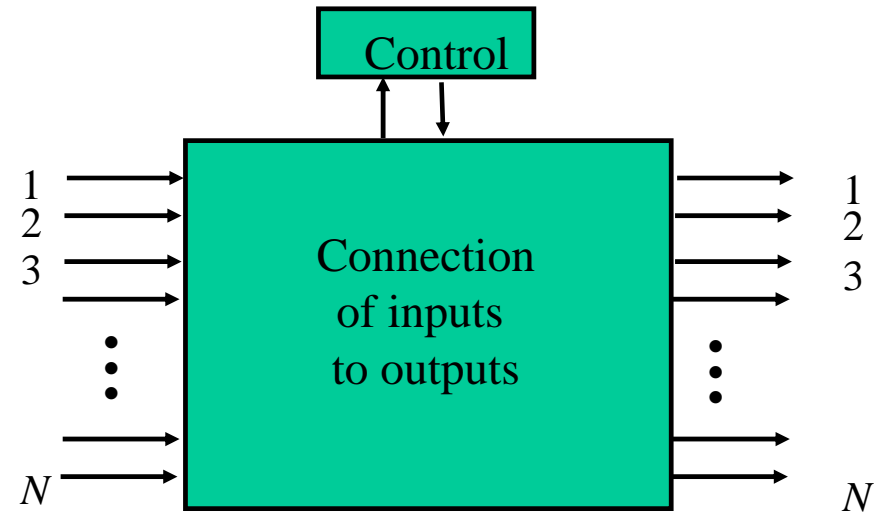
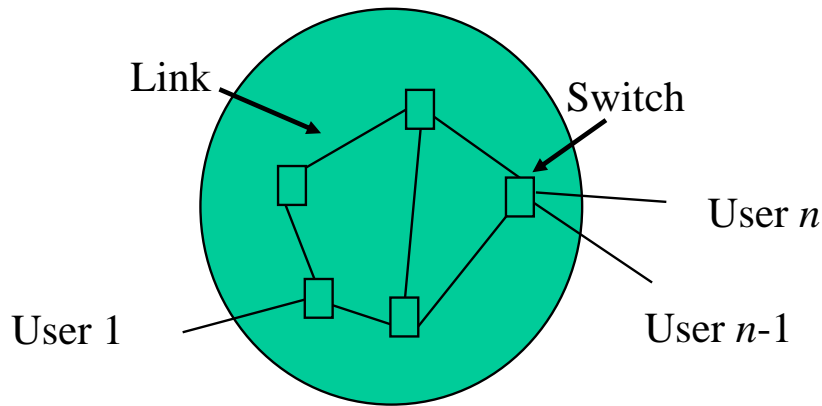


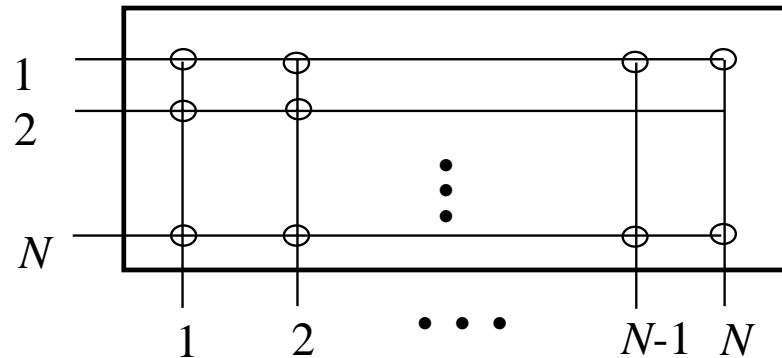
# Circuit Switches

- Providing connectivity between users across a network



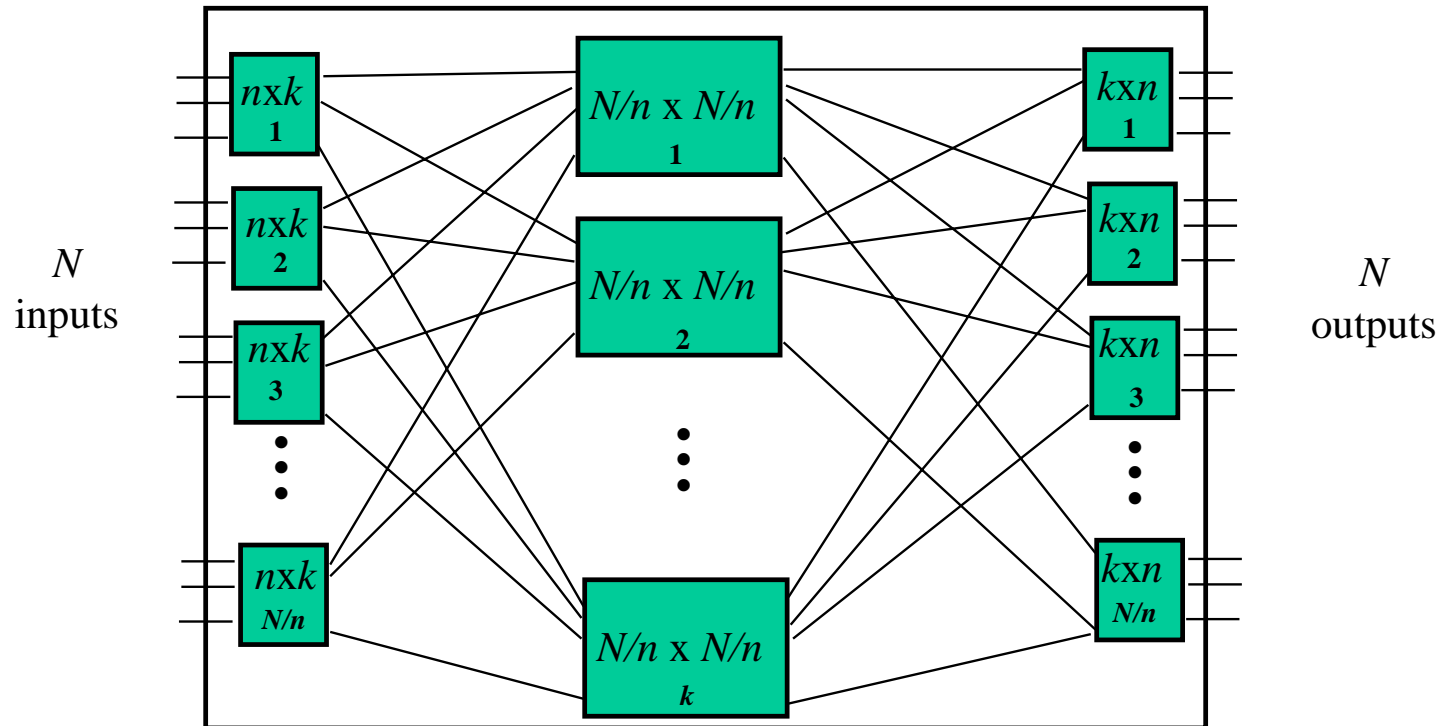
- a sequence of switches must be set across the network to set up a circuit
- cases:
  - » switching one input to one output
  - » switching a flow of multiplexed signals
    - must be demultiplexed first, in principle
    - but switching time-division multiplexed signals possible on-the-fly

- Space-Division switches : a Crossbar switch:



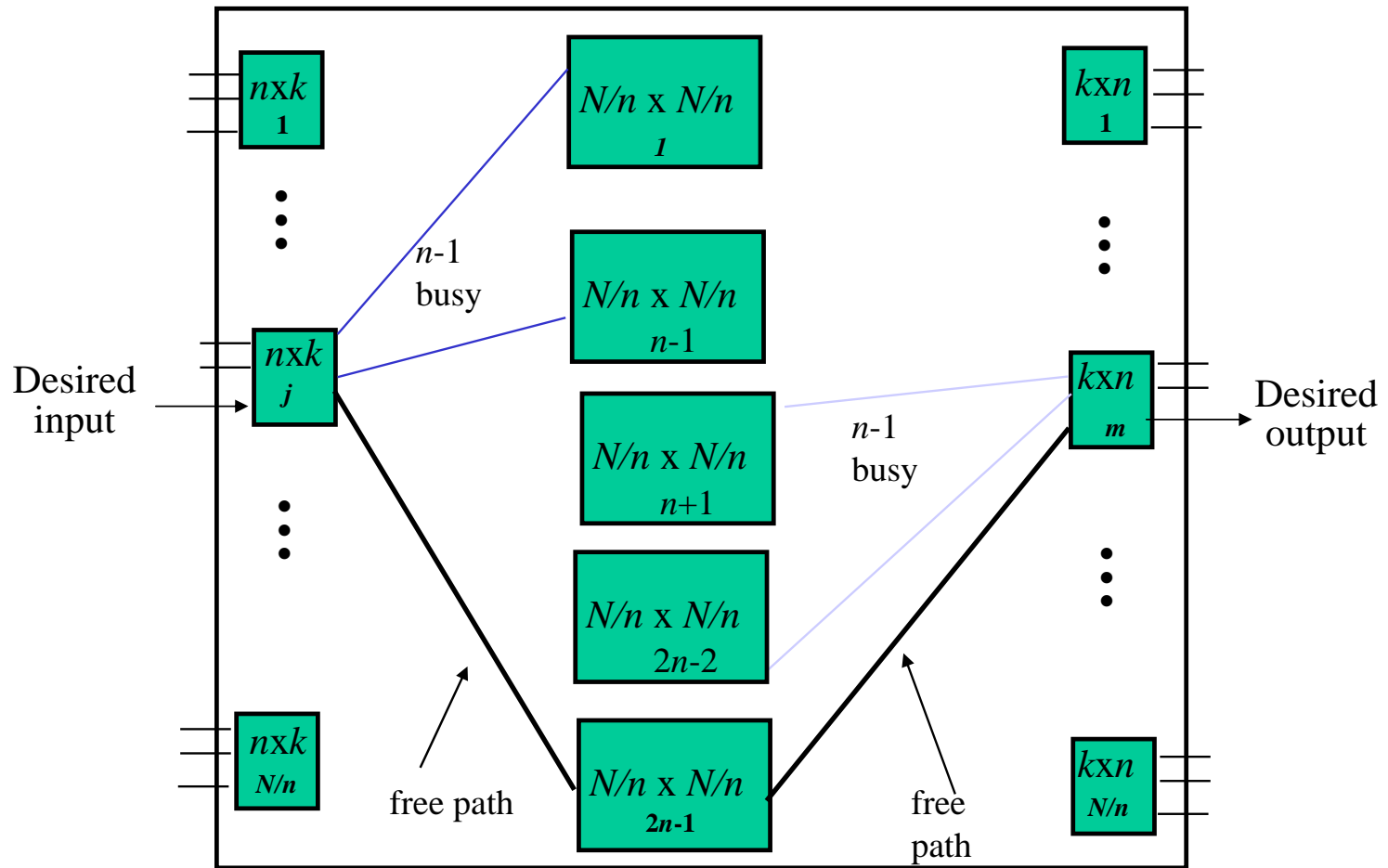
- » an  $N \times N$  array of crosspoints
- » can connect any input to any available output
- » when a connection request received, correspond crosspoint closed
- » *nonblocking* – connection requests never need to be denied through lack of connectivity resources
  - only denied when outgoing line already busy
- » complexity in terms of the number of crosspoints =  $N^2$ 
  - grows very quickly with number of input and output ports
  - complexity can be reduced by using multistage switches

- Multistage switches
  - e.g. a 3-stage switch:



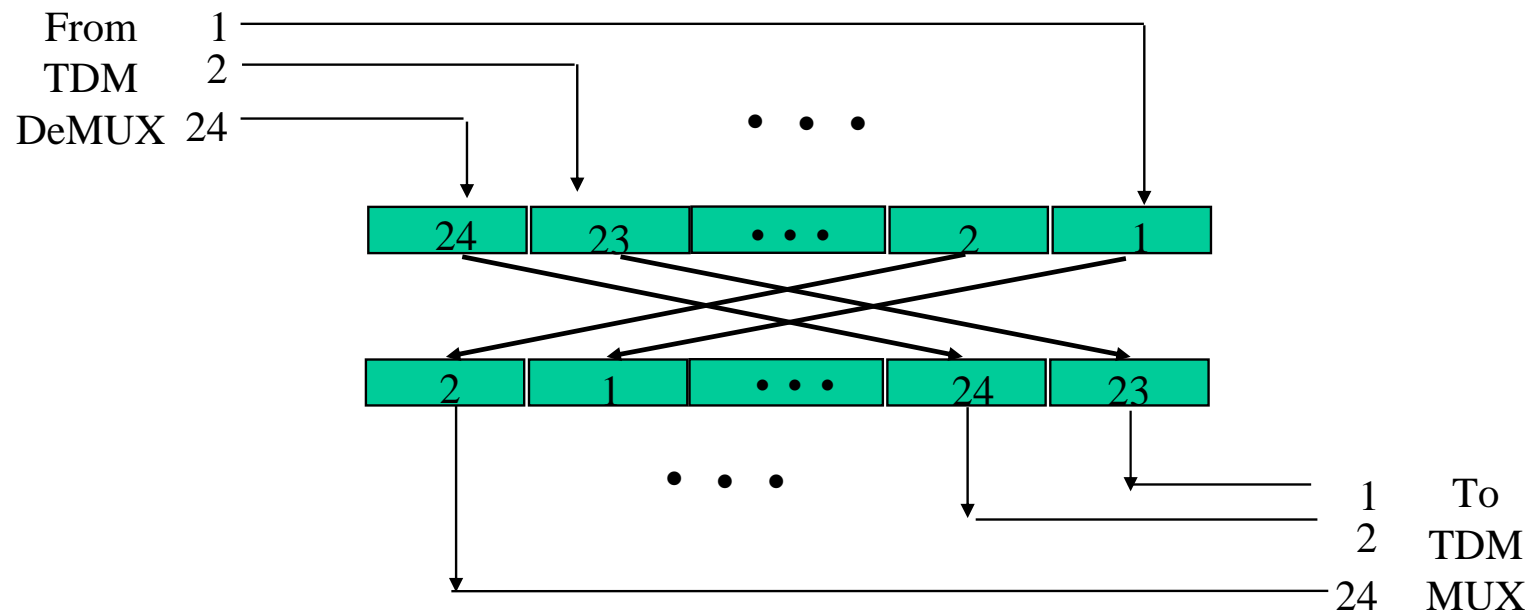
- $N$  inputs grouped into  $N/n$  groups of  $n \times k$  switches
- each of the  $n \times k$  first-stage switch output ports is connected to an input port of one of the  $N/n \times N/n$  intermediate switches
- the output ports of an intermediate-level switch are connected to each of the  $k \times n$  third-stage switches

- in effect, each set of  $n$  input lines shares  $k$  possible paths to any one of the switches at the last stage
  - » 1<sup>st</sup> path goes through 1<sup>st</sup> intermediate switch
  - » 2<sup>nd</sup> path goes through 2<sup>nd</sup> intermediate switch etc.
- not necessarily non-blocking
  - » if  $k < n$ , as soon as a first-stage switch has  $k$  connections, its other connections will be blocked
- when is a multistage switch non-blocking?
  - » consider any desired input and any desired output
  - » worst case for the desired input is when all the other  $(n-1)$  inputs in its group have already been connected
  - » worst case for the desired output is when all the other  $(n-1)$  outputs in its group have already been connected
  - » the worst-case situation that maximises the number of intermediate-level switches is when each existing connection uses a different intermediate switch
  - » i.e. maximum number of intermediate switches not available to connect the desired input to the desired output is  $2(n-1) = 2n-2$
  - » if  $k = 2n-1$ , a single path is still available for the desired connection
    - i.e.  $k = 2n-1$  is non-blocking



- number of crosspoints in a three-stage switch:
  - »  $N/n$  input switches x  $nk$  crosspoints/switch
  - »  $k$  intermediate switches x  $(N/n)^2$  crosspoints/switch
  - »  $N/n$  output switches x  $nk$  crosspoints/switch
  - »  $= 2Nk + k(N/n)^2$
- to make switch non-blocking,  $k = 2n-1$ 
  - » i.e. number of crosspoints =  $2N(2n-1) + (2n-1)(N/n)^2$
- number of crosspoints can be minimised by choice of group size  $n$ 
  - » differentiating wrt  $n$  gives a minimum when  $n \approx (N/2)^{1/2}$
  - » then minimum number of crosspoints =  $4N((2N)^{1/2} - 1)$
  - » grows at a rate proportional to  $N^{1.5}$
  - » i.e. less than the  $N^2$  of the crossbar switch
- when  $k < 2n-1$ , a nonzero probability that a connection request will be blocked

- Time-Division switches : Time-Slot interchange (TSI)
  - replaces crosspoint switches with reading and writing of a slot in memory
  - consider a T-1 time-division multiplexed signal carrying 24 voice channels
    - » 8000 bytes/sec per channel interleaved byte-by-byte
  - suppose voice assigned to slot 1 is talking to voice assigned to slot 23
    - » need to route incoming slot 1 to outgoing slot 23 and incoming slot 23 to outgoing slot 1, in each frame of 24 slots
    - » similarly if another pair of speakers assigned to slots 2 and 24



- each incoming byte is written into an array in memory as it arrives
- call setup procedure sets up a *permutation* table that controls the order in which bytes are read out of the array
- outgoing frame begins by reading contents of slot 23, then slot 24 and so on until slots 1 and 2 are read
- this procedure can connect any input to any available output
- frames come 8000 times/sec
- the time-slot interchange for the whole frame requires *one* write and *one* read per slot
- hence, maximum number of slots that can be handled is:

$$\frac{125\mu\text{s}}{2 \times \text{memory cycle time}}$$

- e.g. cycle time of 50ns gives a maximum of 1250 slots i.e. 625 connections
- telephone exchanges initially used space-division switches
  - » introduction of TSI switches led to significant cost savings
  - » crucial in the development of all-digital telephone switching networks



- Time-Space-Time switches

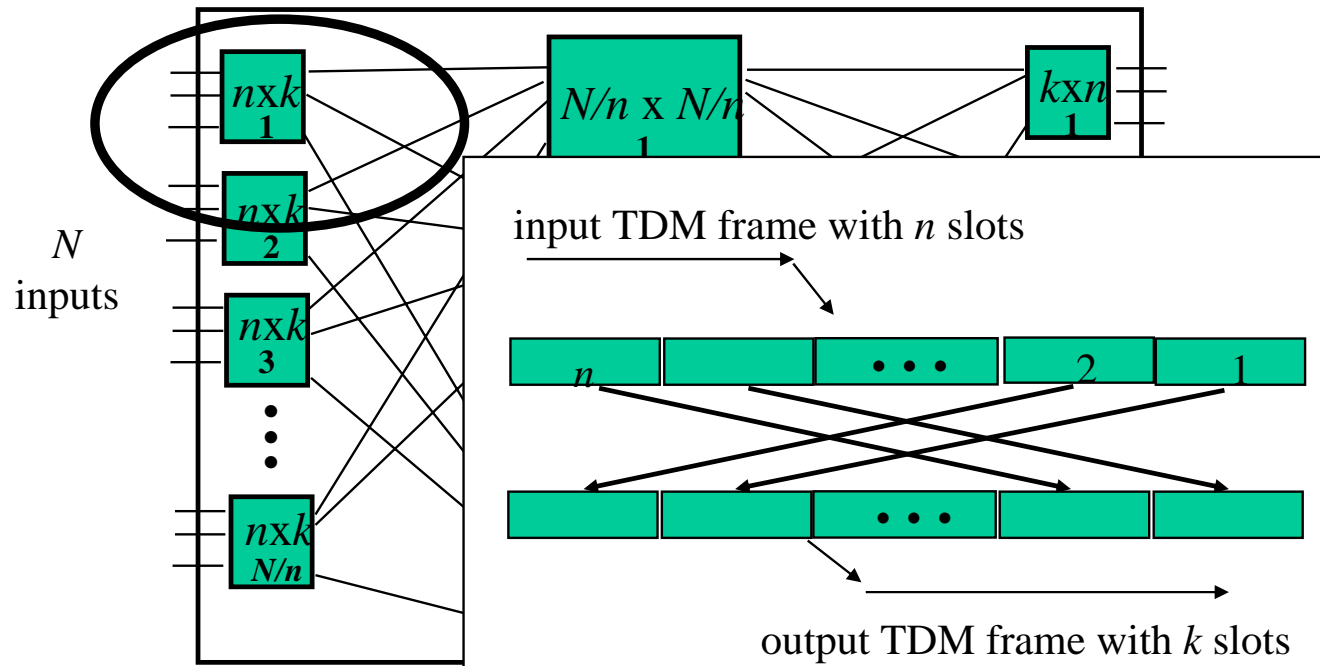
- a hybrid design of multistage switch

- » TSI switches at input and output stages

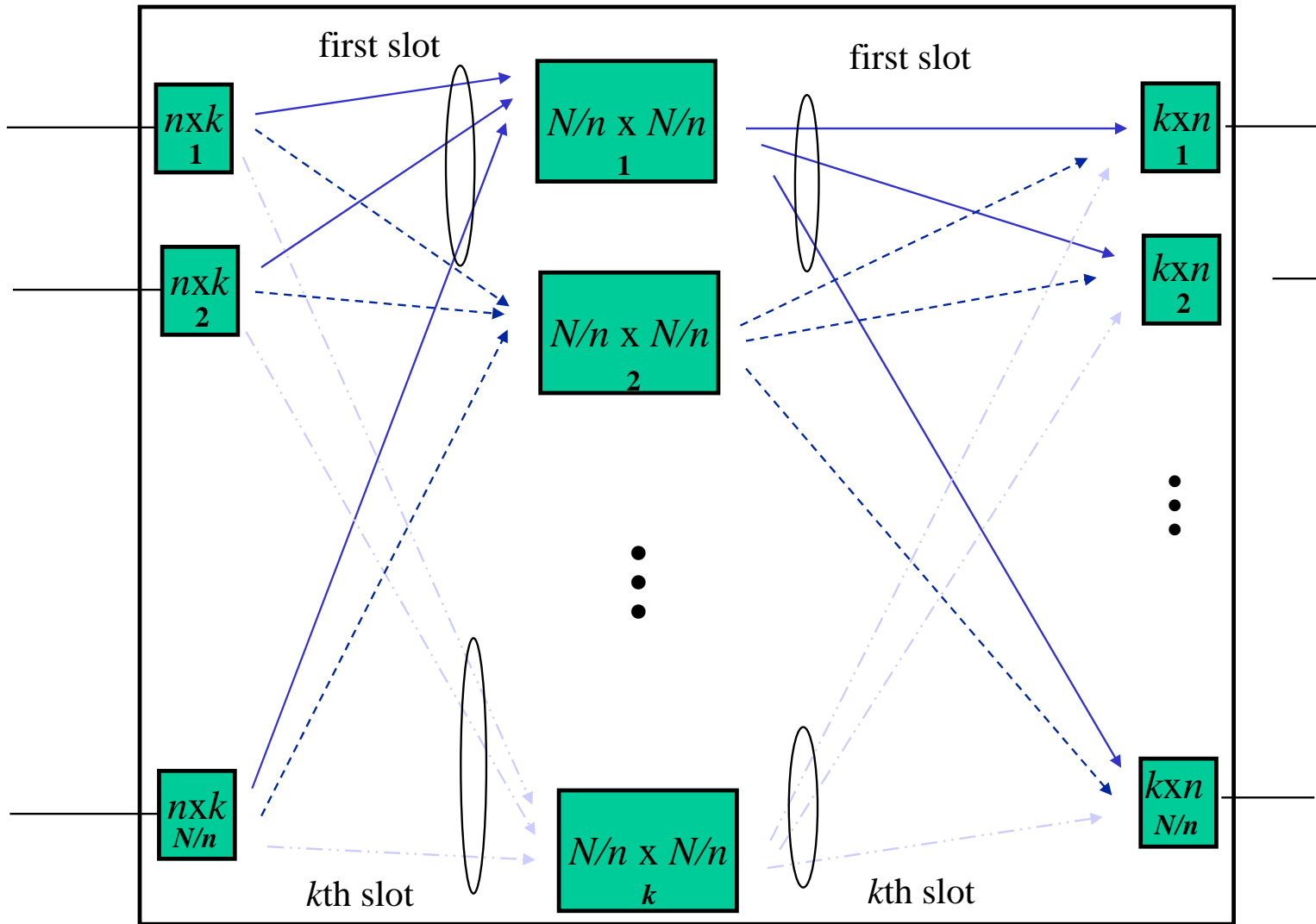
- » a crossbar space switch at the intermediate stage

- design approach is to establish an exact correspondence between the input lines in a space-division switch in the first stage and time-slots in a TSI switch

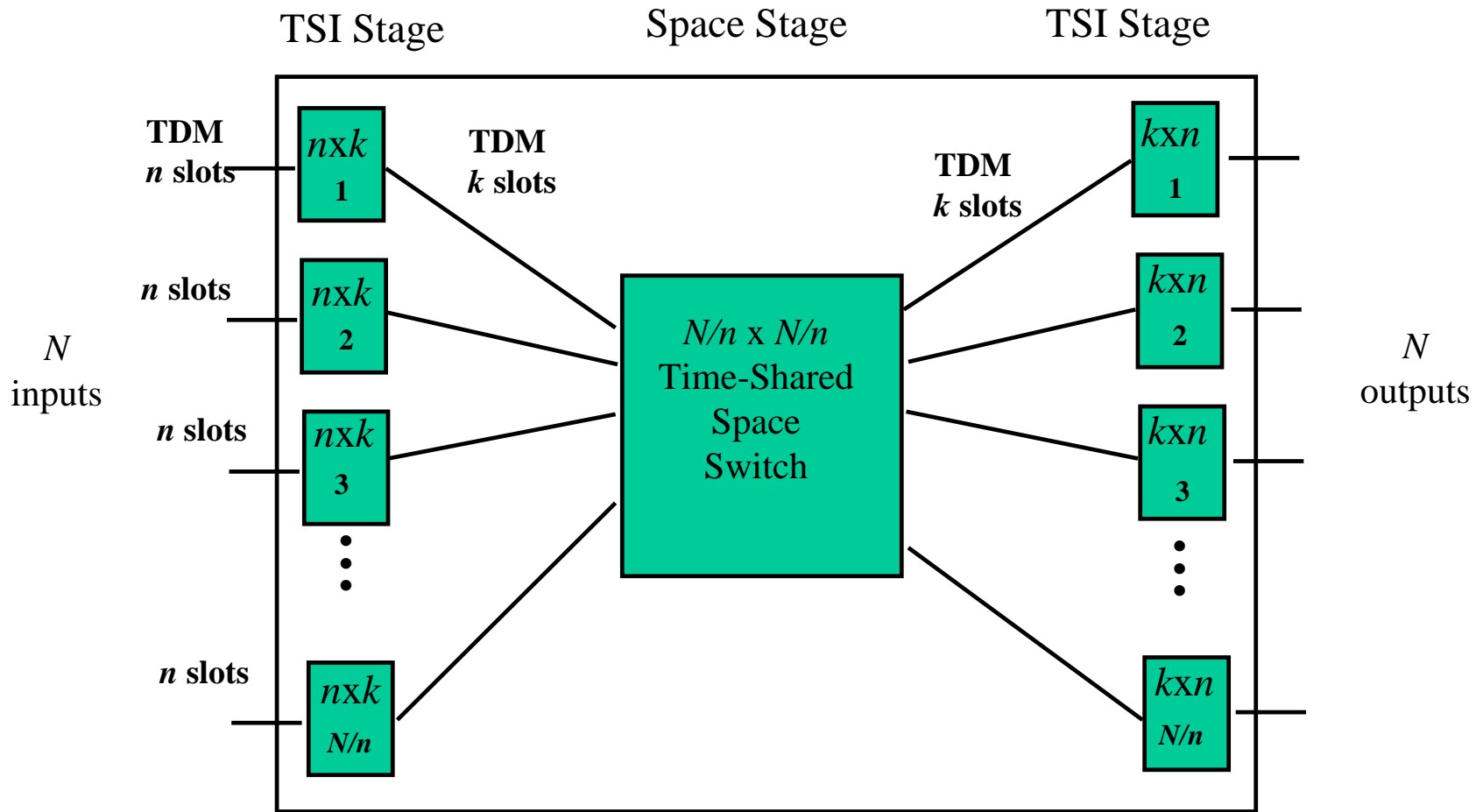
- each  $n \times k$  first-stage switch is replaced by an  $n \times k$  TSI switch:



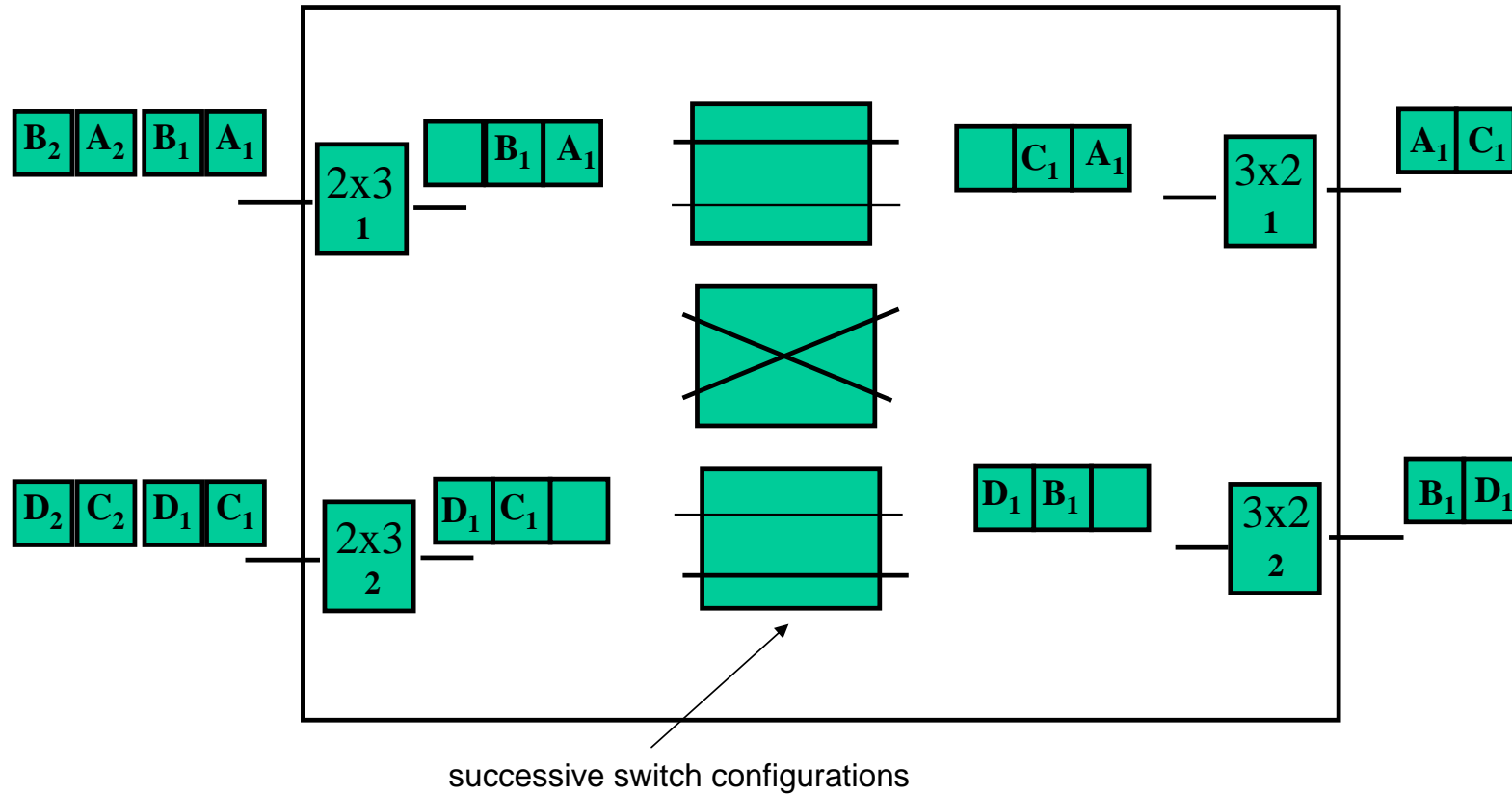
- each input line corresponds to a slot
  - » i.e. the TSI has input frames of size  $n$  and  $k$  output slots
- operation of the the TSI switch:
  - » writes the  $n$  slots of the incoming frame into memory
  - » reads them out from memory into a frame of size  $k$  slots according to some preset permutation
- for the system to operate in a synchronous fashion:
  - » transmission time of an input frame must equal that of the output frame
  - » e.g. if  $k = 2n-1$ , internal speed is nearly double that of the incoming signal
- consider the flow of slots between the first stage and the intermediate stage:
  - » assume TSI frames in the first stage are all *synchronised*
- first time-slot corresponds to first output line of each of the first-stage switches
- recall that the first output line of each first-stage switch is connected to the inputs of the first intermediate switch
  - » the *first intermediate switch* therefore operates on the *first time-slot outputs* from the first-stage switches
  - » the other intermediate switches are *idle* while the first one is busy



- the intermediate switch is a crossbar switch
  - » transfers  $N/n$  inputs to  $N/n$  outputs according to the crosspoint settings
  
- the second output line of each first-stage switch is connected to the inputs of the second intermediate switch
  - » the *second intermediate switch* therefore operates on the *second time-slot outputs* from the first-stage switches
  - » the other intermediate switches are *idle* while the second one is busy
  
- the third output line/time-slot connected to third intermediate switch
  - » the other intermediate switches are *idle* while the third one is busy
  
- only one of the intermediate switches is busy during any one time-slot
- the  $k$  intermediate switches can therefore be replaced by one switch
  - » time-shared among the  $k$  slots in the frame
  - » the single intermediate switch must be reconfigured for each time-slot
  - » known as *time-division* switching

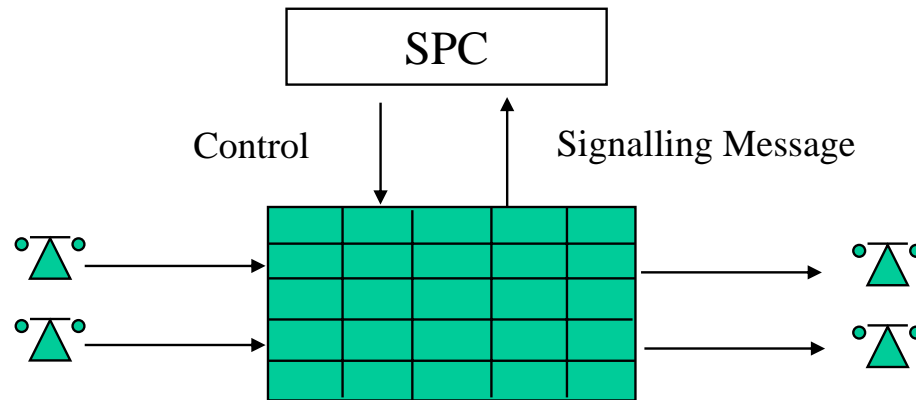


- example: a 4 x 4 switch configured for the connection pattern (A, B, C, D) to (C, A, D, B)
  - » using 2 x 3 input TSI switches
  - » i.e. 2-slot input frames and 3-slot output frames at the first stage switches

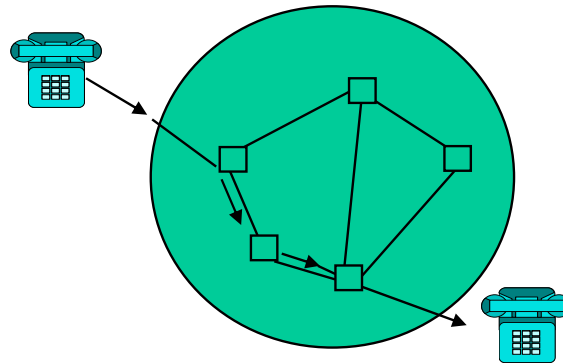


- Switching in the Telephone Network

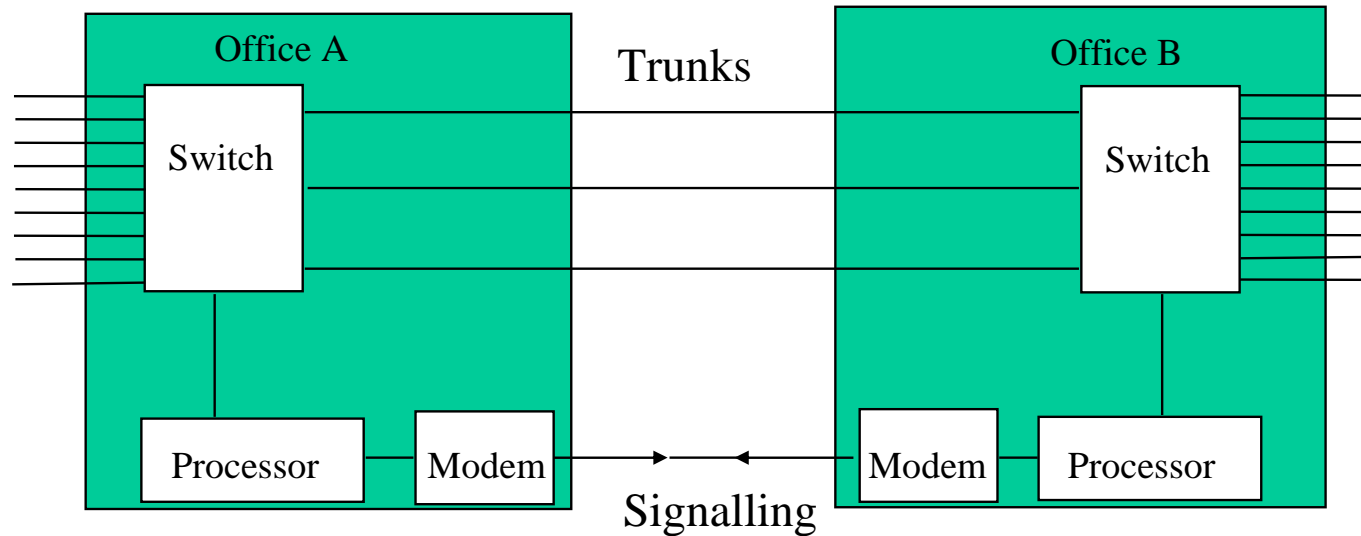
- switches between users must be configured to make the desired connections
  - » under Stored Program Control (SPC) i.e. computer control, in digital exchanges



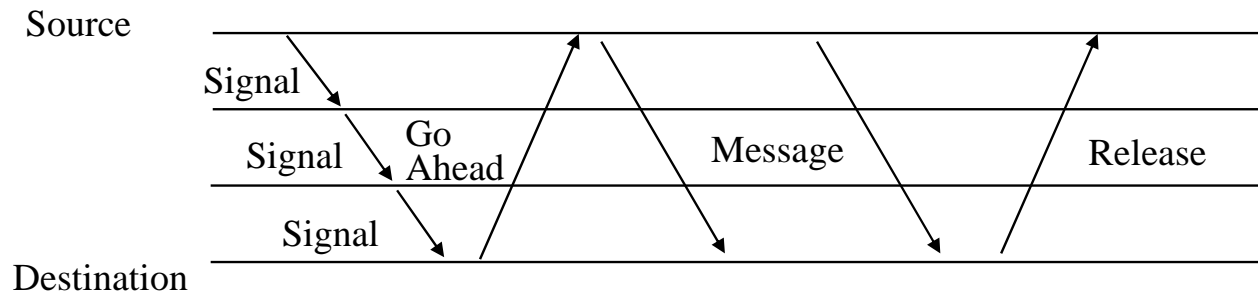
- straightforward if switching in the same local exchange
- switch reconfiguration needed in several exchanges *en route* for remote calls



- a signalling system between exchanges is used to set switch configurations
- a *separate* `out-of-band' signalling network connecting exchanges
  - » designed for high reliability



- e.g. call setup:

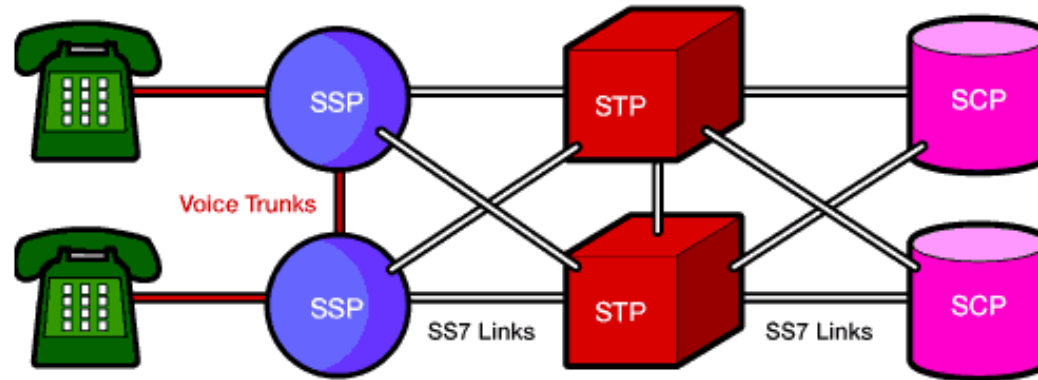




- The Common Channel Signalling System No. 7 (SS#7)
  - ref. <http://www.pt.com/tutorials/ss7>
  - an ITU global standard for defining the protocol for telephone network elements to exchange information
    - » used for: call setup, `tear-down', routing and control
    - » enhanced services: freephone (0800), premium lines, call forwarding, calling party name/number display, credit-card calls, conferencing etc.
    - » wireless mobile roaming
  - out-of-band signalling claims to give:
    - » faster call setup times compared with in-band multi-tone signalling
    - » more efficient use of voice circuits
    - » support for intelligent network services
      - e.g. requiring database accesses
    - » better control over fraudulent use

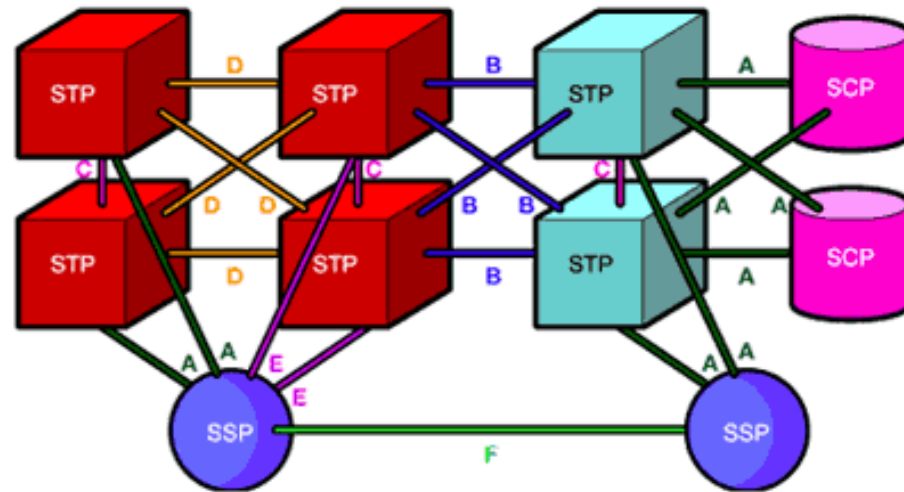
- Signalling points and networks

- each signalling point is uniquely identified by a numeric *point code*
  - » source and destination point codes carried in signalling messages



- SSP : Service Switching Point
  - » switches that originate and terminate calls
  - » sends messages to other SSPs to setup, manage and release voice circuits
  - » may also send query messages to a centralised database (SCP) to determine how to route a call e.g. 0800 calls, or alternate call number if busy/unanswered
- SCP : Service Control Point
- STP : Signal Transfer Point
  - » message packet routing between SSPs, STPs and SCPs

- STPs and SCPs deployed in mated pairs for resilience in event of failure
  - » housed at physically separate places
  - » links between signalling points also deployed in pairs
- alternate pairs of STPs also possible if cost of extra resilience justifiable



- various link types for interconnecting different types of signalling points
  - » A : Access link, B : Bridge link, C : Cross link, D : Diagonal link etc.
- links are bidirectional

- SS#7 Protocol Stack

- comparison with OSI 7-layer model :

- Message Transfer Part (MTP) level 1

- » equivalent to OSI physical layer

- » defines physical & electrical characteristics of the link

- includes interfaces for DS-1 (1.5Mbps), DS-0 (64kbps), etc.

- MTP level 2

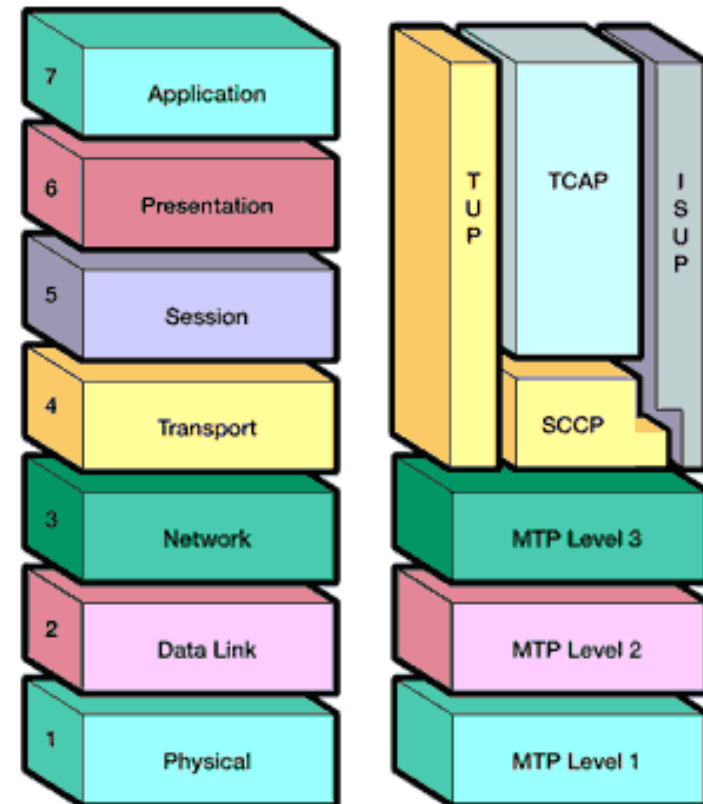
- » ensures accurate end-to-end transmission of a message across a link

- » implements flow control. message sequence validation, error checking

- message retransmission on error

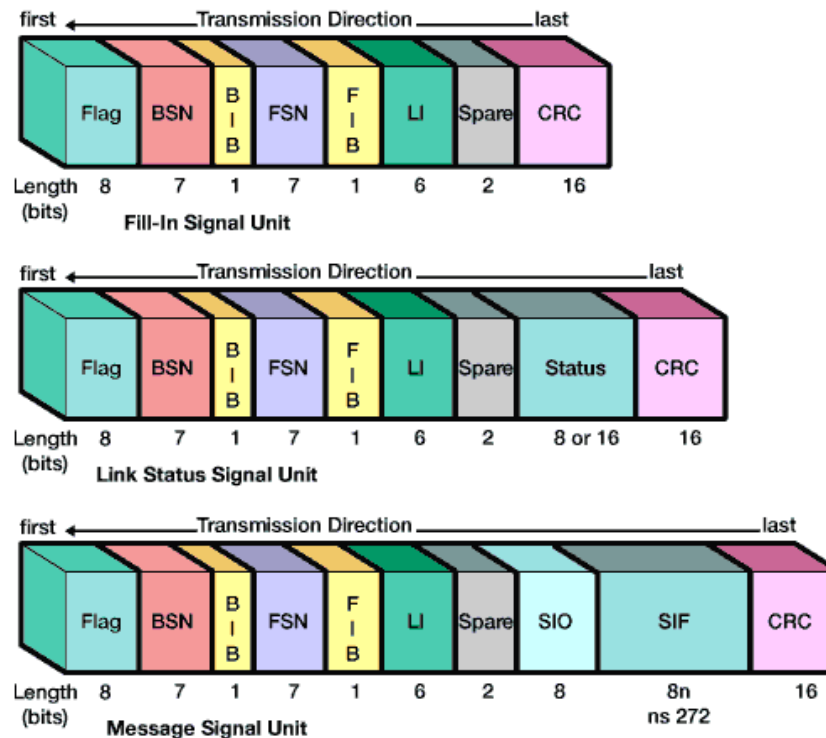
- » nominally equivalent to OSI Data Link Layer

- but also performs higher level functions



– an SS#7 message is called a *signal unit*

- » Fill-in Signal Units (FISUs), Link Status Signal Units (LSSUs), Message Signal Units (MSUs),



- » FISUs are transmitted continuously in both directions in absence of other messages
- » can include acknowledgment of signal unit receipt by a remote signalling point
- » CRC checksum included, so link quality continuously checked

- LSSUs carry status information regarding signalling points at ends of a link
- MSUs carry all call control, database query/response, network management and maintenance data
  - » in the Signalling Information Field (SIF), 1-63 bytes long
- Flag field indicates start of a message = 01111110
  - » 0 bit-stuffing after successive 11111 sequences in rest of message
- BSN : Backward Sequence Number
  - » used to acknowledge receipt of a message
  - » contains the sequence number of the message being acknowledged
- BIB : Backward Indicator Bit
  - » the negative acknowledgment indicator
- FSN : Forward Sequence Number
  - » the sequence number of the message being sent
- FIB : Forward Indicator Bit
  - » error indicator, like BIB
- uses a *Go-Back-N ARQ* (Automatic Repeat Request) protocol (*later lecture*)

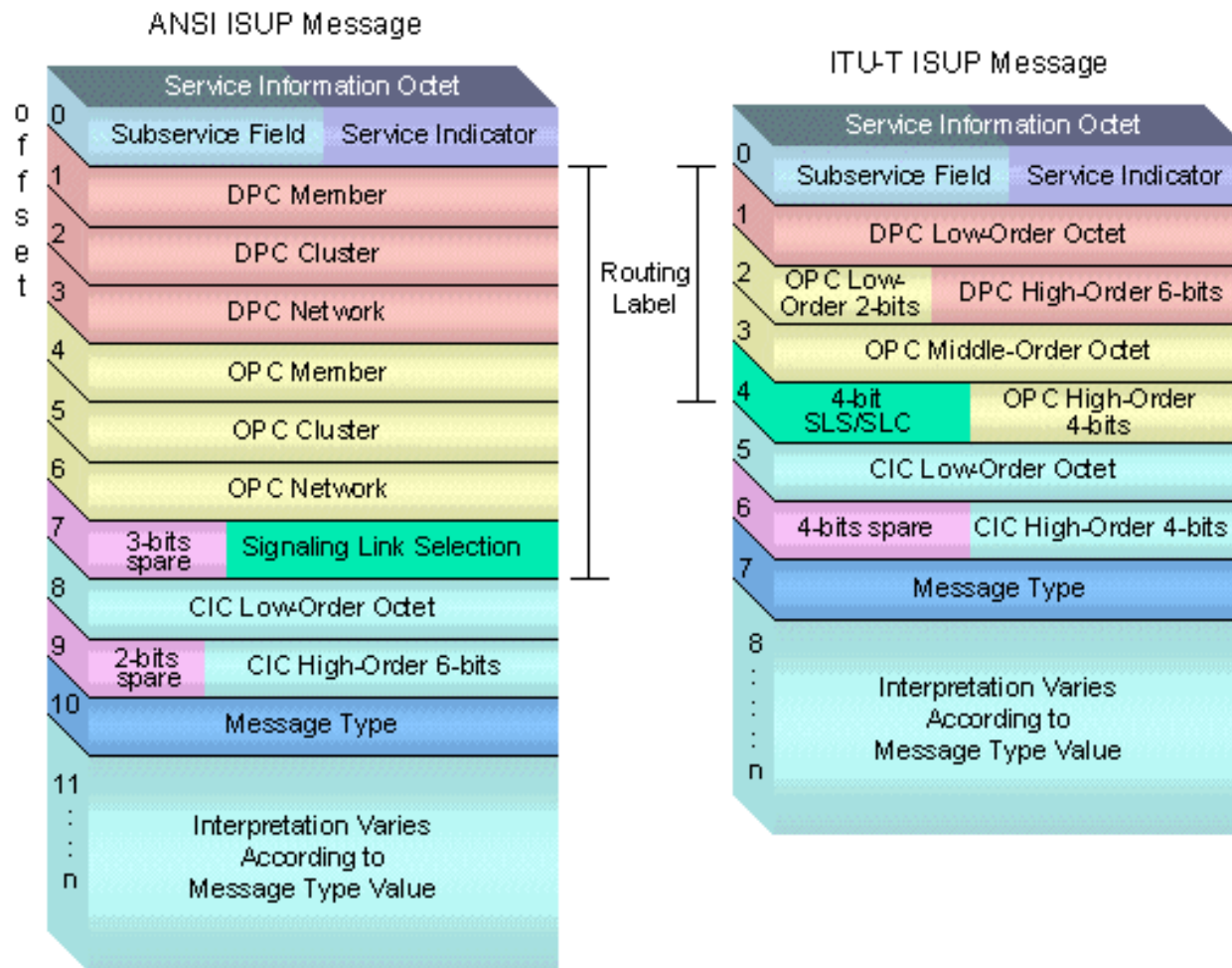
- MTP levels 1 and 2 may be replaced by the ATM (Asynchronous Transfer Mode) protocol in future implementations

- MTP level 3

- » provides message routing between signalling points
- » equivalent to OSI Network layer
- » source and destination routing labels (point codes) in messages
- » message fed up to higher level protocol handler at its destination
- » passed on along the network if not destined for this point

- ISUP : ISDN User Part

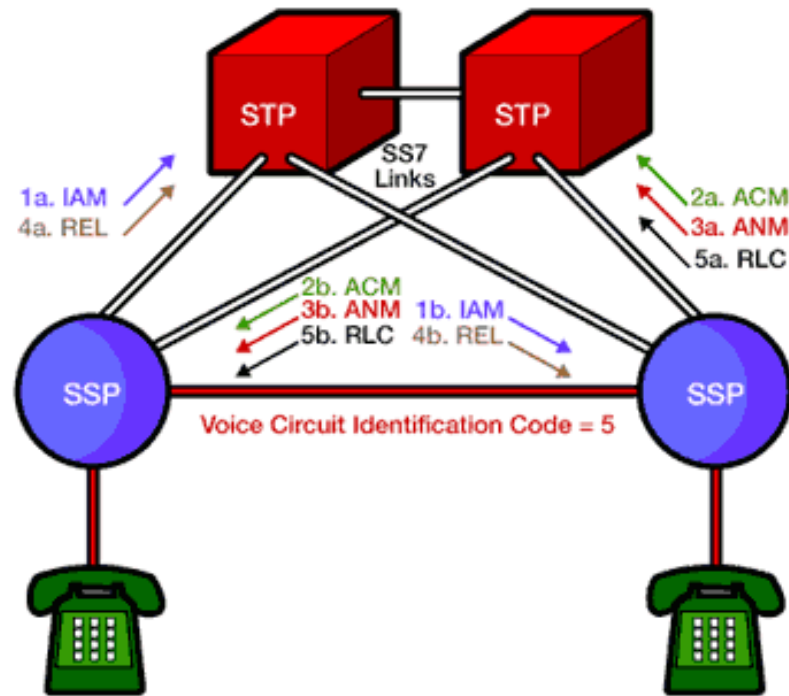
- » used for call setup, management and call release
- » used for both ISDN (Integrated Services Digital Network) and non-ISDN
  - ISDN provides 2 x 64 kbps digital interface at the home/small business level
  - predecessor of broadband



- hierarchical point codes
  - » as in IP addresses
  - » different variants in US and rest of world



- Basic ISUP Call Control : example :



1. Originating SSP transmits an ISUP Initial Address Message (IAM)
  - to reserve an idle trunk circuit from origin to destination switch (1a)
  - message includes circuit identification code, dialled digits and (optionally) the calling party number and name
  - IAM routed, via home STP of originating SSP, to destination SSP (1b)
  - same signalling links used for call duration unless a failure occurs

## 2. Destination switch examines dialled number

- checks that called line is available for ringing
  - puts ring-tone onto called party's line
  - transmits an ISUP Address Complete Message (ACM) back to originating switch (2a), via its home STP
    - to indicate that remote end of trunk circuit has been reserved
  - destination's home STP routes the ACM to the originating switch (2b)
  - originating switch puts ring-tone onto calling party's line
  - and connects it to the trunk to complete the voice circuit
- 
- if originating and destination switches not directly connected with trunks:
    - originating switch transmits an IAM to reserve a trunk to an intermediate switch
    - intermediate switch sends an ACM acknowledging the circuit reservation request
    - then transmits an IAM to reserve a trunk circuit to another switch in the route
    - process continues until all the trunks required to complete the voice circuit from the originating switch to the destination switch are reserved

### 3. When called party picks up the phone

- destination switch terminates the ring-tone
- transmits an ISUP Answer Message (ANM) to originating switch via its home STP (3a)
- STP routes the ANM to originating switch (3b)
  - verifies that calling party's line is connected to the reserved trunk
  - initiates billing!

### 4. If calling party hangs up first

- originating switch sends an ISUP Release Message (REL)
  - to release the trunk circuit between the switches (4a)
- STP routes the REL to the destination switch (4b)

If called party hangs up first, or if the line is busy,

- destination switch sends an REL back to the originating switch
  - indicating the cause of the release e.g. *normal* release of *busy*

## 5. Upon receiving the REL

- destination switch disconnects trunk from called party's line
- sets trunk state to *idle*
- transmits an ISUP Release Complete Message (RLC) to originating switch (5a)
  - acknowledges the release of the remote end of the trunk circuit

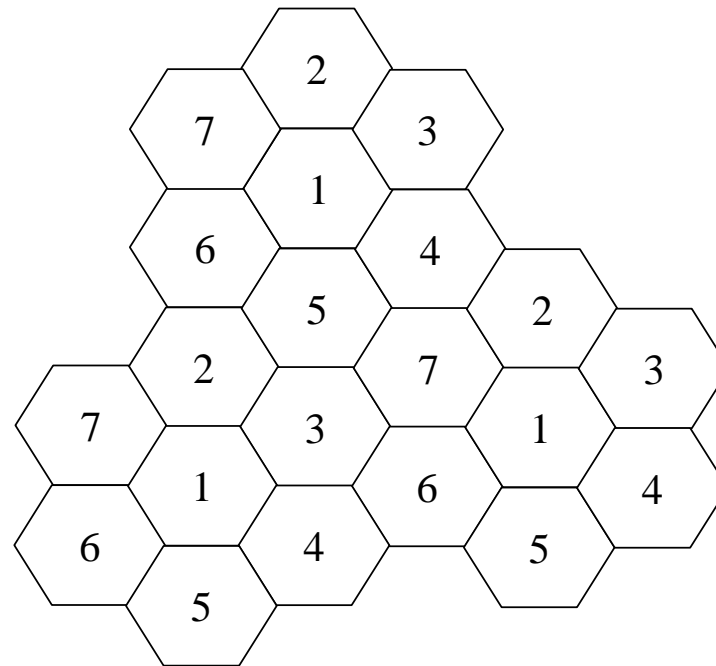
When originating switch receives, or generates, the RLC (5b)

- it terminates the billing
- sets the trunk to *idle* in preparation for the next call

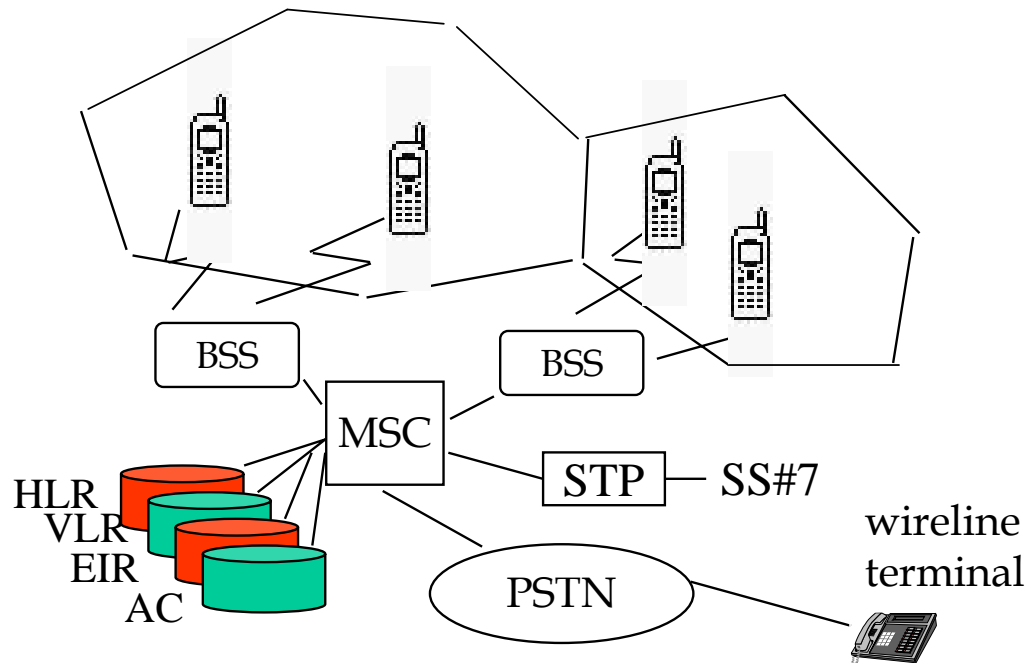
- SCCP : Signalling Connection Control Part
  - provides connectionless and connection-oriented services above MTP level 3
  - allows messages to be addressed to specific applications and services at signalling points
  - used as the *transport layer* for TCAP services
- TCAP : Transaction Capabilities Application Part
  - enables intelligent network services
  - e.g. a query to an SCP to determine the routing address of 0800 numbers and the response from the SCP
  - e.g. when a mobile subscriber roams into a new Mobile Switching Centre area
    - » the Visitor Location Register requests the service profile information from the subscriber's Home Location Register
    - » for validation of service requested etc.

- Cellular Telephone networks

- a region divided up into *cells*
- each cell has a *base station* receiving and transmitting into that cell
  - » in practice, directional transmitters placed at meeting points of three cells
- different radio frequencies used in adjacent cells
  - » frequencies can be re-used in non-adjacent cells



- base stations are connected to the Mobile Switching Centre (MSC) via land-line or microwave links
- MSC handles connections between cells and also to the public switched telephone network via STPs
- also handles *hand-off* as users move from cell to cell

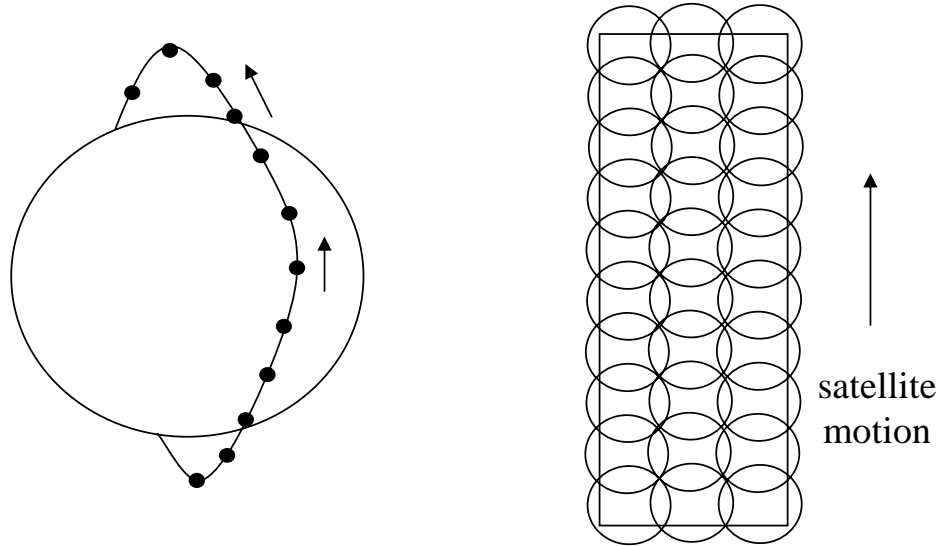


AC = authentication center  
 BSS = base station subsystem  
 EIR = equipment identity register  
 HLR = home location register

MSC = mobile switching center  
 PSTN = public switched telephone network  
 STP = signal transfer point  
 VLR = visitor location register

- Satellite Phones

- geostationary satellites have too long a round-trip time for conversation
- low-earth orbit needed – 750-2000km height
- in polar orbits to cover whole earth surface



- either satellite-fixed (cells defined relative to satellite) or earth-fixed
- ground stations to connect to PSTN + intersatellite links for information relay
- Iridium (Boeing, Motorola, DoD) – 66 satellites in 780Km LEO
- Teledesic (Bill Gates, Craig McCaw) – not off the ground yet