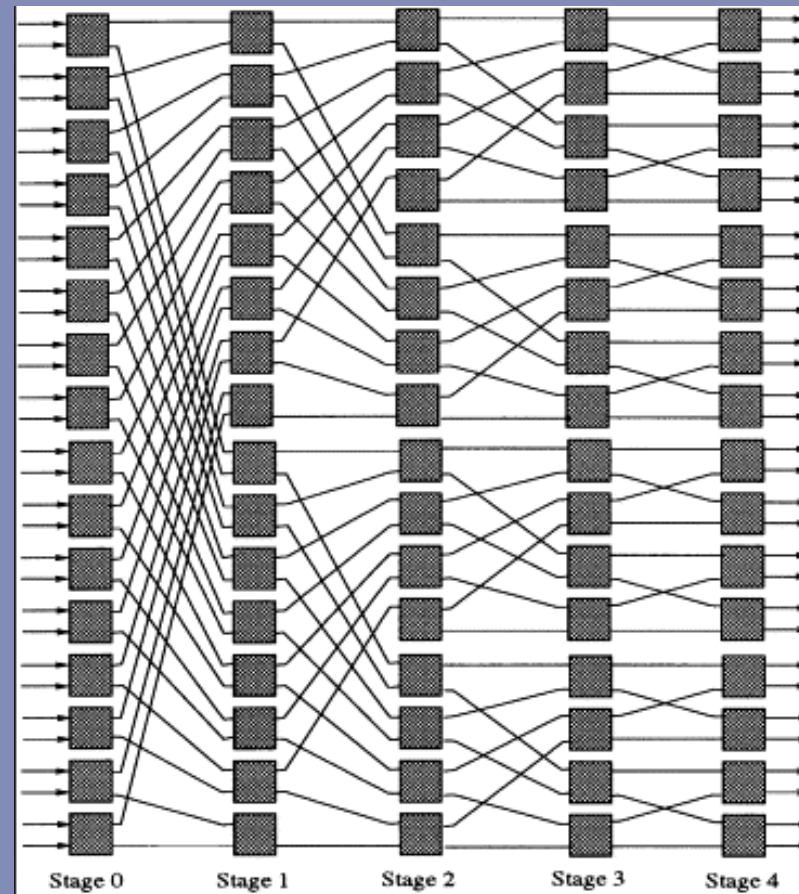
**$2\log N + 1$  stages: non blocking with  $O(N\log N)$  switches)**

# Variants of PS networks

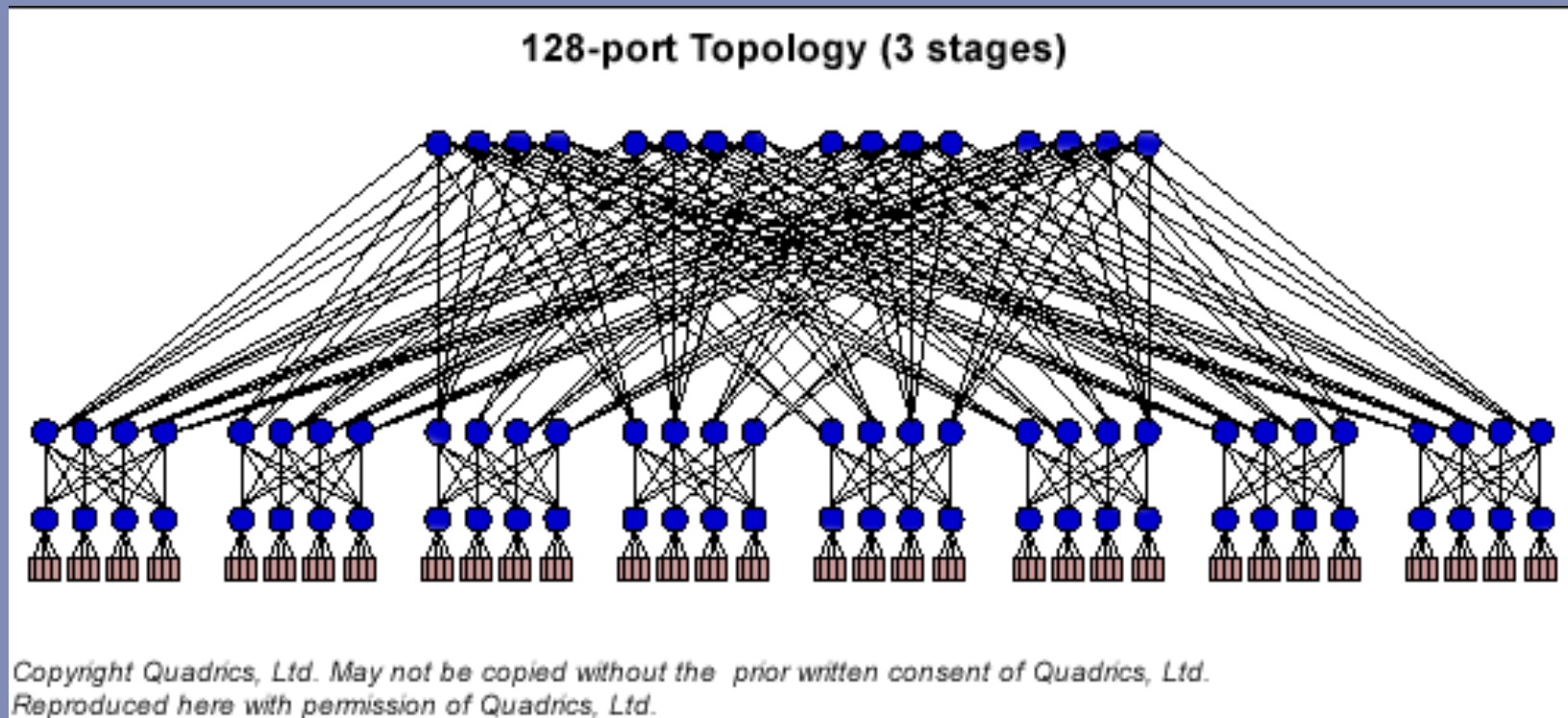
## Cube Network



# Butterfly Network



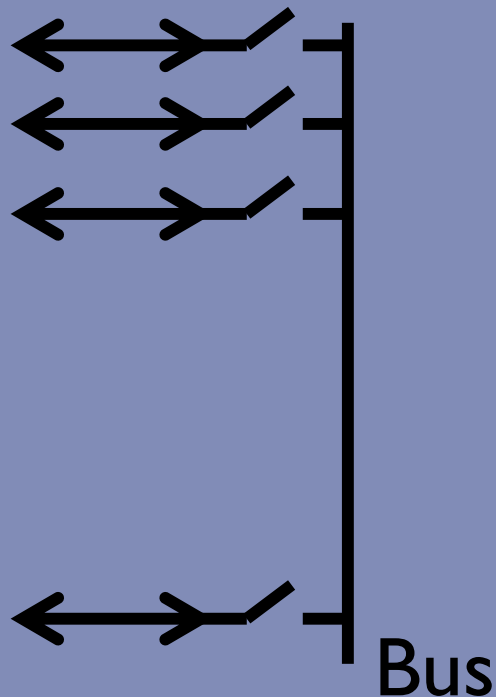
# Fat Tree Network





# Time Division Switching

**Do not confuse with TDM !!!**



Bandwidth of Bus  $>$   
 $\Sigma$  indiv. bandw.  
Then non-blocking!

# Routing in Circuit Switched Networks

- Alternate Routing
  - Each switching node has its own routing table

	First Choice	Second Choice
A to B	Via switch i	Via switch j
A to C	Via switch j	Via switch k

- Fixed Alternate Routing
  - Routing tables do not change
- Dynamic Alternate Routing
  - Depending on time (e.g. time of the day) routing tables will change
- Adaptive Routing
  - Central Controller gets status of all switches and gives routing updates to all switches

# Packet Switching

**Data is sent by packets (usually < 1000 octets),  
Every switching nodes has buffers**

- **Datagram**

Every packet is routed independently

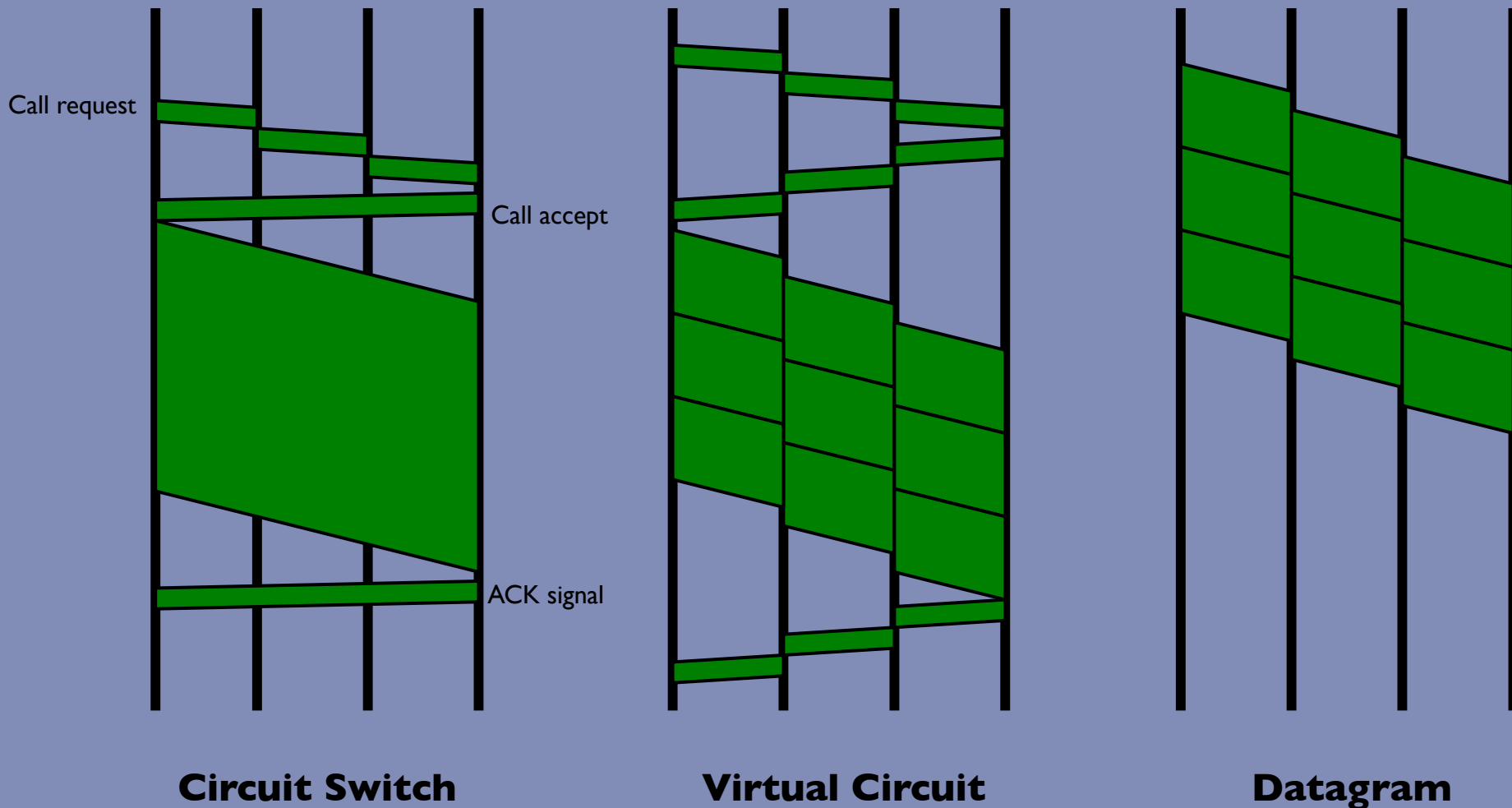
→ As a consequence packets can arrive out of order

- **Virtual Circuit (Wormhole)**

Before communication is initiated a Call-Request packet is sent on the network, which fixates a virtual path between sender and receiver.

→ Packets arrive in order, but it not as flexible as datagram

# Summarizing

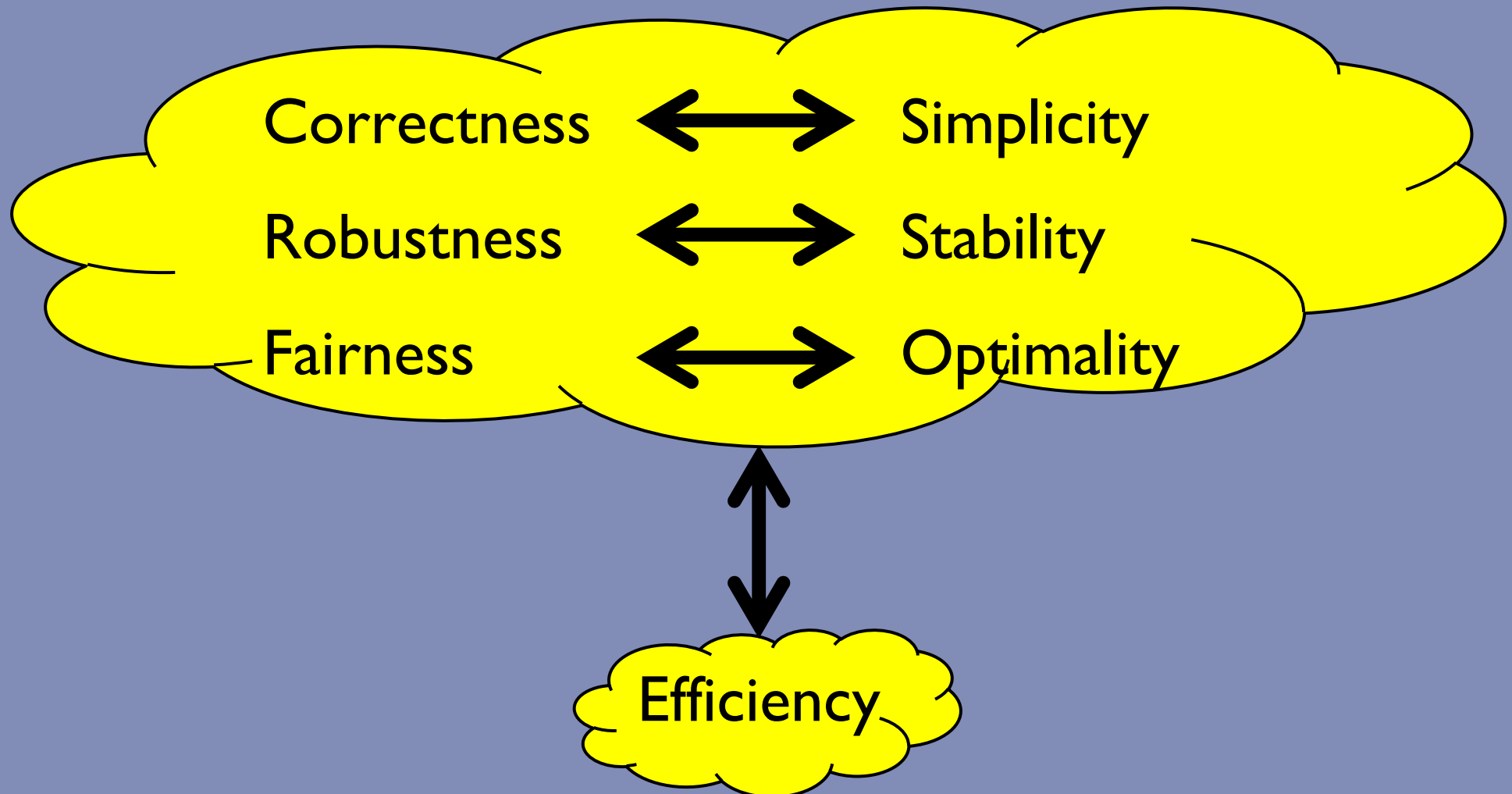


# Different Combinations

- External Virtual Circuit & Internal Virtual Circuit
- External Virtual Circuit & Internal Datagram
- External Datagram & Internal Virtual Circuit
- External Datagram & Internal Datagram

**Which one makes sense?**

# Routing Trade Offs



# Routing for Packet Switched Networks

Like circuit switching can we differentiate between: **Fixed Routing** and **Alternate Routing**

➔ **No difference between datagram and virtual circuit**

## **Random Routing**

Every node chooses randomly outgoing link, based on some prob. Distribution, e.g.

$$P_i = R_i / \sum R_j$$

with  $R_i$  data rate possible on link  $i$ .

**Flooding** (very inefficient, very reliable)

Every node puts incoming packet on every outgoing link, except the incoming link

→ Exponential growth

1. Every node logs all the packets

If packets arrives a second time: **discard**

2. Every packet, contains counter: hop-count

If hop-count > threshold: **discard**

**Adaptive Routing** (Central vs Distributed)

Every node gets network status information

- Local, e.g. queue length of the outgoing links
- Adjacent nodes
- All nodes



# ARPANET

based on Adaptive Routing, Distr. & Adjacent Nodes

## First Version: 1969

Based on **Bellman-Ford Algorithm**

Every node  $i$  has two vectors:

$$D_i = \begin{pmatrix} d_{i1} \\ d_{i2} \\ \vdots \\ d_{iN} \end{pmatrix} \quad S_i = \begin{pmatrix} s_{i1} \\ s_{i2} \\ \vdots \\ s_{iN} \end{pmatrix}$$

$N = \text{\#nodes}$   
 $d_{ij} = \text{estimated delay from node } i \text{ to } j$   
 $s_{ij} = \text{next node on the route } i \text{ to } j$

Every 128 ms every node exchanges delay vector with adjacent nodes.

Then every node  $k$ :  $d_{kj} = \text{Min}_{i \in A} [d_{ij}^{\text{new}} + d_{ki}]$  and  $s_{kj} = i$ ,  
the node  $i$  which minimizes  $d_{kj}$ . Link delays are the queue length for that link.

Disadvantages:      Link delays were not accurate  
                                 Thrashing would occur

## 2<sup>de</sup> Generation (1979)

Every node:

- Timestamp on incoming message (arrival time)
- Departure time recorded
- If pos. ACK is received:  $\text{delay} = (\text{dept. time} - \text{arrival time})$

Every 10 sec: every node computes the average delay per link

If delay is different: **FLOODING** is used to inform all the other nodes

Every node gets status of the whole network!!!!!!!

Dijkstra's shortest path algorithm is used to compute new routing table

## 3<sup>de</sup> Generation (1987)

### When load is heavy:

Observed delay under old routing  $\neq$  delay under new routing

→ Oscillation effects

→ Instead of BEST route: a “good” route

### Smoothing of link costs (delays)

Every 10 seconds:

1. (Queuing theory)  $\rho = 2(s-t)/(s-2t)$ , with

$\rho$  = link utilization

$t$  = measured delay

$s$  = service time

2.  $U(n+1) = 0.5\rho(n+1) + 0.5U(n)$ ,  $U(n)$  average utilization

3. New delays are computed based on  $U(n)$ ,

terrestrial: 1 Hop for  $U(n) < 0.5$ , 2 Hops for  $U(n) > 0.8$ , and  $1 + (U(n) - 0.5) / 0.3$

Hops otherwise

satellite: 2 Hops for  $U(n) < 0.8$

Otherwise the same as 2<sup>de</sup> generation